PHYSIOLOGICAL CONTROL SYSTEMS

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Analysis, Simulation, and Estimation

Second Edition

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To
Pam, Bryant, Mason, and Amber
and in memory of my parents
John and Betty Khoo

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PREFACE

It has been 17 years since the publication of the original edition of this monograph. Over this period, I have taught, almost on a yearly basis, a course at the University of Southern California that is based largely on the contents of this book. This second edition incorporates much of the experience I have gained and student feedback I have received from teaching this class. I have also received input from the many instructors who have used this book for their classes. To all these readers, I am deeply appreciative of their helpful comments and questions. The primary goals of this book remain the same as in the first edition: to highlight the basic techniques employed in control theory, systems analysis, and model identification and to give the biomedical engineering student an appreciation of how these principles can be applied to better understand the processes involved in physiological regulation. As before, my assumption is that much of the contents of this second edition are suitable for use in a one-semester course on physiological control systems or physiological systems analysis taken by junior or senior undergraduates or as an introductory class on physiological systems for first-year graduate students. The more advanced parts of this book and its accompanying software may also prove to be a useful resource for biomedical engineers and interested life science or clinical researchers who have had little formal training in systems or control theory. Throughout this book, I have emphasized the physiological applications of control engineering, focusing in particular on the analysis of feedback regulation. In contrast, the basic concepts and methods of control theory are introduced with little attention paid to mathematical derivations or proofs. For this reason, I would recommend the inclusion of a more traditional, engineering-oriented control theory course as a supplement to the material covered in this volume.

One of the main issues I have had with the first edition was the "gap" between the main concepts in systems and control that were introduced assuming continuous-time systems and some of the more advanced applications that featured discrete-time models. Chapter 7 has been introduced to bridge this gap, and to show the reader how continuous-time systems can be converted into discrete-time systems, as well as the impact of different methods of conversion on stability characteristics of the system in question. This additional background should also be useful since many physiological processes (including cardiac, respiratory, and neural) are naturally oscillatory, and models that employ a cycle-by-cycle (and hence, discrete) time base may be more suitable for characterizing longer term dynamics. In Chapter 9, I have revamped what was previously Chapter 8 to cover the essential aspects of time-varying or nonstationary systems. The chapter on physiological system identification (now Chapter 8) has been expanded to include more techniques, such as nonparametric identification using multivariable autoregressive with exogenous (ARX) models and basis function expansion. Finally, the chapter on nonlinear analysis (now Chapter 10) has been expanded to include the Volterra kernel approach to nonparametric estimation of nonlinear systems as well as an introductory discussion of other methods. I have also added material to update various other sections, as well as new problems to the end of each chapter. The MATLAB/SIMULINK files accompanying the book have also been expanded and existing programs have been updated to be compatible with release version R2016b. I see these programs to be an essential complement to the learning experience, allowing the reader to explore "first-hand" the dynamics underlying the biological mechanisms being studied. I do make the implicit assumption that the reader has some basic familiarity with MATLAB/SIMULINK. For the reader who has not used MATLAB or SIMULINK, it is fortunate that there are currently many "primers" on the subject that can be easily found online or in any academic bookstore.

The completion of this second edition has taken much longer than I had anticipated when I took on the project (and I am quite embarrassed to disclose how long 'long' is!). I thank Wiley-IEEE Press for giving me the opportunity to produce this second edition, editor Mary Hatcher for her infinite patience, and my friend Metin Akay, the book series editor, for his constant encouragement. This second edition would not have been possible without the feedback and insights gained through my interactions with my past and present Ph.D. students over the years. In particular, I am most grateful to my former student and current research associate, P. "Sang" Chalacheva, who so generously gave her spare time and effort to help with the development of the new MATLAB files and the editing of all parts of this second edition. I would be remiss if I did not also mention the supportive environment provided by the NIH-NIBIB-funded Biomedical Simulations Resource (BMSR), which has funded my research on physiological control and modeling for the past three decades. The modeling activities of my colleagues in the BMSR, David D'Argenio, Vasilis Marmarelis, and Ted Berger, have been a great source of intellectual stimulation over the years. I cannot help but end these remarks

by citing my favorite line from the writings of the late Professor Fred Grodins, who recruited me to USC many many moons ago:

"There is nothing magic about Models (or is there?)!"

Michael C.K. Khoo

ABOUT THE COMPANION WEBSITE

This book is accompanied by a companion website:

   http://www.wiley.com/go/khoo/controlsystems2e

The website includes:

- MATLAB and Simulink Files

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INTRODUCTION

1.1 PRELIMINARY CONSIDERATIONS

A *control system* may be defined as a collection of interconnected components that can be made to achieve a desired response in the face of external disturbances. The "desired response" could be the tracking of a specified dynamic trajectory, in which case the control system takes the form of a *servomechanism*. An example of this type of control system is a robot arm that is programmed to grasp some object and to move it to a specified location. There is a second class of control system termed the *regulator*, for which the "desired response" is to maintain a certain physical quantity within specified limits. A simple example of this kind of control system is the thermostat.

There are two basic ways in which a control system can be made to operate. In *open-loop* mode, the response of the system is determined only by the controlling input(s). As an example, let us suppose that we wish to control the temperature of a room in winter with the use of a fan-heater that heats up and circulates the air within the room. By setting the temperature control to "medium," for instance, we should be able to get the room temperature to settle down to an agreeable level during the morning hours. However, as the day progresses and the external environment becomes warmer, the room temperature also will rise, because the rate at which heat is added by the fan-heater exceeds the rate at which heat is dissipated from the room. Conversely, when night sets in and the external temperature falls, the temperature in

the room will decrease below the desired level unless the heater setting is raised. This is a fundamental limitation of open-loop control systems. They can perform satisfactorily as long as the external conditions do not affect the system much. The simple example we have described may be considered a physical analog of thermoregulatory control in *poikilothermic* or "cold-blooded" animals. The design of the thermoregulatory processes in these animals do not allow core body temperature to be maintained at a level independent of external conditions; as a consequence, the animal's metabolism also becomes a function of external temperature.

Coming back to the example of the heating system, one way to overcome its limitation might be to anticipate the external changes in temperature and to "preprogram" the temperature setting accordingly. But how would we know what amounts of adjustment are required under the different external temperature conditions? Furthermore, while the external temperature generally varies in a roughly predictable pattern, there will be occasions when this pattern is disrupted. For instance, the appearance of a heavy cloud cover during the day could limit the temperature increase that is generally expected. These problems can be eliminated by making the heater "aware" of changes in the room temperature, thereby allowing it to respond accordingly. One possible scheme might be to measure the room temperature, compare the measured temperature with the desired room temperature, and adjust the heater setting in proportion to the difference between these two temperatures. This arrangement is known as *proportional feedback control*. There are, of course, other control strategies that make use of the information derived from measurements of the room temperature. Nevertheless, there is a common feature in all these control schemes: They all employ *feedback*. The great mathematician-engineer, Norbert Wiener (1961), characterized feedback control as "a method of controlling a system by reinserting into it the results of its past performance." In our example, the system *output* (the measured room temperature) is "fed back" and used to adjust the *input* (fan speed). As a consequence, what we now have is a control system that operates in *closed-loop* mode, which also allows the system to be *self-regulatory*. This strategy of control is ubiquitous throughout Nature: The physiological analog of the simple example we have been considering is the thermoregulatory control system of *homeothermic* or "warm-blooded" animals. However, as we will demonstrate throughout this book, the exact means through which closed-loop control is achieved in physiological systems invariably turns out to be considerably more complicated than one might expect.

1.2   HISTORICAL BACKGROUND

The concept of physiological regulation dates back to ancient Greece (∼500 BC), where the human body was considered a small replica of the universe. The four basic elements of the universe – air, water, fire, and earth – were represented in the body by blood, phlegm, yellow bile, and black bile, respectively. The interactions among pairs of these elements produced the four irreducible qualities of wetness,

warmth, dryness, and cold. It was the harmonious *balance* among these elements and qualities that led to the proper functioning of the various organ systems. The Greek physician, Galen (about second century AD), consolidated these traditional theories and promoted a physiological theory that was largely held until the end of the sixteenth century. Similar concepts that developed alongside the Taoist school of thought may be traced back to the third century BC in ancient China. Here, the universe was composed of five agents (*Wu Xing*): wood, fire, earth, metal, and water. These elements interacted with one another in two ways – one was a productive relationship, in which one agent would enhance the effects of the other; the other was a limiting or destructive relationship whereby one agent would constrain the effects of the other. As in the Graeco-Roman view, health was maintained by the harmonious balancing of these agents with one another (Unschuld, 1985).

The notion of regulatory control clearly persisted in the centuries that followed, as the writings of various notable physiologists such as Boyle, Lavoisier, and Pflüger demonstrate. However, this concept remained somewhat vague until the end of the nineteenth century when French physiologist Claude Bernard thought about self-regulation in more precise terms. He noted that the cells of higher organisms were always bathed in a fluid medium, for example, blood or lymph, and that the conditions of this environment were maintained with great stability in the face of disturbances to the overall physiology of the organism. The maintenance of these relatively constant conditions was achieved by the organism itself. This observation so impressed him that he wrote: "It is the fixity of the 'milieu interieur' which is the condition of free and independent life." He added further that "all the vital mechanisms, however varied they may be, have only one object, that of preserving constant the conditions of life in the internal environment." In the earlier half of this century, Harvard physiologist Walter Cannon (1939) refined Bernard's ideas further and demonstrated systematically these concepts in the workings of various physiological processes, such as the regulation of adequate water and food supply through thirst and hunger sensors, the role of the kidneys in regulating excess water, and the maintenance of blood acid–base balance. He went on to coin the word *homeostasis* to describe the maintenance of relatively constant physiological conditions. However, he was careful to distinguish the second part of the term, that is, "stasis," from the word "statics," since he was well aware that although the end result was a relatively unchanging condition, the coordinated physiological processes that produce this state are highly dynamic.

Armed with the tools of mathematics, Wiener in the 1940s explored the notion of feedback to a greater level of detail than had been done previously. Mindful that most physiological systems were nonlinear, he laid the foundation for modeling nonlinear dynamics from a Volterra series perspective. He looked into the problem of instability in neurological control systems and examined the connections between instability and physiological oscillations. He coined the word "cybernetics" to describe the application of control theory to physiology, but with the passage of time, this term has come to take on a meaning more closely associated with

robotics. The race to develop automatic airplane, radar, and other military control systems during the Second World War provided a tremendous boost to the development of control theory. In the post-war period, an added catalyst for even greater progress was the development of digital computers and the growing availability of facilities for the numerical solution of the complex control problems. Since then, research on physiological control systems has become a field of study on its own, with major contributions coming from a mix of physiologists, mathematicians, and engineers. These pioneers of "modern" physiological control systems analysis include Adolph (1961), Grodins (1963), Clynes and Milsum (1970), Milhorn (1966), Milsum (1966), Bayliss (1966), Stark (1968), Riggs (1970), Guyton et al. (1973), and Jones (1973).

1.3 SYSTEMS ANALYSIS: FUNDAMENTAL CONCEPTS

Prior to analyzing or designing a control system, it is useful to define explicitly the major variables and structures involved in the problem. One common way of doing this is to construct a *block diagram*. The block diagram captures in schematic form the relationships among the variables and processes that comprise the control system in question. Figure 1.1 shows block diagrams that represent open- and closed-loop control systems in canonical form. Consider first the open-loop system
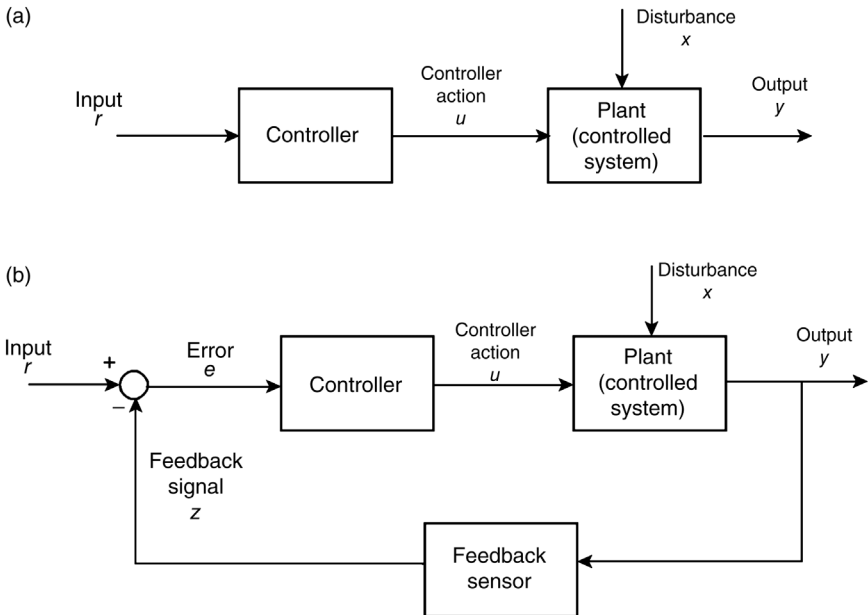
FIGURE 1.1 Block diagrams of an open-loop control system (a) and a closed-loop control system (b).

(Figure 1a). Here, the controller component of the system translates the input (r) into a controller action (u), which affects the controlled system or "plant," thereby influencing the system output (y). At the same time, however, external disturbances (x) also affect plant behavior; thus, any changes in y reflect contributions from both the controller and the external disturbances. If we consider this open-loop system in the context of our previous example of the heating system, the heater would be the controller and the room would represent the plant. Since the function of this control system is to regulate the temperature of the room, it is useful to define a *set point* that would correspond to the desired room temperature. In the ideal situation of no fluctuations in external temperature (i.e., $x = 0$), a particular input voltage setting would place the room temperature exactly at the set point. This input level may be referred to as the *reference input value*. In linear control systems analysis, it is useful (and often preferable from a computational viewpoint) to consider the system variables in terms of *changes* from these reference levels instead of their absolute values. Thus, in our example, the input (r) and controller action (u) would represent the deviation from the reference input value and the corresponding change in heat generated by the heater, respectively, while the output (y) would reflect the resulting change in room temperature. Due to the influence of changes in external temperature (x), r must be adjusted continually to offset the effect of these disturbances on y.

As mentioned earlier, we can circumvent this limitation by "closing the loop." Figure 1.1b shows the closed-loop configuration. The change in room temperature (y) is now measured and transduced into the feedback signal (z) by means of a feedback sensor, that is, the thermostat. The feedback signal is subsequently subtracted from the reference input and the error signal (e) is used to change the controller output. If room temperature falls below the set point (i.e., y becomes negative), the feedback signal (z) would also be negative. This feedback signal is subtracted from the reference input setting ($r = 0$) at the *mixing point* or *comparator* (shown as the circular object in Figure 1.1), producing the error signal (e) that is used to adjust the heater setting. Since z is negative, e will be positive. Thus, the heater setting will be raised, increasing the flow of heat to the room and consequently raising the room temperature. Conversely, if room temperature becomes higher than its set point, the feedback signal now becomes positive, leading to a negative error signal, which in turn lowers the heater output. This kind of closed-loop system is said to have *negative feedback*, since any changes in system output are compensated for by changes in controller action in the *opposite* direction.

Negative feedback is the key attribute that allows closed-loop control systems to act as regulators. What would happen if, rather than being subtracted, the feedback signal were to be *added* to the input? Going back to our example, if the room temperature were to rise and the feedback signal were to be added at the comparator, the error signal would become positive. The heater setting would be raised and the heat flow into the room would be increased, thereby increasing the room temperature further. This, in turn, would increase the feedback signal and the error signal, and thus produce even further increases in room temperature. This kind of situation represents the *runaway* effect that can result from *positive* feedback. In lay

language, one would refer to this as a *vicious cycle* of events. Dangerous as it may seem, positive feedback is actually employed in many physiological processes. However, in these processes, there are constraints built in that limit the extent to which the system variables can change. Nevertheless, there are also many positive feedback processes, for example, circulatory shock, that in extreme circumstances can lead to the shutdown of various system components, leading eventually to the demise of the organism.

1.4 PHYSIOLOGICAL CONTROL SYSTEMS ANALYSIS: A SIMPLE EXAMPLE

One of the simplest and most fundamental of all physiological control systems is the *muscle stretch reflex*. The most notable example of this kind of reflex is the *knee jerk*, which is used in routine medical examinations as an assessment of the state of the nervous system. A sharp tap to the patellar tendon in the knee leads to an abrupt stretching of the extensor muscle in the thigh to which the tendon is attached. This activates the muscle spindles, which are stretch receptors. Neural impulses, which encode information about the magnitude of the stretch, are sent along afferent nerve fibers to the spinal cord. Since each afferent nerve synapses with one motorneuron in the spinal cord, the motorneurons get activated and, in turn, send efferent neural impulses back to the same thigh muscle. These produce a contraction of the muscle that acts to straighten the lower leg. Figure 1.2 shows the basic components of this
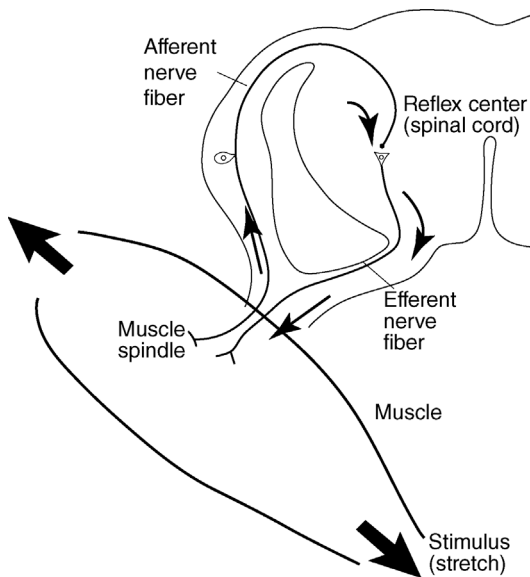
FIGURE 1.2 Schematic illustration of the muscle stretch reflex. (Adapted from Vander et al. (1997).)

reflex. A number of important features of this system should be highlighted. First, this and other stretch reflexes involve reflex arcs that are monosynaptic, that is, only two neurons and one synapse are employed in the reflex. Other reflexes have at least one interneuron connecting the afferent and efferent pathways. Second, this closed-loop regulation of muscle length is accomplished in a completely involuntary fashion, as the name "reflex" suggests.

A third important feature of the muscle stretch reflex is that it provides a good example of negative feedback in physiological control systems. Consider the block diagram representation of this reflex, as shown in Figure 1.3. Comparing this configuration with the general closed-loop control system of Figure 1.1, one can see that the thigh muscle now corresponds to the plant or controlled system. The disturbance x is the amount of initial stretch produced by the tap to the knee. This produces a proportionate amount of stretch y in the muscle spindles, which act as the feedback sensor. The spindles translate this mechanical quantity into an increase in afferent neural traffic (z) sent back to the reflex center in the spinal cord, which corresponds to our controller. In turn, the controller action is an increase in efferent neural traffic (u) directed back to the thigh muscle, which subsequently contracts in order to offset the initial stretch. Although this closed-loop control system differs in some details from the canonical structure shown in Figure 1.1, it is indeed a *negative feedback* system, since the initial disturbance (tap-induced stretch) leads to a controller action that is aimed at *reducing* the effect of the disturbance.
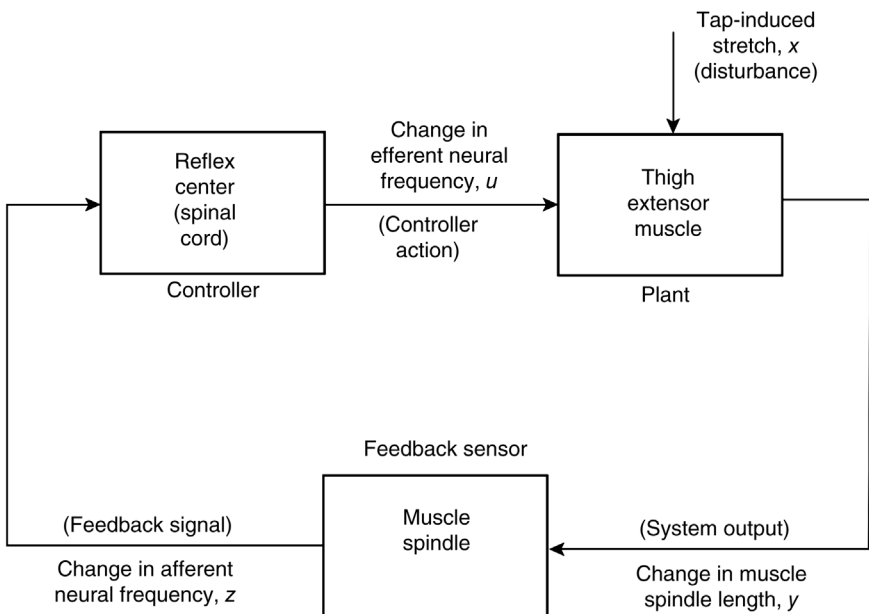
FIGURE 1.3 Block diagram representation of the muscle stretch reflex.

1.5   DIFFERENCES BETWEEN ENGINEERING AND PHYSIOLOGICAL CONTROL SYSTEMS

While the methodology of systems analysis can be applied to both engineering and physiological control systems, it is important to recognize some key differences:

(a) An engineering control system is designed to accomplish a defined task, and frequently the governing parameters would have been fine-tuned extensively so that the system will perform its task in an "optimal" manner (at least under the circumstances in which it is tested). In contrast, physiological control systems are built for versatility and may be capable of serving several different functions. For instance, although the primary purpose of the respiratory system is to provide gas exchange, a secondary but also important function is to facilitate the elimination of heat from the body. Indeed, some of the greatest advances in physiological research have been directed at discovering the functional significance of various biological processes.

(b) Since the engineering control system is synthesized by the designer, the characteristics of its various components are generally known. On the other hand, the physiological control system usually consists of components that are unknown and difficult to analyze. Thus, we are confronted with the need to apply system identification techniques to determine how these various subsystems behave before we are able to proceed to analyze the overall control system.

(c) There is an extensive degree of *cross-coupling* or interaction among different physiological control systems. The proper functioning of the cardiovascular system, for instance, is to a large extent dependent on interactions with the respiratory, renal, endocrine, and other organ systems. In the example of the muscle stretch reflex considered earlier, the block diagram shown in Figure 1.3 oversimplifies the actual underlying physiology. There are other factors involved that we had omitted and these are shown in the modified block diagram shown in Figure 1.4. First, some branches of the afferent nerves also synapse with the motorneurons that lead to other extensor muscles in the thigh that act synergistically with the primary muscle to straighten the lower leg. Second, other branches of the afferent nerves synapse with interneurons, which, in turn, synapse with motorneurons that lead to the flexor or antagonist muscles. However, here the interneurons introduce a polarity change in the signal so that an increase in afferent neural frequency produces a decrease in the efferent neural traffic that is sent to the flexor muscles. This has the effect of relaxing the flexor muscles so that they do not counteract the activity of the extensor muscles.
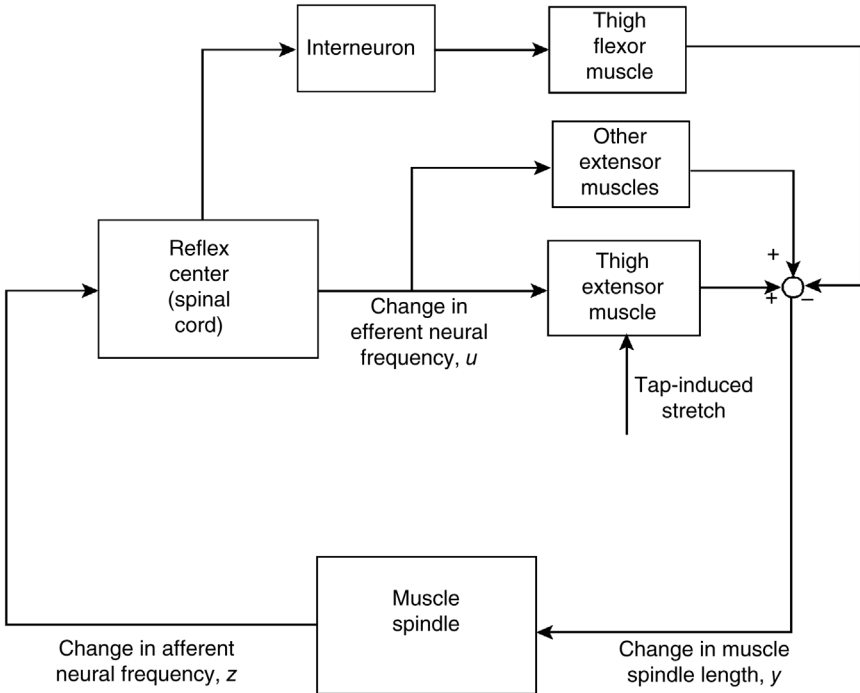
FIGURE 1.4   Contributions of interrelated systems to the muscle stretch reflex.

(d) Physiological control systems, in general, are *adaptive*. This means that the system may be able to offset any change in output not only through feedback but also by allowing the controller or plant characteristics to change. As an example of this type of feature, consider again the operation of the muscle stretch reflex. While this reflex plays a protective role in regulating muscle stretch, it also can hinder the effects of voluntary control of the muscles involved. For instance, if one voluntarily flexes the knee, the stretch reflex, if kept unchanged, would come into play and this would produce effects that oppose the intended movement. Figure 1.5 illustrates the solution chosen by Nature to circumvent this problem. When the higher centers send signals down the alpha motorneurons to elicit the contraction of the flexor muscles and the relaxation of the extensor muscle, signals are sent simultaneously down the efferent gamma nerves that innervate the muscle spindles. These gamma signals produce in effect a resetting of the operating lengths of the muscle spindles so that the voluntarily induced stretch in the extensor muscles is no longer detected by the spindles. Thus, by employing this clever, adaptive arrangement, the muscle stretch reflex is basically neutralized.
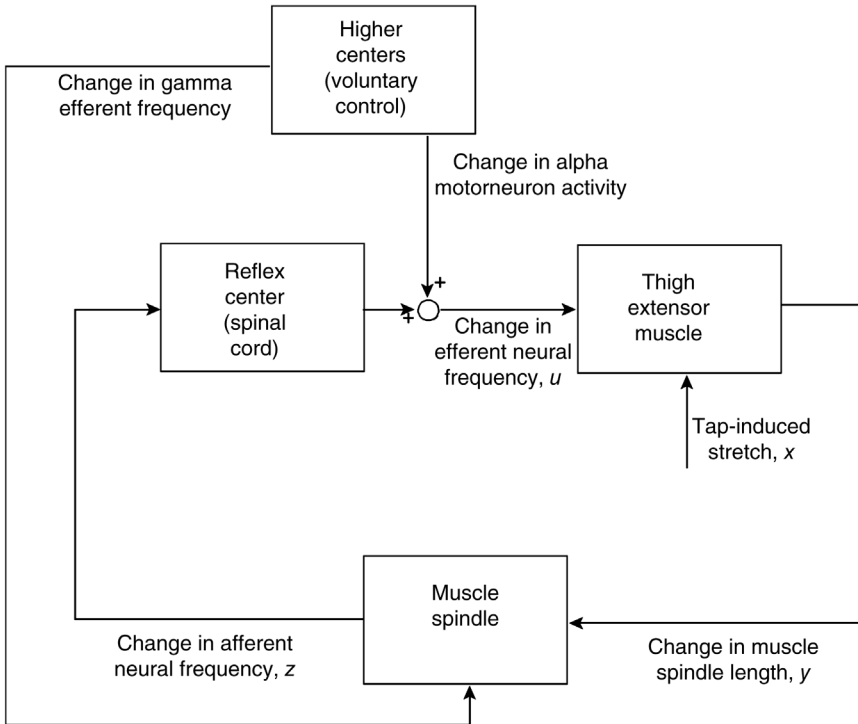
FIGURE 1.5   Adaptive characteristics of the muscle stretch reflex.

(e) At the end of Section 1.4, we alluded to another difference that may be found between physiological control systems and simpler forms of engineering control systems. In Figure 1.1, the feedback signal is explicitly subtracted from the reference input, demonstrating clearly the use of negative feedback. However, in the stretch reflex block diagram of Figure 1.3, the comparator is nowhere to be found. Furthermore, muscle stretch leads to an *increase* in both afferent and efferent neural traffic. So, how is negative feedback achieved? The answer is that negative feedback in this system is "built into" in the plant characteristics: Increased efferent neural input produces a *contraction* of the extensor muscle, thereby acting to counteract the initial stretch. This kind of *embedded* feedback is highly common in physiological systems.

(f) One final difference is that physiological systems are generally nonlinear, while engineering control systems can be linear or nonlinear. Frequently, the engineering designer prefers the use of linear system components since they have properties that are well-behaved and easy to predict. This issue will be revisited many times over in the chapters to follow.

1.6 THE SCIENCE (AND ART) OF MODELING

As we have shown, the construction of block diagrams is useful in helping us clarify in our own minds what key variables best represent the system under study. It is also helpful in allowing us to formalize our belief (which is usually based partly on other people's or our own observations and partly on intuition) of how the various processes involved are causally related. The block diagram that emerges from these considerations, therefore, represents a *conceptual model* of the physiological control system under study. However, such a model is limited in its ability to enhance our understanding or make predictions, since it only allows qualitative inferences to be made.

To advance the analysis to the next level involves the upgrading of the conceptual model into a *mathematical model*. The mathematical model allows us to make hypotheses about the contents in each of the "boxes" of the block diagram. For instance, in Figure 1.3, the box labeled "controller" will contain an expression of our belief of how the change in afferent neural frequency may be related to the change in efferent neural frequency. Is the relationship between afferent frequency and efferent frequency linear? If the changes in afferent frequency follow a particular time-course, what would the time-course of the response in efferent frequency be like? One way of answering these questions would be to isolate this part of the physiological control system and perform experiments that would allow us to measure this relationship. In this case, the relationship between the controller input and controller output is derived purely on the basis of observations, and therefore it may take the form of a table or a curve best fitted to the data. Alternatively, these data may already have been measured, and one could simply turn to the literature to establish the required input–output relationship. This kind of model assumes no internal structure and has been given a number of labels in the physiological control literature, such as *black-box*, *empirical*, or *nonparametric* model. Frequently, on the basis of previous knowledge, we also have some idea of what the underlying physical or chemical processes are likely to be. In such situations, we might propose a hypothesis that reflects this belief. On the basis of the particular physical or chemical laws involved, we would then proceed to derive an algebraic, differential, or integral equation that relates the "input" to the "output" of the system component we are studying. This type of model is said to possess an internal structure, that is, it places some constraints on how the input may affect the output. As such, we might call this a *structural* or *gray-box* model. In spite of the constraints built into this kind of model, the range of input–output behavior that it is capable of characterizing can still be quite extensive, depending on the number of free parameters (or coefficients) it incorporates. For this reason, this type of model is frequently referred to as a *parametric* model.

Mathematical modeling may be seen as the use of a "language" to elaborate on the details of the conceptual model. However, unlike verbal languages, mathematics provides descriptions that are unambiguous, concise, and self-consistent. By

being unambiguous, different researchers are able to use and test the same model without being confused about the hypotheses built into the model. Since the equations employed in the model are based, at least in large part, on existing knowledge of the physiological processes in question, they also serve the useful purpose of archiving past knowledge and compressing all that information into a compact format. The inherent self-consistency of the model derives from the operational rules of mathematics, which provide a logical accounting system for dealing with the multiple system variables and their interactions with one another. On the other hand, the hypotheses embedded in some components of the model are only *hypotheses*, reflecting our best belief regarding the underlying process. More often than not, these are incorrect or oversimplistic. As a consequence, the behavior of the model may not reflect the corresponding reality. Yet, the power of the modeling process lies in its replication of the scientific method: The discrepancy between model prediction and physiological observation can be used as "feedback" to alert us to the inadequacies of one or more component hypotheses. This allows us to return to the model development stage once again in order to modify our previous assumptions. Subsequently, we would retest the revised model against experimental observations. And so, the alternating process of induction and deduction continues until we are satisfied that the model can "explain" most of the observed behavior. Then, having arrived at a "good" model, we could venture to use this model to predict how the system might behave under experimental conditions that have not been employed previously. These predictions would serve as a guide for the planning and design of future experiments.

1.7 "SYSTEMS PHYSIOLOGY" VERSUS "SYSTEMS BIOLOGY"

We would be remiss if we did not mention the currently widespread application of mathematical modeling and control theory to biological systems over a much broader spectrum of spatial and temporal scales. "Systems biology" has come to be recognized as a mainstay of biological science, rather than an isolated discipline. To understand the compelling need for systems biology, one must look back into the 1950s when Watson and Crick (1953) published their two-page, landmark paper in *Nature*, entitled "Molecular Structure of Nucleic Acids: A Structure for Deoxyribose Nucleic Acid." This paper provided a jump-start to the nascent field of molecular biology at the time. In subsequent lectures and papers, Crick introduced the "sequence hypothesis" that evolved into the "central dogma": This laid out the two-step process, transcription and translation, through which genetic information flows from DNA to mRNA to protein. These and other concurrent developments in molecular biology heralded the golden age of modern biology. The rush was on to develop more reliable and higher throughput methods of DNA sequencing, which ushered in the field of genomics. In turn, technologies were also developed to detect gene mutations using SNP methods. Attention then turned to the development of other technologies (e.g., gene chips, microarrays) for transcriptomics – the

cataloging of the complete set of RNA molecules produced by the genome – and, subsequently, proteomics and metabolomics.

As these developments progressed at exponentially increasing speeds, it became clear that the original "reductionist" approach of attempting to "explain" a biological system as the sum of its various components was woefully inadequate. Instead, it was necessary to consider the networks that bind these disparate components and to study the dynamic interactions among these components. Sequential reasoning and intuitive thinking, which worked well for the classical physiologists and biologists, fell by the wayside as high-throughput techniques and new tools from the "-omics" generated avalanche after avalanche of data. As such, it has become necessary to adopt the rigorous framework with the necessary computational tools to select out features from the data that bear relevance to the questions being posed, arrive at a mathematical framework for capturing the dynamic relationships among the interacting dynamic variables, and subsequently use the model structure to predict what would likely be observed under a variety of experimental conditions. The basic workflow cycle of observation, feature extraction, model building, parameter estimation, and prediction using the model lies at the core of systems biology. The same fundamental principles (with perhaps different specific techniques and tools) apply to systems physiology as well. A key difference is that systems biology, as the term is used now, requires information at the molecular and cellular levels and as such requires the development of models that transcend multiple levels of spatial and temporal scales – what is commonly referred to as "multiscale modeling." However, our focus, in this book, is on the application of model building and control theory to physiological systems at the organ systems level. Nevertheless, we believe that the principles and techniques presented here provide a useful foundation for the reader who is interested in pursuing a more comprehensive grasp of systems biology. There are a number of textbooks and review papers that focus on system biology: for example, Kitano (2002), Ideker et al. (2006), Voit (2013) and Klipp et al. (2016). For less "textbookish" reading, one is referred to the elegantly written introduction to systems biology by Noble (2006).

PROBLEMS

Based on the verbal descriptions of the following physiological reflex systems, construct block diagrams to represent the major control mechanisms involved. Clearly identify the physiological correlates of the controller, the plant, and the feedback element, as well as the controlling, controlled, and feedback variables. Describe how negative (or positive) feedback is achieved in each case.

P1.1. The Bainbridge reflex is a cardiac reflex that aids in the matching of cardiac output (the flow rate at which blood is pumped out of the heart) to venous return (the flow rate at which blood returns to the heart). Suppose there is a

transient increase in the amount of venous blood returning to the right atrium. This increases blood pressure in the right atrium, stimulating the atrial stretch receptors. As a result, neural traffic in the vagal afferents to the medulla is increased. This, in turn, leads to an increase in efferent activity in the cardiac sympathetic nerves as well as a parallel decrease in efferent parasympathetic activity. Consequently, both heart rate and cardiac contractility are increased, raising cardiac output. In this way, the reflex acts like a servomechanism, adjusting cardiac output to track venous return.

P1.2. The pupillary light reflex is another classic example of a negative feedback control system. In response to a decrease in light intensity, receptors in the retina transmit neural impulses at a higher rate to the pretectal nuclei in the midbrain, and subsequently to the Edinger–Westphal nuclei. From the Edinger–Westphal nuclei, a change in neural traffic down the efferent nerves back to the eyes leads to a relaxation of the sphincter muscles and contraction of the radial dilator muscles that together produce an increase in pupil area, which increases the total flux of light falling on the retina.

P1.3. The regulation of water balance in the body is intimately connected with the control of sodium excretion. One major mechanism of sodium reabsorption involves the renin–angiotensin–aldosterone system. Loss of water and sodium from the body, for example, due to diarrhea, leads to a drop in plasma volume, which lowers mean systemic blood pressure. This stimulates the venous and arterial baroreflexes that cause an increase in activity of the renal sympathetic nerves, which in turn stimulates the release of renin by the kidneys into the circulation. The increase in plasma renin concentration leads to an increase in plasma angiotensin, which stimulates the release of aldosterone by the adrenal cortex. Subsequently, the increased plasma aldosterone stimulates the reabsorption of sodium by the distal tubules in the kidneys, thereby increasing plasma sodium levels.

P1.4. The control system that regulates water balance is intimately coupled with the control of sodium excretion. When sodium is reabsorbed by the distal tubules of the kidneys, water will also be reabsorbed if the permeability of the tubular epithelium is lowered. This is achieved in the following way. When there is a drop in plasma volume, mean systemic pressure decreases, leading to a change in stimulation of the left atrial pressure receptors. The latter send signals to a group of neurons in the hypothalamus, increasing its production of vasopressin or antidiuretic hormone (ADH). As a result, the ADH concentration in blood plasma increases, which leads to an increase in water permeability of the kidney distal tubules and collecting ducts.

P1.5. Arterial blood pressure is regulated by means of the baroreceptor reflex. Suppose arterial blood pressure falls. This reduces the stimulation of the baroreceptors located in the aortic arch and the carotid sinus, which lowers

the rate at which neural impulses are sent along the glossopharyngeal and vagal afferents to the autonomic centers in the medulla. Consequently, sympathetic neural outflow is increased, leading to an increase in heart rate and cardiac contractility, as well as vasoconstriction of the peripheral vascular system. At the same time, a decreased parasympathetic outflow aids in the heart rate increase. These factors together act to raise arterial pressure.

P1.6. A prolonged reduction in blood pressure due to massive loss of blood can lead to "hemorrhagic shock" in which the decreased blood volume lowers mean systemic pressure, venous return, and thus cardiac output. Consequently, arterial blood pressure is also decreased, leading to decreased coronary blood flow, reduction in myocardial oxygenation, loss in the pumping ability of the heart, and therefore further reduction in cardiac output. The decreased cardiac output also leads to decreased oxygenation of the peripheral tissues, which can increase capillary permeability, thereby allowing fluid to be lost from the blood to the extravascular spaces. This produces further loss of blood volume and mean systemic pressure, and therefore, further reduction in cardiac output.

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2

MATHEMATICAL MODELING

2.1 GENERALIZED SYSTEM PROPERTIES

In this chapter, we will review the basic concepts and methods employed in the development of "gray box" models. Models of very different systems often contain properties that can be characterized using the same mathematical expression. The first of these is the *resistive* property. Everyone is familiar with the concept of electrical resistance (R), which is defined by Ohm's law as

$$V = RI \tag{2.1}$$

where V is the voltage or driving potential across the resistor and I represents the current that flows through it. Note that V is an "across"-variable and may be viewed as a measure of "effort." On the other hand, I is a "through"-variable and represents a measure of "flow." Thus, if we define the generalized "effort" variable ψ and the generalized "flow variable, ζ, Ohm's law becomes

$$\psi = R\zeta \tag{2.2}$$

where R now represents a generalized resistance. Figure 2.1 shows the application of this concept of generalized resistance to different kinds of systems. In the mechanical dashpot, when a force F is applied to the plunger (and, of course, an

(a)

$$F = v\, R_m$$

(b)

$$\Delta P = Q\, R_f$$

$$\Delta P = P_1 - P_2$$

(c)

θ_1 θ_2

Q

$$\Delta\theta = Q\, R_t$$

$$\Delta\theta = \theta_1 - \theta_2$$

(d)

ϕ_1 ϕ_2

Q

$$\Delta\phi = Q\, R_c$$

$$\Delta\phi = \phi_1 - \phi_2$$
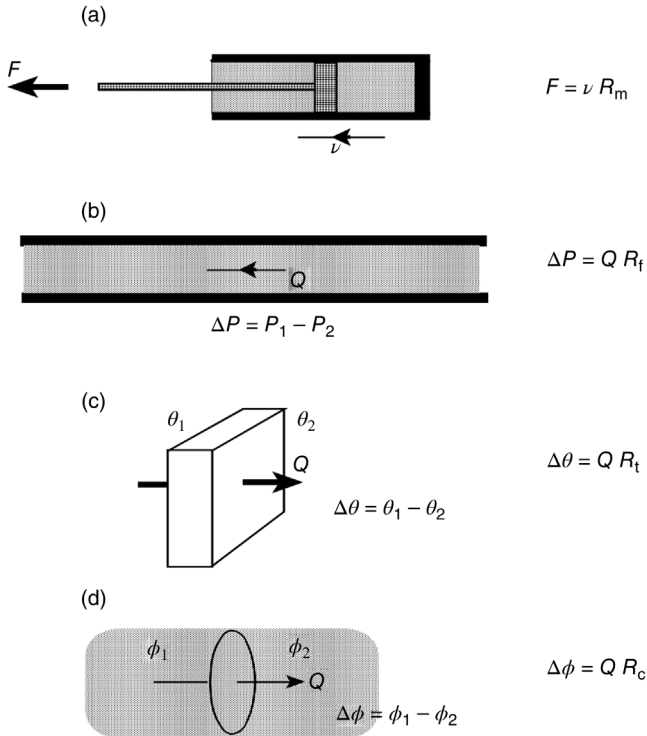
FIGURE 2.1 "Resistance" in (a) mechanical, (b) fluidic, (c) thermal, and (d) chemical systems.

equal and opposite force is applied to the dashpot casing), it will move with a velocity v that is proportional to F. As illustrated in Figure 2.1a, this relationship takes on the same form as the generalized Ohm's law (Equation 2.2), when F and v are made to correspond to ψ and ζ, respectively. The constant of proportionality, R_m, which is related to the viscosity of the fluid inside the dashpot, provides a measure of "mechanical resistance." In fact, R_m determines the performance of the dashpot as a shock absorber and is more commonly known as the "damping coefficient." In fluid flow, the generalized Ohm's law assumes the form of Poiseuille's law, which states that the volumetric flow of fluid (Q) through a rigid tube is proportional to the pressure difference (ΔP) across the two ends of the tube. This is illustrated in Figure 2.1b. Poiseuille further showed that the fluid resistance R_f is directly related to the viscosity of the fluid and the length of the tube, and inversely proportional to the square of the tube cross-sectional area. In Fourier's law of thermal transfer, the flow of heat conducted through a given material is directly proportional to the temperature difference that exists across the material (Figure 2.1c). Thermal resistance R_t can be shown to be inversely related to the thermal conductivity of the material. Finally, in chemical systems, the flux Q of a

given chemical species across a permeable membrane separating two fluids with different species concentrations is proportional to the concentration difference $\Delta\varphi$ (Figure 2.1d). This is known as Fick's law of diffusion. The diffusion resistance R_c is inversely proportional to the more commonly used parameter, the membrane diffusivity.

The second generalized system property is that of *storage*. In electrical systems, this takes the form of capacitance, defined as the amount of electrical charge (q) stored in the capacitor per unit voltage (V) that exists across the capacitor:

$$C = \frac{q}{V} \tag{2.3}$$

Note that q represents the accumulation of all electric charge delivered via current flow to the capacitor, so the following relationship exists between q and I:

$$q = \int_0^t I\,dt \tag{2.4}$$

Thus, using Equation 2.4 in Equation 2.3 and rewriting the result in generalized form, we obtain the following expression:

$$\psi = \frac{1}{C}\int_0^t \zeta\,dt \tag{2.5}$$

For mechanical systems, this storage property takes the form of "compliance," as in the case of the elastic spring shown in Figure 2.2a. For a given applied force, the mechanical compliance determines the extent to which the spring will be extended or compressed. This property is also inversely related to the stiffness or elastic modulus of the spring: The more compliant the spring, the less stiff it would be, and the more it would extend for a given applied tension. Similarly, in fluidic systems, as represented in the example of the fluid-filled balloon (Figure 2.2b), compliance determines the volume by which the balloon will expand or contract per unit change in applied pressure. The compliance here is determined primarily by the elasticity of the balloon material: The stiffer the material, the less "compliant" the balloon. A much smaller contribution to the compliance arises from the compressibility of the fluid inside the balloon. In thermal systems, the "thermal mass" represents the amount of heat stored in a certain medium per unit difference in temperature that exists between the interior and exterior (Figure 2.2c). This thermal capacitance depends on the dimensions and specific heat of the medium in question. Finally, in chemical systems, the storage property is represented by the total volume of the fluid in which the chemical species exists (Figure 2.2d), that is, for given volume, the total mass of the chemical species

(a)

$x = \int \nu \, dt$

$x = F \, C_m$

(b)

ΔV (volume change)

ΔP

$\Delta V = \Delta P \, C_f$

(c)

θ_2

θ_1

Heat stored, $q = \Delta\theta \, C_t$

$\Delta\theta = \theta_1 - \theta_2$

(d)

Mass of
chemical $q = \phi \, C_c$
species
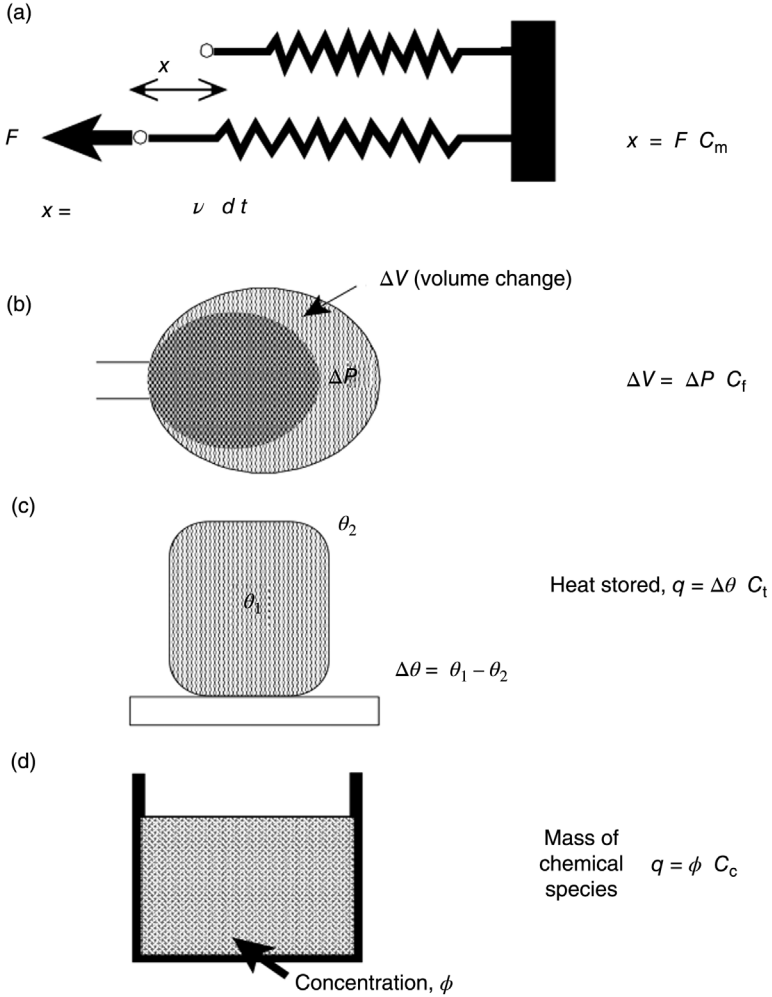
Concentration, ϕ

FIGURE 2.2   "Capacitance" in (a) mechanical, (b) fluidic, (c) thermal, and (d) chemical systems.

present is proportional to its concentration. This property, in fact, is used in the *definition* of concentration.

The resistive and storage properties also represent elements through which energy is dissipated or stored, respectively. In the context of electrical resistance, note that the product of voltage and current yields power. In mechanical systems, force multiplied by velocity also yields power. Similarly, in fluidic systems, power is defined as the product of pressure and flow rate. Thus, in any general system, energy is *dissipated* when effort is applied to produce flow through the resistive element. On the other hand, the storage element allows the accumulation of "static"

or *potential energy*. For instance, in the mechanical spring, application of force F that produces an extension of x length units will lead to the storage of $F \cdot x$ units of potential energy. Similarly, in the balloon system, potential energy is stored when the balloon is expanded with the application of internal pressure.

The final generalized system property, *inertance*, allows the storage of *kinetic energy* in electrical systems. This property is also known as *inductance* (L), which is defined as the voltage required to produce a given rate of change of electrical current:

$$V = L\frac{dI}{dt} \tag{2.6}$$

Replacing V and I by the corresponding generalized system variables ψ and ζ, we have

$$\psi = L\frac{d\zeta}{dt} \tag{2.7}$$

Note that in the context of mechanical systems, ψ becomes F (force) and ζ becomes velocity; so Equation 2.7 becomes Newton's second law of motion: Force equals mass times acceleration. Thus, in this case, the inertance is simply the mass of the system. Inertance is also present in fluidic systems; so fluid acceleration is proportional to the pressure differential applied. On the other hand, there is no element that represents inertance in thermal and chemical systems; kinetic energy storage does not exist in these systems.

2.2   MODELS WITH COMBINATIONS OF SYSTEM ELEMENTS

An assumption implicit in the previous section is that the system properties are time-invariant and independent of the values of the generalized system variables. In other words, the three basic model elements are *linear*. In reality, this will not be the case. An electrical resistor will heat up as the current that passes through it increases, thereby raising its resistance. Similarly, fluid resistance remains relatively constant only under conditions of steady, laminar flow; as flow increases and becomes turbulent, the resistance becomes a function of flow itself. Thus, as the effect of nonlinearities increases, the similarities in behavior among these different systems will be progressively reduced. Nevertheless, in scientific exploration, we always have to start somewhere – and past scientific history has shown that it is wise to begin with the simplest model. Therefore, linear analysis plays an important role in physiological systems modeling by allowing us to obtain a first approximation to the underlying reality.

We will proceed with our discussion of linear analysis by considering how we can derive the overall model equations from various combinations of the three basic
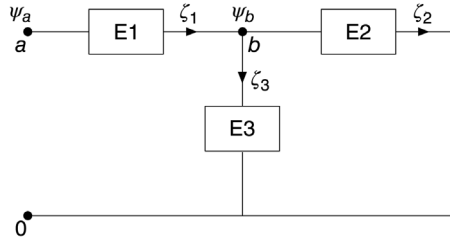
FIGURE 2.3   Simple model consisting of a network of generalized system elements.

types of system elements. Figure 2.3 shows a simple "circuit" linking three generalized system elements in series and parallel. The node labeled "0" represents the reference level of the "effort" variable ψ with which all other nodes are compared; this is set equal to zero. In electrical systems, this node would be called "electrical ground." ψ_a and ψ_b represent the values of the *across-variables* at nodes a and b, respectively, relative to node 0. ζ_1, ζ_2, and ζ_3 represent the values of the *through-variables* that pass through the elements E_1, E_2, and E_3, respectively. The mathematical relationships among these variables can be derived by applying two fundamental physical principles:

(1) The algebraic sum of the "across-variable" values around any closed circuit must equal zero. Thus, in the loop a–b–0–a in Figure 2.3, we have

$$(\psi_a - \psi_b) + (\psi_b - 0) + (0 - \psi_a) = 0 \tag{2.8}$$

(2) The algebraic sum of all "through-variable" values into a given node must equal zero. In Figure 2.3, the only node where this rule will apply is node b:

$$\zeta_1 + (-\zeta_2) + (-\zeta_3) = 0 \tag{2.9}$$

The preceding two generalized principles take the form of Kirchoff's laws when applied to electrical systems. The first (voltage) law appears trivial in the form presented, but in order to apply this law, each of the component terms on the left-hand side of Equation 2.8 has to be expressed as a function of the corresponding through-variables and the system elements. The second (current) law is essentially a statement of the conservation of mass principle. In Equation 2.9, ζ_2 and ζ_3 are given negative signs since, in Figure 2.3, they assume a flow direction that points away from node b. Although each of the boxes labeled E1, E2, and E3 was intended to represent one of the three basic system elements, each in general could also contain a network within itself; within each network, the above two laws would still apply. Thus, starting from the basic system elements, we can construct progressively more complex models by connecting these elements in either series or parallel configurations. And by using the generalized Kirchoff's laws, it is possible to deduce

(a)

$$R = \left(\frac{1}{R_1} + \frac{1}{R_2}\right)^{-1}$$

(c)

$$C = C_1 + C_2$$

(b)

$$R = R_1 + R_2$$

(d)

$$C = \left(\frac{1}{C_1} + \frac{1}{C_2}\right)^{-1}$$
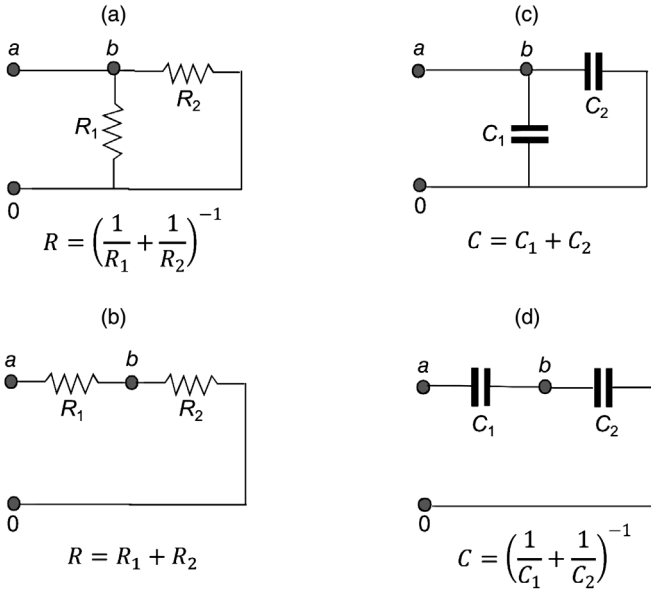
FIGURE 2.4   Model properties that emerge from simple networks of system elements in electrical, fluidic, thermal, and chemical systems.

mathematical expressions that characterize the overall properties of these synthesized networks. Figure 2.4 illustrates the resultant system properties that emerge from simple combinations of resistive and storage elements for electrical, fluidic, thermal, and chemical systems. These results are easily derived using basic circuit analysis.

The expressions that relate combined resistances and compliances to their respective component elements are somewhat different for mechanical systems, as Figure 2.5 shows. Consider Figure 2.5a, where the two mechanical dashpots (resistances) are placed in parallel. This parallel configuration constrains the motion of the two dashpot plungers by requiring them to move at the same velocity. Therefore, the total force F required to extend the two dashpots with resistances R_{m1} and R_{m2} at velocity v is

$$F = v(R_{m1} + R_{m2}) \tag{2.10}$$

But the combined resistance R is defined as F/v. Thus, Equation 2.10 yields

$$R = R_{m1} + R_{m2} \tag{2.11}$$

This relationship for the two mechanical resistances in parallel is different from the corresponding expression for parallel combinations of the other types of resistances.

(a)

R_{m1}

R_{m2}

$$R = R_{m1} + R_{m2}$$

(b)

R_{m1} R_{m2}

$$R = \left(\frac{1}{R_{m1}} + \frac{1}{R_{m2}} \right)^{-1}$$

(c)

C_{m1}

C_{m2}

$$C = \left(\frac{1}{C_{m1}} + \frac{1}{C_{m2}} \right)^{-1}$$

(d)

C_{m1} C_{m2}

$$C = C_{m1} + C_{m2}$$
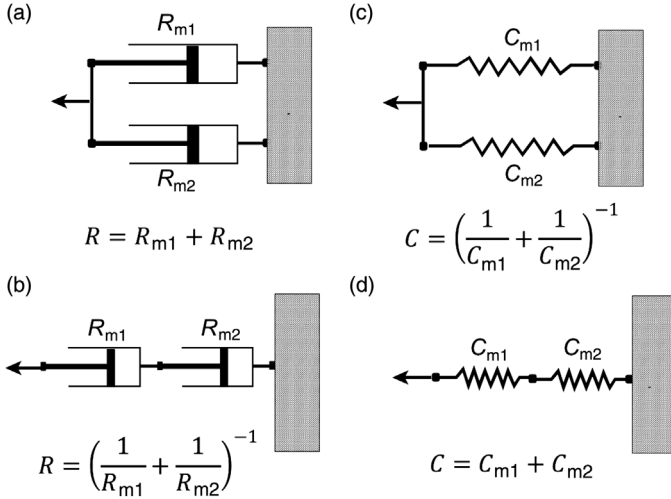
FIGURE 2.5   Models of parallel and series combinations of mechanical dashpots (resistances) and springs (compliances).

Figure 2.5c represents mechanical springs with compliances C_{m1} and C_{m2} connected in parallel. As in Figure 2.5a, the parallel mechanical arrangement constrains the springs so that they must extend by equal amounts ($= x$). Thus,

$$x = F_1 C_{m1} = F_2 C_{m2} \tag{2.12}$$

where F_1 and F_2 represent the corresponding tensions developed in the two springs. But the sum of F_1 and F_2 yields the total force F required to extend the spring combination by x, and since $C = x/F$ by definition, Equation 2.12 leads to

$$\frac{1}{C} = \frac{1}{C_{m1}} + \frac{1}{C_{m2}} \tag{2.13}$$

which again differs from the corresponding situation for capacitances placed in parallel in other systems. Similar considerations apply to series combinations of mechanical dashpots and springs (Figure 2.5b and d). As such, one has to be cautious in converting models of mechanical systems into their electrical analogs.

2.3   LINEAR MODELS OF PHYSIOLOGICAL SYSTEMS: TWO EXAMPLES

In this section, we will derive the mathematical formulations that characterize the input–output properties of two simple physiological models. The first model provides a linearized description of lung mechanics (Figure 2.6). The airways
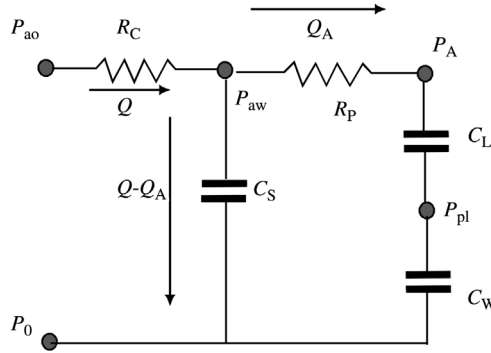
FIGURE 2.6 Linear model of respiratory mechanics.

are divided into two categories: the larger or central airways and the smaller or peripheral airways, with fluid mechanical resistances equal to R_C and R_P, respectively. Air that enters the alveoli also produce an expansion of the chest wall cavity by the same volume. This is represented by the connection of the lung (C_L) and chest wall (C_W) compliances in series. However, a small fraction of the volume of air that enters the respiratory system is shunted away from the alveoli as a result of the compliance of the central airways and gas compressibility. This shunted volume is very small under normal circumstances at regular breathing frequencies, but becomes progressively more substantial if disease leads to peripheral airway obstruction (i.e., increased R_P) or a stiffening of the lungs or chest wall (i.e., decreased C_L or C_W). We account for this effect by placing a shunt compliance C_S in parallel with C_L and C_W. The pressures developed at the different points of this lung model are P_{ao} at the airway opening, P_{aw} in the central airways, P_A in the alveoli, and P_{pl} in the pleural space (between the lung parenchyma and chest wall). These pressures are referenced to P_0, the ambient pressure, which we can set to zero. Suppose the volume flow rate of air entering the respiratory system is Q. Then, the objective here is to derive a mathematical relationship between P_{ao} and Q.

From Kirchoff's second law (applied to the node P_{aw}), if the flow delivered to the alveoli is Q_A, then the flow shunted away from the alveoli must be $Q - Q_A$. Applying Kirchoff's first law to the closed circuit containing C_S, R_P, C_L, and C_W, we have

$$R_P Q_A + \left(\frac{1}{C_L} + \frac{1}{C_W} \right) \int Q_A \, dt = \frac{1}{C_S} \int (Q - Q_A) \, dt \qquad (2.14)$$

Applying Kirchoff's first law to the circuit containing R_C and C_S, we have

$$P_{ao} = R_C Q + \frac{1}{C_S} \int (Q - Q_A) \, dt \qquad (2.15)$$

Differentiating Equation 2.14 and Equation 2.15 with respect to time, and subsequently, reducing the two equations to one by eliminating Q_A, we obtain the equation relating P_{ao} to Q:

$$\frac{d^2 P_{ao}}{dt^2} + \frac{1}{R_P C_T}\frac{dP_{ao}}{dt} = R_C \frac{d^2 Q}{dt^2} + \left(\frac{1}{C_S} + \frac{R_C}{R_P C_T}\right)\frac{dQ}{dt} + \frac{1}{R_P C_S}\left(\frac{1}{C_L} + \frac{1}{C_W}\right)Q$$

(2.16)

where C_T is defined by

$$C_T = \left(\frac{1}{C_L} + \frac{1}{C_W} + \frac{1}{C_S}\right)^{-1}$$

(2.17)

The second example that we will consider is the linearized physiological model of skeletal muscle, as illustrated in Figure 2.7. F_0 represents the force developed by the active contractile element of the muscle, while F is the actual force that results after taking into account the mechanical properties of muscle. R represents the viscous damping inherent in the tissue, while C_P (parallel elastic element) and C_S (series elastic element) reflect the elastic storage properties of the sacrolemma and the muscle tendons, respectively. First, consider the mechanical constraints placed on the model components as a result of the parallel configuration. If spring C_P is stretched by an incremental length x, the entire series combination of R and C_S will also extend by the same length. Furthermore, the sum of the force transmitted through the two branches of the parallel configuration must equal F. Although the sum of the extensions of C_S and R will have to equal x, the individual length contributions from C_S and R need not be equal. Thus, if we assume C_S is stretched a length x_1, then the extension in the parallel combination of R and F_0 will be $x - x_1$. The velocity with which the dashpot represented by R is extending is obtained by differentiating $x - x_1$ with respect to time, that is, $d(x - x_1)/dt$.
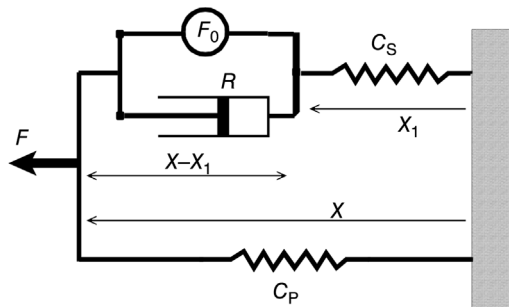
FIGURE 2.7   Linear model of muscle mechanics.

Using the principle that the force transmitted through C_S must be equal to the force transmitted through the parallel combination of F_0 and R, we obtain the following equation:

$$\frac{x_1}{C_S} = R\left(\frac{dx}{dt} - \frac{dx_1}{dt}\right) + F_0 \tag{2.18}$$

Then, using the second principle, that is, the total force from both limbs of the parallel combination must sum to F, we have

$$F = \frac{x_1}{C_S} + \frac{x}{C_P} \tag{2.19}$$

Eliminating x_1 from Equation 2.18 and Equation 2.19 yields the following differential equation relating F to x and F_0:

$$\frac{dF}{dt} + \frac{1}{RC_S}F = \left(\frac{1}{C_S} + \frac{1}{C_P}\right)\frac{dx}{dt} + \frac{1}{RC_S C_P}x + \frac{F_0}{RC_S} \tag{2.20}$$

Note that in Equation 2.20, under steady-state, isometric conditions (i.e., the muscle length is constrained to be constant), $x = 0$, $dx/dt = 0$, and $dF/dt = 0$, which lead to the result $F = F_0$. Therefore, under steady-state isometric conditions, the force developed by the muscle model will reflect the force developed by the active contractile element of the muscle.

2.4   CONVERSIONS BETWEEN ELECTRICAL AND MECHANICAL ANALOGS

In the previous section, we developed the ordinary differential equations that characterize two dynamic linear systems, one representing the pressure–airflow dynamics of the respiratory system and the other representing mechanical behavior of muscle. Along with the differential equations, we introduced the electrical analog of the respiratory mechanics model and the mechanical analog of the muscle model. As alluded to earlier in this chapter, we could just as easily have represented the respiratory mechanics model as a mechanical analog consisting of springs and dashpots, instead of an electrical analog with capacitors and resistor, or the muscle model as an electrical circuit. All analogs of each of these models would still be characterized by the same underlying ordinary differential equations derived in Section 2.3. Here, we will demonstrate how one can convert an electrical analog into an equivalent mechanical analog, and vice versa, using the two examples considered in the previous section. Note that once a given analog of a model has been correctly constructed, we do not need to go back to "first principles" to construct a different analog of the same model. We only need to keep track of how the various
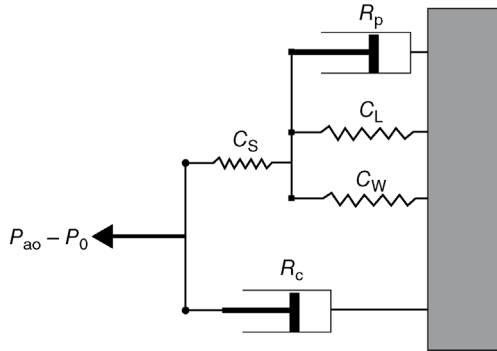
FIGURE 2.8 Mechanical analog of electrical circuit model of respiratory mechanics in Figure 2.6.

"elements" of the model are related to one another through the generalized Kirchoff's laws.

Figure 2.8 displays the mechanical analog of the respiratory model that was originally characterized as an electrical circuit. First, the overall "voltage" applied to the electrical circuit (Figure 2.6) is $P_{ao} - P_0$, and therefore this translates into the overall force applied to the equivalent mechanical analog (Figure 2.8). The elements R_P, C_L, and C_W are in series electrically, meaning that they share the same "current" that flows through them. Mechanically, this translates to their equivalent dashpot and springs being connected together in parallel (i.e., mechanically, these three elements are constrained to be stretched or compressed by the same displacement). On the other hand, C_S is electrically in parallel with the R_P–C_L–C_W series combination, meaning that the two "branches" share the same potential difference. This translates mechanically to their sharing the same force – hence, mechanically, C_s must be in series with the R_P–C_L–C_W parallel combination. Finally, in the electrical analog, R_C is in series with the rest of the circuit, meaning that it shares the same current. Thus, in the mechanical equivalent, R_C should share the same displacement as the C_S–R_P–C_L–C_W combination. Note that in the mechanical analog, $P_A - P_0$ is given by the sum of the forces transmitted through C_L and C_W.

In this next example, we convert the mechanical analog of the muscle model to an electrical analog. In the muscle model (Figure 2.7), the element C_P is mechanically in parallel with the rest of the elements, meaning that it is constrained to share the same displacement or velocity. Thus, electrically, C_P should be in series with the rest of the circuit, since the current that flows through C_P will flow through the rest of the circuit. In the mechanical analog, C_S is in series with the parallel F_0–R combination, and thus the same force is transmitted through C_S and the F_0–R combination. This means that electrically, C_S must be in parallel with the F_0–R combination. In Figure 2.7, the force generator F_0 represents the contractile element of muscle – thus, it acts against the tensile force created
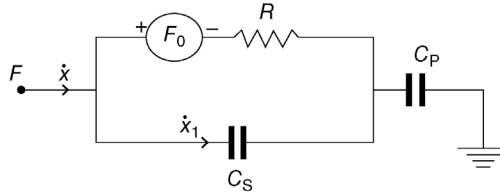
FIGURE 2.9   Electrical analog of linear muscle mechanics model in Figure 2.7.

when the external F is applied to the mechanical system. In the electrical analog, F_0 takes the form of a voltage generator whose positive voltage terminal opposes the voltage resulting from the applied external F. As such, the potential difference across the resistor R is reduced by F_0. The complete electrical analog of the muscle model is displayed in Figure 2.9.

2.5   DISTRIBUTED-PARAMETER VERSUS LUMPED-PARAMETER MODELS

The models that we have considered up to this point are known as *lumped-parameter models*. A given property of the model is assumed to be "concentrated" into a single element. For example, in the lung mechanics model (Figure 2.6), the total resistance of the central airways is "lumped" into a single quantity, R_C, even though in reality the central airways are comprised of the trachea and a few branching generations of airways, each of which has very different fluid mechanical resistance. Similarly, a single constant C_L is assumed to represent the compliance of the lungs, even though the elasticity of lung tissue varies from region to region. In order to provide a more realistic characterization of the *spatial* distribution of system properties, it is often useful to develop a *distributed-parameter model*. This kind of model generally takes the form of one or more partial differential equations with time and some measure of space (e.g., length or volume) as independent variables.

A distributed-parameter model can be viewed as a network of many infinitesimally small lumped-parameter submodels. To illustrate this relationship, we will derive the governing differential equation of a distributed-parameter model of the passive cable characteristics of an unmyelinated nerve fiber. As shown in Figure 2.10, the nerve fiber is modeled as a network containing serially connected multiple subunits, each with circuit elements r_x, r_M, and c_M. r_x represents the axial resistance of 1 cm of nerve tissue per cm^2 of cross-sectional area, and is given in Ω-cm. r_M and c_M represent the resistance and capacitance of 1 cm^2 of nerve membrane surface area, respectively. We assume that the extracellular medium bathing the nerve fiber represents the electrical ground in this model. How do we relate the current passing through to the voltage found at any point in this "cable?"
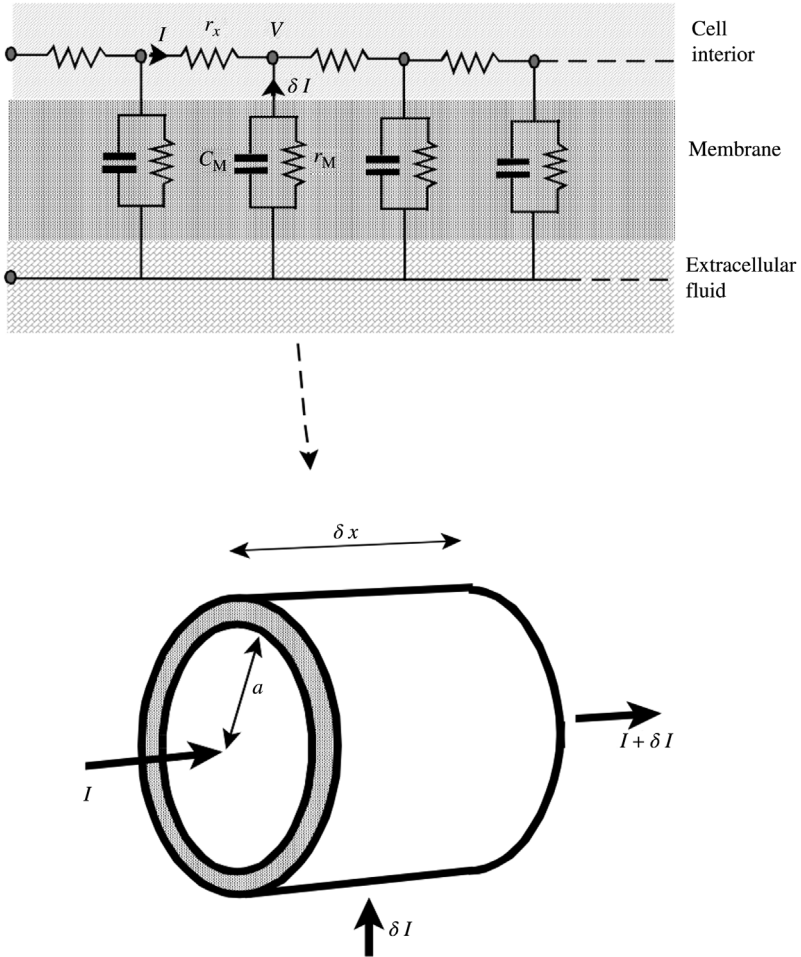
FIGURE 2.10 Relationship between the lumped-parameter and distributed-parameter models of the passive cable characteristics of an unmyelinated nerve fiber.

In the distributed-parameter model, the nerve fiber takes the form of a long cylindrical conductor of radius a. We focus our analysis on a small length δx of this cable that contains one of these resistance–capacitance subunits. We assume the intracellular voltage to increase by δV and the axial current to increase by δI over this small segment of cable. The assumption of "increases" instead of "decreases" in V and I merely establishes a sign convention. In reality, the voltage along the nerve fiber drops as current flows out of the nerve fiber through a leaky membrane. If we adopt the stated sign convention in a consistent manner, the final results will show the change in V or I with length to be negative. The voltage increase δV occurring

over an axial distance of δx is related to the axial current by

$$\delta V = -\frac{Ir_x \delta x}{\pi a^2} \tag{2.21}$$

The right-hand-side of Equation 2.21 bears a negative sign since the "increase" in voltage should be associated with a current flowing in a direction *opposite* to that assumed in the figure. Dividing both sides of Equation 2.21 by δx, and taking the limit as δx is made to approach zero, we obtain

$$\frac{\partial V}{\partial x} = -\frac{Ir_x}{\pi a^2} \tag{2.22}$$

The membrane current δI is related to the intracellular voltage V by

$$\delta I = -\left(\frac{V}{r_M} \cdot 2\pi a \cdot \delta x + c_M \cdot 2\pi a \cdot \delta x \cdot \frac{\partial V}{\partial t} \right) \tag{2.23}$$

Again, dividing both sides of Equation 2.23 by δx and taking the limit as δx approaches zero, we obtain

$$\frac{\partial I}{\partial x} = -\left(\frac{V}{r_M} \cdot 2\pi a + c_M \cdot 2\pi a \cdot \frac{\partial V}{\partial t} \right) \tag{2.24}$$

Finally, Equations 2.22 and 2.24 can be combined into one equation by differentiating Equation 2.22 with respect to x and substituting for $\partial I/\partial x$ in Equation 2.24:

$$\frac{\partial^2 V}{\partial x^2} = \frac{2r_x}{a} \left(c_M \cdot \frac{\partial V}{\partial t} + \frac{V}{r_M} \right) \tag{2.25}$$

Equation 2.25 is known as the one-dimensional cable equation, and describes intracellular voltage along the nerve fiber as a continuous function of length and time.

2.6   LINEAR SYSTEMS AND THE SUPERPOSITION PRINCIPLE

All the models we have considered up to this point are *linear systems*. We have shown that these linear systems can be characterized by linear ordinary or partial differential equations. A differential equation is linear when all its terms that contain the output and input variables and its derivatives are of the first degree. For example, the model differential equations that we have derived are of the general form:

$$a_2 \frac{d^2 y}{dt^2} + a_1 \frac{dy}{dt} + y = b_2 \frac{d^2 x}{dt^2} + b_1 \frac{dx}{dt} + b_0 x \tag{2.26}$$

Since all terms in y and its derivatives, as well as all terms in x and its derivatives, are raised to only the first power (degree), Equation 2.26 is linear. If the coefficients a_1, a_2, b_0, b_1, and b_2 are constants, then Equation 2.26 is also *time-invariant*. On the other hand, if one or more of these coefficients are functions of time, for example, $a_1 = t$, Equation 2.26 remains linear but becomes *time-varying*. However, if one or more of the coefficients are functions of y or x, Equation 2.26 becomes *nonlinear*.

Let us suppose that the input x of the system described by Equation 2.26 takes on a particular time-course, for example, $x = x_1(t)$, and the resulting time-course for y is $y_1(t)$. Then, it follows that

$$a_2 \frac{d^2 y_1}{dt^2} + a_1 \frac{dy_1}{dt} + y_1 = b_2 \frac{d^2 x_1}{dt^2} + b_1 \frac{dx_1}{dt} + b_0 x_1 \qquad (2.27)$$

When x takes on a different time-course $x_2(t)$ and the resulting time-course for y is $y_2(t)$, the following relationship also holds:

$$a_2 \frac{d^2 y_2}{dt^2} + a_1 \frac{dy_2}{dt} + y_2 = b_2 \frac{d^2 x_2}{dt^2} + b_1 \frac{dx_2}{dt} + b_0 x_2 \qquad (2.28)$$

By adding terms on both sides of Equations 2.27 and 2.28 with the same coefficients, we have

$$a_2 \frac{d^2(y_1 + y_2)}{dt^2} + a_1 \frac{d(y_1 + y_2)}{dt} + (y_1 + y_2) = b_2 \frac{d^2(x_1 + x_2)}{dt^2}$$
$$+ b_1 \frac{d(x_1 + x_2)}{dt} + b_0(x_1 + x_2) \qquad (2.29)$$

What Equations 2.27 through 2.29 together imply is that the response of a linear system to the sum of two different inputs is equal to the sum of the responses of the system to the individual inputs. This result can be extended to more than two inputs and is known as the *principle of superposition*. It is a defining property of linear systems and is frequently used as a test to determine whether a given system is linear.

The principle of superposition also implies that the complete solution (i.e., response in y) to Equation 2.26 can be broken down into two components:

$$y(t) = y_c(t) + y_p(t) \qquad (2.30)$$

where $y_c(t)$ is known as the *complementary function* and $y_p(t)$ is called the *particular solution*. $y_c(t)$ is the response of the linear system in Equation 2.26 when the input forcing $x(t)$ is set equal to zero:

$$a_2 \frac{d^2 y_c}{dt^2} + a_1 \frac{dy_c}{dt} + y_c = 0 \qquad (2.31)$$

Thus, $y_c(t)$ reflects the component of $y(t)$ that remains the same independent of the form of input $x(t)$. However, the time-course of $y_c(t)$ depends on the initial values taken by y and its derivatives immediately prior to input stimulation. On the other hand, $y_p(t)$ is the response of the same linear system when the input forcing $x(t)$ is a particular function of time. Thus, $y_p(t)$ satisfies the differential equation:

$$a_2 \frac{d^2 y_p}{dt^2} + a_1 \frac{dy_p}{dt} + y_p = b_2 \frac{d^2 x}{dt^2} + b_1 \frac{dx}{dt} + b_0 x \qquad (2.32)$$

Unlike $y_c(t)$, $y_p(t)$ depends only on the coefficients of Equation 2.32 and the time-course of the input $x(t)$. If the linear system in question is *stable*, then the component of $y(t)$ characterized by $y_c(t)$ will eventually decay to zero while the overall response of the system will be increasingly dominated by $y_p(t)$. For this reason, $y_c(t)$ is also called the *transient response*; $y_p(t)$ is known as the *steady-state response* when $x(t)$ is an input forcing function that persists over time.

2.7 ZERO-INPUT AND ZERO-STATE SOLUTIONS OF ODEs

A parallel approach to solving linear ordinary differential equations, used more widely in the engineering community, is to consider the complete solution to be the sum of the *zero-input* and *zero-state* solutions. The zero-input solution is the response of the linear time-invariant system to nonzero initial conditions but no input. The zero-state solution is the response of the system to the external input but starting with zero initial conditions. Thus, if we again consider the system represented by Equation 2.26, we have the following expression for the complete solution for $y(t)$:

$$y(t) = y_{zi}(t) + y_{zs}(t) \qquad (2.33)$$

where $y_{zi}(t)$ is the zero-input solution and $y_{zs}(t)$ is the zero-state solution. Considering the zero-input solution first, we set $x(t)$ to zero in Equation 2.26 and thus we obtain

$$a_2 \frac{d^2 y_{zi}}{dt^2} + a_1 \frac{dy_{zi}}{dt} + y_{zi} = 0 \qquad (2.34)$$

The zero-input solution is obtained by solving Equation 2.34 assuming nonzero values for $y_{zi}(0)$ and $dy_{zi}/dt(0)$. Equation 2.34 is essentially the same as the homogeneous equation in Equation 2.31. However, a key difference is that in Equation 2.31, the initial conditions are specified at time $t = 0^+$, that is, immediately after the input (if any) has been applied. In Equation 2.34, the initial conditions are specified at time $t = 0^-$. Thus, the complementary function (solution to the

homogeneous equation) and the zero-input solution can, in general, be different from one another.

The zero-state solution is obtained by solving the following differential equation for the particular input $x(t)$ in question, assuming zero initial conditions:

$$a_2\frac{d^2y_{zs}}{dt^2} + a_1\frac{dy_{zs}}{dt} + y_{zs} = b_2\frac{d^2x}{dt^2} + b_1\frac{dx}{dt} + b_0x \tag{2.35}$$

Unlike the "steady-state response" or "particular solution" (Section 2.6), the zero-state solution will in general contain a transient component. However, the overall solution, obtained by adding together the zero-input and zero-state solutions, turns out to be exactly the same as what one would obtain using the more conventional way of solving the same ordinary differential equation for the complementary function and particular solution. Both Equations 2.34 and 2.35 are most conveniently solved using the Laplace transform, as demonstrated in the next section.

2.8   LAPLACE TRANSFORMS AND TRANSFER FUNCTIONS

The Laplace transformation, denoted by $\mathcal{L}[\cdot]$, and its inverse, denoted by $\mathcal{L}^{-1}[\cdot]$, are mathematical operations defined as follows:

$$\mathcal{L}[y(t)] = Y(s) = \int_0^\infty y(t)e^{-st}dt \tag{2.36a}$$

$$y(t) = \mathcal{L}^{-1}[Y(s)] = \frac{1}{2\pi j}\int_{\sigma-j\infty}^{\sigma+j\infty} Y(s)e^{st}ds \tag{2.36b}$$

where the Laplace variable s is complex, that is, $s = \sigma + j\omega$, and $j = \sqrt{-1}$. By employing the mathematical operation defined in Equation 2.36a, the function of time $y(t)$ is converted into an equivalent function $Y(s)$ in the complex s-domain. If we apply this mathematical operation to the time-derivative of y, dy/dt, we can evaluate the result by performing an integration by parts procedure:

$$\mathcal{L}\left[\frac{dy(t)}{dt}\right] = \int_0^\infty \frac{dy}{dt}e^{-st}dt = s\int_0^\infty y(t)e^{-st}dt + \left[y(t)e^{-st}\right]_0^\infty \tag{2.37a}$$

Thus,

$$\mathcal{L}\left[\frac{dy}{dt}\right] = sY(s) - y(0) \tag{2.37b}$$

By applying the same transformation to d^2y/dt^2, it can be shown that

$$\mathcal{L}\left[\frac{d^2y}{dt^2}\right] = s^2Y(s) - sy(0) - \left(\frac{dy}{dt}\right)_{t=0} \tag{2.38}$$

Conversely, the Laplace transform of the time-integral of $y(t)$, assuming $y(t) = 0$ for $t < 0$, is

$$\mathcal{L}\left[\int_0^t y(t)dt\right] = \frac{Y(s)}{s} \tag{2.39}$$

Laplace transforms of various "standard" functions in time have been evaluated and are available in the form of tables, such as that presented in Appendix A. These same tables are also used to convert Laplace expressions back into time-domain functions.

If we apply the Laplace transformation to the linear system represented by Equation 2.26, and use the above expressions in Equations 2.37b and 2.38 to evaluate the transforms of the derivatives, we will obtain the following result:

$$a_2s^2Y(s) + a_1sY(s) + Y(s) = b_2s^2X(s) + b_1sX(s) + b_0X(s) \tag{2.40a}$$

Equation 2.40a assumes that the values of x and y and their first time-derivatives are all equal to zero at time $t = 0$. Even if any one of these initial values are actually nonzero, they do not affect the functional nature of the dynamics of the linear system being characterized. Equation 2.40a can be rearranged and presented in the following form:

$$\frac{Y(s)}{X(s)} = \frac{b_2s^2 + b_1s + b_0}{a_2s^2 + a_1s + 1} \tag{2.40b}$$

Equation 2.40b describes in very compact format how the input to the linear system in question is transformed into its output. Starting with the time-domain representation of the input $x(t)$, we first determine its Laplace transform $X(s)$. Then, multiplying $X(s)$ by the function displayed on the right-hand-side of Equation 2.40b, we obtain the Laplace transform $Y(s)$ of the system response. Finally, by determining the inverse Laplace transform of $Y(s)$, we recover the time-course of the response $y(t)$. Of course, we could have derived $y(t)$ by solving the differential equation in Equation 2.26 in the time domain. However, application of the Laplace transformation converts the differential equation into an algebraic equation, which is generally easier to solve. The ratio of $Y(s)$ to $X(s)$ in Equation 2.40b is called the *transfer function* of the system in question. Employing this approach allows the convenient representation of the input–output characteristics
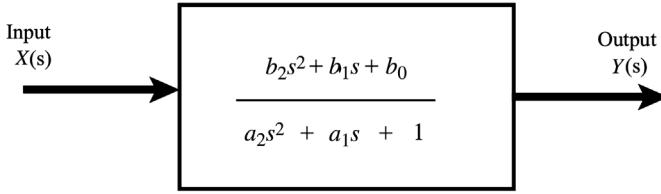
FIGURE 2.11   Transfer function representation of the linear system described in Equation 2.6.

of any linear system in block diagram form, as illustrated in Figure 2.11 for the example that we have been discussing.

2.8.1   Solving ODEs with Laplace Transforms

We consider below an example that highlights the utility of the Laplace transform for solving ordinary differential equations. Assume, in Equation 2.26, the following values for its coefficients: $b_2 = 0$, $b_1 = 0.5$, $b_0 = 1.25$, $a_2 = 0.25$, and $a_1 = 1.25$. Also, assume the initial conditions are $y(0) = 1$ and $\dot{y}(0) = 1$. If the input were to take the form of a unit step (i.e., $x(t) = 1$ for $t > 0$ and $x(t) = 0$ for $t < 0$), what would be the response $y(t)$?

To determine the solution to Equation 2.26, we employ the approach outlined in Section 2.7, which applies the principle of superposition and considers the complete solution $y(t)$ to be the sum of the zero-input and zero-state solutions. First, we consider the zero-input solution, which is the solution to the case where the input is assumed to be absent, that is,

$$0.25\frac{d^2 y_{zi}}{dt^2} + 1.25\frac{dy_{zi}}{dt} + y_{zi} = 0 \tag{2.41}$$

but subject to the initial conditions: $y_{zi}(0) = 1$ and $\dot{y}_{zi}(0) = 1$. Taking the Laplace transform of both sides of Equation 2.41, we get

$$0.25s^2 Y_{zi}(s) - 0.25s - 0.25 + 1.25s Y_{zi}(s) - 1.25 + Y_{zi}(s) = 0 \tag{2.42a}$$

Rearranging and simplifying terms in Equation 2.42a, we obtain the following result:

$$Y_{zi}(s) = \frac{s+6}{s^2 + 5s + 4} = \frac{s+6}{(s+1)(s+4)} \tag{2.42b}$$

In order to simplify the process for determining the inverse Laplace transform of $Y_{zi}(s)$, we express the right-hand-side of Equation 2.42b in terms of partial fractions:

$$Y_{zi}(s) = \frac{5/3}{s+1} + \frac{-2/3}{s+4} \tag{2.43}$$

By looking up the Laplace transform table in Appendix A, we can solve for $y_{zi}(t)$ in the time domain. By doing this, we obtain the zero-input solution of $y_{zi}(t)$ as follows:

$$y_{zi}(t) = \left(\frac{5}{3}e^{-t} - \frac{2}{3}e^{-4t}\right)u(t) \tag{2.44}$$

where $u(t)$ is the unit step (or Heaviside) function, so that $u(t) = 1$ for $t > 0$ and $u(t) = 0$ for $t < 0$. The presence of $u(t)$ in Equation 2.44 underscores the point that $y_{zi}(t)$ is a causal response, that is, $y_{zi}(t)$ is zero prior to the imposition of the (nonzero) initial conditions.

Next, we turn to determining the zero-state solution. This is for the case where initial conditions are assumed to be zero but the external input is taken into account. Thus, inserting the values of the coefficients into Equation 2.35, we have

$$0.25\frac{d^2 y_{zs}}{dt^2} + 1.25\frac{dy_{zs}}{dt} + y_{zs} = 0.5\frac{dx}{dt} + 1.25x \tag{2.45}$$

Applying the Laplace transformation to both sides of Equation 2.45 and multiplying throughout by 4, we obtain

$$(s^2 + 5s + 4)Y_{zs}(s) = (2s + 5)X(s) \tag{2.46}$$

Since we are assuming $x(t)$ to be the unit step input ($= u(t)$), from Appendix A, we find that $X(s) = 1/s$. Thus, after rearranging terms, Equation 2.46 becomes

$$Y_{zs}(s) = \frac{2s + 5}{s(s + 1)(s + 4)} \tag{2.47}$$

Breaking Equation 2.47 into partial fractions, we have

$$Y_{zs}(s) = \frac{1.25}{s} - \frac{1}{s + 1} + \frac{0.25}{s + 4} \tag{2.48}$$

We obtain the zero-state solution $y_{zs}(t)$ by applying the inverse Laplace transformation to the partial fractions on the right-hand side of Equation 2.48 (Appendix A):

$$y_{zs}(t) = \left(1.25 - e^{-t} - 0.25e^{-4t}\right)u(t) \tag{2.49}$$

Finally, we add the zero-input and zero-state solutions to arrive at the complete solution to Equation 2.26:

$$y(t) = \left(1.25 + \frac{2}{3}e^{-t} - \frac{11}{12}e^{-4t}\right)u(t) \tag{2.50}$$

More details on the methodological aspects of solving differential equations via Laplace transforms can be found in any standard text on applied mathematics or linear systems theory, for example, Lathi (2004) and Mitra (2015).

2.9   THE IMPULSE RESPONSE AND LINEAR CONVOLUTION

Suppose we have a linear system with unknown transfer function $H(s)$. By definition, $Y(s) = H(s) X(s)$. So, if we can find an input $x(t)$ that has the Laplace transform $X(s) = 1$, then $Y(s) = H(s)$, that is, the Laplace transform of the resulting response would reveal the transfer function of the unknown system. Appendix A shows that the time-function that corresponds to $X(s) = 1$ is $\delta(t)$, the Dirac delta function or the *unit impulse*. The unit impulse may be considered a rectangular pulse of infinite amplitude but infinitesimal duration. Consider the function $p(t)$ defined such that $p(t) = a$ for $0 \le t \le 1/a$, and $p(t) = 0$ for $t < 0$ and $t > 1/a$, where a is positive constant. Applying the Laplace transformation to $p(t)$ yields

$$\int_0^\infty p(t)e^{-st}dt = \int_0^{1/a} ae^{-st}dt = \frac{1 - e^{-s/a}}{s/a} \tag{2.51}$$

Expanding the exponential term as an infinite series, we obtain

$$\frac{1 - e^{-s/a}}{s/a} = \frac{1 - \left\{1 - \dfrac{s}{a} + \dfrac{\left(\frac{s}{a}\right)^2}{2} + \cdots \right\}}{s/a} = 1 - \frac{s}{2a} + \cdots \tag{2.52}$$

Now, if $a \to \infty$, then $p(t) \to \delta(t)$, and Equation 2.51 will yield the Laplace transform of $\delta(t)$:

$$\int_0^\infty \delta(t)e^{-st}dt = 1 \tag{2.53}$$

Thus, when $X(s) = 1$, the inverse Laplace transformation of $Y(s) = H(s)$ yields the result: $y(t) = h(t)$. The system output resulting from the unit impulse input, that is, the system *impulse response* $h(t)$, is also the inverse Laplace transform of the transfer function $H(s)$.

In the case where $H(s)$ is a known transfer function and $X(s)$ is the Laplace transform of some arbitrary input $x(t)$, we have shown through the example presented in Section 2.7 that the corresponding output $y(t)$ is deduced by performing

the inverse Laplace transformation of $Y(s) = H(s)X(s)$. In general, for any $H(s)$ and $X(s)$, we have

$$y(t) = \mathcal{L}^{-1}[H(s)X(s)] = \frac{1}{2\pi j} \int_{\sigma-j\infty}^{\sigma+j\infty} H(s)X(s)e^{st} ds$$

$$= \frac{1}{2\pi j} \int_{\sigma-j\infty}^{\sigma+j\infty} \int_{0}^{\infty} h(\tau)e^{-s\tau} d\tau \cdot X(s)e^{st} ds \qquad (2.54)$$

$$= \int_{0}^{\infty} \frac{1}{2\pi j} \int_{\sigma-j\infty}^{\sigma+j\infty} X(s)e^{s(t-\tau)} ds \cdot h(\tau) d\tau$$

Note that in Equation 2.54, we have reversed the order of integration so that the outer integral is based on the time-variable τ. Consider the inner integral, which is a contour integral made with respect to the complex variable s. By definition of the inverse Laplace transform,

$$\frac{1}{2\pi j} \int_{\sigma-j\infty}^{\sigma+j\infty} X(s)e^{s(t-\tau)} ds = x(t - \tau) \qquad (2.55)$$

Thus, Equation 2.54 simplifies to the following:

$$y(t) = \mathcal{L}^{-1}[H(s)X(s)] = \int_{0}^{\infty} h(\tau)x(t - \tau) d\tau \qquad (2.56)$$

The last term in Equation 2.56 represents the *linear convolution* of $h(t)$ and $x(t)$; this could be considered the most important and most fundamental of all mathematical operations in linear systems and signals analysis. Equation 2.56 forms the basis of the following key notions:

(a) The multiplication operation in the time domain is equivalent to the convolution operation in the s-domain. Similarly, it can be shown that the convolution operation in the time domain is equivalent to the multiplication operation in the s-domain.

(b) The impulse response $h(t)$ provides a complete characterization of the dynamic behavior of a given linear system; once $h(t)$ is known, the time-response $y(t)$ of this system to *any* arbitrary input $x(t)$ can be deduced by *convolving* $h(t)$ with $x(t)$.

$$y(t) = \int_0^\infty h(\tau)\, x(t-\tau)\, d\tau$$

$$Y(s) = H(s)\, X(s)$$
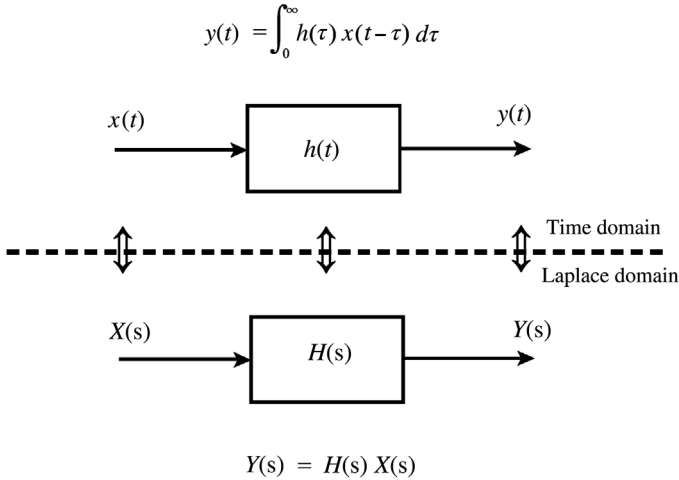
FIGURE 2.12 The equivalence between block diagram representations in time and Laplace domains.

(c) Alternatively, the dynamics of a linear system can also be completely characterized in terms of its transfer function $H(s)$. Once $H(s)$ is known, the Laplace transform of the system response $Y(s)$ to any arbitrary input can be deduced by *multiplying* $H(s)$ with the Laplace transform of the input $X(s)$.

These concepts are illustrated in Figure 2.12.

In the special case where $x(t)$ takes the form of a unit step input (i.e., $x(t) = u(t)$, where $u(t) = 1$ for $t > 0$ and $u(t) = 0$ for $t \leq 0$), Equation 2.56 yields the following result:

$$y(t) = g(t) = \int_0^t h(\tau)d\tau \tag{2.57}$$

which implies that the step response $g(t)$ of a linear system can be obtained by integrating its impulse response with respect to time.

2.10   STATE-SPACE ANALYSIS

To characterize the complexities that are generally found in physiological systems, we often have to resort to the use of differential equations of high order. Analytical solutions of such complicated equations generally are not available, and, furthermore, the numerical solution of these high-order differential equations is often fraught with problems of instability. The Laplace transform approach is useful, but this method of solution can become complicated when some of the initial

conditions have to assume nonzero values. In such circumstances, *state-space modeling* offers an attractive alternative. A very significant advantage of this approach is that the state-space model can quite easily be extended to characterize *time-varying* and *nonlinear* systems. In addition, problems that are formulated as state-space models are readily amenable to standard parameter estimation techniques, such as Kalman filtering.

A key premise in state-space analysis is that the dynamics of a given system can be completely characterized by a minimal set of quantities known collectively as the *state*. This means that if the equations describing the system dynamics are known, one can predict the future state of the system given the present state and the complete time-course of the inputs to the system. Suppose we have a linear system that can be described in terms of the following Nth-order differential equation:

$$\frac{d^N y}{dt^N} + a_{N-1} \frac{d^{N-1} y}{dt^{N-1}} + \cdots + a_1 \frac{dy}{dt} + a_0 y = b_0 x(t) \tag{2.58}$$

We define a set of *N-state variables* $z_1(t)$, $z_2(t)$, . . . , $z_N(t)$ such that

$$z_1(t) = y(t)$$

$$z_2(t) = \frac{dy(t)}{dt} = \frac{dz_1(t)}{dt}$$

$$\vdots \tag{2.59}$$

$$z_N(t) = \frac{d^{N-1} y(t)}{dt^{N-1}} = \frac{dz_{N-1}(t)}{dt}$$

Then, using Equation 2.59, we can recast Equation 2.58 into the following form:

$$\frac{dz_N}{dt} = -a_0 z_1 - a_1 z_2 - \cdots - a_{N-1} z_N + b_0 x \tag{2.60}$$

The above equations can be combined to yield the following first-order matrix differential equation:

$$\frac{d\underline{z}(t)}{dt} = F\underline{z}(t) + Gx(t) \tag{2.61}$$

where

$$\underline{z}(t) = \begin{bmatrix} z_1(t) \\ z_2(t) \\ \vdots \\ z_{N-1}(t) \\ z_N(t) \end{bmatrix}, \quad F = \begin{bmatrix} 0 & 1 & 0 & \cdots & \cdots & 0 \\ 0 & 0 & 1 & 0 & \cdots & 0 \\ \vdots & \vdots & \vdots & \vdots & \vdots & \vdots \\ 0 & 0 & 0 & \cdots & 0 & 1 \\ -a_0 & -a_1 & \cdots & \cdots & -a_{N-2} & -a_{N-1} \end{bmatrix}, \quad \text{and} \quad G = \begin{bmatrix} 0 \\ 0 \\ \vdots \\ 0 \\ b_0 \end{bmatrix} \tag{2.62}$$

It is clear that, basically, what the state-space approach does is to convert the Nth-order differential equation (Equation 2.58) into a set of N first-order differential equations (Equation 2.62).

Finally, we can relate the state vector $\underline{z}(t)$ to the output $y(t)$ of the system in question through

$$y(t) = C\underline{z}(t) + Dx(t) \tag{2.63}$$

where in this case

$$C = \begin{bmatrix} 1 & 0 & \cdots & 0 & 0 \end{bmatrix} \quad \text{and} \quad D = 0 \tag{2.64}$$

Equation 2.63 allows the possibility that one might not be able to measure the state variables directly, although in the particular example that we have considered, we are able to observe the first state variable $z_1(t)$. Equation 2.63 is commonly referred to as the *observation equation*, while Equation 2.61 is called the *state equation*.

In linear systems, it is a relatively simple matter to convert a model represented as a transfer function (i.e., in Laplace transform description) into the state-space form, and vice versa. For the sake of illustration, consider the transfer function given by Equation 2.40b. We define a new intermediate variable $U(s)$ such that this transfer function $H(s)$ can be expressed as the product of two components:

$$(s) = \frac{U(s)}{X(s)} \cdot \frac{Y(s)}{U(s)} = \frac{1}{a_2 s^2 + a_1 s + 1} \cdot \frac{b_2 s^2 + b_1 s + b_0}{1} \tag{2.65}$$

Now, the first component of this product yields

$$a_2 s^2 U(s) + a_1 s U(s) + U(s) = X(s) \tag{2.66}$$

The inverse Laplace transform of Equation 2.66 is

$$a_2 \frac{d^2 u(t)}{dt^2} + a_1 \frac{du(t)}{dt} + u(t) = x(t) \tag{2.67}$$

As in Equation 2.59, we define the state variables $z_1(t)$ and $z_2(t)$ such that

$$z_1(t) = u(t) \tag{2.68a}$$

$$z_2(t) = \frac{dz_1(t)}{dt} = \frac{du(t)}{dt} \tag{2.68b}$$

Then, Equation 2.67 can be rewritten:

$$\frac{dz_2(t)}{dt} = -\frac{a_1}{a_2}z_2(t) - \frac{1}{a_2}z_1(t) + \frac{1}{a_2}x(t) \tag{2.69}$$

Equations 2.68a, 2.68b, and 2.69 can be combined to form the following matrix state equation:

$$\begin{bmatrix} \dfrac{dz_1(t)}{dt} \\ \dfrac{dz_2(t)}{dt} \end{bmatrix} = \begin{bmatrix} 0 & 1 \\ -\dfrac{1}{a_2} & -\dfrac{a_1}{a_2} \end{bmatrix} \begin{bmatrix} z_1(t) \\ z_2(t) \end{bmatrix} + \begin{bmatrix} 0 \\ {}^{1}/{a_2} \end{bmatrix} x(t) \tag{2.70}$$

We turn next to the other component of $H(s)$. Here, we have

$$Y(s) = b_2 s^2 U(s) + b_1 s U(s) + b_0 U(s) \tag{2.71}$$

Taking the inverse Laplace transform of Equation 2.71, we obtain

$$y(t) = b_2 \frac{d^2 u(t)}{dt^2} + b_1 \frac{du(t)}{dt} + b_0 u(t) \tag{2.72}$$

Then, using Equations 2.68a, 2.68b, and 2.69, we obtain the following result:

$$y(t) = \left[\left(b_0 - \frac{b_2}{a_2} \right) \left(b_1 - b_2 \frac{a_1}{a_2} \right) \right] \begin{bmatrix} z_1 \\ z_2 \end{bmatrix} + \frac{b_2}{a_2}x(t) \tag{2.73}$$

Thus, conversion of the transfer function in Equation 2.40b into state-space form leads to the state equation given by Equation 2.70 and the observation equation given by Equation 2.73.

2.11 COMPUTER ANALYSIS AND SIMULATION: MATLAB AND SIMULINK

Although the use of Laplace transforms and state-space modeling greatly simplifies the mathematical characterization of linear systems, models that provide adequate representations of realistic dynamical behavior are generally too complicated to deal with analytically. Therefore, in such complex situations, the logical approach is to translate the system block representation into a computer model and to solve the corresponding problem numerically. Traditionally, one would derive the differential equations that represent the model and develop a program in some basic

programming language, for example, C or Fortran, to solve these equations. However, a variety of software tools are available that further simplify the task of model simulation and analysis. One of these, named SIMULINK, is currently used by a large segment of the scientific and engineering community. SIMULINK provides a graphical environment that allows the user to easily convert a block diagram into a network of blocks of mathematical functions. It runs within the interactive, command-based environment called MATLAB. In the discussions that follow, it is assumed that the reader has access to, at least, the Student Versions of MATLAB and SIMULINK, both of which are products of The Mathworks, Inc. (Natick, MA). We also assumed that the reader is familiar with the most basic functions in MATLAB. One advantage of employing MATLAB and SIMULINK is that these tools are platform-independent; thus, the same commands apply whether one is using SIMULINK on a Windows-based machine or on a Unix-based computer. In the rest of this section, our aim is to give the reader a brief "hands-on" tutorial on the use of SIMULINK by demonstrating in a step-by-step manner how one would go about simulating the linear lung mechanics model discussed in Section 2.3. For more details and advanced topics, the reader is referred to the User's Guides of both SIMULINK (Dabney and Harman, 2004) and MATLAB (Hanselman and Littlefield, 2012).

Let us suppose that we would like to find out how much tidal volume is delivered to a patient in the intensive care unit when the peak pressure of a ventilator is set at a prescribed level. Obviously, the solution of this problem requires a knowledge of the patient's lung mechanics. We assume that this patient has relatively normal mechanics, and the values of the various pulmonary parameters are as follows: $R_C = 1$ cm H_2O s L^{-1}, $R_P = 0.5$ cm H_2O s L^{-1}, $C_L = 0.2$ L cm H_2O^{-1}, $C_W = 0.2$ L cm H_2O^{-1}, and $C_S = 0.005$ L cm H_2O^{-1} (see Figure 2.6). We will consider two ways of approaching this problem. The first and most straightforward method is to derive the transfer function for the overall system and use it as a single "block" in the SIMULINK program. The differential equation relating total airflow Q to the applied pressure at the airway opening P_{ao} was derived using Kirchoff's laws and presented in Equation 2.16. Substituting the above parameter values into this differential equation and taking its Laplace transform yields, after some re-arrangement of terms, the following expression:

$$\frac{Q(s)}{P_{ao}(s)} = \frac{s^2 + 420s}{s^2 + 620s + 4000} \tag{2.74}$$

To implement the above model, run SIMULINK from within the MATLAB command window (i.e., type `simulink` at the MATLAB prompt). The SIMULINK *Start Page* will be displayed in a new window. Next, select *Blank Model* in the *New* tab. This will open another window named `untitled` – This is the working window in which we will build our model. To open *Library Brower* where we can select standard block functions from for model development, go to *View* menu and select *Library Browser*. In our case, we would like to choose a block

that represents the transfer function shown in Equation 2.74. Open the *Continuous* sub-library – This will display a number of block functions in this sub-library. Select `Transfer Fcn` and drag this block into the working window. Double-click this block to input the parameters of the transfer function. Enter the coefficients of the polynomials in *s* in the numerator and denominator of the transfer function in the form of a row vector in each case. Thus, for our example, the coefficients for the numerator will be [1 420 0]. These coefficients are ordered in descending powers of *s*. Note that, even though the constant term does not appear in the numerator of Equation 2.74, it is still necessary to include explicitly as zero the "coefficient" corresponding to this term. Once the transfer function block has been set up, the next step is to include a generating source for P_{ao}. This can be found in the *Sources* sub-library. In our case, we will select and drag the `Sine Wave` function generator into the working window. Double-click this icon in order to modify the amplitude and frequency of the sine wave: We will set the amplitude to 2.5 cm H_2O (i.e., peak-to-peak swings in P_{ao} will be 5 cm H_2O), and the frequency to 1.57 rad s^{-1}, which corresponds to $1.57/(2^*\pi) = 0.25$ Hz or 15 breaths min^{-1}. Connect the output port of the sine wave block to the input port of the transfer function block with a line. To view the resulting output *Q*, open the *Sinks* sub-library and drag a `Scope` block into the working window. Double-click the `Scope` block – This will open the *Scope* window where you can view the plot and modify the plot layout/range. To change the range of *y*-axis, go to *View* menu, select *Configuration Properties*, click the *Display* tab, and then enter *y*-axis limits. It is always useful to view the input simultaneously. So, drag another `Scope` block into the working window and connect the input of this block to the line that "transmits" P_{ao}. At this point, the model is complete, and one can proceed to run the simulation. However, it is useful to add a couple of features. We would also like to view the results in terms of volume delivered to the patient. This is achieved by integrating *Q*. From the *Continuous* sub-library, select the `integrator` block. Send *Q* into the input of this block and direct the output (*Vol*) of the block to a third `Scope`. It is also advisable to save the results of each simulation run into a data file that can be examined later. From the *Sinks* sub-library, select and drag the `To File` block into the working window. This output file will contain a matrix of numbers. A name has to be assigned to this output file as well as to the matrix variable. In our case, we have chosen to give the file the name `respm1.mat`, and the matrix variable the name `respm1`. Note that "mat" files are the standard (binary) format in which Matlab variables and results are saved. The first row of matrix `respm1` will contain the times that correspond to each iteration step of the simulation procedure. In our case, we would like to save the time-histories of P_{ao}, *Q*, and *Vol*. Thus, the `To File` block will have to be adjusted to accommodate three inputs. Since this block expects the input to be in the form of a vector, the three-scalar variables will have to be transformed into a three-element vector prior to being sent to the `To File` block. This is achieved with the use of the `Mux` block, found in the *Signal Routing* sub-library. After dragging it into the working

window, enter "3" for the *Number of inputs* and press Enter. Note that there are now three input ports on the Mux block. Connect the Mux output with the input of the To File block. Connect the input ports of the Mux block to P_{ao}, Q, and *Vol*. The block diagram of the completed SIMULINK model is shown in Figure 2.13a. The final step is to run the simulation. Go to the *Simulation* menu and select *Model Configuration Parameters*. This allows the user to specify the duration over which the simulation will be conducted, as well as the minimum and maximum step sizes for each computational step. The latter will depend on the dynamics of the system in question – In our case, we have chosen both the Max and Min step sizes to be 0.01 s. The algorithm (*Solver*) for performing integration is selected automatically, but the user can also select the algorithm from the *Solver* list depending on the problem at hand. The user is encouraged to experiment with different algorithms
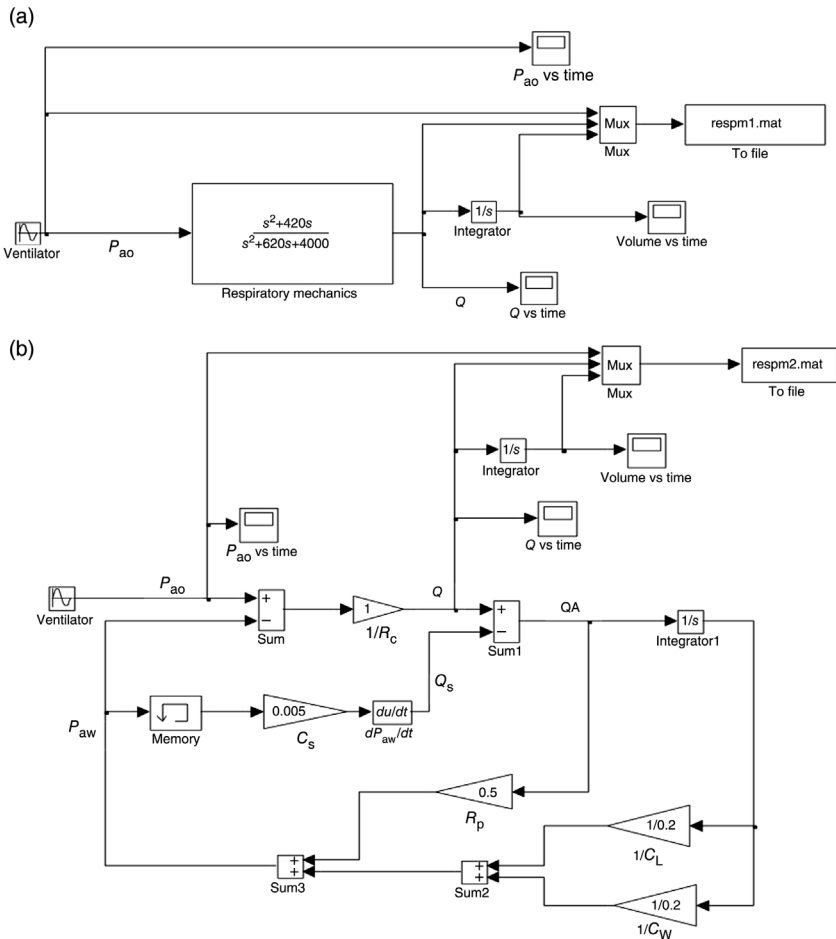
FIGURE 2.13   Two SIMULINK models of simple lung mechanics.

for a given problem. Some algorithms may produce numerically unstable solutions, while others may not. Finally, go to *Simulation* menu and select *Run* to proceed with the simulation.

In the SIMULINK model of Figure 2.13a, the bulk of our effort in solving the problem was expended on deriving the analytical expression for the transfer function $Q(s)/P_{ao}(s)$ (Equation 2.16). This clearly is not the approach that we would want to take in general, since one of the reasons for computer simulation is to simplify the task of modeling. Obtaining the solution for $Q(t)$, for given P_{ao} and lung mechanical parameters, leads also to the simultaneous solution of other "internal" variables that might be of interest. These include the alveolar pressure P_A and the airflow actually delivered to the alveoli, Q_A. However, the implementation of the model in Figure 2.13a does not allow access to these internal variables since the system dynamics are "lumped" into a single transfer function block. The SIMULINK implementation of the same model in Figure 2.13b displays a more "open structure." Various segments in the block diagram shown correspond directly with the basic circuit equations (Equations 2.14 and 2.15) derived from applying Kirchoff's laws to the model of Figure 2.6. For instance, the double loop containing the gains $1/C_L$, $1/C_W$, R_P, and C_S in the SIMULINK diagram of Figure 2.13b represents Equation 2.14. This kind of open architecture also makes it easier to determine how alterations in the parameters, such as what might occur with different lung diseases, are expected to affect overall lung mechanics. However, a common limitation with this approach is the creation of *algebraic loops*. This problem arises when blocks with direct feedthrough are connected together in a loop. For example, in Figure 2.13b, the gains R_P and C_S are connected together with the derivative block in a closed loop. These functions are all feedthrough blocks; so at any integration step, the simultaneous solution of the two equations represented by this loop is required. This is accomplished iteratively, but numerical ill-conditioning could lead to no convergent solution. For each algebraic loop, SIMULINK reports an error when more than 200 iterations are expended per integration step. We eliminate this problem in our example by adding a `Memory` block to the closed loop in question. The `Memory` block simply adds a delay of one integration time-step to the circuit. However, one should be cautioned that this "fix" does not always work and, under certain circumstances, could lead to numerical instability.

Figure 2.14 shows sample simulation results produced by either of the model implementations. As indicated earlier, we assume the ventilator generates a sinusoidal P_{ao} waveform of amplitude 2.5 cm H_2O. In Figure 2.14a, the ventilator frequency is set at 15 breaths min^{-1}, which is approximately the normal frequency of breathing at rest. At this relatively low frequency, the volume waveform is more in phase with P_{ao}; the airflow Q shows a substantial phase lead relative to P_{ao}. This demonstrates that lung mechanics is dominated by compliance effects at such low frequencies. The peak-to-peak change in volume (i.e., tidal volume) is approximately 0.5 L, while peak Q is $\sim$0.4 L s^{-1}. When the ventilator frequency is increased fourfold to 60 breaths min^{-1} with amplitude kept unchanged, peak Q clearly
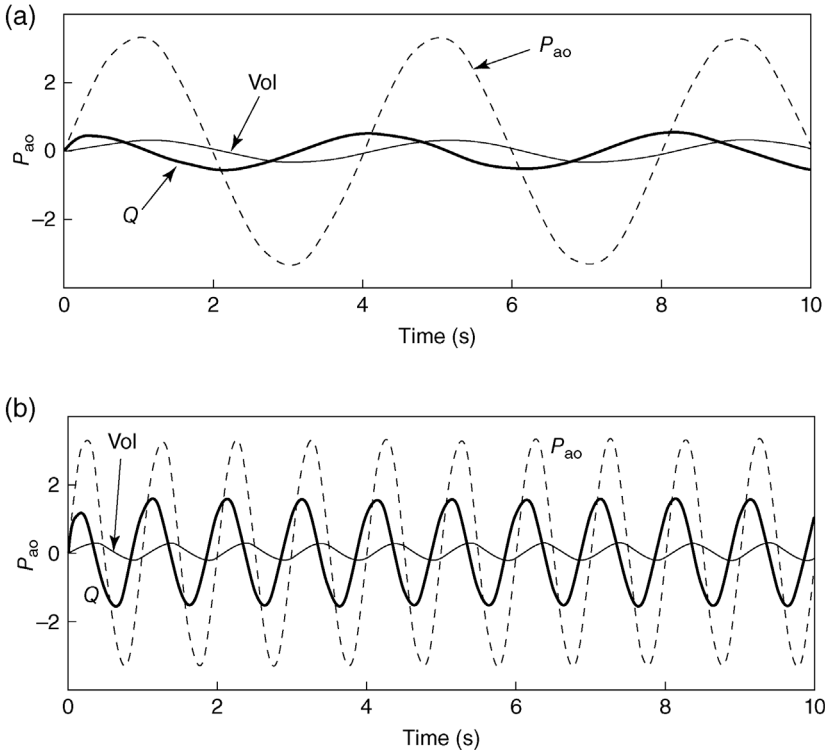
FIGURE 2.14 Sample simulation results from SIMULINK implementation of lung mechanics model. (a) Predicted dynamics of airflow Q and volume Vol in response to sinusoidal forcing of P_{ao} (amplitude $= 2.5$ cm H_2O) at 15 breaths min^{-1}. (b) Predicted dynamics of Q and Vol in response to sinusoidal forcing of P_{ao} (amplitude $= 2.5$ cm H_2O) at 60 breaths min^{-1}.

increases (to ~ 1.2 L s^{-1}), while tidal volume is decreased (to ~ 0.4 L). Now, Q has become more in phase with P_{ao} while volume displays a significant lag. Thus, resistive effects have become more dominant at the higher frequency. The changes in peak Q and tidal volume with frequency demonstrate the phenomenon pulmonologists refer to as *frequency dependence* of pulmonary resistance and compliance, that is, the lungs appear stiffer and less resistive as frequency increases from resting breathing.

The SIMULINK programs displayed in Figure 2.13a and b have been saved as the model files `respm1.slx` and `respm2.slx`, respectively. These have been included in the library of MATLAB script files (m-files) and SIMULINK model files (slx-files) accompanying this book. For a full compilation of all script files, model files, and MATLAB and SIMULINK functions employed in this book, the reader is referred to Appendix B.

PROBLEMS

P2.1. **(a)** Show that $\mathcal{L}\{x(t - T)\} = e^{sT}X(s)$, where $\mathcal{L}\{\ \}$ is the Laplace transform operator, $x(t - T) = 0$ for $t \leq T$, and $X(s) = \mathcal{L}\{x(t)\}$.

(b) Show that the Laplace transform of $\int_0^t x(\tau)d\tau$ is equal to $X(s)/s$.

(c) Show that multiplication in the s-plane corresponds to convolution in the time domain: $\mathcal{L}^{-1}\{X_1(s) \cdot X_2(s)\} = \int_0^t x_1(\tau)x_2(t - \tau)d\tau$

P2.2. **(a)** Determine the impulse response for the linear system characterized by the following differential equation (assuming $x(t)$ to be the input, and $y(t)$ to be the output) by solving the equation in the time domain:

$$\frac{d^2y}{dt^2} + 4\frac{dy}{dt} + 3y = x(t)$$

(b) Determine the transfer function for the above equation, and by taking the inverse Laplace transform, obtain an expression for the impulse response (time domain). The impulse response determined in (b) should be the same as that determined in (a).

P2.3. Figure P2.1 shows a schematic diagram of the five-element Windkessel model that has been used to approximate the hemodynamic properties of the arterial tree. The model consists of a distensible (as illustrated by the two-ended arrows) aorta and a lumped representation of the rest of the arterial vasculature. The latter is modeled as a simple parallel combination of peripheral resistance R_P and peripheral compliance C_P. The mechanical parameters pertinent to the aortic portion are (a) the compliance of the aortic wall C_{ao}, (b) the viscous resistance of the aortic wall R_{ao}, and (c) the inertance to flow through the aorta, L_{ao}. Note that resistance to flow in the aorta is considered negligible compared to R_P. Construct the electrical
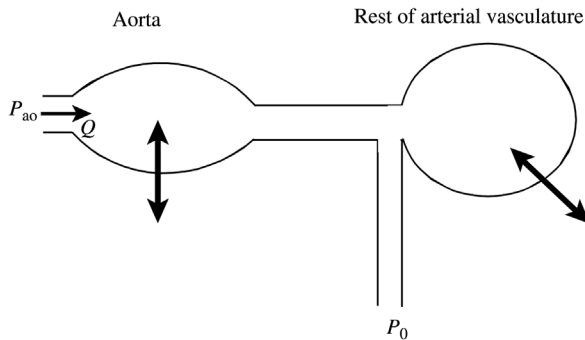
FIGURE P2.1 Schematic representation of the five-element Windkessel model of aortic and arterial hemodynamics.

analog of this model and derive the transfer function and equivalent state-space model relating aortic pressure P_{ao} to aortic flow Q.

P2.4. A somewhat different version of linear muscle mechanics from that displayed in Figure 2.7 is shown in Figure P2.2. Here, the elastic element C_P is placed in parallel to the viscous damping element R and the contractile element, and the entire parallel combination is placed in series with the elastic element C_S and the lumped representation of the muscle mass m. Derive an expression for the transfer function relating the extension of the muscle x to an applied force F. Convert this transfer function description into the equivalent state-space model.
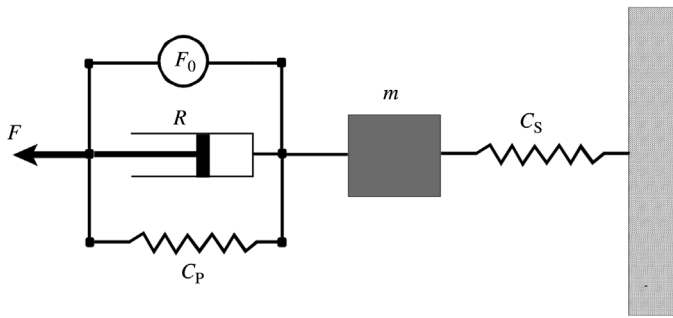
FIGURE P2.2   Alternative model of muscle mechanics that includes the effect of muscle mass.

P2.5. Figure P2.3 displays the equivalent circuit of a short length of squid axon according to the Hodgkin–Huxley model of neuronal electrical activity. The elements shown as circles represent voltage sources that correspond to the Nernst potentials for sodium, potassium, and chloride ions. The resistances are inversely proportional to the corresponding membrane
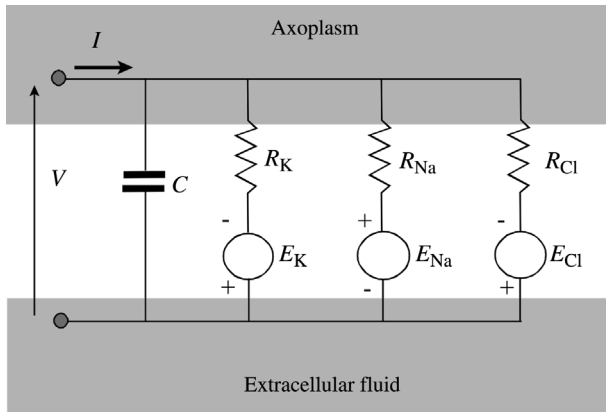
FIGURE P2.3   Hodgkin–Huxley model of the electrical properties of nerve membrane.

conductances for these three types of ions, while C represents membrane capacitance.

(a) Derive the Hodgkin–Huxley equation, that is, the differential equation that relates the net current I flowing through the membrane to the applied voltage V across the membrane.

(b) Develop the mechanical analog of this electrical circuit. Briefly explain why the various system elements are placed in series or parallel.

P2.6. The time-domain response of a mechanoreceptor to stretch, applied in the form of a step of magnitude x_0 (in arbitrary length units), is

$$V(t) = x_0\left(1 - e^{-5t}\right)u(t)$$

where the receptor potential V is given in millivolts and $u(t)$ is the unit step function ($u(t) = 1$ for $t > 0$ and $u(t) = 0$ for $t < 0$) and time t from the start of the step is given in seconds. Assuming the system to be linear:

(a) Derive an expression for the transfer function of this system.

(b) Determine the response of this system to a unit impulse.

(c) Determine the response of this system to a unit ramp.

P2.7. A linear dynamic system is represented by the following state-space model:

$$\left.\begin{array}{c} \dfrac{dz}{dt} = A\underline{z} + Bx \\ y = C\underline{z} \end{array}\right\}, \quad \text{where} \quad A = \begin{pmatrix} -2 & -9 \\ 1 & 0 \end{pmatrix},$$

$$B = \begin{pmatrix} 5 \\ 0 \end{pmatrix}, \quad C = \begin{pmatrix} 0 & 1 \end{pmatrix}, \quad \underline{z} = \begin{pmatrix} z_1 \\ z_2 \end{pmatrix}$$

where y is the observed output, x is the input, and $\underline{z}$ is the state vector. Derive the transfer function (input $= x$, output $= y$) for this system.

P2.8. Figure P2.4 displays a lumped-parameter mechanical model (known as the "Burger" model) of a strip of viscoelastic material. The model consists of two linear springs (with compliances C_S and C_P) and two linear dashpots (with mechanical resistances R_S and R_P). If F is the tensile force applied to this strip of material, the model is extended by a displacement x, the left plate that couples R_P and C_P is displaced by x_1, and the right plate that couples R_P and C_P is displaced by x_2 (all displacements are relative to the immovable wall).

(a) Derive the three equations of motion (i.e., one for C_S, one for C_P and R_P, and the third for R_S) that characterize the dynamics of this system.

(b) Assuming force to be equivalent to voltage and displacement (extension) to be equivalent to electrical charge, develop the electrical analog of the Burger model. Provide brief explanations of why certain
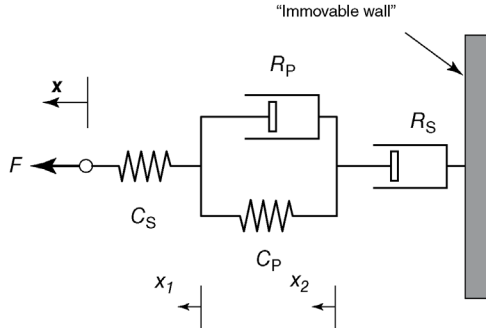
FIGURE P2.4 Lumped-parameter mechanical model of a strip of viscoelastic material.

components are in series or parallel. Label clearly the electrical equivalents of C_S, C_P, R_S, R_P, F, x, x_1 and x_2.

P2.9. Figure P2.5 shows the lumped-parameter mechanical model of a block of tissue extracted from a cancerous mass (mass m_m) that is placed on an elastic platform (platform has mass m_t). The cancerous tissue is being tested for its mechanical response to a variable force F applied to it. k_m and b represent the elastic stiffness and viscous damping, respectively, of the tissue, whereas k_t represents the elastic stiffness of the platform. x_m and x_t represent the vibrational displacements of the tissue and platform, respectively, in response to F. *Note:* stiffness $= 1/$compliance. Ignore the effects of gravity.

(a) Derive the ordinary differential equations that characterize the dynamics of m_m and m_t.
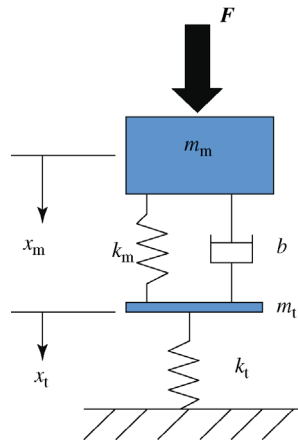
FIGURE P2.5 Lumped-parameter mechanical model of a block of tissue extracted from a cancerous mass and placed on an elastic platform.

(b) Develop the electrical analog of this model. Briefly explain why the various components are in parallel or series. Label each of the elements (e.g., k_m and b).

P2.10. Using the SIMULINK program `respm2.slx` as the starting point, incorporate into the respiratory mechanics model the effect of inertance to gas flow in the central airways (L_C). Assume a value of 0.01 cm $H^2O\,s^2\,L^{-1}$ for L_C. Keep the values of the other parameters unchanged. Through model simulations, determine how airflow and tidal volume would vary in response to sinusoidal pressure waveforms at the airway opening (P_{ao}) applied at 15, 60, 120, 240, 480, and 960 breaths min^{-1}. To simulate a subject with emphysema, increase lung compliance (C_L) to 0.4 L cm H_2O^{-1} and peripheral airway resistance (R_P) to 7.5 cm $H_2O\,s\,L^{-1}$. Repeat the simulations at the frequencies listed above. Compare the frequency dependence of the resulting airflow and volume waveforms to those obtained for the normal subject.

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3

STATIC ANALYSIS OF PHYSIOLOGICAL SYSTEMS

3.1 INTRODUCTION

Although the primary interest in most studies of physiological control systems is in the dynamic aspects, a preliminary investigation of steady-state behavior invariably leads to useful insights. This information can be used subsequently as the basis for further dynamic analysis. Steady-state measurements are generally easier to make in physiological systems, and, therefore, this knowledge is usually more complete than knowledge about the dynamics. As such, it is useful to conduct an analysis that can demonstrate how the static characteristics of the various components lead to the steady-state behavior of the overall system. This allows us to verify our working hypothesis of how these components are interconnected and to determine whether there may be other factors that need to be included. Steady-state analysis also allows us to compare the operating characteristics of the system with and without feedback.

Under normal circumstances, physiological control systems generally operate within a relatively narrow range. For instance, body temperature hovers around 37 °C, resting arterial blood P_{CO2} close to 40 mmHg, and cardiac output is generally about 5 L min^{-1}. These are just a few examples of Walter Cannon's *homeostatic principle* at work. How are these *equilibrium* or steady-state values determined? As we had discussed in Section 1.3, in an engineering control system, one can always introduce a *reference input* that then determines the *set point* of the system (see

Figure 1.1). However, in most physiological systems, it is difficult to identify any explicit "reference input" signal. We also mentioned previously that the equivalent of the "comparator" in engineering control systems is generally not found explicitly in physiology. Instead, negative feedback is *embedded* into the properties of one of the system components. It turns out that it is this "embedded" negative feedback that allows the closed-loop physiological system to determine its steady-state operating level.

3.2   OPEN-LOOP VERSUS CLOSED-LOOP SYSTEMS

In Chapter 1, we pointed out in a qualitative way the advantage of employing negative feedback to regulate some selected variable. Here, we will examine this notion quantitatively. We turn back to the example in Chapter 1 of the simple control scheme to regulate the temperature of a room in winter with the use of a fan-heater. The open-loop control scheme is shown in Figure 3.1a. Let us first assume that the environmental temperature is $0\,°C$, and that if the fan-heater is not turned on, the room temperature will also equilibrate to $0\,°C$. We will also assume that the two subsystems involved, the fan-heater and the room, have linear characteristics. For the fan-heater, this takes the form of a constant gain G_C, so that if a reference voltage
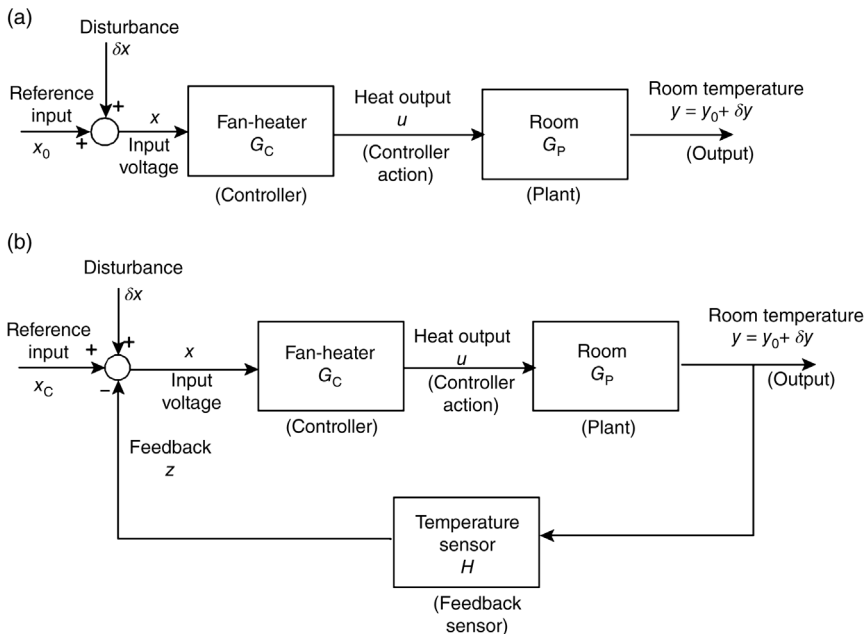
FIGURE 3.1   Analysis of the influence of input fluctuations on the control of room temperature using (a) open-loop and (b) closed-loop schemes.

x_0 (in volts) is supplied to the fan-heater, the rate at which heat is produced is u_0 (in cal s^{-1}) will be related to x_0 by

$$u_0 = G_C x_0 \tag{3.1}$$

The addition of heat at this rate to the room in the steady state will equal the rate at which heat is lost to the cold exterior at the equilibrium temperature of y_0 (in °C). y_0 will be related to u_0 by

$$y_0 = G_P u_0 \tag{3.2}$$

where G_P is the "gain" of the room. Note that G_C and G_P will have units of cal s^{-1} V^{-1} and °C s cal^{-1}, respectively. Thus, for the overall open-loop system, we have

$$y_0 = G_C G_P x_0 \tag{3.3}$$

Now, assume that there is an unexpected change in input voltage to the fan-heater of magnitude δx. Ignoring transient effects, the resulting room temperature will become

$$y = G_C G_P (x_0 - \delta x) \tag{3.4}$$

Subtracting Equation 3.3 from Equation 3.4, we find that the change in room temperature, δy, is

$$\delta y = G_C G_P \delta x \tag{3.5}$$

If we extend this result to the case of the general linear open-loop control system, Equation 3.5 states that the change in the output or regulated variable is proportional to the magnitude of the input disturbance. The constant of proportionality is known as the *open-loop gain* (OLG), and

$$OLG = G_C G_P \tag{3.6}$$

We turn now to the closed-loop control scheme, illustrated in Figure 3.1b. Here the room temperature is measured and converted into a feedback voltage (z), which is subtracted from the reference input. The resulting voltage is used to drive the fan-heater. We assume the reference input required to support a temperature set point of y_0 °C is x_C V. Under set point conditions, the driving voltage will be $x_C - z_0$ and, consequently, the heat output rate of the fan-heater will be

$$u_0 = G_C (x_C - z_0) \tag{3.7}$$

As in the open-loop case, the temperature set point will be related to u_0 through Equation 3.2. Therefore, combining Equations 3.7 and 3.2, the relationship between y_0 and the driving voltage of the fan-heater will be

$$y_0 = G_C G_P (x_C - z_0) \tag{3.8a}$$

But z_0 is linearly proportional to y_0 through the feedback gain H:

$$z_0 = H y_0 \tag{3.9}$$

By eliminating z_0 from Equations 3.8a and 3.9 and rearranging terms, we obtain the following result:

$$y_0 = \frac{G_C G_P}{1 + G_C G_P H} x_C \tag{3.8b}$$

Comparing Equations 3.8b and 3.3, it is clear that the reference voltage in the open-loop case, x_0, will be different from the reference input voltage in the closed-loop case, x_C. Now, consider the effect of a disturbance of magnitude δx and assuming this leads to a change in the room temperature by δy, we obtain the following:

$$y_0 + \delta y = \frac{G_C G_P}{1 + G_C G_P H}(x_C + \delta x) \tag{3.10}$$

Subtracting Equation 3.8b from Equation 3.10 yields

$$\delta y = \frac{G_C G_P}{1 + G_C G_P H} \delta x \tag{3.11}$$

Thus, by definition, the *closed-loop gain* (CLG) of the feedback system is

$$CLG = \frac{G_C G_P}{1 + G_C G_P H} \tag{3.12a}$$

Comparing Equations 3.6 and 3.11, it follows that

$$CLG = \frac{OLG}{1 + OLG \cdot H} \tag{3.12b}$$

Since G_C, G_P, and H are all positive quantities, Equation 3.12b implies that CLG is always smaller than OLG. This means that the incorporation of negative feedback into a control system can lead to a *reduction* of the effect of disturbances on the system. As a result, a closed-loop regulator has a greater ability to maintain the regulated variable within narrower limits than its open-loop counterpart. In

addition, a closed-loop servomechanism possesses greater inherent capability of tracking its prescribed trajectory in the presence of external noise.

In Equation 3.11, note that the degree to which feedback reduces the effect of disturbances depends on the factor $G_C G_P H$. The larger this term, the smaller the effect of disturbances on the system output. Since $G_C G_P H$ represents the product of gains of all system components along the closed loop, it is also known as the *loop gain* (LG). Therefore, an increased LG enhances the effectiveness of the negative feedback. Notice also that, to increase LG, we are not necessarily limited to increasing the feedback gain H; LG can be increased by increasing one or more of the gains G_C, G_P, and H. On the other hand, Equation 3.10 also implies that although the effect of a constant input disturbance on the operating level of the controlled variable (y) is attenuated, it is never completely eliminated unless LG is infinitely high. Therefore, in closed-loop systems with proportional feedback, there will always be a *steady-state error* between the new steady-state operating level (in the presence of the disturbance) and the previous set point.

3.3   DETERMINATION OF THE STEADY-STATE OPERATING POINT

In the previous example, the temperature set point was a direct function of a reference voltage input. We turn now to physiological control systems that generally do not have an explicitly controlled set point. On the other hand, if the primary function of the control system is to regulate some physiological variable, this controlled variable normally will fluctuate within a relatively narrow range. So, although there is no explicit set point, there is generally a *steady-state operating point*. One could also use the term "equilibrium" loosely to refer to the steady-state operating point. However, in reality, the regulated variable is subjected to many cyclical influences, such as circadian rhythms, as well as influences resulting from coupling with other physiological organ systems. Therefore, a true static equilibrium never really exists.

We consider again the simple model of the muscle stretch reflex discussed in Section 1.4. However, this time around, we will ignore all dynamic aspects of the model and assume that we know the steady-state characteristics of all the three component blocks. These are as illustrated in Figure 3.2. At the level of the spinal cord, afferent neural discharge frequency f_a is converted into efferent discharge frequency f_e through the linear relationship:

$$f_e = G_C f_a \tag{3.13}$$

Since a larger increase in efferent neural frequency leads to a greater contraction of the extensor muscle, the gain of the plant component must be negative. We have indicated in Chapter 1 that this is where the negative feedback of the closed-loop
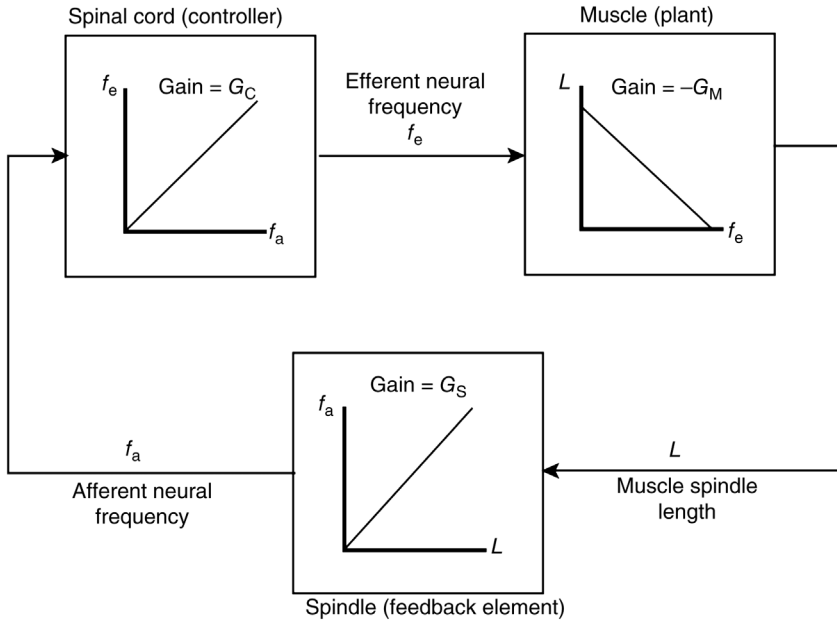
FIGURE 3.2 Block diagram displaying the steady-state characteristics of the muscle stretch reflex model components.

control system is "embedded," since there is no physiological component that can be identified as an explicit "comparator." Assuming the amount of contraction is proportional to the increase in efferent frequency, we have the following steady-state muscle characteristic:

$$L = L_0 - G_M f_e \tag{3.14}$$

where L_0 is the (hypothetical) muscle length if the efferent nerve is completely silenced. Finally, we assume that the muscle spindle sends afferent neural impulses back to the spinal cord in proportion to the length of the muscle, so that afferent traffic increases when the muscle is stretched:

$$f_a = G_S L \tag{3.15}$$

Given these characteristics and the fact that there is no explicit reference input in this case, what would be the steady-state operating point of this system?

Figure 3.3 shows in graphical form how this equilibrium level is arrived at. Here, we have rearranged the graphs of the static characteristics of the three system components such that they share common axes. The spinal cord (f_e versus f_a) graph shares the same f_e axis as the muscle (L versus f_e) graph and is rotated 90° clockwise, so the f_a axis points downward. The spindle (f_a versus L) graph is
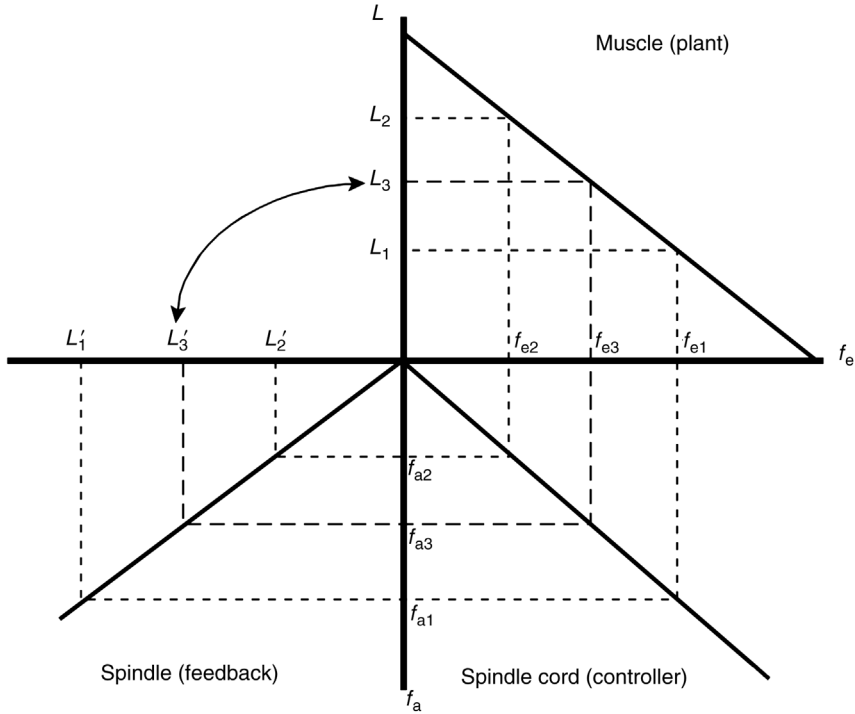
FIGURE 3.3 Graphical determination of the steady-state operating point.

rotated 180° clockwise and shares the same axis as the spinal cord graph. If the spindle were to be somehow isolated from the muscle and stretched by some artificial means, then both afferent and efferent neural frequencies would increase with increasing stretch. Thus, if length sensed by the spindle is equivalent to a muscle length of L_1' (> equilibrium muscle length), the afferent neural discharge rate sent to the spinal cord would be f_{a1} (> equilibrium afferent discharge rate). This induces an efferent neural discharge rate of f_{e1} (> equilibrium efferent discharge rate). This increased efferent discharge rate would produce a contraction of the extensor muscle to length L_1 (< equilibrium muscle length). On the other hand, if the length sensed by the spindle is equivalent to a muscle length of L_2' (< equilibrium muscle length), the afferent neural discharge rate sent to the spinal cord would be f_{a2} (< equilibrium discharge rate) and the resultant efferent neural discharge rate would be f_{e2} (< equilibrium discharge rate). This decreased efferent discharge rate would produce a relaxation of the extensor muscle to length L_2 (> equilibrium muscle length). It follows that at some point in between these two extremes, these opposing effects will come to a balance, establishing the steady-state operating level that is the equilibrium muscle length ($L_3' = L_3 \rightarrow f_{a3} \rightarrow f_{e3}$).
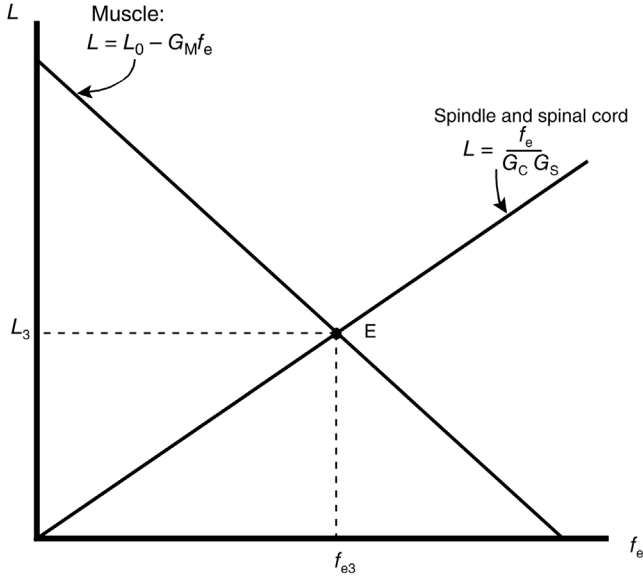
FIGURE 3.4 The point of intersection (E) between the "muscle" and "spindle+spinal cord" plots yields the steady-state operating point of the muscle stretch reflex system.

A simpler way of determining the steady-state operating level is shown in Figure 3.4. Here, we combine Equations 3.13 and 3.15 so that f_e is expressed as a function of L:

$$f_e = G_C G_S L \qquad (3.16)$$

This, in essence, collapses two graphs (f_e versus f_a and f_a versus L) into one. In Figure 3.4, Equation 3.16 is plotted on the same axes as Equation 3.14. However, since L is plotted on the vertical axis, the slope of the line corresponding to Equation 3.16 is $1/G_C\,G_S$. Figure 3.4 shows that the equilibrium level (labeled E) is determined by the intersection between the two plots, since it is only at the intersection point that both muscle and spindle+spinal cord relationships are simultaneously satisfied.

The above considerations point to a third but nongraphical way of arriving at the same answer: simultaneous solution of the algebraic equations represented by Equations 3.13 through 3.15. Eliminating f_a and f_e from these three equations, we obtain the equilibrium solution (L_3) for L:

$$L_3 = \frac{L_0}{1 + G_M G_C G_S} \qquad (3.17)$$

Substituting Equation 3.17 into Equation 3.16 yields the equilibrium solutions (f_{a3} and f_{e3}) for the afferent and efferent neural discharge frequencies, respectively:

$$f_{a3} = \frac{G_S L_0}{1 + G_M G_C G_S} \tag{3.18}$$

$$f_{e3} = \frac{G_C G_S L_0}{1 + G_M G_C G_S} \tag{3.19}$$

3.4 STEADY-STATE ANALYSIS USING SIMULINK

In systems that contain several components, graphical solution of the steady-state operating point may prove to be somewhat laborious. Furthermore, if these components are nonlinear, simultaneous solution of the corresponding algebraic equations could be difficult. In such situations, it may be useful to solve the problem through numerical means. In this section, we illustrate how a steady-state analysis may be conducted using SIMULINK.

We turn once again to our simple model of the muscle stretch reflex. In the present example, however, we will assume a nonlinear relationship to represent L versus f_e for the muscle component:

$$L = 1 - \frac{f_e^5}{0.5^5 + f_e^5} \tag{3.20}$$

For simplicity, we have scaled all variables involved (L, f_a, and f_e) to their corresponding maximum values, so that scaled variables range between 0 and 1. To represent the spindle, we assume another nonlinear expression:

$$f_a = 0.6 L e^{0.5L} \tag{3.21}$$

Finally, for the spinal cord, we will assume a simple linear gain of unity between f_a and f_e:

$$f_e = f_a \tag{3.22}$$

For reference purposes, we first deduce the steady-state operating point of this system using the graphical procedure described in the previous section. Figure 3.5 shows, on the same set of axes, a plot of the muscle characteristics (Equation 3.20) superimposed against a plot of the combined characteristics of the spindle and spinal cord (Equations 3.21 and 3.22 combined). The intersection, labeled E, represents the steady-state operating point of this system; here, $L = 0.58$ and $f_e = 0.47$.
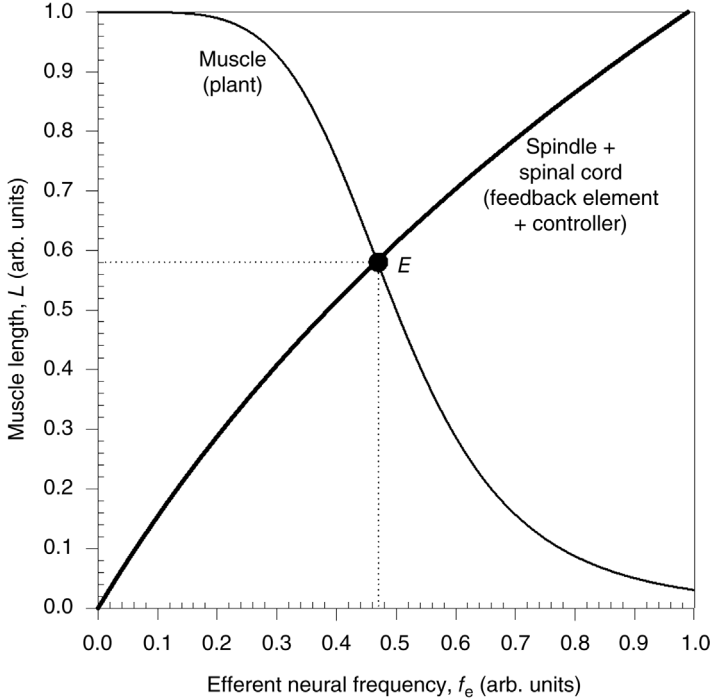
FIGURE 3.5   Steady-state solution for muscle stretch reflex model with nonlinear characteristics. The equilibrium point is labeled E.

The SIMULINK implementation of the above model, named `msrflx.slx`, is shown in Figure 3.6. The three system components, representing the spinal cord, muscle, and spindle, are linked together in a closed loop. The spinal cord component takes the form of a simple linear gain function (with gain $= 1$). To represent the nonlinear muscle characteristics, we employ the `Fcn` block from the *User-Defined Functions* sub-library. This block allows us to custom-design any mathematical relation using MATLAB-styled expressions. The MATLAB expression corresponding to Equation 3.20 is

$$1 - u^5 / (0.5^5 + u^5)$$

where u, the default input variable name, represents f_e. The same kind of `Fcn` block is used to represent the spindle characteristics. Here, the corresponding MATLAB expression is

$$0.6 \cdot u \cdot \exp(0.5 \cdot u)$$

where u (again, the default input variable name) now represents L. Since this SIMULINK implementation solves the system of equations, Equations 3.20
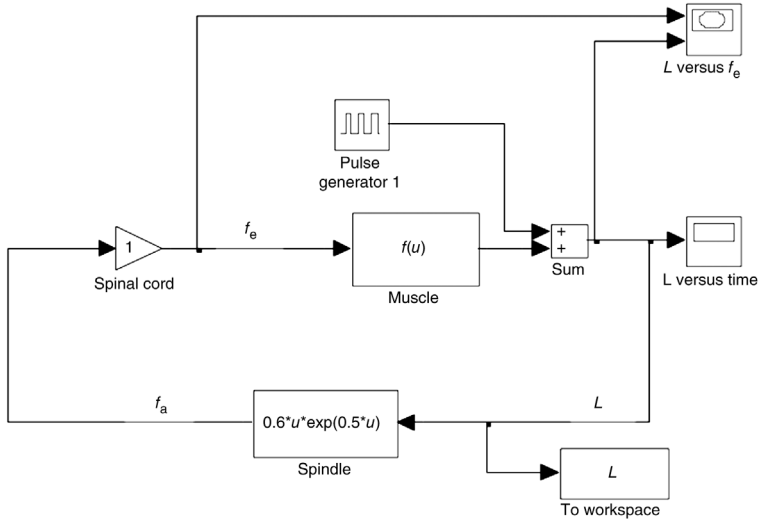
FIGURE 3.6 SIMULINK program `msrflx.slx` used for steady-state analysis of the muscle stretch reflex model.

through 3.22, in an iterative fashion, an "initial guess" of the solution has to be made. We achieve this by introducing an initial "impulse" into the closed loop. This is done by using a `Pulse Generator` block from the *Sources* sub-library. The period of the pulse is set to a value (15 s) larger than the duration (i.e., simulation time = 10 s) for which the simulation will be run. The "impulse" is approximated by using a very short pulse duration by setting the duty cycle in the `Pulse Generator` block to 0.75%. The magnitude of the pulse can be set to some arbitrary number; we have chosen a value of unity in this case. A `Scope` block is positioned to display how L would behave as a function of "time" (which translates into iteration number, since this is a steady-state and not dynamic analysis). Since we are interested in the final steady-state operating point, we introduce an `XY Graph` block to plot L versus f_e at every iteration. Figure 3.7 shows the results of one simulation, using a total simulation duration of "10 s" and time-step of "0.01 s". In Figure 3.7a, L can be seen to start off at its initial condition of "1" and very rapidly converge to its steady-state solution of 0.58. In Figure 3.7b, where L is plotted against f_e, the solution begins at point (1,1) and follows a straight line trajectory to end at the final steady-state point where $f_e = 0.47$ and $L = 0.58$. This solution is consistent with the reference solution obtained graphically (see Figure 3.5).

In the sections that follow, we will perform steady-state analyses of three physiological control models. The purpose is not only to show further examples of the analysis procedures that we have been discussing but also to demonstrate to the reader that steady-state analysis can yield important insights into the integrative physiology of the system in question.

(a)

(b)
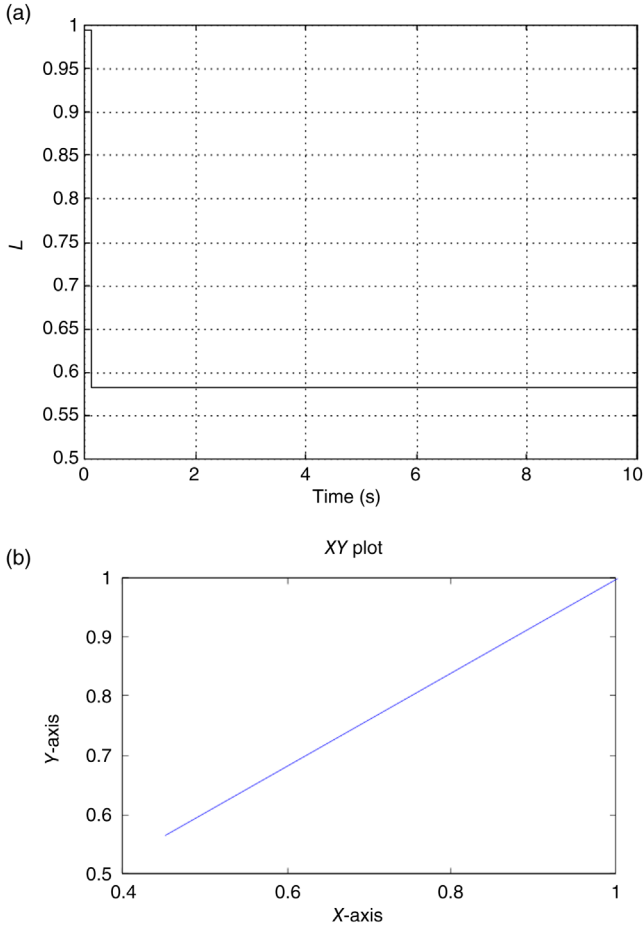
FIGURE 3.7 Sample simulation results from the SIMULINK model of Figure 3.6. (a) After short initial transient, L settles quickly to its steady-state value of 0.58. (b) L (y-axis) is plotted against f_e (x-axis). Starting from initial conditions at (1,1), the system heads for its final steady-state operating point at $f_e = 0.47$ and $L = 0.58$.

3.5   REGULATION OF CARDIAC OUTPUT

The fundamental notion underlying cardiac output regulation was best summarized by Patterson et al. (1914) in what is commonly called *Starling's law*. They stated in their classic paper that

"the output of the heart is equal to and determined by the amount of blood flowing into the heart, and may be increased or diminished within very wide limits according to the inflow."

In other words, in the steady state, the *venous return*, which is determined primarily by the mechanical properties of the systemic circulation, is always equal to the *cardiac output*, which is a function of many factors affecting the pumping ability of the heart.

3.5.1 The Cardiac Output Curve

The simplest possible model of the heart and systemic circulation is shown in Figure 3.8. In this simplified model, the component that we will label the "heart"
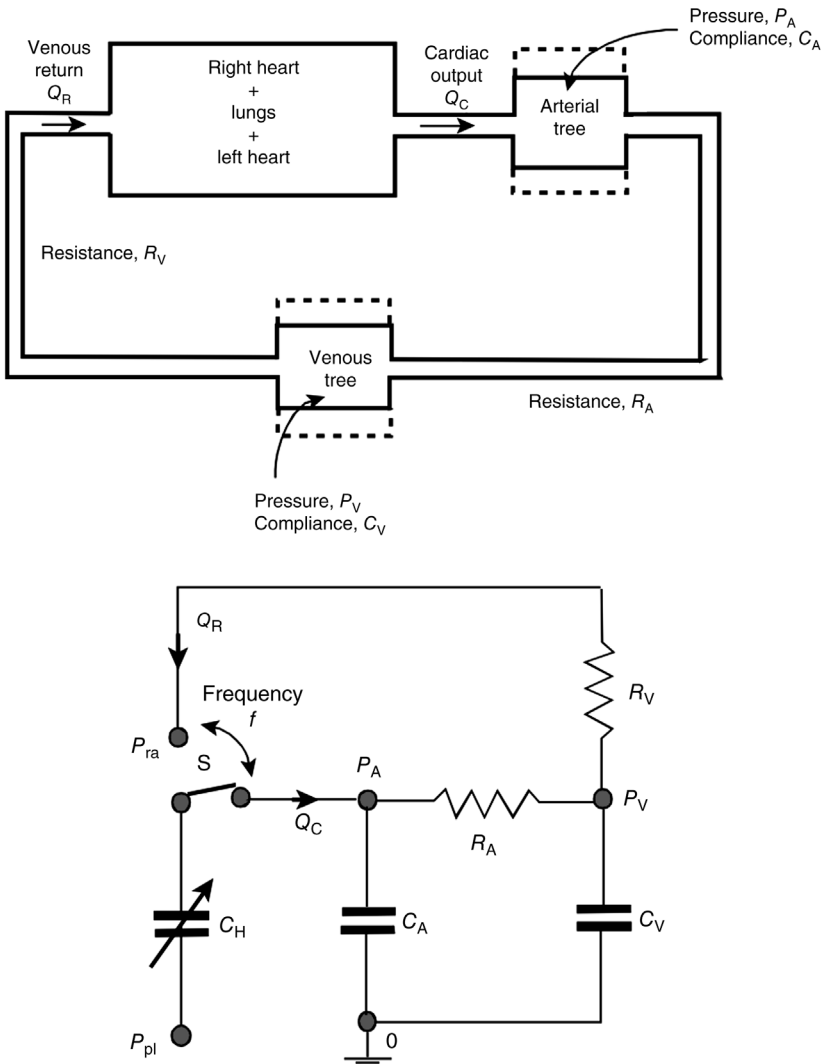
FIGURE 3.8 Simplified model of cardiac output regulation.

actually incorporates the combined functional characteristics of the right heart, the pulmonary circulation, and the left heart. The "heart" is modeled by assuming its capacitance C_H to vary between two levels. During diastole (the phase of ventricular relaxation), $C_H = C_D$; while during systole (ventricular contraction), $C_H = C_S$, where C_D is about an order of magnitude larger than C_S. During diastole, the heart model is connected to the venous side of the circuit, so that C_H ($= C_D$) is "charged up" by the filling pressure, which is equal to the right atrial pressure (referenced to atmospheric pressure), P_{ra}, minus the pleural pressure P_{pl} (which, in the intact subject, is negative relative to the atmosphere). Thus, at the end of diastole, the volume of blood in the heart would be

$$V_{HD} = C_D\left(P_{ra} - P_{pl}\right) \tag{3.23}$$

During systole, the switch S takes on its other position, connecting the variable capacitor to the arterial side of the circuit, allowing the capacitor to "discharge" into the systemic circulation. Therefore, at the end of systole, the volume of blood in the heart becomes

$$V_{HS} = C_S\left(P_A - P_{pl}\right) \approx C_S P_A \tag{3.24}$$

The approximation in Equation 3.24 is valid because P_A is much larger than P_{pl} in magnitude. The difference between the end-diastolic volume and the end-systolic volume is the amount of blood ejected in one beat, that is, the stroke volume SV:

$$SV = V_{HD} - V_{HS} = C_D\left(P_{ra} - P_{pl}\right) - C_S P_A \tag{3.25}$$

But the volume of blood pumped out in each beat multiplied by the number of beats that occurs per unit time (f), that is, the heart rate, must equal the cardiac output Q_C:

$$Q_C = f \cdot SV \tag{3.26}$$

Substituting Equation 3.25 into Equation 3.26, we obtain the following relationship:

$$Q_C = f C_D \left(P_{ra} - \frac{C_S P_A}{C_D} - P_{pl} \right) \tag{3.27}$$

Equation 3.27 states that cardiac output increases proportionally with right atrial pressure (known as "preload") but decreases with increasing arterial pressure (known as "afterload"). Also, note that, since Q_C cannot be negative, cardiac output becomes zero when

$$P_{ra} \leq \frac{C_S P_A}{C_D} + P_{pl} \tag{3.28}$$

Since $C_S \ll C_D$ and P_{pl} is negative, it follows that in the intact subject, cardiac output decreases to zero when P_{ra} takes on slightly negative values. This also means that even when $P_{ra} = 0$, there remains a substantial cardiac output. In Equation 3.27, Q_C increases linearly with P_{ra} without bound. This, of course, cannot be possible physiologically. The degree of diastolic filling is limited beyond a certain point by factors such as increasing stiffening of heart connective tissue and the size restrictions imposed by the pericardial sac. Thus, we impose a threshold limitation on Q_C so that

$$Q_C \leq Q_{C_{max}} = f C_D \left(P_{ra}^* - \frac{C_S P_A}{C_D} - P_{pl} \right) \tag{3.29}$$

where P_{ra}^* is the value of P_{ra} above which Q_C cannot increase any further. The plot of Q_C against P_{ra} is known as the *cardiac output curve* or *cardiac function curve*. Figure 3.9 displays the form of the cardiac output curve, as predicted by our simple model. Note that an increase in heart rate or sympathetic stimulation leads to an elevation in the slope of the curve as well as an increase in Q_{Cmax}. Conversely, parasympathetic stimulation, a decrease in heart rate, or the presence of myocardial damage due to heart disease leads to a reduction in slope and a decreased Q_{Cmax} (Figure 3.9a). Opening the chest wall eliminates the negative intrapleural pressure and therefore shifts the cardiac output curve to the right, without increasing Q_{Cmax}. On the other hand, breathing into a chamber held at negative pressure reduces intrapleural pressure further, thereby shifting the cardiac output curve to the left (Figure 3.9b). Again, in this case, Q_{Cmax} is not affected.

3.5.2 The Venous Return Curve

We now turn our attention to the systemic circulation part of the model. The resistance and capacitance of the arterial vasculature are modeled as lumped elements R_A and C_A, respectively. Similarly, we also lump the resistance and capacitance of the venous vasculature into the elements R_V and C_V, respectively. The total resistance of the capillaries is incorporated into R_A, while the capacitance of the capillaries is neglected since it is much smaller than C_A or C_V. We first consider the hypothetical situation where there is no blood flow. Under this condition, pressures throughout the systemic circulation would be equilibrated. However, due to the finite capacitance of the combined vasculature, this equilibrium pressure would not be zero but would take on a positive value ($\sim$7 mmHg). This is called the *mean systemic pressure* (P_{ms}) or *mean circulatory pressure*. If the volume of blood in the arterial vasculature is V_A and the volume of blood in the venous vasculature is V_V, then

$$P_{ms} = \frac{V_A + V_V}{C_A + C_V} \tag{3.30}$$
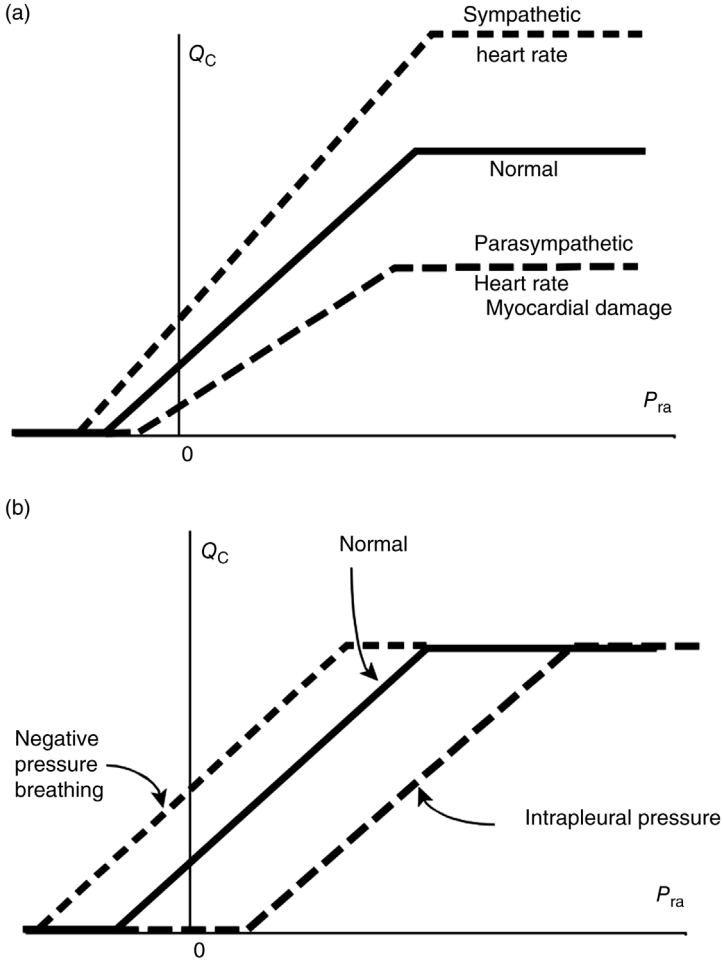
FIGURE 3.9 Cardiac output curves. (a) Factors that affect slope and position. (b) Factors that affect only position.

Now, consider the situation where blood is flowing through the systemic circulation at the volumetric rate Q_R. Under steady-state conditions, the arterial pressure P_A and venous pressure P_V will be related to Q_R through Ohm's law (see Figure 3.8):

$$P_A = Q_R(R_A + R_V) + P_{ra} \tag{3.31}$$

$$P_V = Q_R R_V + P_{ra} \tag{3.32}$$

However, since V_A and V_V remain the same regardless of whether there is blood flow or not, these volumes will be related to P_A and P_V, respectively, through the

following equations:

$$V_V = C_V P_V \tag{3.33}$$

$$V_A = C_A P_A \tag{3.34}$$

Substituting Equation 3.31 through 3.34 into Equation 3.30 yields the result:

$$P_{ms} = \frac{C_A Q_R (R_A + R_V) + C_A P_{ra} + C_V Q_R R_V + C_V P_{ra}}{C_A + C_V} \tag{3.35a}$$

Rearranging terms in Equation 3.35a, we obtain the following expression that relates Q_R to P_{ra}:

$$Q_R = \frac{P_{ms} - P_{ra}}{R_V + \dfrac{R_A C_A}{C_A + C_V}} \tag{3.35b}$$

One should be reminded that Equation 3.35b describes how Q_R would vary with P_{ra} in the systemic circulation only: This is equivalent to a hypothetical situation in which we have "disconnected" the heart from the systemic circulation and where we are now concerned only with the input–output characteristics of the latter. The significance of Equation 3.35b is that it tells us what the cardiac output would be, given the mechanical properties of the systemic circulation, the total blood volume (which determines P_{ms}), and right atrial pressure. We could also use Equation 3.31 to make this prediction, but we would need to know arterial blood pressure also. For example, if we wanted to know how Q_R would change if R_A were doubled, it would be more difficult to deduce the answer from Equation 3.31, since P_A would also be changed. On the other hand, in Equation 3.35b, P_{ms} is independent of this change; so if we know P_{ra} and the other mechanical properties of the circulation, Q_R can be determined simply.

The *venous return curve*, described by Equation 3.35b and illustrated in Figure 3.10, shows that Q_R varies linearly with P_{ra} but with a negative slope, so that as P_{ra} becomes more positive, Q_R decreases. However, the range of P_{ra} over which Equation 3.35b remains valid is limited. When P_{ra} becomes equal to or higher than P_{ms}, no flow occurs. At the other end of the spectrum, when P_{ra} decreases to approximately −4 mmHg or below, Q_R does not increase any further. This is due to the collapse of the veins in the thoracic cavity when the intramural pressures become lower than intrathoracic pressure. The slope of the linear part of the venous return curve is a function of the mechanical properties of the circulation. Systemic vasoconstriction, which increases peripheral resistance, lowers this slope, while vasodilation increases it (Figure 3.10a). The effect of an arteriovenous (A-V) fistula, which represents a "short-circuiting" of the systemic circulation, is to produce a large increase in slope of the venous return curve. On the other hand, factors that alter P_{ms} act only to shift the venous return curve to the right (increased P_{ms}) or left
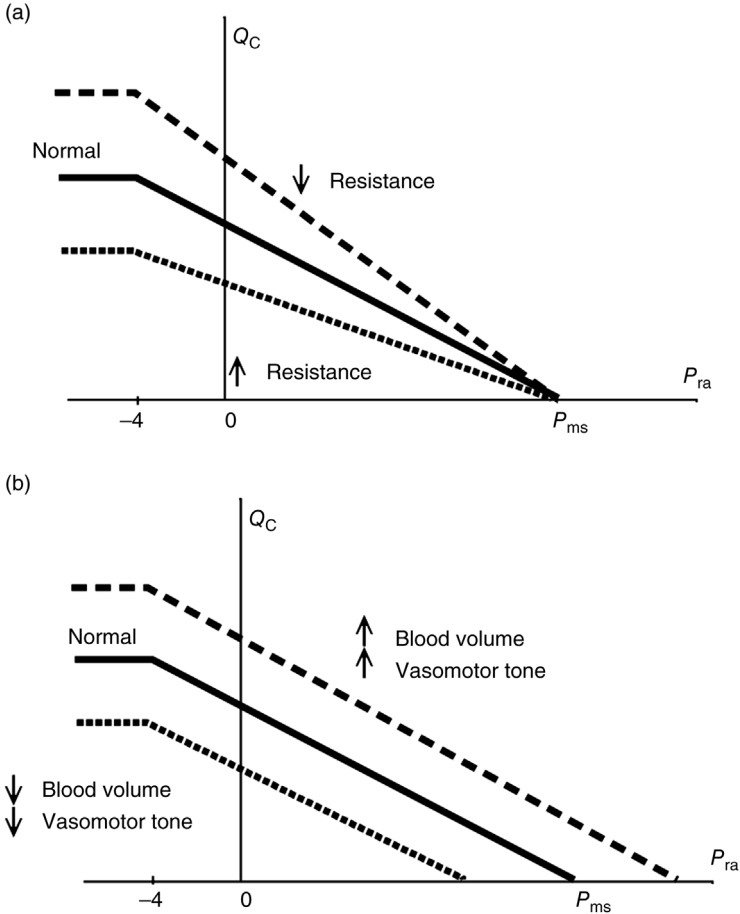
FIGURE 3.10 Venous return curves. (a) Factors that affect slope. (b) Factors that affect position.

(decreased P_{ms}) without altering its slope (Figure 3.10b). An increase in total blood volume (e.g., due to transfusion) would raise P_{ms} and thus shift the venous return curve to the right, while a decrease in blood volume due to hemorrhage would have the opposite effect. Increased vasomotor tone, which decreases the arterial and venous compliances, would also raise P_{ms}.

Equation 3.35b points to an important model prediction that initially may appear counterintuitive. Since C_V is generally about 18 times as large as C_A, Equation 3.35b may be approximated by

$$Q_R = \frac{P_{ms} - P_{ra}}{R_V + \dfrac{R_A}{19}} \qquad (3.36)$$

The above equation implies that a change in R_V would have a much larger effect on venous return than a change in R_A of the same magnitude, whereas one would generally think otherwise, since R_A is much larger than R_V. The reason for this apparent paradox is that when venous resistance is raised, P_V does not increase much because of the large venous capacitance; thus, the increase in driving pressure is small compared to the increase in resistance and, consequently, blood flow is dramatically reduced. On the other hand, when R_A is increased, P_A also increases substantially because of the relatively small arterial compliance. As a result, venous return is not decreased as much. This kind of initially counterintuitive result is encountered frequently in modeling and underscores the fact that, when several variables are involved in a problem, only the systematic approach inherent in a mathematical model will allow us to make predictions that are consistent with our underlying assumptions.

3.5.3 Closed-Loop Analysis: Heart and Systemic Circulation Combined

Starling's law is the consequence of connecting the "heart" and "systemic circulation" components of the model together and allowing the system to operate in closed-loop mode (Guyton et al., 1973). We assume the following parameter values, which have been chosen so that the model provides a first approximation to the human cardiovascular system under normal resting conditions: $f = 72$ beats $\min^{-1}$, $C_D = 0.035\,\mathrm{L\,mmHg^{-1}}$, $C_S = 0.0007\,\mathrm{L\,mmHg^{-1}}$, $P_{pl} = -4\,\mathrm{mmHg}$, $R_A = 19.2\,\mathrm{mmHg\,min\,L^{-1}}$, $R_V = 0.4\,\mathrm{mmHg\,min\,L^{-1}}$, $C_A = 0.028\,\mathrm{L\,mmHg^{-1}}$, $C_V = 0.5\,\mathrm{L\,mmHg^{-1}}$, $P_{ms} = 7\,\mathrm{mmHg}$, $P_A = 100\,\mathrm{mmHg}$. Under such conditions, the cardiac output and venous return curves are as shown in Figure 3.11a. The intersection between the two curves yields the steady-state operating point labeled N. This is established at a cardiac output of $5\,\mathrm{L\,min^{-1}}$ and P_{ra} of $0\,\mathrm{mmHg}$.

Using this model, can we predict what cardiac output would be during *moderate exercise*? At the onset of exercise, the tensing of the muscles involved plus an increase in venomotor tone produces a decrease in venous compliance, thereby raising P_{ms}. Sympathetic stimulation leads subsequently to an increase in heart rate, which elevates the slope of the cardiac output curve and increased vasomotor tone. Then, local vasodilation of the muscular vascular beds produces a marked decrease in peripheral resistance. In this example, we have assumed C_A, C_V, R_A, and R_V each to decrease by 40% and f to increase by 40%. These changes affect the cardiac output and venous return curves in the manner shown in Figure 3.11b, with the new condition being represented in the form of dashed curves. The new steady-state cardiac output is now increased to $\sim 10.5\,\mathrm{L\,min^{-1}}$, twice the resting value, while P_{ra} remains relatively unchanged (point E). These predictions are consistent with empirical evidence. Note, however, that the bulk of the increase in cardiac output has come about as a result of changes in the systemic circulation: If the latter were to remain at its original state, cardiac output would increase only fractionally by less than $1\,\mathrm{L\,min^{-1}}$ (point E^*). This somewhat unexpected result is yet another excellent example of how mathematical modeling can lead us to a conclusion that would have been difficult to predict otherwise.
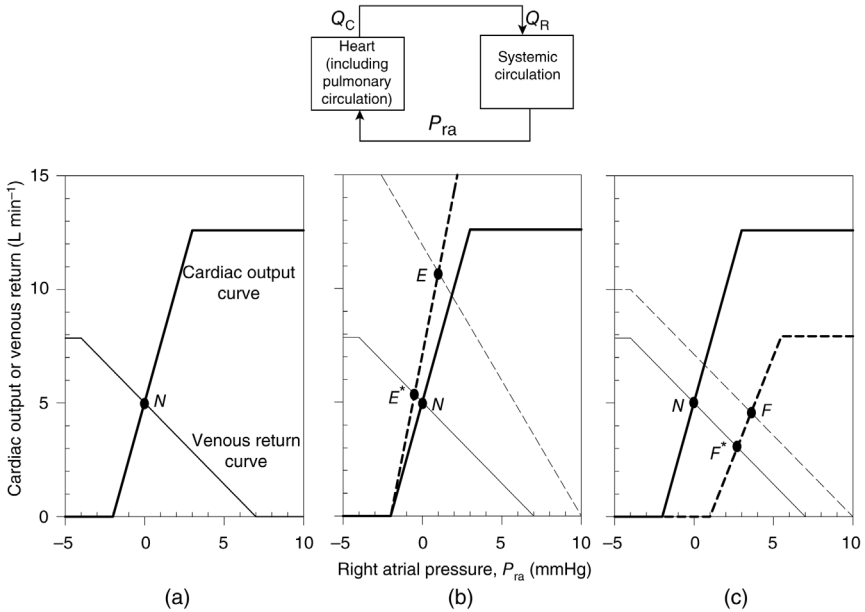
FIGURE 3.11 Steady-state closed-loop analysis of cardiac output regulation during (a) normal resting conditions, (b) moderate exercise, and (c) compensated heart failure.

In Figure 3.11c, the model is used to predict the steady-state values of cardiac output and P_{ra} in heart failure following *myocardial infarction*. To represent the reduction in effectiveness of the heart as a pump, we assume that C_S is increased and C_D is decreased by 30%. Thus, the ratio C_S/C_D is increased from 0.02 to 0.05. According to Equation 3.27, this decreases the slope of the cardiac output curve and shifts zero-flow intercept to the right by approximately 3 mmHg. If this were the only effect of heart failure, cardiac output would decrease by 40% to about $3\,L\,min^{-1}$ (point F^*). At the same time, P_{ra} would rise by ~3 mmHg. Fortunately, the body generally compensates for this decreased cardiac output by reducing urine output and thus retaining body fluid. This raises blood volume so that P_{ms} increases, thereby shifting the venous return curve to the right. The net effect of these compensatory changes, which usually occur over a week, is to restore cardiac output back toward its normal level. In this example, the steady-state cardiac output following compensation falls just short of $5\,L\,min^{-1}$. P_{ra} is now ~4 mmHg higher than normal.

3.6 REGULATION OF GLUCOSE INSULIN

We turn to another example of a physiological control system with negative feedback: The system that regulates blood glucose levels. When plasma glucose levels are elevated, insulin secretion is stimulated. This raises the level of insulin in
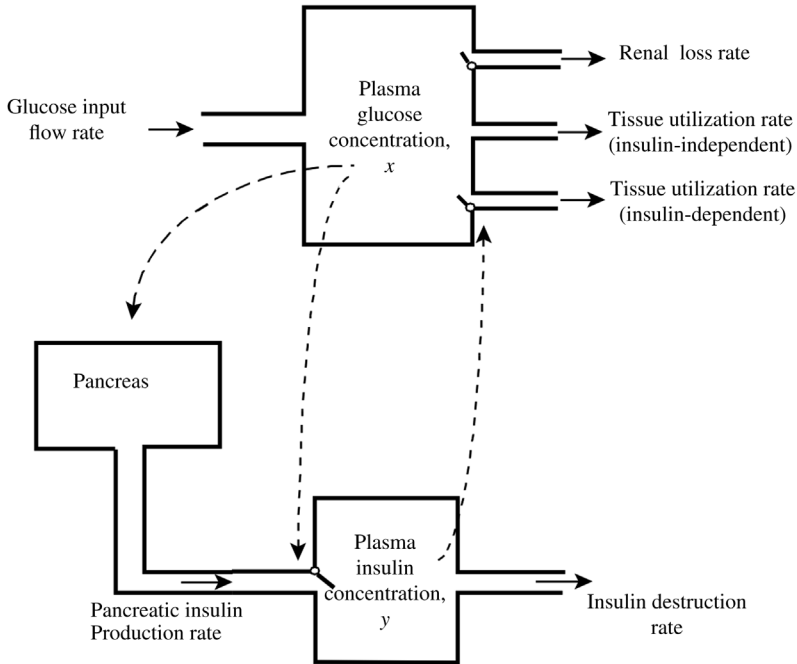
FIGURE 3.12   Schematic representation of the processes involved in the regulation of glucose and insulin.

the blood, which increases the uptake of blood glucose by the tissues. The increased outflow of glucose from the blood and interstitial fluid leads to a decrease in glucose concentration, which subsequently produces a reduction in insulin secretion.

The model we will introduce in this section was first proposed by Stolwijk and Hardy (1974). The model assumes that the total volume of blood and interstitial fluids is represented by a single large compartment, and that the steady-state concentration of glucose in this compartment is x (in units of mg mL^{-1}). For this level of x to remain constant, the total inflow of glucose into the compartment must equal the total outflow from the compartment. Figure 3.12 shows a schematic representation of the main processes that affect this balance. Under normal circumstances, glucose enters the blood through absorption from the gastrointestinal tract or through production from the liver. We assume this input flow rate to be Q_L (in mg h^{-1}). There are three major ways through which glucose is eliminated from the blood:

a) When x is elevated beyond a certain threshold (θ), glucose is excreted by the kidneys at a rate proportional to the gradient between x and θ:

$$\text{Renal loss rate} = \mu(x - \theta), \quad x > \theta \qquad (3.37a)$$

$$\text{Renal loss rate} = 0, \quad x \leq \theta \qquad (3.37b)$$

b) Glucose leaves the blood to enter most cells through facilitated diffusion. In some tissues, the rate of glucose utilization depends only on the extracellular-to-intracellular concentration gradient. In most circumstances, we can ignore the intracellular concentration. Thus, we have

$$\text{Tissueutilizationrate (insulin-independent)} = \lambda x \qquad (3.38)$$

c) In certain types of cells, such as those in muscle and adipose tissue, insulin helps to stimulate this facilitated diffusion process. Therefore, the rate at which glucose is taken up by these cells is proportional to x as well as to the blood insulin concentration y:

$$\text{Tissueutilizationrate (insulin-dependent)} = vxy \qquad (3.39)$$

In the above equations, μ, λ, and ν are constant proportionality factors.

Equating the inflow to the sum of the three outflows, we obtain the following mass balance equations for blood glucose:

$$Q_L = \lambda x + vxy, \quad x \le \theta \qquad (3.40a)$$

$$Q_L = \lambda x + vxy + \mu(x - \theta), \quad x > \theta \qquad (3.40b)$$

Note that in the above equation, a strong *nonlinearity* in the form of the product of x and y is introduced, along with the *thresholding nonlinearity* that defines one regime above θ and one below it. Also, the negative feedback in this control system is clearly embedded in the characteristics described by Equations 3.40a and 3.40b: Since Q_L is a constant, an increase in x must lead to a corresponding decrease in y, and vice versa.

A similar mass balance can be established for blood insulin. Insulin is produced by the pancreas at a rate dependent on the plasma glucose level. However, if x falls below a certain threshold (φ), insulin production ceases. Thus, we have

$$\text{Insulinproductionrate} = 0, \quad x \le \varphi \qquad (3.41a)$$

$$\text{Insulinproductionrate} = \beta(x - \varphi), \quad x > \varphi \qquad (3.41b)$$

Insulin is destroyed through a reaction involving the insulinase enzyme, at a rate proportional to its concentration in blood:

$$\text{Insulindestructionrate} = \alpha y \qquad (3.42)$$

Combining Equations 3.41 and 3.42, we obtain the following equations relating the steady-state level of y to that of x:

$$y = 0, \quad x \le \varphi \qquad (3.43a)$$

$$y = \frac{\beta}{\alpha}(x - \varphi), \quad x > \varphi \qquad (3.43b)$$

Therefore, aside from the threshold nonlinearity, the insulin response to glucose is basically linear.

The steady-state level of glucose and insulin in the blood under a given set of conditions can be predicted from this model by solving Equations 3.40 and 3.43 simultaneously. As we have shown in the regulation of cardiac output example, graphical analysis is useful in providing not only the steady-state solution but also substantial insight into the overall problem. In Figure 3.13a, the steady-state insulin concentration (in milliUnits per mL blood) is plotted against the steady-state blood glucose concentration (in mg mL^{-1}). The insulin response to glucose is shown as
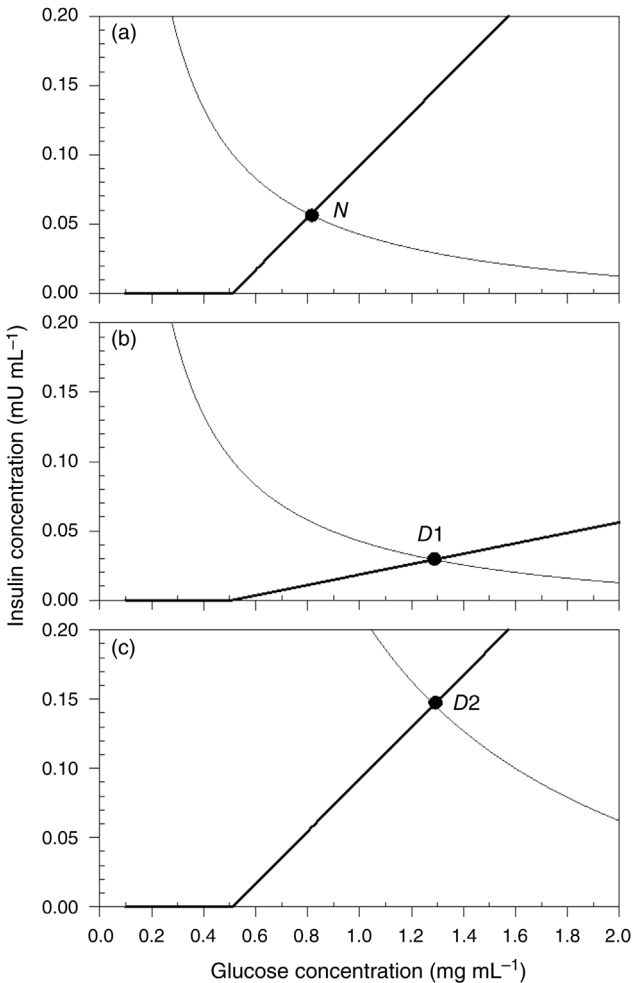
FIGURE 3.13 Steady-state analysis of glucose regulation under (a) normal conditions, (b) type-1 diabetes, and (c) type-2 diabetes.

the bold curve, while the lighter curve reflects the glucose mass balance equation. The parameter values employed in this calculation correspond to the normal adult: $\theta = 2.5\,\mathrm{mg\,mL^{-1}}$, $\mu = 7200\,\mathrm{mL\,h^{-1}}$, $\lambda = 2470\,\mathrm{mL\,h^{-1}}$, $\nu = 139{,}000\,\mathrm{mU^{-1}\,h^{-1}}$, $\varphi = 0.51\,\mathrm{mg\,mL^{-1}}$, $\beta = 1430\,\mathrm{mU\,ml\,mg^{-1}\,h^{-1}}$, $\alpha = 7600\,\mathrm{mL\,h^{-1}}$, and $Q_\mathrm{L} = 8400$ $\mathrm{mg\,h^{-1}}$. The intersection of the glucose and insulin curves yields the steady-state operating point labeled N, where the glucose concentration is $0.81\,\mathrm{mg\,mL^{-1}}$ and the insulin concentration is $0.055\,\mathrm{mU\,mL^{-1}}$.

The model is used next to predict the steady-state operating levels of glucose and insulin that would arise from diabetes. In *type-1* or *insulin-dependent diabetes*, the main defect is in the inability of the islet cells in the pancreas to produce sufficient insulin. The most common form of this disorder begins in childhood and, for this reason, is frequently called *juvenile-onset* diabetes. The other form begins in adulthood and is known as *ketone-prone* diabetes. We can model this condition by lowering the sensitivity of the insulin response to glucose, β. Figure 3.13b demonstrates the effect of reducing β to 20% of its normal value. The new steady-state operating point is now established at $D1$, resulting in a highly elevated blood glucose concentration of $1.28\,\mathrm{mg\,mL^{-1}}$ and a depressed plasma insulin concentration of $0.029\,\mathrm{mU\,mL^{-1}}$.

Type-2 diabetes is also referred to as *noninsulin-dependent* diabetes, since here the pancreas may be making normal amounts of insulin. However, for reasons that remain unclear, there is a drastic reduction in the ability of insulin to stimulate glucose uptake by the body tissues. We model this condition by changing the value of the parameter ν, which is the constant that multiplies the product of x and y in the glucose mass balance equation. The insulin response to glucose may remain normal or may decrease. In Figure 3.13c, however, we have reduced ν to 20% of its original value while leaving the insulin curve unchanged. This change produces a shift of the glucose curve away from the origin as well as a steepening in local slopes. The new equilibrium point is established at $D2$, where the glucose concentration is elevated to $1.29\,\mathrm{mg\,mL^{-1}}$. A somewhat counterintuitive result is that the steady-state insulin concentration now is actually almost three times higher than normal, at a level of $0.146\,\mathrm{mU\,mL^{-1}}$. Thus, in this case, treatment with insulin clearly would not be useful.

3.7   CHEMICAL REGULATION OF VENTILATION

The final example that we will consider in this chapter is the chemoreflex regulation of respiration. In normoxic conditions, breathing is controlled almost exclusively by the level of CO_2 in the arterial blood. In fact, ventilation is highly sensitive to P_aCO2, the partial pressure of CO_2 in arterial blood. A rise in P_aCO2 by 1 mmHg from its normal level of approximately 40 mmHg may increase the ventilatory output by a third of its resting level. However, upon ascent to altitude or during inhalation of a gas mixture containing low O_2 content, there is an additional drive to breathe due to hypoxia. This hypoxic drive becomes noticeable when the partial pressure of O_2 in

arterial blood, P_{aO2}, drops below 70 mmHg. Since the metabolic consumption rate of O_2 and the metabolic elimination rate of CO_2 are relatively constant in the steady state, a higher level of ventilation would lead to an increase in P_{aO2} and a decrease in P_{aCO2}, which in turn would lower ventilation. Therefore, the "negative" part of this negative-feedback system is embedded in the gas exchange characteristics of the lungs. The simple model that we will analyze is depicted in block diagram form in Figure 3.14a. The ventilatory control system is divided into two components: the gas exchanging portion and the respiratory controller. An important distinction between this model and the previous models that we have analyzed is that the
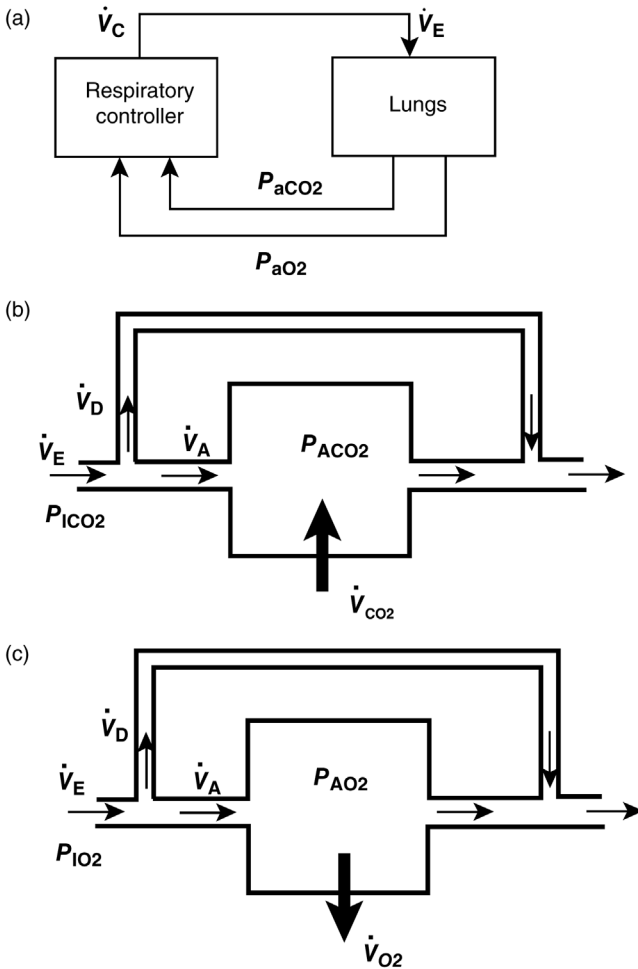
FIGURE 3.14 (a) Steady-state model of the chemical regulation of ventilation. (b) Model of steady-state CO_2 exchange in the lungs. (c) Model of steady-state O_2 exchange in the lungs.

components are either dual-input-single-output (controller) or single-input-dual-output (lungs) systems.

3.7.1 The Gas Exchanger

The gas exchanging component involves a combination of many processes that take place in the lungs, vasculature, and body tissues. However, as a first approximation, we will restrict our attention only to gas exchange occurring in the lungs. The operating characteristics of the gas exchanger are obtained by deriving the mass balance equations for CO_2 and O_2. We begin by considering CO_2 exchange, which is depicted schematically in Figure 3.14b. We assume the metabolic CO_2 production rate to be $\dot{V}_{CO_2}$; this is the rate at which CO_2 is delivered to the lungs from the blood that is perfusing the pulmonary circulation. In the steady state, this must equal the *net* flow of CO_2 exiting the lungs in gas phase. The latter is equal to the difference in volumetric fraction (or concentration) of CO_2 in the air entering (F_{ICO2}) and leaving (F_{ACO2}) the alveoli multiplied by the *alveolar ventilation* $\dot{V}_A$. The alveolar ventilation represents that portion of the *total ventilation*, $\dot{V}_E$, that actually participates in the gas exchange process. Part of $\dot{V}_E$ is "wasted" on ventilating the non-gas-exchanging airways in the lungs; this flow is known as "dead space ventilation," $\dot{V}_D$. Thus, we have

$$\dot{V}_A = \dot{V}_E - \dot{V}_D \tag{3.44}$$

and the CO_2 mass balance:

$$\dot{V}_{CO_2} = k\dot{V}_A \left(F_{ACO_2} - F_{ICO_2} \right) \tag{3.45}$$

In Equations 3.44 and 3.45, the ventilatory flow rates are generally measured in *BTPS* (*body temperature pressure saturated*) units, while the CO_2 metabolic production rate is usually expressed in *STPD* (*standard temperature pressure dry*, that is, at 273 K and 760 mmHg) units. The constant k allows volumes and flows measured in BTPS units to be converted into STPD units. This conversion is achieved by using the ideal gas equation:

$$\frac{V_{STPD}760}{273} = \frac{V_{BTPS}(P_B - 47)}{310} \tag{3.46a}$$

The above equation assumes body temperature to be 37 °C or 310 K and a saturated water vapor partial pressure of 47 mmHg at that temperature. P_B represents the barometric pressure under which the gas exchange process is taking place; at sea level, this is 760 mmHg, but the value decreases with ascent to high altitude. Upon rearranging Equation 3.46a, we obtain the following expression for k:

$$k = \frac{V_{STPD}}{V_{BTPS}} = \frac{P_B - 47}{863} \tag{3.46b}$$

The volumetric fractions F_{ICO2} and F_{ACO2} can be converted into their corresponding partial pressures P_{ICO2} and P_{ACO2} using Dalton's law:

$$P_{ICO_2} = F_{ICO_2}(P_B - 47) \tag{3.47a}$$

$$P_{ACO_2} = F_{ACO_2}(P_B - 47) \tag{3.47b}$$

Therefore, using Equations 3.46b, 3.47a, and 3.47b in Equation 3.45 yields the following result:

$$P_{ACO_2} = P_{ICO_2} + \frac{863\dot{V}_{CO_2}}{\dot{V}_A} \tag{3.48}$$

Equation 3.48 shows a hyperbolic relation between P_{ACO2} and $\dot{V}_A$, and for this reason it is commonly referred to as the *metabolic hyperbola*. By employing the same kind of mass balance analysis (see Figure 3.14c), a similar "metabolic hyperbola" can be deduced for O_2:

$$P_{AO_2} = P_{IO_2} - \frac{863\dot{V}_{O_2}}{\dot{V}_A} \tag{3.49}$$

The negative sign in Equation 3.49 accounts for the fact that O_2 is removed from the lungs by the perfusing blood and, therefore, the alveolar O_2 content (P_{ACO2}) will always be lower than the inhaled O_2 content (P_{IO2}).

A further assumption that we will make in this model is that the alveolar partial pressures are completely equilibrated with the corresponding arterial blood gas partial pressures, that is,

$$P_{aCO_2} = P_{ACO_2} \tag{3.50a}$$

$$P_{aO_2} = P_{AO_2} \tag{3.50b}$$

This is approximately true in normals, although for O_2 there is an alveolar–arterial gradient of 5 mmHg or more. However, in patients with lung disease, ventilation–perfusion mismatch can give rise to rather substantial gradients between the alveolar and arterial partial pressures.

Apart from the shared value of $\dot{V}_A$, Equations 3.48 and 3.49 appear to suggest that CO_2 and O_2 exchange are independent of each other. This, however, is a consequence of limiting our considerations only to the exchange processes that occur in gas phase. For more realistic modeling, it is essential to incorporate the blood–gas dissociation relationships for CO_2 and O_2, as well as considerations of gas exchange at the level of the body tissues. For instance, CO_2 affects the affinity with which O_2 is bound to hemoglobin (Bohr effect), and the level of oxygenation affects the blood CO_2 concentration at any given partial pressure (Haldane effect).

At the level of cellular metabolism, the rate at which CO_2 is produced for a given O_2 consumption rate depends on the type of nutrient being oxidized. Fortunately, the effects of these complications on the final predictions of the alveolar or arterial partial pressures are not very large.

3.7.2   The Respiratory Controller

The controller part of the system includes chemoreceptors, neuronal circuits in the lower brain involved in the generation of the respiratory rhythm as well as the neural drive to breathe, and respiratory muscles. The controller response to CO_2 has been shown to be linear over the physiological range. In the absence of vigilance, such as during sleep, the controller output falls rapidly to zero (i.e., central apnea occurs) when P_{aCO_2} decreases slightly below normal awake resting levels. Exposure to hypoxia (i.e., when P_{aO_2} decreases below $100\,mmHg$) leads to an increase in the CO_2 response slope as well as in the ventilatory controller output. Hence, there is a strong interaction between CO_2 and O_2 at the level of the controller. Cunningham (1974) modeled the ventilatory controller output ($\dot{V}_C$) as the sum of an O_2-independent term and a term in which there is a multiplicative interaction between hypoxia and hypercapnia:

$$\dot{V}_C = \left(1.46 + \frac{32}{P_{aO_2} - 38.6}\right)\left(P_{aCO_2} - 37\right), \quad P_{aCO_2} > 37 \tag{3.51a}$$

$$\dot{V}_C = 0, \quad P_{aCO_2} \leq 37 \tag{3.51b}$$

Note that the above expression becomes progressively less valid as P_{aO2} approaches the asymptote value of 38.6, in which case $\dot{V}_C$ would become infinitely large. As pointed below, precautions have to be taken to ensure that P_{aO2} does not fall below a physiologically realistic range.

3.7.3   Closed-Loop Analysis: Lungs and Controller Combined

In the closed-loop situation, the controller output $\dot{V}_C$ would equal the ventilation $\dot{V}_E$, driving the gas exchange processes for CO_2 and O_2, as shown in Figure 3.14a. To obtain the steady-state operating point for the closed-loop system, Equations 3.48 through 3.51 must be solved simultaneously. As we have done previously, it is possible to arrive at the solution through graphical analysis. However, since three variables ($\dot{V}_E$, P_{ACO2}, and P_{AO2}) are involved, both graphical and algebraic methods of solution can be quite laborious. Thus, in this case, we resort to a numerical approach using SIMULINK.

Figure 3.15 displays the layout of the SIMULINK model file `respss.slx` that allows the solution of the steady-state ventilatory control equations. Basically, the program simulates the closed-loop system in "open-loop mode." A `Repeating Sequence` block (labeled "`VdotEin Input Ramp`") is used to generate a linearly

FIGURE 3.15 SIMULINK program `respss.slx` to determine the steady-state operating point of the ventilatory control system.

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increasing sequence of $\dot{V}_E$ values. Each $\dot{V}_E$ value is fed into Equations 3.48 and 3.49 so that corresponding P_{ACO2} and P_{AO2} values are generated. Each pair of P_{ACO2} and P_{AO2} values is subsequently used in Equations 3.50 and 3.51 to generate the corresponding ventilatory controller output, $\dot{V}_C$ (labeled "VdotEout" in Figure 3.15). The initially low $\dot{V}_E$ values would produce high P_{ACO2} and low P_{AO2} levels, which would act on the controller to produce high $\dot{V}_C$ values. However, as $\dot{V}_E$ increases, chemical drive levels would decrease, in turn, decreasing $\dot{V}_C$. The steady-state equilibrium point is established at that combination of P_{ACO2} and P_{AO2} values where $\dot{V}_E$ level becomes equal to $\dot{V}_C$. A relational operator block is incorporated to check for this condition and to stop the simulation when the condition is satisfied. The steady-state values of $\dot{V}_E$, P_{ACO2}, and P_{AO2} are saved to the MATLAB workspace in the scalar variables "vent", "paco2," and "pao2," respectively. An important point to note is that we included a "saturation block to limit the allowable range for P_{AO2}. This ensures that P_{AO2} would not fall to a point where the O_2-dependent term in the controller becomes infinite or negative,
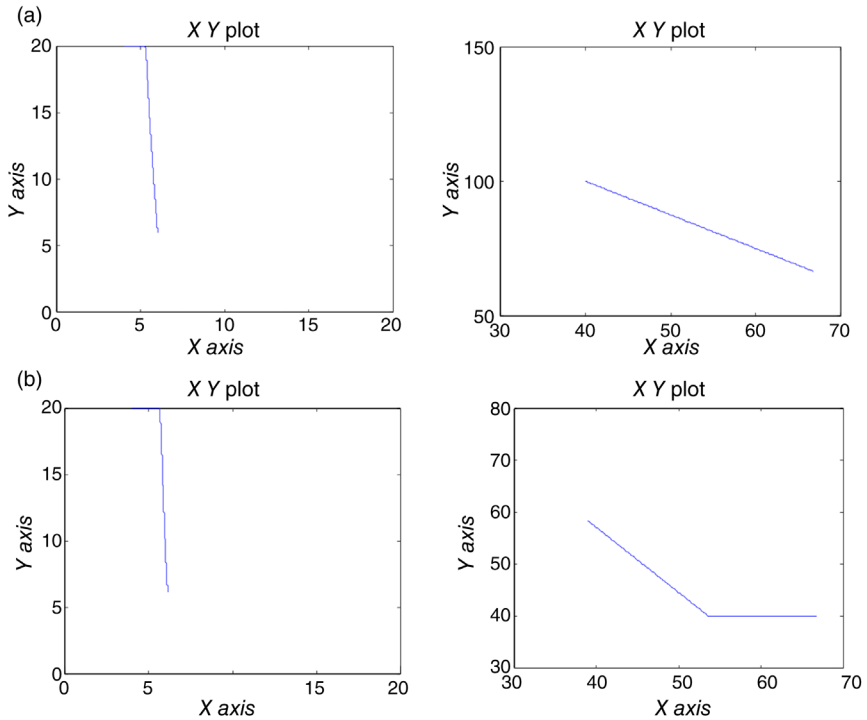
FIGURE 3.16   Results of SIMULINK simulations to determine the steady-state operating point during (a) normoxia ($P_{IO2} = 150$ mmHg) and (b) inhalation of 15% O_2 mixture ($P_{IO2} = 107$ mmHg). *Left panels:* Ventilatory controller output versus ventilation in L min^{-1} (simulation is terminated when they become equal). *Right panels:* Corresponding trajectory of P_{AO2} versus P_{ACO2} in mmHg.

The results of two SIMULINK simulations are shown in Figure 3.16. In case A, P_{IO2} is set equal to 150 mmHg (i.e., 21% room air), while P_{ICO2} is set equal to 0. Due to the initially low $\dot{V}_E$ value, P_{ACO2} and P_{AO2} are initially ~67 and ~65 mmHg, respectively, while $\dot{V}_C$ is higher than 20 L min^{-1}. As $\dot{V}_E$ increases, P_{ACO2} decreases, while P_{AO2} rises and $\dot{V}_C$ falls. The simulation is terminated when $\dot{V}_C$ becomes equal to $\dot{V}_E$. This occurs at $\dot{V}_E = \dot{V}_C = 6$ L min^{-1}, $P_{ACO2} = 40$ mmHg, and $P_{AO2} = 100$ mmHg. In case B, we simulate a subject inhaling a gas mixture containing only 15% O_2 or, equivalently, a subject ascending to an altitude of 8500 ft. Thus, P_{IO2} is set equal to 107 mmHg, while P_{ICO2} is left at 0. As before, the initial value of P_{ACO2} is in the high 60s, while P_{AO2} is consistent with a value lower than 40 mmHg. However, due to the effect of the saturation block, P_{AO2} is not allowed to fall below 40 mmHg. The final equilibrium point is established at $\dot{V}_E = 6.1$ L min^{-1}, $P_{ACO2} = 39$, and $P_{AO2} = 58.3$. These two examples demonstrate quite clearly the negative feedback nature of respiratory control. Although exposure to hypoxia tends to produce an additional drive to breathe, the added ventilation blows off CO_2 and, consequently, the lower P_{ACO2} acts to offset the hypoxic-induced drive. As a result, ventilation remains close to its original normoxic level.

The equivalent graphical analyses of cases A and B are presented in Figure 3.17a and b. The controller responses are depicted as bold curves, while the gas exchange responses are shown as light curves. The steady-state operating points for normoxia and hypoxia are labeled N and H, respectively. The two-dimensional plots do not provide a good sense of the three-dimensional nature of the problem.
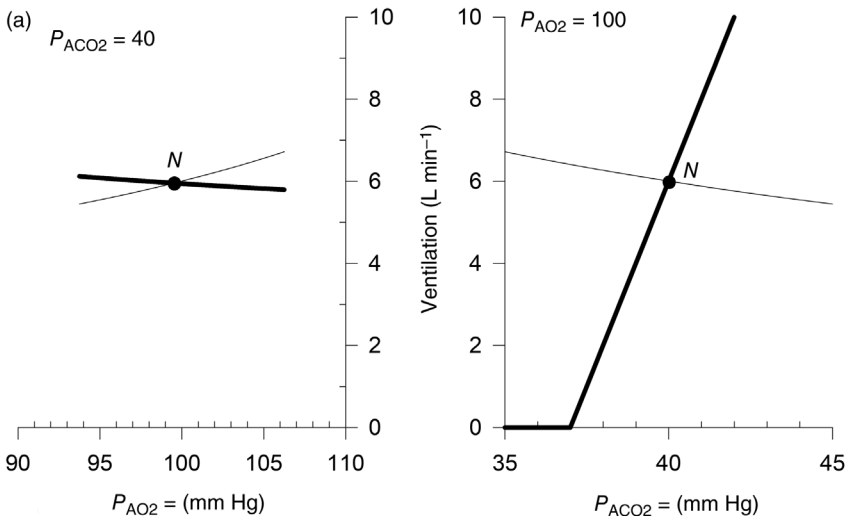
FIGURE 3.17 Graphical analysis of the steady-state regulation of ventilation during (a) normoxia ($P_{IO2} = 150$ mmHg) and (b) exposure to mild hypoxia through inhalation of 15% O_2 mixture or ascent to altitude (~8500 ft). The steady-state operating points are labeled N in case (a) and H in case (b).

(b)
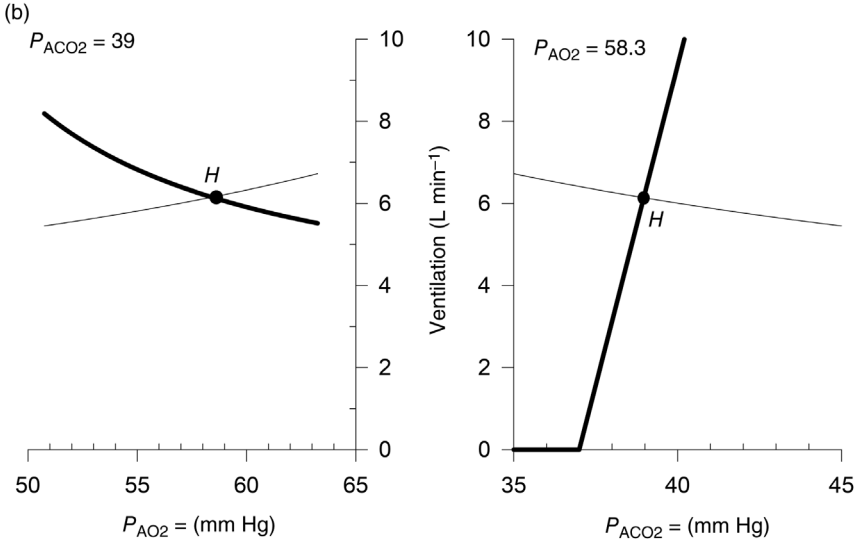
FIGURE 3.17   (*Continued*)

For instance, the controller plot shown in the ventilation–P_{ACO2} graph in case A represents only the P_{AO2} value of 100 mmHg. Similarly, the controller plot shown in the ventilation–P_{AO2} graph in case A assumes P_{ACO2} to be 40 mmHg. The same comments apply to the graphs in case B.

PROBLEMS

P3.1. Assume that the block diagram of a temperature-regulating space-suit to be worn by an astronaut for a mission to Mars is as shown in Figure P3.1. The variable x represents the external temperature, while y represents the temperature inside the space-suit. G_C is the steady-state gain of the heating/
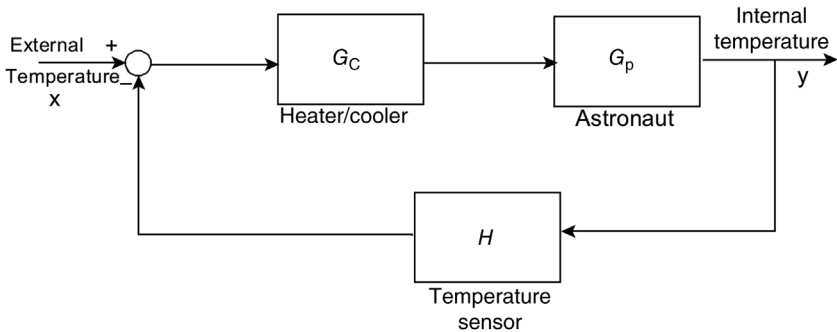
FIGURE P3.1   Block diagram of the temperature control system of a space-suit.

cooling device (controller) built into the space-suit, while G_P represents the steady-state gain associated with the thermal characteristics of the astronaut. H is the gain with which the internal temperature is fed back to the controller. The operating internal temperature (y) is allowed to range from 60 to 100°F.
   Assume that $G_C = 2$, $G_P = 1$, and $H = 7$.

   (a) What range of external temperatures can this space-suit be used for, if it is deployed in open-loop mode?

   (b) What is the permissible range of external temperatures when the space-suit is deployed in closed-loop mode?

   (c) Based on the results obtained in (a) and (b), what can you conclude about the effect of negative feedback in this device.

P3.2. Figure P3.2 shows the block diagram of a sophisticated biomedical device for regulating the dosage of anesthetic gases being delivered to a patient during surgery. Note that the plant and controller are themselves feedback control systems.

   (a) Derive an expression for the *open-loop gain* of the overall control system.

   (b) Derive an expression for the *closed-loop gain* of the overall control system.

   (c) If $G_1 = 1$, $G_2 = 2$, $H_1 = 1$, and $H_2 = 2$, what is the loop-gain of the overall system?
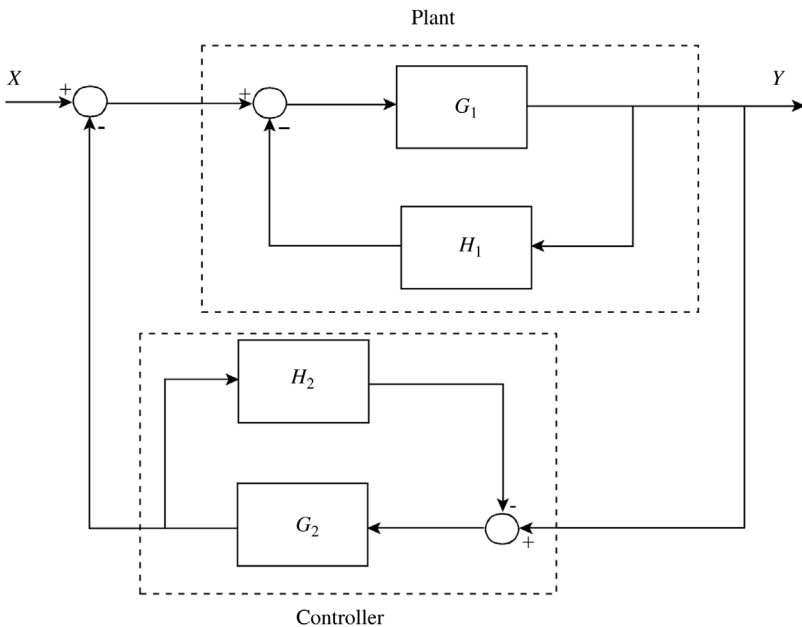
FIGURE P3.2   Block diagram of the control system of a hypothetical biomedical device.

P3.3. The cardiac output curve of a heart that has been transplanted into a patient is given in the following tabular form:

Right atrial pressure (mmHg)	0	1	2	3	4	5	6	7	8	9	10	11	12	13
Cardiac output (L min^{-1})	0	0.3	1.0	2.5	4.8	7.0	9.0	10.5	11.0	11.3	11.5	11.6	11.6	11.6

Suppose the venous return characteristics of the patient's systemic circulation can be expressed in the form of the following equations:

$$Q_R = 14, \quad P_{ra} \le 0$$
$$Q_R = 14 - 2P_{ra}, \quad 0 < P_{ra} < 7$$
$$Q_R = 0, \quad P_{ra} \ge 7$$

(a) Deduce the patient's steady-state right atrial pressure (P_{ra}) and cardiac output, assuming the transplant operation has been successful.

(b) What would be the steady-state values for cardiac output and P_{ra} if the total circulatory resistance were to be doubled?

(c) To counteract the increased circulatory resistance in (b), suppose a sufficient quantity of blood is transfused into the patient so that mean systemic pressure is raised by 5 mmHg. What would be the new steady-state values for P_{ra} and cardiac output?

P3.4. Assume the metabolic hyperbola for CO_2 given by Equation 3.48, where the steady-state CO_2 production rate is $200\,\text{mL min}^{-1}$ and the inspired CO_2 concentration is 0. Also, assume a dead-space ventilation rate of $1\,\text{L min}^{-1}$. Now, suppose the steady-state ventilatory response to CO_2 is given by Equation 3.51, where P_{aO2} is set equal to $100\,\text{mmHg}$.

(a) What are the steady-state values of ventilation and P_{aCO2}?

(b) The onset of sleep shifts the CO_2 response curve to the right, so that the apneic threshold is increased from 37 to 42 mmHg. How would this affect the steady-state values of ventilation and P_{aCO2}?

(c) How would inhalation of a gas mixture containing 7% CO_2 in air affect the steady-state ventilation and P_{aCO2} during sleep?

P3.5. Rising suddenly from a reclining to standing position sometimes causes a feeling of faintness due to a decrease in blood flow to the brain. However, in the normal person, this is quickly compensated for by adjustments in the circulation. Although cardiac output and venous return curves reflect *steady-state responses*, they remain useful for providing a qualitative picture of the sequence of events accompanying the change in posture.

Explain how the cardiac output and venous return curves are affected at each stage of the response. Also, describe how cardiac output and right atrial pressure are changed:

(a) Rising suddenly causes extra blood to be stored in the veins of the legs.

(b) The drop in blood pressure is sensed by the baroreceptors, which lead to an increase in generalized sympathetic outflow. This increases heart rate and cardiac contractility as well as peripheral resistance.

(c) Finally, venoconstriction restores mean systemic pressure back toward its normal level.

P3.6. In Figure P3.3, the thin curve represents the glucose response to insulin, while the bold line represents the insulin response to glucose. In this case, the subject represented by these curves is a type-1 diabetic patient (one whose pancreas is unable to produce sufficient insulin), who has high glucose and low insulin levels.

(a) Deduce what would be the new steady-state operating point for glucose-insulin, if this patient were to wear an insulin pump that continuously infuses insulin at a constant rate into his body (independent of blood glucose level) – Would the new insulin and glucose levels be lower/ higher compared to the operating levels indicated by the filled (dark) circle in Figure P3.3? *Hint:* To arrive at your answer, sketch on the figure the new "insulin response to glucose" line and/or the new "glucose response to insulin" curve, consistent with the installation of the insulin pump, and determine the new equilibrium point.

(b) Next, assume that this patient does not receive insulin therapy. Instead, he is prescribed a drug that increases his insulin resistance, thus reducing
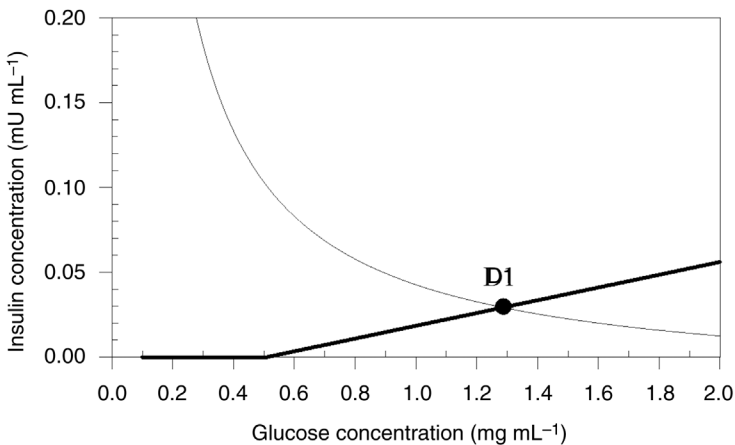
FIGURE P3.3   Plots showing insulin dependence on glucose concentration (thick line) and glucose dependence on insulin (thin curve).

the ability of insulin to stimulate glucose uptake by the body tissues. What would the new insulin and glucose levels be now, relative to the original operating levels indicated by the filled (dark) circle? Explain how you arrived at your answer.

P3.7 Figure P3.4 displays a schematic block diagram of a simplified closed-loop model of cardiac baroreflex regulation. When blood pressure decreases, the baroreflexes increase heart rate and cardiac contractility, the combined effect of which increases cardiac output. The increase in cardiac output subsequently raises blood pressure via Ohm's law: cardiac output × systemic vascular resistance = arterial blood pressure. When blood pressure increases, the opposite effect occurs. Note that in Figure P3.4a, systemic vascular resistance is represented by the block labeled "circulatory mechanics." Figure P3.4b displays the steady-state characteristics of the baroreflex and circulatory mechanics.

(a) Which component of the model ("baroreflex" or "circulatory mechanics") does each of the lines A, B, and C represent?

(b) Suppose administering a drug intravenously at a constant infusion rate leads to vasodilation of the peripheral blood vessel, lowering systemic vascular resistance. Using the information provided in Figure P3.4b, determine the preinfusion and postinfusion values of cardiac output and arterial blood pressure (be sure to indicate which set of values represents "pre" and which set represents "post"). Briefly explain how you obtained your answers.

(c) Using Figure P3.4b, estimate what the drop in arterial blood pressure would be if the infused drug not only lowers vascular resistance but also blocks the baroreceptors from responding to changes in blood pressure? Briefly explain how you arrived at your answer.
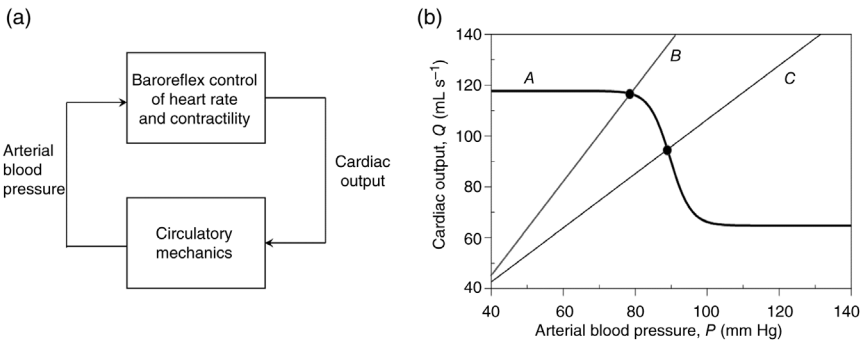
FIGURE P3.4 Block diagram of simplified closed-loop model of cardiac baroreflex regulation.

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4

TIME-DOMAIN ANALYSIS OF LINEAR CONTROL SYSTEMS

4.1 LINEARIZED RESPIRATORY MECHANICS: OPEN-LOOP VERSUS CLOSED-LOOP

In the previous chapter, we considered how feedback can change the steady-state behavior of physiological systems. In this chapter, we will explore the basic concepts and analytical techniques used to quantify the dynamics of *linearized* physiological models. We will perform extensive mathematical analyses of models with first- and second-order dynamics. These are models that one can employ as "first approximations" to a number of physiological systems. They are useful in demonstrating the methods of analysis and concepts that can be applied, while allowing the mathematics to remain at a manageable, nondistracting level.

We consider a simplified version of the linearized lung mechanics model discussed in Section 2.3. Instead of the several regional resistances and compliances, this model contains only one resistance (R) and one compliance (C) element that represent, respectively, the overall mechanical resistive and storage properties of the respiratory system. Thus, R represents a combination of resistance to airflow in the airways, lung tissue resistance, and chest wall resistance. C represents the combined compliance of lung tissue, chest wall, and airways. In addition, however, we will also add an inductance element L that represents fluid inertance in the airways. The electrical analog of this model is displayed in Figure 4.1. Our task is to
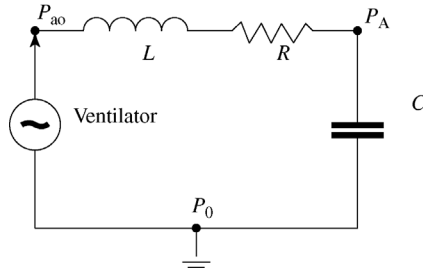
FIGURE 4.1   Electrical analog of lung mechanics model.

predict how the alveolar pressure P_A will respond dynamically to different pressure waveforms (P_{ao}) applied at the airway opening.

Applying Kirchoff's first law (see Chapter 2) to the model, we find that the pressure drop across the entire model must be equal to the sum of all the pressure drops across each of the circuit elements. Thus,

$$P_{ao} - P_0 = L\frac{dQ}{dt} + RQ + \frac{1}{C}\int Qdt \tag{4.1}$$

In Equation 4.1, Q represents the airflow rate. A similar expression can be derived to relate P_A to Q:

$$P_A - P_0 = \frac{1}{C}\int Qdt \tag{4.2}$$

We will reference all pressures to the ambient pressure (i.e., set $P_0 = 0$). Combining Equations 4.1 and 4.2 and eliminating Q from both equations, we obtain

$$P_{ao} = LC\frac{d^2 P_A}{dt^2} + RC\frac{dP_A}{dt} + P_A \tag{4.3}$$

Equation 4.3 describes the dynamic relationship between P_{ao} and P_A. Applying the Laplace transform to this second-order differential equation yields the transfer function of the model:

$$\frac{P_A(s)}{P_{ao}(s)} = \frac{1}{LCs^2 + RCs + 1} \tag{4.4}$$

This transfer function is displayed schematically in Figure 4.2a. Note that since P_A is entirely dependent on P_{ao}, this depicts an *open-loop* configuration.

Let us now consider an alternative situation where we would like to be able to attenuate the changes in P_A as much as possible, for a given set of lung mechanical parameters and a given imposed change in P_{ao}. In the clinical setting, this is desirable, since large fluctuations in P_A can cause pulmonary barotrauma or damage

(a)

(b)
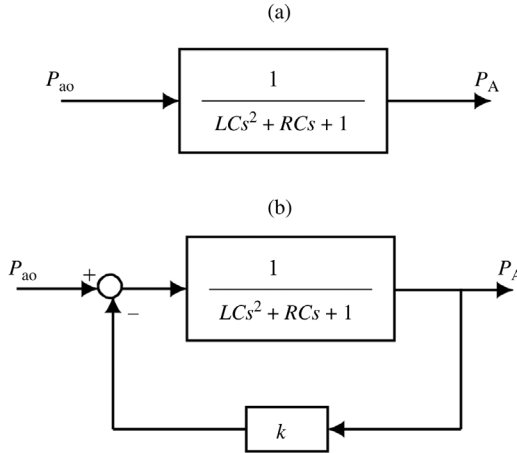
FIGURE 4.2 (a) Lung mechanics model – open loop configuration. (b) Lung mechanics model – closed-loop configuration.

to lung tissue. In order to "control" P_A, it is necessary to measure this variable and feed the measurement back to the controller. In practice, this can be achieved by measuring the pressure in the mid-esophagus with the use of an esophageal balloon, since fluctuations in esophageal pressure have been demonstrated to closely reflect fluctuations in alveolar pressure. Thus, we assume the arrangement shown in Figure 4.2b, where P_A is measured and a scaled representation of this measurement is fed back and subtracted from the input P_{ao}. This clearly is a *closed-loop* configuration and the type of control scheme is known as *proportional feedback*, since the feedback variable is proportional to the system output. Reanalysis of the new block diagram yields the following result:

$$\frac{P_A(s)}{P_{ao}(s) - kP_A(s)} = \frac{1}{LCs^2 + RCs + 1} \tag{4.5a}$$

By rearranging terms in Equation 4.5a, we can derive the following expression for the overall transfer function of the closed-loop system:

$$\frac{P_A(s)}{P_{ao}(s)} = \frac{1}{LCs^2 + RCs + (1 + k)} \tag{4.5b}$$

Equations 4.4 and 4.5b can be generalized to represent both the open- and closed-loop conditions:

$$\frac{P_A(s)}{P_{ao}(s)} = \frac{1}{LCs^2 + RCs + \lambda} \tag{4.6}$$

where $\lambda = 1$ for the open-loop case, and $\lambda = 1 + k$ for the closed-loop case.

4.2 OPEN-LOOP VERSUS CLOSED-LOOP TRANSIENT RESPONSES: FIRST-ORDER MODEL

In the range of spontaneous breathing frequencies, studies with reasonably realistic models of respiratory mechanics, such as that by Jackson and Milhorn (1973), have demonstrated that airway fluid inertance plays a virtually insignificant role in determining lung pressures and airflow. Thus, under these conditions, we can ignore inertance effects by setting L to zero. The transfer function in Equation 4.6 then becomes

$$\frac{P_A(s)}{P_{ao}(s)} = \frac{1}{\tau s + \lambda} \tag{4.7}$$

where $\tau = RC$.

4.2.1 Impulse Response

We can obtain the impulse response $h_1(t)$ of the first-order system in Equation 4.7 by setting $P_{ao}(s)$ to 1, since we are assuming the input to take the form of a unit impulse. We also multiply both numerator and denominator of the right-hand side of Equation 4.7 by $1/\tau$ to reduce it to the standard form:

$$P_A(s) = \frac{1/\tau}{s + \lambda/\tau} \tag{4.8}$$

Using the table of Laplace transforms in Appendix A, it can be seen that the impulse response is

$$h_1(t) = \frac{1}{\tau} e^{-(\lambda/\tau)t} \tag{4.9}$$

Thus, the impulse response under both open- and closed-loop conditions is a simple exponential. Note that the peak of the impulse response is a function only of τ, which depends on the system parameters R and C, but not of λ, that is, it is the same value under open- and closed-loop conditions. However, the time constant of the exponential is τ/λ. Without proportional feedback, this time constant is τ, since λ is unity. However, with proportional feedback, $\lambda > 1$ and, therefore, the closed-loop impulse response decays faster. Theoretically, the "response time" of the system can be made infinitely fast if the feedback gain k is raised to an infinitely high level. A comparison of open- and closed-loop responses is shown in Figure 4.3a for the case where $R = 1$ cm H_2O s L^{-1}, $C = 0.1$ L cm H_2O^{-1}, and $\lambda = 2$ (i.e., $k = 1$).
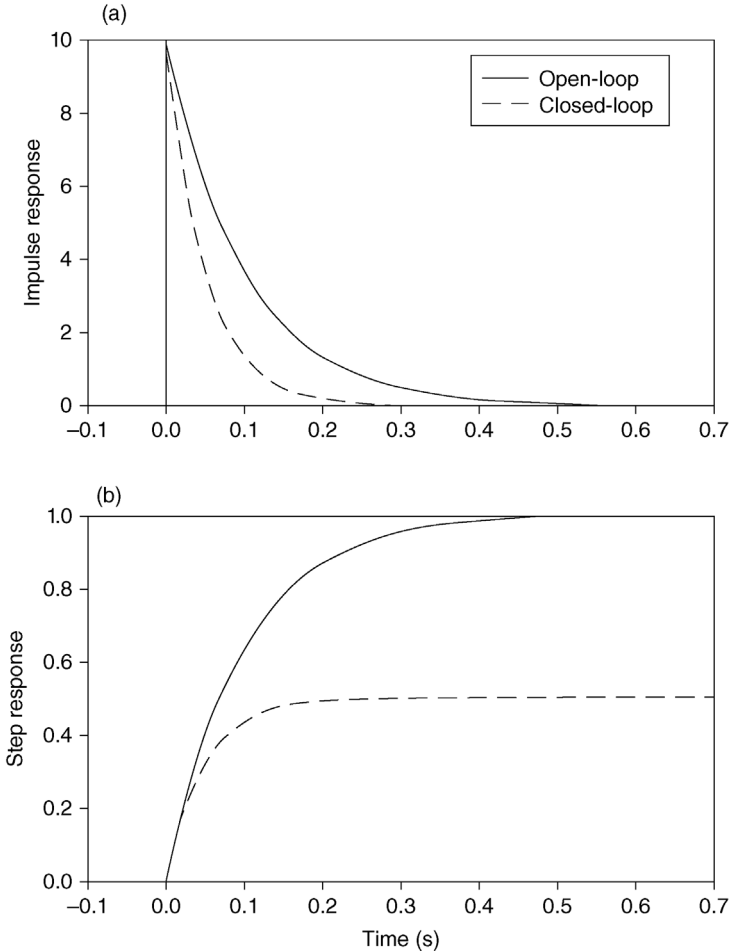
FIGURE 4.3 (a) Response of first-order lung mechanics model to a unit impulse. (b) Response of first-order lung mechanics model to a unit step. Solid and dashed lines represent the model responses in open-loop and closed-loop modes. Parameter values used: $R = 1$ cm H_2O s L^{-1}, $C = 0.1$ L cm H_2O^{-1}, $\lambda = 2$.

4.2.2 Step Response

To deduce the response $g_1(t)$ of the first-order model to a unit step, we set $P_{ao}(s)$ to $1/s$ and rearrange Equation 4.7 to obtain

$$P_A(s) = \frac{1/\tau}{s(s + \lambda/\tau)} \tag{4.10}$$

Again, the corresponding response in the time domain can be found by taking the inverse Laplace transform of Equation 4.10 using the table in Appendix A:

$$g_1(t) = \frac{1}{\tau}\left(1 - e^{-(\lambda/\tau)t}\right) \tag{4.11}$$

As in the case for the impulse response, the time constant for the step response is decreased when proportional feedback is introduced, that is, the closed-loop system responds faster. Whereas the peak amplitude of the impulse response is not affected by feedback, the steady-state magnitude of the closed-loop step response is inversely proportional to λ. Thus, the greater the feedback gain k, the smaller the steady-state value of the closed-loop step response. This can also be expressed as the *steady-state error* ε_1, defined as the final ($t \to \infty$) difference between the input (which is the unit step function) and the closed-loop step response. Thus, in this case, ε_1 increases as k and λ increase:

$$\varepsilon_1|_{t\to\infty} = 1 - \frac{1}{\lambda} \tag{4.12}$$

Figure 4.3b compares the step responses of the first-order open- and closed-loop respiratory mechanics model for the same parameter values as in Figure 4.3a. In the open-loop case, there is no steady-state error.

4.3   OPEN-LOOP VERSUS CLOSED-LOOP TRANSIENT RESPONSES: SECOND-ORDER MODEL

We now turn to the more general situation that covers a larger range of respiratory frequencies. As the rates of change of airflow become larger, so will the effects derived from fluid inertance L. This brings us back to the second-order model represented by Equation 4.6.

4.3.1   Impulse Responses

To deduce the impulse response $h_2(t)$ of the second-order system, we set $P_{ao}(s) = 1$ in Equation 4.4, which becomes

$$P_A(s) = \frac{1/LC}{s^2 + (R/L)s + \lambda/LC} \tag{4.13}$$

To determine the inverse Laplace transform of Equation 4.13, it is necessary to evaluate the roots of the quadratic function in s. If we denote the roots by α_1 and α_2, then

$$\alpha_{1,2} = -\frac{R}{2L} \pm \sqrt{\frac{R^2}{4L^2} - \frac{\lambda}{LC}} \tag{4.14}$$

Depending on the values of the model parameters, the roots α_1 and α_2 can be imaginary, complex, real and equal, or real and different. As we will demonstrate below, these roots determine whether the model behavior is oscillatory, under-damped, critically damped, or overdamped. We will consider each of these cases individually.

4.3.1.1 Undamped Behavior The roots α_1 and α_2 are imaginary when $R = 0$, so Equation 4.13 becomes

$$P_A(s) = \frac{1/LC}{s^2 + \lambda/LC} \tag{4.15}$$

Thus, the roots are

$$\alpha_1 = j\sqrt{\frac{\lambda}{LC}} \quad \text{and} \quad \alpha_2 = -j\sqrt{\frac{\lambda}{LC}} \tag{4.16}$$

The inverse Laplace transform of Equation 4.15 is

$$h_2(t) = \frac{1}{\sqrt{\lambda LC}} \sin\left(\sqrt{\frac{\lambda}{LC}}t\right) \tag{4.17}$$

Equation 4.17 implies that the response of the model to an impulsive change in P_{ao} is a sustained oscillation. In the *open-loop* configuration ($\lambda = 1$), the amplitude and the angular frequency of this oscillation are equal in magnitude, with both assuming values of $(LC)^{-1/2}$. However, in the *closed-loop* situation, $\lambda > 1$, which lowers the amplitude of the oscillation but increases its frequency. These responses are shown graphically in Figure 4.4a. In the example displayed, we have assumed the following parameter values: $L = 0.01$ cm $H_2O\,s^2\,L^{-1}$ and $C = 0.1$ L cm H_2O^{-1}. Under open-loop conditions, these parameter values produce an oscillation of frequency $(1000)^{1/2}/(2\pi)$ Hz, or approximately, 5 Hz. As in Section 4.2.1, we again assume the feedback gain k is set equal to unity, so that $\lambda = 2$. Then, the closed-loop oscillation amplitude will be $1/\sqrt{2}$ times, or approximately 71%, the oscillation amplitude in the open-loop case. At the same time, the oscillation frequency will be $\sqrt{2}$ times the corresponding value under open-loop conditions, or approximately 7 Hz.

4.3.1.2 Underdamped Behavior The sustained oscillatory responses in the previous section are, of course, highly unrealistic, since they require that R be reduced to zero. Consider now the situation when R is nonzero but small, so that

$$\frac{R^2}{4L^2} < \frac{\lambda}{LC} \tag{4.18}$$
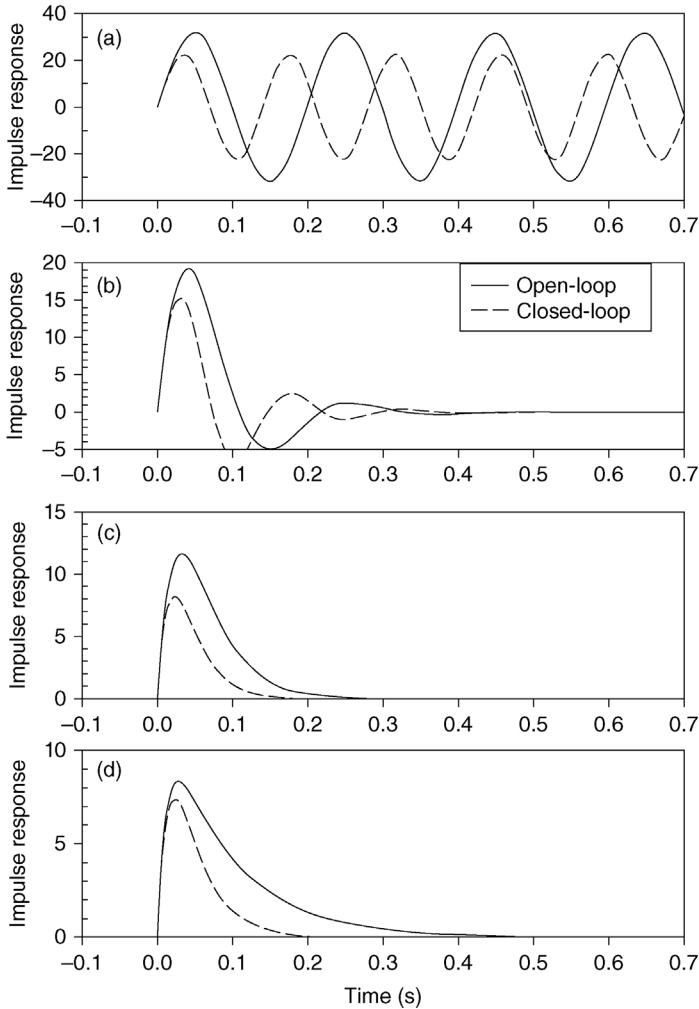
FIGURE 4.4 Responses of the second-order lung mechanics model to a unit impulse under open-loop (solid lines) and closed-loop (dashed lines) modes. (a) Undamped responses. (b) Underdamped responses. (c) Critically damped responses. (d) Overdamped responses.

The term within the square-root operation in Equation 4.14 will become negative, and, consequently, the characteristic roots α_1 and α_2 will be complex. Equation 4.13 then becomes

$$P_A(s) = \frac{1/LC}{\left(s + \dfrac{R}{2L}\right)^2 + \left(\dfrac{\lambda}{LC} - \dfrac{R^2}{4L^2}\right)} \qquad (4.19)$$

which is easily converted to the standard form:

$$P_A(s) = \frac{1/LC}{\left(s + \dfrac{R}{2L}\right)^2 + \gamma^2} \tag{4.20}$$

where

$$\gamma = \frac{R}{2L}\sqrt{\left(\frac{4L\lambda}{R^2C} - 1\right)} > \frac{R}{2L} \tag{4.21}$$

From Equation 4.14, the characteristic roots of the denominator are clearly

$$\alpha_{1,2} = -\frac{R}{2L} \pm j\gamma \tag{4.22}$$

Applying the inverse Laplace transform to Equation 4.20, we obtain the following impulse response:

$$h_2(t) = \frac{1}{LC\gamma} e^{-(R/2L)t} \sin(\gamma t) \tag{4.23}$$

The above result shows that, in the underdamped situation, the model responds to a unit impulse with dynamics that can be described as a damped sinusoid. Note that, in the limit when R decreases to zero, Equation 4.23 degenerates into the sustained oscillation represented by Equation 4.17.

How does the incorporation of negative feedback affect this underdamped response? In the closed-loop situation, λ becomes larger than unity, which increases γ relative to the open-loop case. This, in turn, reduces the amplitude of the damped oscillations but increases their frequency. However, the exponential decay term is unaffected by λ. A graphical comparison of underdamped impulse responses under open-loop versus closed-loop conditions is displayed in Figure 4.4b. In this example, the values of L and C are the same as those employed in Section 4.3.1.1. The value of R used here is $0.5\,\text{cm H}_2\text{O s L}^{-1}$. With these parameter values, $\gamma = 19.4$ in the open-loop case; thus, the frequency of the damped oscillation is approximately 3 Hz. With the incorporation of negative feedback ($k = 1$, so that $\lambda = 2$), $\gamma \approx 37.1$ so that the damped oscillation frequency becomes approximately 6 Hz. At the same time, the amplitude of the damped oscillation in the closed-loop case is approximately half as large as that in the open-loop case.

4.3.1.3 Critically Damped Behavior   If R is increased further until the following condition becomes valid:

$$\frac{R^2}{4L^2} = \frac{\lambda}{LC} \tag{4.24}$$

γ will become zero and Equation 4.13 will reduce to

$$P_A(s) = \frac{1/LC}{\left(s + \dfrac{R}{2L}\right)^2} \tag{4.25}$$

Thus, in this case, the characteristic roots will be real and equal, as shown below:

$$\alpha_{1,2} = -\frac{R}{2L} \tag{4.26}$$

The inverse Laplace transform of Equation 4.25 yields the impulse response of the model:

$$h_2(t) = \frac{1}{LC} t e^{-t/\tau_c} \tag{4.27}$$

where

$$\tau_c = \frac{2L}{R} = \sqrt{\frac{LC}{\lambda}} \tag{4.28}$$

Note that the second part of Equation 4.28 follows directly from the equality condition expressed in Equation 4.24.

The above results demonstrate that, in the critically damped mode, all oscillatory behavior disappears. How is the response affected by the introduction of negative feedback? Equation 4.28 shows quite clearly that, in the closed-loop configuration where $\lambda > 1$, the single time constant for the exponential decay is shorter compared to the open-loop case when $\lambda = 1$. Thus, as was the case for the first-order model, proportional feedback increased the speed of response of the system. This comparison is displayed graphically in Figure 4.4c. However, compared to the open-loop case, we see from Equation 4.24 that R has to be increased to a higher value before the damped oscillatory behavior disappears and critical damping is achieved in the closed-loop system.

4.3.1.4 Overdamped Behavior   When R increases above the point at which critical damping occurs, the following inequality will take effect:

$$\frac{R^2}{4L^2} > \frac{\lambda}{LC} \tag{4.29}$$

Under these circumstances, the characteristic roots of Equation 4.14 become real and different:

$$\alpha_{1,2} = -\frac{R}{2L}(1 \pm \mu) \tag{4.30}$$

where

$$\mu = \sqrt{1 - \frac{4L\lambda}{R^2 C}} \qquad (4.31)$$

It follows from the inequality expressed in Equation 4.29 that μ must lie between zero and unity in Equation 4.31. The resulting expression for $P_A(s)$ becomes

$$P_A(s) = \frac{1/LC}{\left(s + \frac{R}{2L}(1 - \mu)\right)\left(s + \frac{R}{2L}(1 + \mu)\right)} \qquad (4.32)$$

Consequently, the inverse transform of Equation 4.32 yields

$$h_2(t) = \frac{1}{\mu RC}\left(e^{-t/\tau_1} - e^{-t/\tau_2}\right) \qquad (4.33)$$

where

$$\tau_1 = \frac{2L}{R(1 - \mu)} \qquad (4.34a)$$

$$\tau_2 = \frac{2L}{R(1 + \mu)} \qquad (4.34b)$$

Thus, in the overdamped system, the impulse response is composed of two exponential decay contributions with larger time constant τ_1 and smaller time constant τ_2.

To compare the overdamped impulse responses in the closed-loop versus open-loop cases, we assume the values of L and C employed previously: $L = 0.01$ cm $H_2O\, s^2\, L^{-1}$ and $C = 0.1$ L cm H_2O^{-1}. To ensure that the condition described by Equation 4.29 is met in both open- and closed-loop conditions, we set $R = 1$ cm $H_2O\, s\, L^{-1}$. Since the "tails" of the impulse responses will be dominated by the contribution with the longer time constant, we will compare only the values of $\tau 1$ for open-loop versus closed-loop conditions. Applying Equation 4.34a, we find that in the open-loop situation, $\tau 1$ is approximately 0.09 s, while in the closed-loop condition, it is approximately 0.04 s. This comparison is shown in Figure 4.4d. Therefore, as the previous cases considered, closing the loop here also increases the speed of response of the system.

4.3.2 Step Responses

To determine the response of our lung mechanics model to a unit step change in P_{ao}, we could apply the same approach that was employed for calculating the step

response of the first-order model. However, for illustrative purposes, we will proceed along a somewhat different path by making use of the results that were derived for the impulse response of the second-order model. The basic principle employed here is the equivalence between multiplication in the Laplace domain and convolution in the time domain (Section 2.7). Thus, the step response, represented in the Laplace domain as

$$P_A(s) = \frac{1}{LCs^2 + RCs + \lambda} \cdot \frac{1}{s} \tag{4.35}$$

can be evaluated in the time domain from

$$g_2(t) = \int_0^t h_2(\sigma)u(t - \sigma)d\sigma \tag{4.36}$$

where

$$u(t) = 1, \quad t > 0 = 0, \quad t \le 0 \tag{4.37}$$

and $h(t)$ represents the impulse response of the model. Inserting Equation 4.37 into Equation 4.36, the step response can be evaluated as follows:

$$g_2(t) = \int_0^t h_2(\sigma)d\sigma \tag{4.38}$$

The expression shown in Equation 4.38 implies that the step response can be evaluated by integrating the impulse response with respect to time.

4.3.2.1 *Undamped Behavior*   Integrating Equation 4.17 with respect to time, we obtain

$$g_2(t) = -\frac{1}{\lambda} \cos\left(\sqrt{\frac{\lambda}{LC}}t\right) + A \tag{4.39a}$$

where A is an arbitrary constant. Imposing the initial condition $P_A(0) = 0$ on Equation 4.39a, we obtain the step response for undamped conditions:

$$g_2(t) = \frac{1}{\lambda}\left(1 - \cos\left(\sqrt{\frac{\lambda}{LC}}t\right)\right) \tag{4.39b}$$
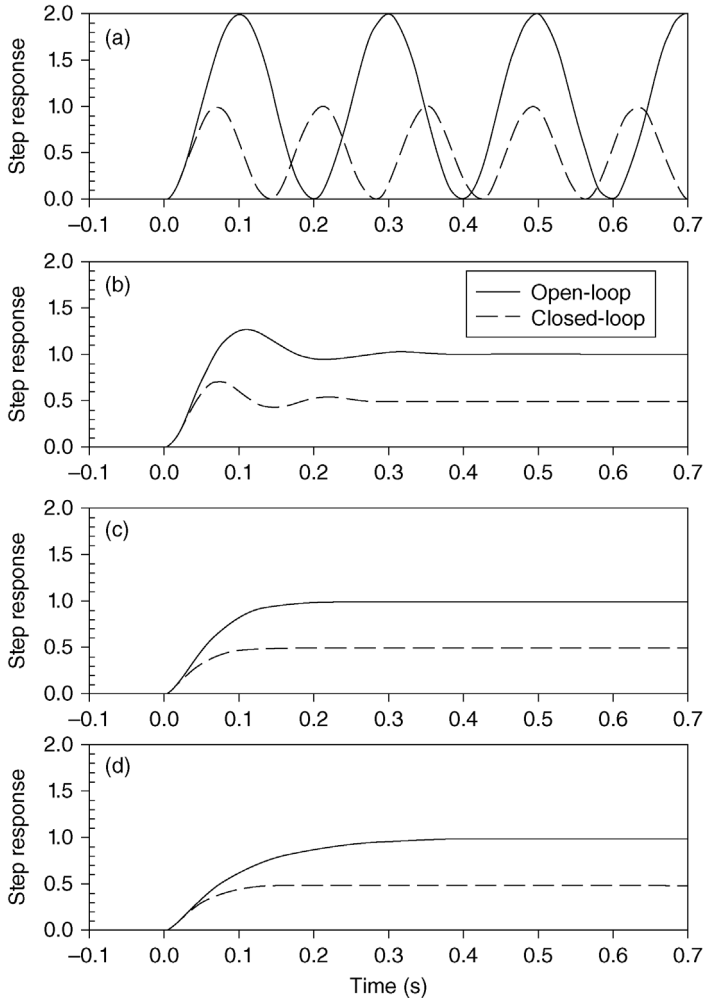
FIGURE 4.5 Responses of the second-order lung mechanics model to a unit step under open-loop (solid lines) and closed-loop (dashed lines) modes. (a) Undamped responses. (b) Underdamped responses. (c) Critically damped responses. (d) Overdamped responses.

As in the case for the impulse response, the step input elicits a sustained oscillation when there is no resistance in the system. Closing the loop increases the frequency of the oscillation but decreases its amplitude. These responses are shown in Figure 4.5a.

4.3.2.2 Underdamped Behavior As in the undamped case, we obtain the step response here by convolving the impulse response described in Equation 4.23 with a unit step. This turns out to be the same as integrating the impulse response

with respect to time:

$$g_2(t) = \frac{1}{LC\gamma} \int_0^t e^{-(R/2L)\sigma} \sin(\gamma\sigma)d\sigma \tag{4.40}$$

where γ is given by Equation 4.21. Performing integration by parts and imposing the initial condition that $P_A(0) = 0$, we obtain the following expression for the under-damped step response:

$$g_2(t) = \frac{1}{\lambda}\left(1 - e^{-(R/2L)t}\cos(\gamma t) - \frac{R}{2L\gamma}e^{-(R/2L)t}\sin(\gamma t)\right) \tag{4.41}$$

As can be seen from Figure 4.5b, the damped oscillatory characteristics of this response are the same as those of the impulse response in both open- and closed-loop modes. However, in the steady state, the oscillations become fully damped out and the response settles to the constant level given by

$$g_2(t \to \infty) = \frac{1}{\lambda} \tag{4.42}$$

In Equation 4.42, note that in the open-loop case where $\lambda = 1$, the response in P_A settles down to a value of 1, that is, the same as the unit step in P_{ao}. With the loop closed, however, where $\lambda > 1$, the steady-state value of P_A is less than unity. Thus, as it was for the first-order model, the underdamped step response for the second-order model shows a *steady-state error* ε_2 given by

$$\varepsilon_2|_{t\to\infty} = 1 - \frac{1}{\lambda} \tag{4.43}$$

4.3.2.3 *Critically Damped Behavior*   We obtain the critically damped response to the unit step by integrating Equation 4.27 with respect to time. After imposing the initial condition $P_A(0) = 0$, we have

$$g_2(t) = \frac{1}{LC}\left(\tau_c^2 - \tau_c(\tau_c + t)e^{-t/\tau_c}\right) \tag{4.44}$$

where τ_c is defined by Equation 4.28. The step responses for open- and closed-loop conditions are displayed in Figure 4.5c. As in the case for the corresponding impulse responses, closing the loop leads to a smaller τ_c and thus faster speed of response. In the steady state, as $t \to \infty$, Equation 4.44 becomes

$$g_2(t \to \infty) = \frac{\tau_c^2}{LC} = \frac{1}{\lambda} \tag{4.45}$$

where the second part of the above equation is derived by using Equation 4.28 to substitute for τ_c. Thus, the steady-state response to a unit step and the corresponding steady-state error in the critically damped mode are the same as those in the underdamped mode.

4.3.2.4 *Overdamped Behavior*   For the overdamped response to the unit step, we integrate Equation 4.27 with respect to time and impose the initial condition $P_A(0) = 0$ to obtain

$$g_2(t) = \frac{1}{\mu RC} \left(\tau_1 \left(1 - e^{-t/\tau_1} \right) - \tau_2 \left(1 - e^{-t/\tau_2} \right) \right) \tag{4.46}$$

As in the critically damped case, introducing negative feedback increases the speed of response. In the steady state, as $t \to \infty$, we obtain the following result:

$$g_2(t \to \infty) = \frac{\tau_1 - \tau_2}{\mu RC} \tag{4.47a}$$

By substituting for τ_1 and τ_2 in Equation 4.47a and employing the definition of μ, it can be shown that this equation reduces to

$$g_2(t \to \infty) = \frac{1}{\lambda} \tag{4.47b}$$

Closing the loop gives rise to a steady-state error of the same magnitude as in the previous step responses. Open- and closed-loop overdamped responses to the unit step are compared in Figure 4.5d.

4.4   DESCRIPTORS OF IMPULSE AND STEP RESPONSES

4.4.1   Generalized Second-Order Dynamics

The impulse and step responses of both first- and second-order lung mechanics models have demonstrated that when proportional feedback is introduced, alveolar pressure changes resulting from perturbations in P_{ao} (the input) are attenuated. The resulting fluctuations in P_A also respond more quickly to changes in P_{ao} under closed-loop conditions.

   In the various impulse and step responses derived for the lung mechanics model in Section 4.3, it should be pointed out that although the model contained three physiological parameters (L, C, and R), these parameters always appeared in combination with one another, for example, LC and RC. Indeed, the transfer function $P_A(s)/P_{ao}(s)$ contains only two free parameters for a given value of k, the feedback gain; thus, more than one combination of R, C, and L may produce the

same dynamics. In this section, we will present the system equations for a generalized second-order model that is characterized by the same dynamics as the lung mechanics model. In the generalized model, the second-order dynamics are governed by two independent parameters. A third parameter, the steady-state input–output gain, G_{SS} is also introduced. In the particular example of the lung mechanics model, G_{SS} turned out to be unity. In addition, we generalize the input and output to be $x(t)$ and $y(t)$. Then, denoting the Laplace transforms of $x(t)$ and $y(t)$ by $X(s)$ and $Y(s)$, respectively, we can convert Equation 4.5a to

$$\frac{Y(s)}{X(s) - kY(s)} = \frac{G_{SS}}{LCs^2 + RCs + 1} \tag{4.48}$$

We can generalize Equation 4.48 further by introducing two new parameters to substitute for the three redundant parameters: R, L, and C. It will soon become obvious that these two new parameters provide a highly intuitive description of the dynamic properties of the model. We begin by considering the undamped open-loop system ($k = 0$). As we had shown earlier, the responses to unit impulse or step took the form of a sustained oscillation. In fact, the angular frequency of the oscillation represents the highest frequency at which the system will "resonate." This frequency is commonly referred to as the *natural frequency* ω_n. From Equation 4.17, we find that ω_n is defined by

$$\omega_n = \frac{1}{\sqrt{LC}} \tag{4.49}$$

The second new parameter that we will introduce is ζ, defined as

$$\zeta = \frac{R}{2}\sqrt{\frac{C}{L}} \tag{4.50}$$

Substituting Equations 4.49 and 4.50 into Equation 4.48 and rearranging terms, it can be easily shown that the overall transfer function for the model now becomes

$$\frac{Y(s)}{X(s)} = \frac{G_{SS}\omega_n^2}{s^2 + 2\zeta\omega_n s + (1 + kG_{SS})\omega_n^2} \tag{4.51}$$

The open- and closed-loop versions of this generalized system are depicted schematically in Figure 4.6a and b, respectively.

Whether the resulting impulse or step responses are undamped, underdamped, critically damped, or overdamped depends on the roots of the denominator in Equation 4.51. It can be seen that this, in turn, depends on the value of the parameter ζ. Note that when $\zeta = 0$, the impulse or step response will be a sustained

(a)

(b)
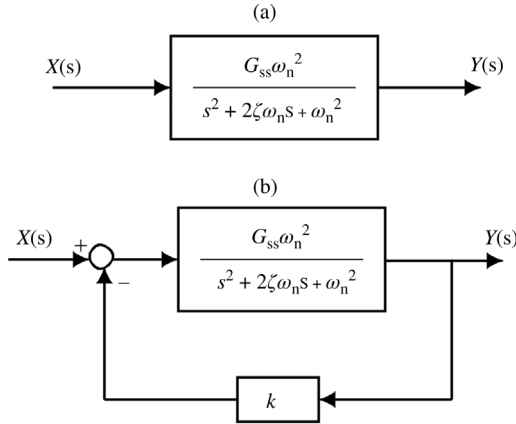
FIGURE 4.6 (a) Generalized second-order open-loop model. (b) Generalized second-order closed-loop model.

oscillation. In the open-loop case, when $0 < \zeta < 1$, the impulse and step responses will show damped oscillatory behavior. However, when $\zeta \geq 1$, these responses will assume an exponential form. It is clear that ζ represents the amount of "damping" inherent in the system, and for this reason, it is commonly referred to as the *damping factor* or *damping ratio*.

4.4.1.1 Undamped Dynamics In the case when $\zeta = 0$, Equation 4.51 becomes

$$\frac{Y(s)}{X(s)} = \frac{G_{SS}\omega_n^2}{s^2 + (1 + kG_{SS})\omega_n^2} \tag{4.52}$$

The inverse Laplace transform of Equation 4.52 yields an oscillatory solution for the impulse response $h_2(t)$:

$$h_2(t) = \frac{G_{SS}\omega_n}{\sqrt{1 + kG_{SS}}} \sin\left(\sqrt{1 + kG_{SS}}\,\omega_n t\right) \tag{4.53}$$

The step response, $g_2(t)$, which is also oscillatory, is obtained by integrating $h_2(t)$ with respect to time:

$$g_2(t) = \frac{G_{SS}}{1 + kG_{SS}} \left(1 - \cos\left(\sqrt{1 + kG_{SS}}\,\omega_n t\right)\right) \tag{4.54}$$

Note from Equation 4.54 that the step response oscillates around the constant level $G_{SS}/(1 + kG_{SS})$.

4.4.1.2   *Underdamped Dynamics*   In the underdamped mode, when $\zeta^2 < 1 + kG_{SS}$, the denominator in Equation 4.51 can be rearranged so that the following form is obtained:

$$\frac{Y(s)}{X(s)} = \frac{G_{SS}\omega_n^2}{(s + \zeta\omega_n)^2 + \omega_n^2(1 + kG_{SS} - \zeta^2)} \tag{4.55}$$

The impulse response corresponding to Equation 4.55 is

$$h_2(t) = \frac{G_{SS}\omega_n}{\sqrt{1 + kG_{SS} - \zeta^2}} e^{-\omega_n\zeta t} \sin\left(\sqrt{1 + kG_{SS} - \zeta^2}\,\omega_n t\right) \tag{4.56}$$

while the step response is

$$g_2(t) = \frac{G_{SS}}{1 + kG_{SS}}\left(1 - \frac{e^{-\zeta\omega_n t}}{\sqrt{1 + kG_{SS} - \zeta^2}}\sin\left(\omega_n\sqrt{1 + kG_{SS} - \zeta^2}\,t + \theta\right)\right) \tag{4.57}$$

where

$$\theta = \tan^{-1}\left(\frac{\sqrt{1 + kG_{SS} - \zeta^2}}{\zeta}\right)$$

4.4.1.3   *Critically Damped Dynamics*   The roots of the denominator become real and equal when $\zeta^2 = 1 + kG_{SS}$. At this point, all oscillatory dynamics disappear and the system becomes "critically damped." Equation 4.51 becomes

$$\frac{Y(s)}{X(s)} = \frac{G_{SS}\omega_n^2}{(s + \zeta\omega_n)^2} \tag{4.58}$$

The impulse response that corresponds to Equation 4.58 is

$$h_2(t) = G_{SS}\omega_n^2 t e^{-\zeta\omega_n t} \tag{4.59}$$

while the step response is

$$\begin{aligned}
g_2(t) &= \frac{G_{SS}}{\zeta}\left(\frac{1}{\zeta} - \left(\frac{1}{\zeta} + \omega_n t\right)e^{-\zeta\omega_n t}\right) \\
&= \frac{G_{SS}}{\sqrt{1 + kG_{SS}}}\left(\frac{1}{\sqrt{1 + kG_{SS}}} - \left(\frac{1}{\sqrt{1 + kG_{SS}}} + \omega_n t\right)e^{-\zeta\omega_n t}\right)
\end{aligned} \tag{4.60}$$

4.4.1.4 *Overdamped Dynamics* When ζ^2 exceeds $1 + kG_{SS}$, the roots of the denominator of Equation 4.51 become real and different; so the corresponding impulse and step responses now become

$$h_2(t) = \frac{G_{SS}\omega_n}{2\sqrt{\zeta^2 - 1 - kG_{SS}}} \left(e^{-\omega_n\left(\zeta - \sqrt{\zeta^2 - 1 - kG_{SS}}\right)t} + e^{-\omega_n\left(\zeta + \sqrt{\zeta^2 - 1 - kG_{SS}}\right)t} \right)$$

(4.61)

$$g_2(t) = \frac{G_{SS}}{2\sqrt{\zeta^2 - 1 - kG_{SS}}} \left(\frac{1 - e^{-\omega_n\left(\zeta - \sqrt{\zeta^2 - 1 - kG_{SS}}\right)t}}{\zeta - \sqrt{\zeta^2 - 1 - kG_{SS}}} + \frac{1 - e^{-\omega_n\left(\zeta + \sqrt{\zeta^2 - 1 - kG_{SS}}\right)t}}{\zeta + \sqrt{\zeta^2 - 1 - kG_{SS}}} \right)$$

(4.62)

4.4.1.5 *Steady-State Error* In the underdamped, critically damped, and over-damped modes of the generalized second-order system, the step response attains the same steady-state value. In Equations 4.57 and 4.60, we can deduce this final value easily by letting t tend to infinity. The same can be done for the overdamped mode in Equation 4.62, except that a little algebra will be needed to obtain the following expression for the steady-state response:

$$g_2(t \to \infty) = \frac{G_{SS}}{1 + kG_{SS}}$$

(4.63)

The steady-state error is deduced by subtracting the steady-state response from the input value, which is unity since the unit step was employed. Under open-loop circumstances ($k = 0$), the steady-state error would be

$$\varepsilon_2|_{\text{open-loop}} = 1 - G_{SS}$$

(4.64)

Note that in the special case when $G_{SS} = 1$, as in the example considered in Section 4.3.2, the open-loop steady-state error is zero. However, when G_{SS} assumes other values, the open-loop steady-state error can be quite large. In the closed-loop case, the steady-state error is given by

$$\varepsilon_2|_{\text{closed-loop}} = 1 - \frac{G_{SS}}{1 + kG_{SS}}$$

(4.65)

In the special case where $G_{SS} = 1$, the steady-state error becomes $k/(1+k)$, as was previously shown in the example in Section 4.3.2.

4.4.2 Transient Response Descriptors

The first- and second-order impulse and step responses we have discussed consti-tute the simplest approximations to the corresponding time-domain dynamics of

real physiological systems. To characterize more realistic impulse and step responses, one could in principle extend the modeling analysis to higher order models. But as this process continues, the mathematics rapidly become less and less tractable. Furthermore, the number of parameters needed to describe these responses will also increase. In some situations, it may be necessary to compare the dynamic behavior of one system with that of another. Alternatively, one may need to compare the dynamic characteristics of the same system under different conditions. In order to do this, it is possible to first estimate the impulse and/or step responses, and then extract certain descriptors from these responses empirically. Subsequently, standard statistical analyses, such as the Student *t*-test, can be employed to determine if the two sets of dynamic responses are significantly different from one another. The descriptors discussed in the following sections are among the most commonly used in systems analysis.

4.4.2.1 Impulse Response Descriptors   These descriptive features of the impulse response are illustrated in Figure 4.7a. The most direct feature is the *peak amplitude*, which simply measures the maximum (or minimum, if the response is predominantly negative) value of the impulse response. Thus,

$$\text{Peak amplitude} = \max [h(t)] \quad \text{or} \quad |\min [h(t)]| \qquad (4.66)$$

The *area* under the impulse response function represents the integral of $h(t)$ over time, which in turn yields the *steady-state gain* G_{SS} of the system:

$$G_{SS} = \int_{-\infty}^{\infty} h(t)dt \qquad (4.67)$$

Finally, the *characteristic time* T_c provides a measure of the approximate latency following which the bulk of the impulse response occurs. Alternatively, T_c may also be thought of in the following way. If the impulse response function is represented as a two-dimensional mass, then T_c will be the location (on the time axis) at which the center of mass acts.

$$T_c = \frac{\int_{-\infty}^{\infty} t|h(t)|dt}{\int_{-\infty}^{\infty} |h(t)|dt} \qquad (4.68)$$

4.4.2.2 Step Response Descriptors   The most commonly used descriptors of the step response are shown in Figure 4.7b. As mentioned previously, the *final value* of the response is the steady-state level achieved by the system in question. If the input is a *unit step*, this final value will yield the steady-state gain G_{SS}. If the peak value of
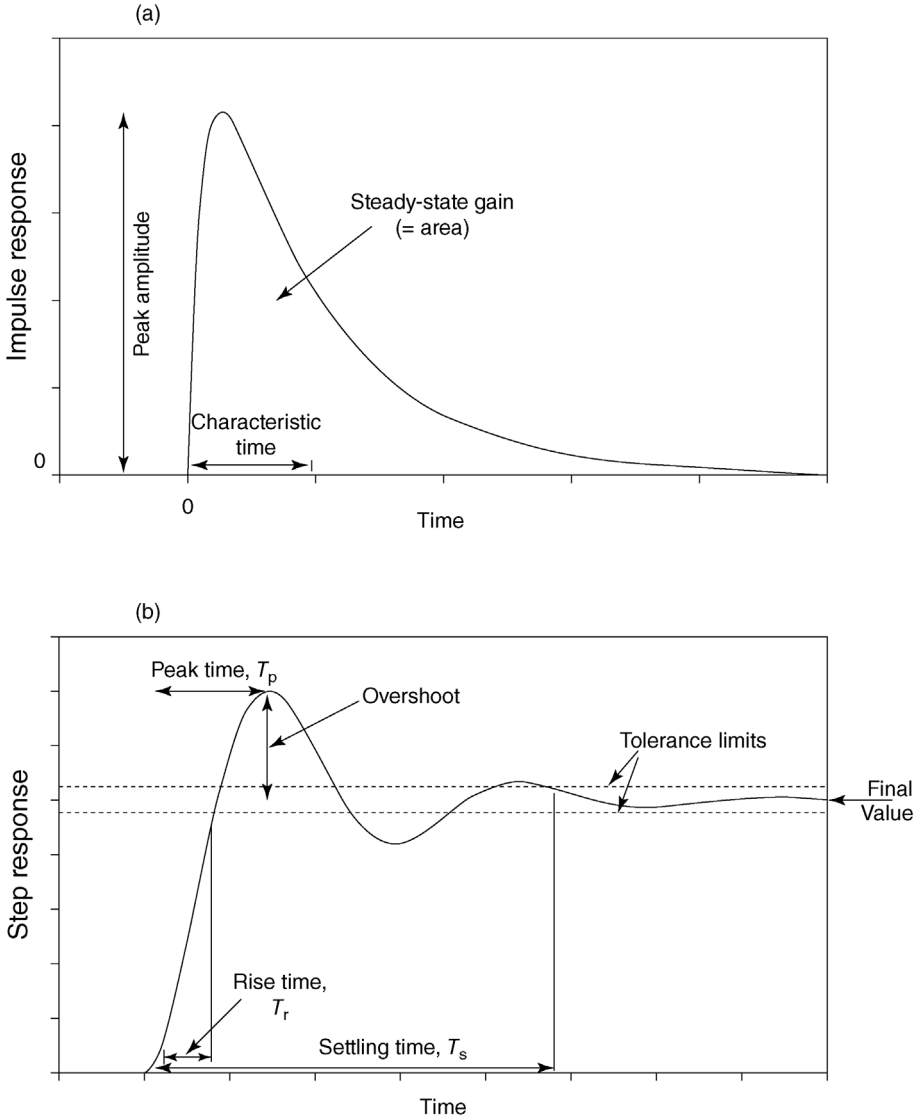
FIGURE 4.7   Descriptors of the (a) impulse and (b) step responses.

the step response is larger than the final value, the *overshoot* will be the difference between this peak value and the final value. Frequently, this overshoot is expressed in percentage terms:

$$\text{Percent overshoot} = \frac{\text{peak response} - \text{final value}}{\text{final value}} \times 100\% \qquad (4.69)$$

The time taken for the step response to achieve its peak value is known as the *peak time* or T_p, as illustrated in Figure 4.7b. Aside from peak time, there are two other measures of speed of response. One is the *rise time* T_r defined as

$$T_r = t_{90\%} - t_{10\%} \tag{4.70}$$

where $t_{90\%}$ and $t_{10\%}$ are the times at which the response first attains 90 and 10% of its final value, respectively.

The other measure of speed of response is the settling time T_s, defined as the time taken for the step response to settle within $\pm\delta\%$ of the final value (Figure 4.7b). The upper and lower levels of this band of values, that is, $100 + \delta\%$ and $100 - \delta\%$ of the final value, define the tolerance limits within which the step response will remain at all times greater than T_s. The values of δ generally employed range from 1 to 5%.

4.5 OPEN-LOOP VERSUS CLOSED-LOOP DYNAMICS: OTHER CONSIDERATIONS

4.5.1 Reduction of the Effects of External Disturbances

In our previous discussions of the first- and second-order models of lung mechanics, we showed that one clear consequence of introducing negative feedback into the control scheme is *an increase in speed of system response*. A second major effect of closing the loop is the reduction in overall system gain. For both first- and second-order models, closing the loop led to a significant reduction of the final values in the unit step responses (see Figures 4.3b and 4.5). This result is consistent with the conclusion that we arrived at in Section 3.2, although those considerations were based entirely on steady-state conditions. As we had pointed out in that section, what is most advantageous about this reduction in overall system gain is the enhanced ability of the closed-loop system to attenuate the impact of external disturbances. To emphasize the importance of this point, we will consider a simple example here.

Figure 4.8a and b illustrates the open- and closed-loop versions of a generalized linear control system. $D(s)$ represents the Laplace transform of an external disturbance that contributes "noise" directly and additively to the output. Thus, in the open-loop case,

$$Y(s) = G(s)X(s) + D(s) \tag{4.71}$$

which clearly shows that 100% of the external disturbance is reflected in the output. However, in the closed-loop case, we have

$$Y(s) = G(s)(X(s) - H(s)Y(s)) + D(s) \tag{4.72a}$$
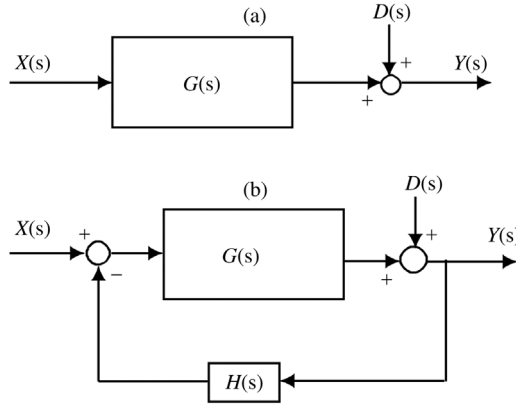
FIGURE 4.8 (a) Generalized linear open-loop system. (b) Generalized linear closed-loop system.

which, upon rearranging terms, becomes

$$Y(s) = \frac{G(s)}{1 + G(s)H(s)} X(s) + \frac{1}{1 + G(s)H(s)} D(s) \qquad (4.72b)$$

Since we have explicitly incorporated negative feedback into the equations, the common denominator in Equation 4.72b satisfies the following condition:

$$|1 + G(s)H(s)| > 1 \qquad (4.73)$$

As such, the effect of $D(s)$ on $Y(s)$ will be attenuated and can be further attenuated as we increase the magnitude of the product $G(s)H(s)$, which is the *loop gain* (LG) of the closed-loop system.

4.5.2   Reduction of the Effects of Parameter Variations

There are situations, particularly when dealing with the artificial control of some physiological variable, where there may be a need to decide upon a range of the input $x(t)$ signal in order to closely regulate variations in the output $y(t)$. This can only be done if we have a very good idea of the characteristics of the feedforward subsystem $G(s)$. However, this may not always be possible, as we may have erroneous estimates of $G(s)$ or $G(s)$ may actually be time-varying. These variations in the system parameters will have an impact on the controlled output.

First, consider the open-loop case. Assume that there is a small change in the transfer characteristics of $G(s)$, which we will denote by $\Delta G(s)$. Then, the effect on the output will be

$$Y(s) + \Delta Y(s) = (G(s) + \Delta G(s))X(s) + D(s) \qquad (4.74)$$

Eliminating the equivalent expression for $Y(s)$, we can derive the following:

$$\Delta Y(s) = \Delta G(s)X(s) \tag{4.75}$$

The above result shows that the variation in $G(s)$ is directly reflected in the output. Now, consider the corresponding result as we apply the same type of analysis to the closed-loop system:

$$Y(s) + \Delta Y(s) = D(s) + (G(s) + \Delta G(s))(X(s) - H(s)(Y(s) + \Delta Y(s))) \tag{4.76}$$

Again, we expand Equation 4.76 and eliminate $Y(s)$ from both sides of the equation. We also eliminate the term containing the product of differences $\Delta G(s)$ and $\Delta Y(s)$, since we have assumed these differences to be small. These steps lead to the following result:

$$\Delta Y(s) = \frac{\Delta G(s)}{(1 + G(s)H(s))^2} X(s) \tag{4.77}$$

Thus, in the closed-loop case, the effect of $\Delta G(s)$ is reduced by a factor of $\{1 + G(s)H(s)\}^2$.

4.5.3 Integral Control

In spite of the many advantages of employing proportional feedback control, one problem that can be highly aggravating in some applications is the existence of the steady-state error. We will demonstrate in this section that the steady-state error can be eliminated completely by employing integral control. To understand how this can be achieved, consider the proportional control and integral control systems shown in Figure 4.9a and b, respectively.

First consider the proportional control system shown. This represents a generalization of the particular first- and second-order models discussed in Sections 4.2 and 4.3. The Laplace transform of the difference (error) between the input and output is given by

$$E(s) = X(s) - Y(s) = \left(1 - \frac{G(s)}{1 + kG(s)}\right)X(s) \tag{4.78}$$

The steady-state error $e(t \to \infty)$ can be deduced from the above equation by using the unit step input (i.e., setting $X(s) = 1/s$) and evaluating the result via the final value theorem for Laplace transforms:

$$e(t \to \infty) = \lim_{s \to 0} sE(s) \tag{4.79}$$
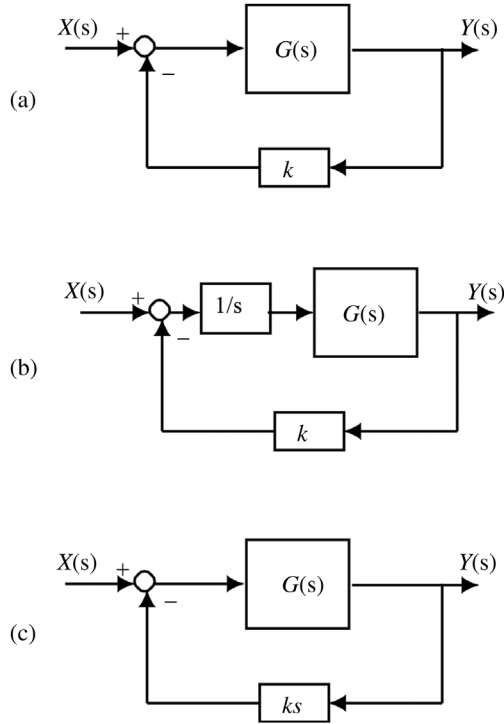
FIGURE 4.9   Different closed-loop control schemes. (a) Proportional feedback control. (b) Integral control. (c) Derivative feedback control.

Applying Equation 4.79 to Equation 4.78, we obtain

$$e(t \rightarrow \infty) = \frac{1 + (k - 1)G_{SS}}{1 + kG_{SS}} \tag{4.80}$$

where G_{SS} represents the steady-state value of $G(s)$. Note that we can minimize the steady-state error by setting k equal to unity, in which case

$$e(t \rightarrow \infty)_{min} = \frac{1}{1 + G_{SS}} \tag{4.81}$$

Now consider the case for integral control, in which the error signal is integrated prior to being used to drive the actuator (plant) portion of the closed-loop system. In this case, the Laplace transform of the difference between input and output is

$$E(s) = X(s) - Y(s) = \left(1 - \frac{G(s)}{s + kG(s)}\right)X(s) \tag{4.82}$$

Using Equation 4.79 and assuming the input to be a unit step ($X(s) = 1/s$), we obtain

$$e(t \to \infty) = \frac{(k-1)G_{SS}}{kG_{SS}} \tag{4.83}$$

In this case, we can eliminate the steady-state error completely by setting k to unity. However, this advantage of not having a steady-state error is derived at the expense of speed of system response.

For a more intuitive explanation as to why there is always a steady-state error in proportional feedback control but not in integral control, consider both cases when $k = 1$. In proportional control, if the steady-state error ($e(t \to \infty) = x(t \to \infty) - y(t \to \infty)$) is zero, this would also make the error signal that drives the actuator/plant zero, which in turn implies that the steady-state output $y(t \to \infty)$ would become zero. This result would be incompatible with the prior assertion that $e(t \to \infty)$ is zero, since $x(t \to \infty)$ equals unity. Thus, for the proportional feedback system, a *steady-state error must exist* in order for the system to produce a nonzero output. Now consider the integral control scheme. Assume that, before time zero, both input and output are zero. When the unit step takes effect at the input, $y(t)$ will initially remain at zero and, consequently, there will be a large error signal that feeds into the integrator. However, with time, as $y(t)$ increases toward its final value, this error signal will diminish. On the other hand, the output of the integrator will remain high since it represents the accumulation of all previous values of the error signal. Finally, when $y(t \to \infty)$ attains the same value as $x(t \to \infty)$, the steady-state error will become zero, and the integrator output, which drives the actuator/plant, will cease increasing but remain at its final positive value so that $y(t \to \infty)$ will be unchanged (and equal to $x(t \to \infty)$).

4.5.4   Derivative Feedback

Instead of feeding back a signal directly proportional to the system output, how would closed-loop dynamics be different if the feedback signal were proportional to the *time-derivative* of the output? Consider the control scheme illustrated in Figure 4.9c and, for the sake of simplicity, let us assume in this example that

$$G(s) = \frac{1}{\tau s + 1} \tag{4.84}$$

Then,

$$\frac{Y(s)}{X(s) - k_d s Y(s)} = \frac{1}{\tau s + 1} \tag{4.85}$$

From Equation 4.85, we derive the following expression for the overall system transfer function:

$$\frac{Y(s)}{X(s)} = \frac{1}{\tau's + 1} \tag{4.86}$$

where

$$\tau' = \tau + k_d \tag{4.87}$$

The unit step response corresponding to Equations 4.86 and 4.87 is

$$g_1(t) = 1 - e^{-(t/\tau + k_d)} \tag{4.88}$$

It is clear from this result that derivative feedback increases the effective time constant and therefore produces a more sluggish response. In other words, derivative feedback increases system damping.

To determine how derivative feedback affects steady-state error, we derive from Equation 4.86 the following expression:

$$E(s) = X(s) - Y(s) = \left(1 - \frac{1}{\tau's + 1}\right) X(s) \tag{4.89}$$

Then, using Equation 4.79, the steady-state error is found to be

$$e(t \to \infty) = \lim_{s \to 0} \left[s\left(1 - \frac{1}{\tau's + 1}\right)\frac{1}{s}\right] = 0 \tag{4.90}$$

Thus, derivative feedback of the kind shown in Figure 4.9c leads to the elimination of the steady-state error.

There is a popular variant of this type of control known as "velocity feedback," in which the feedback signal consists of the sum of a term proportional to the output and a term proportional to the derivative of the output. In this case, there will in general be a steady-state error. However, the steady-state error can be attenuated by increasing the gain of the forward block, G_{SS}. In the limit, when $G_{SS} \to \infty$, the steady-state error will become zero.

4.5.5   Minimizing Effect of External Disturbances by Feedforward Gain

Following up on the discussion of minimizing the effect of external disturbances on the response of a closed-loop system, consider the scheme illustrated in Figure 4.10. Here, the closed-loop system in question consists of the controller $G_c(s)$ and plant $G_p(s)$ blocks. The external disturbance takes the form of $\delta(s)$ that is added as an additional input to $G_p(s)$.
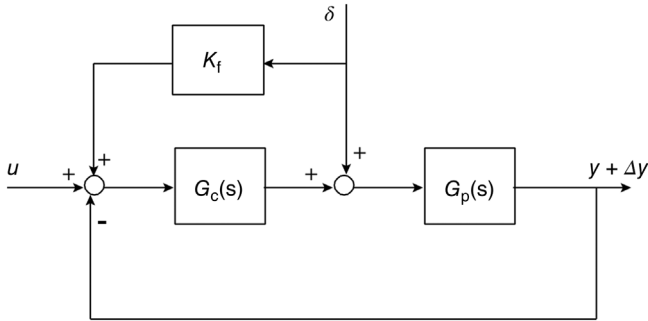
FIGURE 4.10   Scheme for minimizing the effect of external disturbance using feedforward gain.

Here we have added an element that senses the disturbance and produces a proportional output that is "fedforward" into the closed-loop system. This component is known as a *feedforward* element. The system response to the disturbance is now given by

$$\Delta Y(s) = \frac{G_p(s)(1 + K_f G_c(s))}{1 + G_c(s)G_p(s)} \delta(s) \tag{4.91}$$

If $G_c(s)$ is a proportional controller with gain K, and $\delta(s)$ takes the form of a unit step, then the steady-state change in the output becomes

$$\Delta y(t \to \infty) = \lim_{s \to 0} s\Delta Y(s) = \lim_{s \to \infty} \left(\frac{sG_p(s)(1 + K_f K)}{1 + KG_p(s)} \frac{1}{s} \right) = \frac{G_{pss}(1 + K_f K)}{1 + KG_{pss}} \tag{4.92}$$

where G_{pss} is the steady-state gain of the plant $G_p(s)$. The steady-state response to the disturbance can be made to become zero if the feedforward gain K_f is set equal to the negative inverse of the controller gain, that is,

$$K_f = -\frac{1}{K} \tag{4.93}$$

The discussion above illustrates the kind of considerations that a designer of an engineering control system would make if the sole purpose of the design is to eliminate the steady-state influences of external disturbances to the system. However, it is also clear that these considerations represent an oversimplification of what would have been taken into account in the design of a "real" control system. In the examples discussed, we used the steady-state system response to an external disturbance as the measure of "performance": The "better" system would be the one in which the disturbance produces the smaller steady-state change in output. Instead of this criterion, we could have based our performance measure on

the complete (i.e., transient + steady state) response to the disturbance. A different performance measure is likely to have led us to a different conclusion as to which controller design is the most meritorious. Thus, the "optimal" solution is not necessarily a unique one and depends heavily on the criterion used to measure performance.

4.6 TRANSIENT RESPONSE ANALYSIS USING MATLAB

If the form of the transfer function of a given model is known, the response of the system to standard inputs, such as the unit impulse or unit step, as well as any arbitrary input waveform, can be deduced easily in MATLAB. The following MATLAB command lines (also found in the script file `tra_llm.m`) demonstrate how transient response analysis can be applied to the linearized lung mechanics model that we have been discussing.

Assuming that the parameter values of L, R, C, and k in Equation 4.5b have been preassigned, we begin by setting up the transfer function, `Hs`, of the model:

```
>> num = [1];
>> den = [L*C  R*C  1+k];
>> Hs = tf(num,den);
>> t = [0:0.005:0.8];
```

The first two lines assign values to the various terms in the numerator (`num`) and denominator (`den`) of `Hs`. In the case of the denominator, these values are assigned in the order of descending powers of s. The fourth line simply generates a time vector covering the duration of the response that we will examine.

The impulse response is computed and plotted using the following command lines:

```
>> x = impulse(Hs,t);
>> plot(t,x)
```

The command lines that follow produce a plot of the unit step response:

```
>> y = step(Hs,t);
>> plot(t,y)
```

Finally, the response of this system to an input u of arbitrary time-course can be computed using the `lsim` function:

```
>> [u,t]=gensig('square',0.5,5,0.005);
>> y=lsim(Hs,u,t);
>> plot(t,y)
```

In this example, the "arbitrary input" is a square wave of period 0.5 s, lasting up to time $t = 5$ s (in time-steps of 0.005 s), generated with the function `gensig`.

4.7   SIMULINK APPLICATION 1: DYNAMICS OF NEUROMUSCULAR REFLEX MOTION

Up to this point, we have limited our analyses to simple models with only first- or second-order dynamics. This was done intentionally in order to demonstrate the methodology employed in classical time-domain analysis without letting the mathematical details become too intractable and distracting. To extend this kind of analysis to more complex (and more realistic) physiological models, it becomes progressively more convenient to employ the methods of computer simulation. In this section, we will demonstrate an example of time-domain analysis using SIMULINK.

4.7.1   A Model of Neuromuscular Reflex Motion

Examination of the dynamics of neuromuscular reflex motion can yield valuable insight into the status of patients who have neurological disorders. The model that we will consider assumes the following test. The patient is seated comfortably and his shoulder and elbow are held by adjustable supports so that the upper arm remains in a fixed horizontal position throughout the test. The subject's forearm is allowed to move only in the vertical plane. At the start of the experiment, he is made to flex his arm by pulling on a cord that has been attached to a cuff on his wrist. The cord runs around a pulley system and supports a sizable weight. The initial angle between the forearm and upper arm is 135°. The subject is not given any specific instructions about maintaining this angle, except to relax his arm as much as possible while supporting the weight. Then, at time $t = 0$, an electromagnetic catch is switched off so that an additional weight is abruptly added to the original load. Changes in angular motion, $\theta(t)$, of the forearm about the elbow are recorded during and after the quick release of the weight. The mathematical model used to interpret the results of this test is based on the work of Soechting et al. (1971).

4.7.1.1   Limb Dynamics   Figure 4.11a shows a schematic diagram of the forearm, with the black filled circle representing the elbow joint. M_x represents the change in external moment acting on the limb about the elbow joint; in this experiment, M_x would be a step. M represents the net muscular torque exerted in response to the external disturbance. Neglecting the weight of the forearm itself, application of Newton's second law yields the following equation of motion:

$$M_x(t) - M(t) = J\ddot{\theta} \tag{4.94}$$

where J is the moment of inertia of the forearm about the elbow joint.
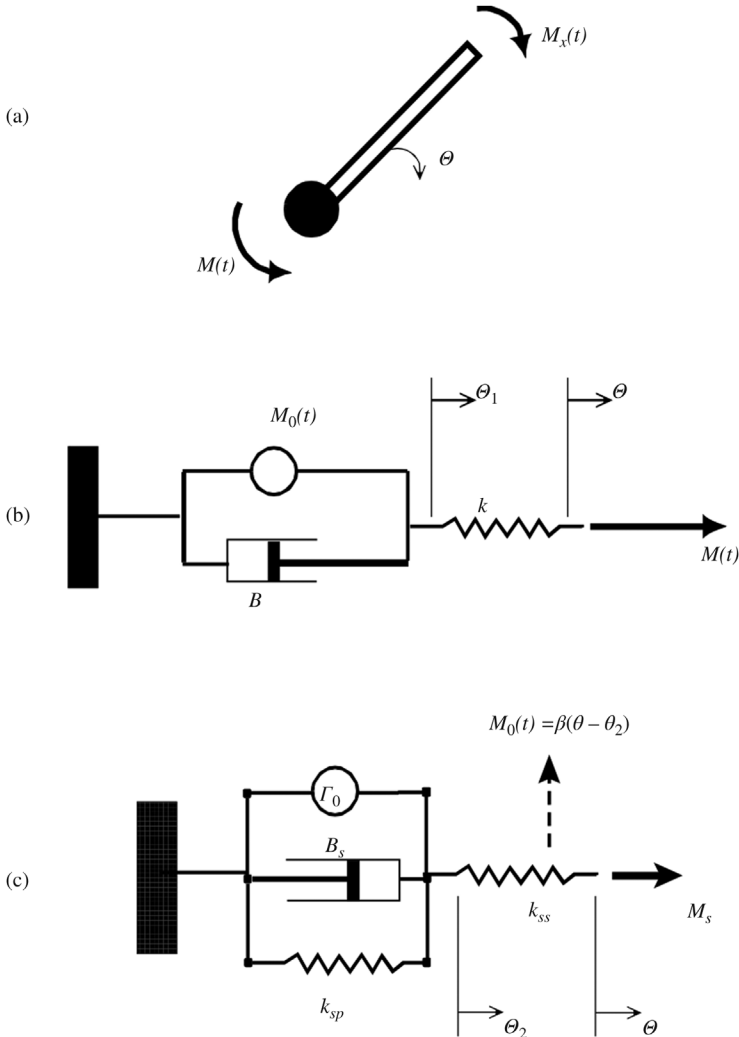
FIGURE 4.11   Components of the neuromuscular reflex model. (a) Limb dynamics. (b) Muscle model. (c) Muscle spindle model.

4.7.1.2   *Muscle Model*   Although this reflex involves both the biceps and triceps muscles, we will assume for simplicity that the net muscular torque in response to M_x is generated by a single equivalent muscle model, illustrated in Figure 4.11b. Note that in this mechanical analog, M is treated as if it were a "force," although it is actually a torque. Accordingly, the "displacements" that result are in fact angular changes, θ and θ_1. As such, the muscle stiffness parameter k and the viscous damping parameter B have units consistent with this representation. The equations

of motion for the muscle model are

$$M(t) = k(\theta - \theta_1) \tag{4.95}$$

and

$$M(t) = M_0(t) + B\dot{\theta}_1 \tag{4.96}$$

where $M_0(t)$ is the torque exerted by the muscle under isometric conditions. $M_0(t)$ is represented as a function of time, since it is dependent on the pattern of firing of the alpha motorneurons.

4.7.1.3 *Plant Equations*

By combining Equations 4.94 through 4.96 and eliminating θ_1 and $M(t)$, we obtain an equation of motion that characterizes the dynamics of the plant, that is, describing how θ would change due to the torque exerted by the external disturbance M_x and the resulting muscular response:

$$\frac{BJ}{k}\dddot{\theta} + J\ddot{\theta} + B\dot{\theta} = \frac{B}{k}\dot{M}_x(t) + M_x(t) - M_0(t) \tag{4.97}$$

4.7.1.4 *Muscle Spindle Model*

This model describes the dynamics by which changes in θ are transduced at the level of the muscle spindles into afferent neural signals. The latter travel to the spinal cord, which sends out efferent signals to the contractile machinery of the muscle to generate $M_0(t)$. We assume that the neural output of the spindle is proportional to the amount by which its nuclear bag region is stretched, so that ultimately

$$M_0(t) = \beta(\theta - \theta_2) \tag{4.98}$$

Figure 4.11c shows the mechanical analog of the muscle spindle model. k_{sp} and B_s are parameters that represent the elastic stiffness and viscous damping properties, respectively, of the pole region of the spindle, while k_{ss} represents the elastic stiffness of the nuclear bag region. Γ_0 represents the contractile part of the pole region, which allows the operating length of the spindle to be reset at different levels, using the gamma motorneuronal pathways. We will assume Γ_0 to be constant at the equilibrium length of the spindle, so that this parameter does not play a role in the dynamics of changes about this equilibrium length. With this consideration in mind, the dynamics of the muscle spindle model may be characterized by the following equations:

$$M_s = k_{ss}(\theta - \theta_2) \tag{4.99}$$

and

$$M_s = B_s\dot{\theta}_2 + k_{sp}\theta_2 \tag{4.100}$$

Another important factor that must be taken into account is the fact that, although θ is sensed virtually instantaneously by the spindle organs, there is a finite delay before this feedback information is finally converted into corrective action at the level of the muscle. This total delay T_d includes all lags involved in neural transmission along the afferent and efferent pathways as well as the delay taken for muscle potentials to be converted into muscular force. Eliminating the intermediate variables, M_s and θ_2, from Equations 4.98 through 4.100, we obtain the following equation for the feedback portion of the stretch reflex model:

$$\dot{M}_0 + \frac{M_0}{\tau} = \beta\left(\dot{\theta}(t - T_d) + \frac{\theta(t - T_d)}{\eta\tau}\right) \tag{4.101}$$

where

$$\tau = \frac{B_s}{k_{ss} + k_{sp}} \tag{4.102}$$

and

$$\eta = \frac{k_{ss} + k_{sp}}{k_{sp}} \tag{4.103}$$

4.7.1.5 Block Diagram of Neuromuscular Reflex Model
Taking the Laplace transforms of Equations 4.97 and 4.101, we obtain the following equations that are represented schematically by the block diagram shown in Figure 4.12:

$$\theta(s) = \frac{((B/k)s + 1)M_x(s) - M_0(s)}{s((BJ/k)s^2 + Js + B)} \tag{4.104}$$

and

$$M_0(s) = \beta\frac{\tau s + 1/\eta}{\tau s + 1}e^{-sT_d}\theta(s) \tag{4.105}$$
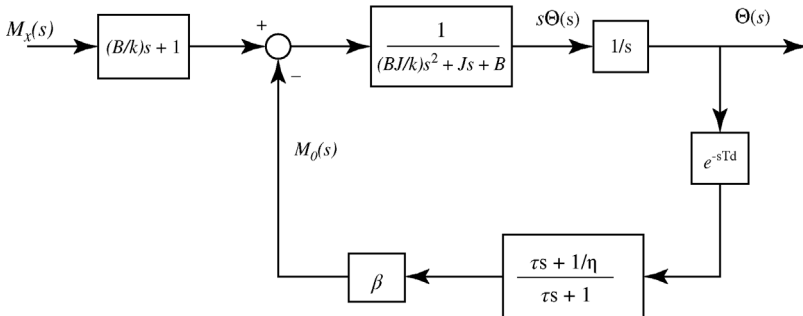
FIGURE 4.12   Block diagram of neuromuscular reflex model.

4.7.2   SIMULINK Implementation

The SIMULINK implementation of the neuromuscular reflex model is depicted in Figure 4.13. This program has been saved as the file `nmreflex.slx`. Note that the model parameters appear in the program as variables and not as fixed constants. This gives us the flexibility of changing the parameter values by entering them in the MATLAB command window or running a MATLAB m-file immediately prior to running the SIMULINK program. In this case, we have chosen the latter path and created an m-file called `nmr_var.m` that specifies the parameter values. The nominal parameter values used in the simulation are as follows: $J = 0.1\,\mathrm{kg\,m^2}$, $k = 50\,\mathrm{N\,m}$, $B = 2\,\mathrm{N\,m\,s}$, $T_\mathrm{d} = 0.02\,\mathrm{s}$, τ (`tau` in Figure 4.13) $= 1/300\,\mathrm{s}$, η (`eta` in Figure 4.13) $= 5$, and β (`beta` in Figure 4.13) $= 100$. These values are consistent with the average physiological equivalents found in normal adult humans.

Figure 4.14 displays the results of three simulation runs with `nmreflex.slx` using the nominal parameter values mentioned above. Figure 4.14a shows the time-course of the external disturbance, M_x, which is a step increase of $5\,\mathrm{N\,m}$ in the moment applied to the forearm. The solid tracing in Figure 4.14b represents the corresponding response in θ, the angular displacement of the forearm, when β was set equal to 100. Note that positive values of θ correspond to increases in the angle of flexion between the forearm and the upper arm. There is a slight overshoot in θ, followed by an almost undetectable oscillation before the steady-state value of approximately 0.25 rad is attained. Note that β represents the overall gain of the reflex arc. When β was increased to 150, the response was a damped oscillation, but the steady-state value achieved by θ became smaller
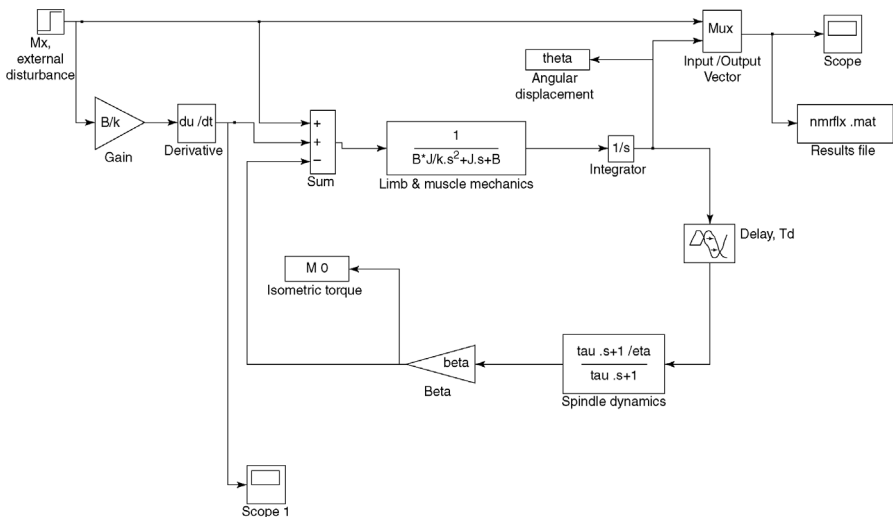
FIGURE 4.13   SIMULINK implementation of neuromuscular reflex model.
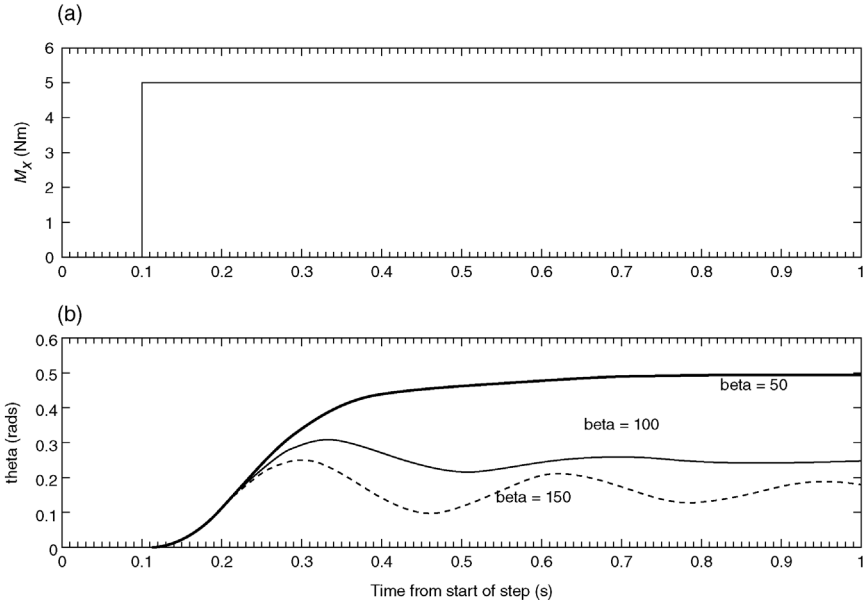
FIGURE 4.14   Sample results of simulations using the SIMULINK implementation of the neuromuscular reflex model.

than that obtained with the nominal value of β. In the third simulation, β was decreased to half the nominal value (i.e., 50). This produced an overdamped response and also resulted in a larger end-value for θ. These results reiterate the point that increased feedback gain leads to better attenuation of the effects of imposed disturbances – Higher values of β produced smaller ending values for θ. On the other hand, the responses also become more oscillatory. This issue of instability will be discussed further in Chapter 6.

4.8   SIMULINK APPLICATION 2: DYNAMICS OF GLUCOSE–INSULIN REGULATION

In Section 3.6, we examined the model of glucose and insulin regulation proposed by Stolwijk and Hardy (1974), but limited the scope of our analysis to the steady state. Here, we present the complete, dynamic version of this model as well as its SIMULINK implementation.

4.8.1   The Model

Employing the same notation as that presented in Section 3.6, Equations 3.40a and 3.40b, which characterize the mass balance of glucose in the blood plasma, may be

extended to incorporate dynamics in the following way:

$$C_G \frac{dx}{dt} = U(t) + Q_L - \lambda x - \nu xy, \quad x \le \theta \tag{4.106a}$$

$$C_G \frac{dx}{dt} = U(t) + Q_L - \lambda x - \nu xy - \mu(x - \theta), \quad x > \theta \tag{4.106b}$$

In the above equations, C_G represents the glucose capacitance in the extracellular space and $U(t)$ represents the time-course with which external glucose is infused into the bloodstream, as part of the "glucose tolerance test." Basically, Equations 4.106a and 4.106b state that the net difference between the rate at which glucose is added to the blood and the rate at which it is eliminated equals the rate at which the glucose concentration x will increase (or decrease). It is important to note that the cross-product term between x and y, the insulin concentration, makes the above equations nonlinear (or strictly speaking, bilinear). The corresponding dynamic mass balance for insulin is simply a straightforward extension of Equations 3.43a and 3.43b:

$$C_1 \frac{dy}{dt} = -\alpha y, \quad x \le \phi \tag{4.107a}$$

$$C_1 \frac{dy}{dt} = -\alpha y + \beta(x - \phi), \quad x > \phi \tag{4.107b}$$

where C_I is the insulin capacitance of the extracellular space.

The SIMULINK implementation of this model (filename: `glucose.slx`) is displayed in Figure 4.15b. The top half of the interconnected block structures represents Equations 4.106a and 4.106b, characterizing the dynamics of blood glucose buildup and elimination, while the bottom half models insulin dynamics, as described by Equations 4.104a and 4.104b. Saturation blocks are employed to function as thresholding operators, with the lower limit set equal to zero and the upper limit set equal to a very large number (so that there is effectively no saturation at the high end). One of these saturation blocks allows for the disappearance of the last term in Equation 4.106b when $x \le \theta$, where θ is the threshold concentration below which all glucose is reabsorbed in the kidneys, which results in no glucose loss in urine. The other saturation block in the insulin portion of the model allows for the disappearance of the last term in Equation 4.107b when $x \le \varphi$, where φ is the threshold glucose concentration for insulin secretion.

The glucose–insulin regulation model is encapsulated into a `Subsystem` block, which is shown in relation to the source and sink blocks in Figure 4.15a. The `Subsystem` block is created by dragging its icon from the *Commonly Used Blocks* sub-library to the model window. By double-clicking on the `Subsystem` block, a subsystem window will appear and the user can proceed to create the model in question within this window. `Inl` blocks are used for all signals entering
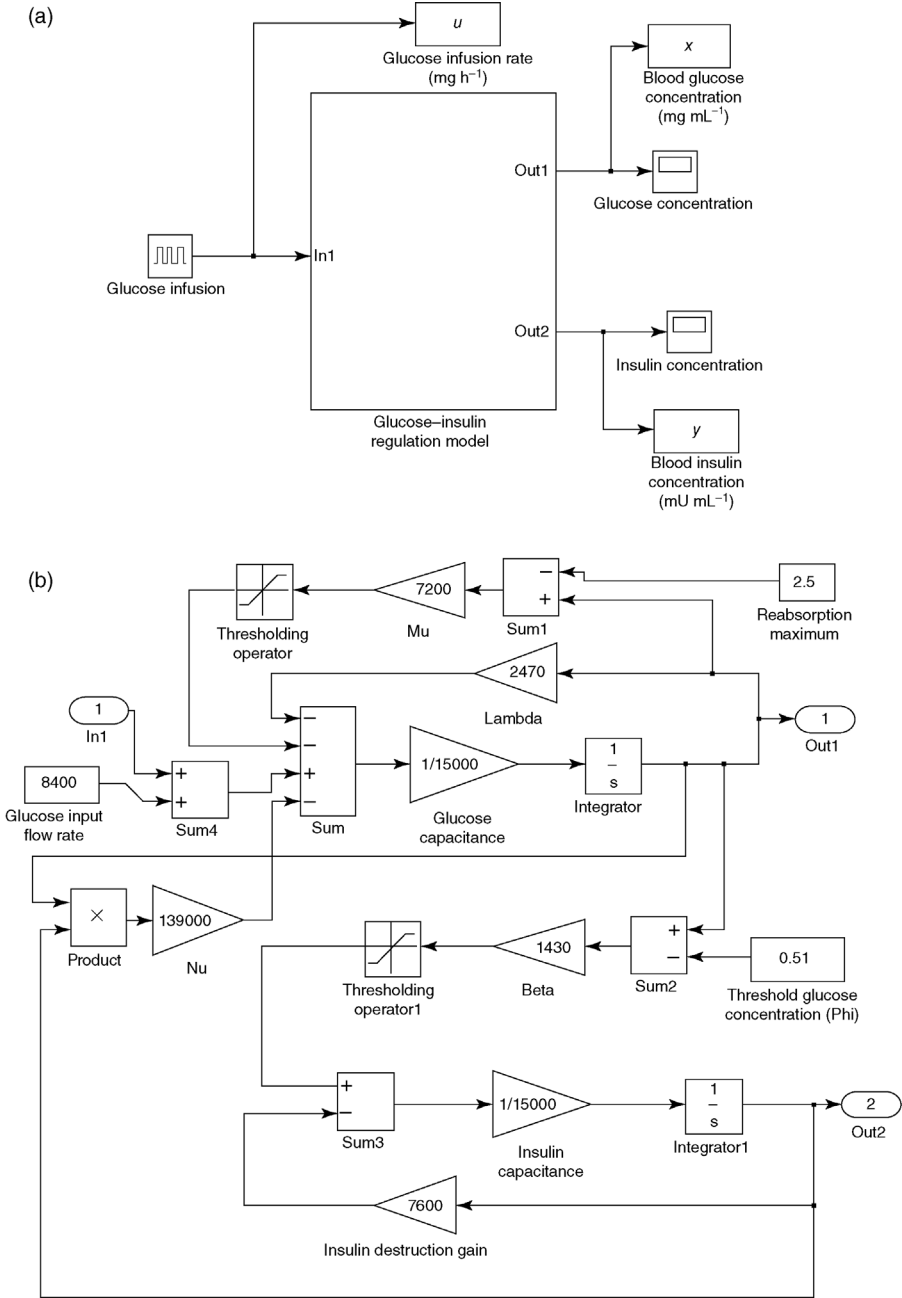
FIGURE 4.15 SIMULINK model of blood glucose–insulin regulation. Part (a) shows the input to and outputs from the model; part (b) shows details of the dynamic structure.
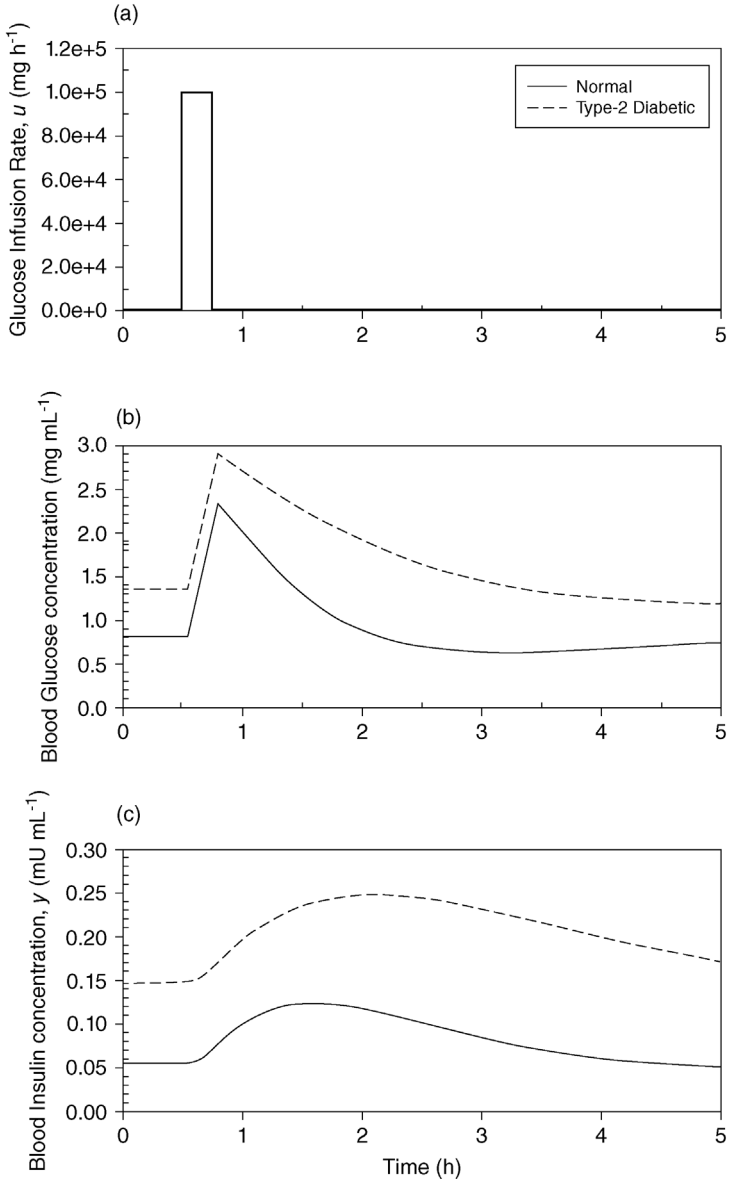
FIGURE 4.16 Response of the glucose–insulin regulation model to a rapid (15 min) infusion of 25 g of glucose in a simulated normal (solid lines) and type-2 diabetic (dashed lines).

the subsystem, while `Out1` blocks form the terminal points for all signals leaving the subsystem.

4.8.2 Simulations with the Model

Glucose infusion into the model is simulated through the use of a `Pulse Generator` block, which produces a rectangular wave. The period of the output waveform is set equal to the simulation time of 5 h, with time-step being 0.01 h. The duty cycle is set to 5% so that glucose infusion occurs over a duration of 0.25 h or 15 min, starting at $t = 0.5$ h. The glucose infusion rate (amplitude of the rectangular wave) is set equal to $100{,}000 \, \text{mg h}^{-1}$.

   Examples of two simulation runs are displayed in Figure 4.16. The input waveform is shown in Figure 4.16a. The resulting time-courses in glucose concentration and insulin concentration are shown in Figure 4.16b and c, respectively. Two classes of subjects are examined here: the normal adult (solid curves) and the type-2 diabetic (dashed curves). Note that in the diabetic, the steady-state levels for glucose and insulin are both higher than corresponding levels in the normal, which confirms what we had found a graphical method of solution in Section 3.6. In addition, in the diabetic, the decay of glucose and insulin concentrations toward steady-state levels following the infusion is noticeably slower compared to the normal subject. Furthermore, the glucose time-course does not show the slight undershoot exhibited by the corresponding time-course in the normal.

PROBLEMS

P4.1. Figure P4.1 shows a simplified linear model of the baroreflex regulation of heart rate and arterial blood pressure. ΔHR and ΔABP represent changes (from the mean) of heart rate and arterial blood pressure, respectively. The "blocks" represent the various physiological components of this reflex. G, k_b, and k_c represent constant gains (all with positive values), while τ represents time constant for the response of the sino-atrial (SA) node (in the heart) to changes in efferent vagal traffic (ΔVg). The expression shown in the box representing the SA node is its transfer function, with s being the Laplace variable. The external stimulus (labeled ΔZ) to this closed-loop system represents the effect of administering a vasoconstrictor drug (e.g., phenylephrine) that acts to raise blood pressure.

   (a) Assuming this model to be correct, describe how you think ΔABP, ΔVg, and ΔHR would change when the external stimulus $\Delta Z \, (>0)$ is applied in the form of a step (i.e., the vasoconstrictor drug, which acts to raise ABP) is infused continuously at a constant dose). Is this model a negative or positive feedback system? Which of the physiological components shown is primarily responsible for the negative or positive feedback?
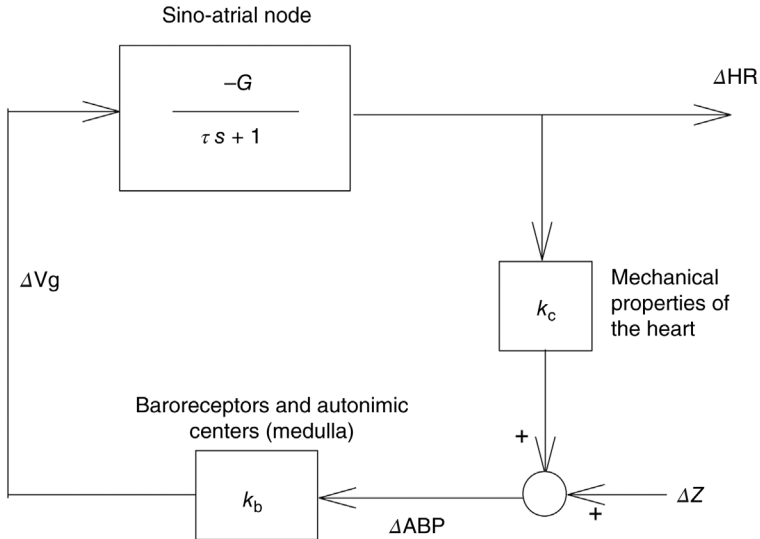
FIGURE P4.1   Simplified linear model of blood pressure and heart rate regulation.

(b) Deduce the transfer function relating ΔABP to ΔZ, with ΔZ as input and ΔABP as output.

(c) Assuming that $\tau = 1$, $G = 1$, $k_b = 1$, and $k_c = 1$, derive an expression for the time-course of ΔABP, following application of a *unit step* in ΔZ. Sketch this time-course as accurately as you can.

(d) Using the step-response derived in (c), deduce the impulse response of this system.

P4.2. Figure P4.2 displays the block diagram of a simplified, linear closed-loop model used to characterize the pupillary light reflex. The input is ΔL (change in light intensity) and the output is ΔA (change in pupil area), and the gains G_1 and G_2 are nonnegative.

(a) Derive expressions for the open- and closed-loop transfer functions of this system.
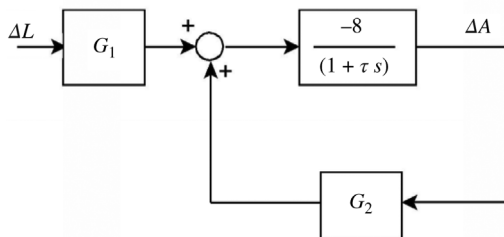
FIGURE P4.2   Simplified linear closed-loop model of pupillary light reflex.

(b) If ΔL takes the form of a unit step, determine how ΔA will respond (in the time domain) when $G_1 = 10$, $G_2 = 0$, and $\tau = 0.1$ s.

(c) If ΔL takes the form of a unit step, determine how ΔA will respond (in the time domain) when $G_1 = 10$, $G_2 = 1$, and $\tau = 0.1$ s.

(d) Is this model (when $G_2 > 0$) a negative-feedback control system? By comparing the open- and closed-loop transfer functions, determine how feedback affects the dynamics of the pupil's response to a step change in ΔL. Specifically, how does the response magnitude and speed of response change? Briefly explain how you arrived at your answer.

P4.3. In the simplified model of eye-movement control, displayed in Figure P4.3, J represents the moment of inertia of the eyeball about the axis of rotation, while B represents the viscous damping associated with the rotational movement of the eye. The target angular position of the eye, θ_{ref}, is set by the higher centers. G is a gain that converts the controlling signal into the torque exerted by the extraocular muscles. Information about the angular position of the eye, θ, is fed back to the controller with unity gain. Velocity information is also fed back with variable gain, K_v (>0). Deduce expressions for the responses of this system to a unit step change in θ_{ref} when

(a) there is no feedback at all;

(b) there is only position feedback ($K_v = 0$);

(c) both position and velocity feedback exist.
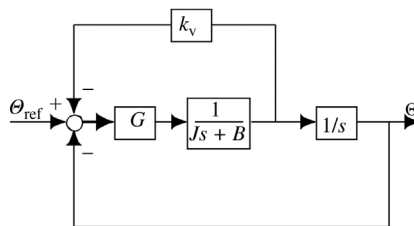
FIGURE P4.3   Simple model of eye-movement control.

P4.4. The following transfer function is one of the simplest linear approximations to the pure time delay T:

$$H(s) = \frac{1 - (Ts/2)}{1 + (Ts/2)}$$

Determine the open- and closed-loop responses for the system shown in Figure P4.4 when the input is a unit step.
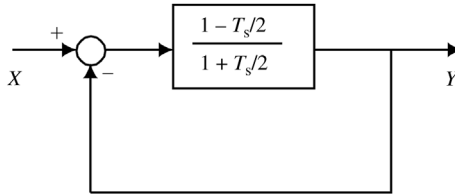
FIGURE P4.4   Closed-loop system containing time-delay approximation.

P4.5. Many types of physiological receptors exhibit the property of rate sensitivity. Carbon dioxide receptors have been found in the lungs of birds and reptiles, although it remains unclear whether such receptors are also found in human lungs. Figure P4.5 shows a highly simplified model of the way in which ventilation may be controlled by these intrapulmonary receptors following denervation of the carotid bodies. The feedforward element in the closed-loop system represents the gas exchange processes of the lungs, while the feedback element represents the dynamic characteristics of the intrapulmonary CO_2 receptors. The parameter α determines how rate-sensitive these receptors are. Determine the responses of this system to a large hyper-ventilatory sigh (which may be approximated by an impulse function) when (a) $\alpha = 0$ (there is no rate sensitivity), (b) $\alpha = 1/2$, and (c) $\alpha = 2$. (Note that the feedback element belongs to a class of systems known as lag-lead (when $\alpha < 1$) or lead-lag (when $\alpha > 1$) systems.)
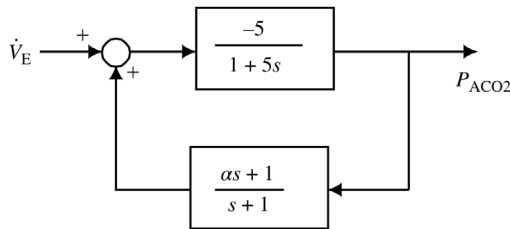
FIGURE P4.5   Simplified model of ventilatory control with intrapulmonary CO_2 receptor feedback.

P4.6. Figure P4.6 shows a highly simplified model of glucose–insulin regulation that has been linearized about its operating point under normal resting
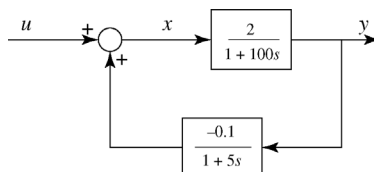
FIGURE P4.6   Simplified linear model of glucose–insulin regulation.

conditions. The variable u represents the amount of glucose entering the body from all sources external to the blood compartment, while x represents changes in blood glucose level and y represents changes in blood insulin level.

(a) Derive the open-loop transfer function of this system.

(b) Derive the closed-loop transfer function of this system.

(c) Following a step increase of 20 units of glucose entering the body, determine how the blood insulin level will change with time, and sketch this time-course.

P4.7. The SIMULINK program `glucose.slx` is a dynamic version of the glucose regulation model discussed in Section 3.6. Determine the time-courses of the concentrations of glucose and insulin in response to the steady infusion of glucose at the rate of 80,000 mg h^{-1} for a period of 1 h. The values of the other parameters are as given in Section 3.6. Compare these time-courses to the corresponding cases where the insulin production parameter β has been reduced to 20% of its nominal value.

P4.8. The degree of spasticity in patients with neuromuscular disorders can be quantified with the use of the "pendulum test." In this clinical procedure, the subject sits relaxed on a table with his lower leg initially supported by the medical examiner so that the knee joint is fully extended. The examiner abruptly releases the lower leg so that it swings freely until it finally comes to rest in the vertical position. The trajectory of the swing, as measured by the change in angle of knee flexion, can reveal information about the neuro-muscular stretch reflex. Modify the SIMULINK program `nmreflex.slx` so that it can be used to simulate this test. Note that the major difference between the pendulum test and the procedure described in Section 4.7 is that, here, the externally applied moment does not remain constant but varies according to the angular displacement of the lower leg, since it is a function of the weight of the lower leg and the moment arm between the center of gravity of the lower leg and the knee joint. Assume the same parameter values used in `nmr_var.m`, except for the following: moment of inertia of the lower leg about the knee joint = 0.25 kg m^2; length of lower leg = 40 cm, weight of lower leg = 5 kg. Determine how the trajectory of the lower leg would change with different values of stretch reflex gain β.

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5

FREQUENCY-DOMAIN ANALYSIS OF LINEAR CONTROL SYSTEMS

5.1 STEADY-STATE RESPONSES TO SINUSOIDAL INPUTS

While the impulse and step functions are useful test signals to use in the characterization of linear systems, it is difficult to find naturally occurring signals that approximate these highly idealized waveforms. Moreover, abrupt steps and impulsive changes are difficult to generate as test signals. On the other hand, periodic phenomena are a common occurrence in physiology. Since it is possible, using the *Fourier series*, to decompose any periodic signal into its sinusoidal components, sine waves represent a highly useful class of basic test inputs. Furthermore, sinusoidal changes are generally much easier to approximate in practice relative to other periodic and most nonperiodic signals.

5.1.1 Open-Loop Frequency Response

Consider the linearized respiratory mechanics model discussed in the previous chapter, expressed in differential equation form:

$$LC\frac{d^2P_A}{dt^2} + RC\frac{dP_A}{dt} + P_A = P_{ao} \tag{5.1}$$

Suppose the input P_{ao} were to assume the form of a sinusoidal waveform of amplitude X_0 and angular frequency ω. Note that ω is related to absolute frequency f by the following relationship:

$$\omega = 2\pi f \tag{5.2}$$

To simplify the mathematics, we employ the generalized sinusoidal (or complex exponential) function instead of the sine or cosine:

$$P_{ao}(t) = X_0 e^{j\omega t} \tag{5.3}$$

where X_0 is a real constant. Solution of the inhomogeneous differential equation, Equation 5.1, in which the right-hand side takes the particular form shown in Equation 5.3, yields a solution for P_A that contains two parts, as discussed previously in Section 2.6. The complementary function represents the transient part of the response, while the particular solution characterizes the steady-state response. In this discussion, we will be concerned only with the steady-state response in P_A. With the input given by Equation 5.3, the only way for equality to hold for arbitrary values of t (time) between the left- and right-hand sides of Equation 5.1 is for the particular solution of P_A to contain the function $e^{j\omega t}$. Thus, we assume the following form for $P_A(t)$:

$$P_A(t) = Z e^{j\omega t} \tag{5.4}$$

which states that the output of the system defined by Equation 5.1 must also be sinusoidal with the same frequency as the input signal. In Equation 5.4, we allow the function Z to be complex. Substituting Equation 5.4 into Equation 5.1, we obtain, after canceling $e^{j\omega t}$ from both sides of the equation and rearranging terms,

$$Z = H_0(\omega) X_0 \tag{5.5}$$

where

$$H_0(\omega) = \frac{1}{(1 - LC\omega^2) + jRC\omega} \tag{5.6}$$

$H_0(\omega)$ is a complex function of the frequency of the input and can be expressed in polar form as

$$H_0(\omega) = |H_0(\omega)| e^{j\phi_0(\omega)} \tag{5.7}$$

where the magnitude is

$$|H_0(\omega)| = \frac{1}{\sqrt{(1 - LC\omega^2)^2 + R^2 C^2 \omega^2}} \tag{5.8}$$

and the phase component is

$$\phi_0(\omega) = -\tan^{-1}\left(\frac{RC\omega}{1 - LC\omega^2}\right) \tag{5.9}$$

The complex function $H_o(\omega)$ represents the relationship between the sinusoidal input $P_{ao}(t)$ and the sinusoidal output $P_A(t)$. Substituting Equation 5.5 and 5.7 back into Equation 5.4, we obtain the following expression for $P_A(t)$:

$$P_A(t) = |H_0(\omega)|X_0 e^{j(\omega t + \phi_0(\omega))} \tag{5.10}$$

Since $P_{ao}(t)$ is $X_0 e^{j\omega t}$, Equation 5.10 implies that, although the output $P_A(t)$ remains sinusoidal at the same angular frequency ω, its amplitude and phase are different from those of $P_{ao}(t)$. The ratio between the output and input amplitudes, or the *gain*, is given by $|H_0(\omega)|$, while the *phase difference* is represented by $\phi_0(\omega)$. It is important to note that both gain and phase difference are functions of the forcing frequency ω. Figure 5.1 shows predictions of $P_A(t)$ produced by the lung mechanics model when $P_{ao}(t)$ assumed the form of sinusoidal waves of *unit amplitude* at absolute frequencies of 1, 4, and 8 Hz. The values of the parameters employed here were $R = 0.3$ cm H_2O s L^{-1}, $C = 0.1$ L cm H_2O^{-1}, and $L = 0.01$ cm H_2O s^2 L^{-1}. At very low frequencies, P_A oscillates virtually in synchrony with P_{ao} and is of the same amplitude. At very high frequencies, P_A lags substantially behind P_{ao} and is significantly attenuated. Using this set of parameters, however, there is a range of frequencies over which P_A is amplified and becomes larger in amplitude than P_{ao}. The amplification is greatest at approximately 4 Hz. If one determines the impulse or step response for this model with these parameter values, one would find an underdamped response with an oscillation frequency of about 4 Hz. Thus, exciting the system with an external sinusoidal input at this frequency produces *resonance*, since the applied forcing acts to reinforce the natural vibrations of the system.

The complex function $H_0(\omega)$ contains all the information shown in Figure 5.1 and much more. It predicts how the lung mechanics model will respond to sinusoidal inputs of unit amplitude and all possible frequency values. As such, it is also called the *frequency response* of the system. Figure 5.2 illustrates one method of graphically representing the frequency response of the lung mechanics model. At each absolute frequency f, we evaluate the gain and phase of $H_0(\omega)$. Two frequency responses are shown in this diagram. The first represents the under-damped system, with $R = 0.3$ cm s L^{-1}; this frequency response encompasses the results shown in Figure 5.1. The second frequency response shown represents the overdamped system where $R = 1$ cm s L^{-1}. In this case it is clear that there is no resonance peak, so the gain continually decreases with increasing frequency.

Note from Equation 4.4 that the transfer function corresponding to the lung mechanics model described by Equation 5.1 is

$$H_0(s) \equiv \frac{P_A(s)}{P_{ao}(s)} = \frac{1}{LCs^2 + RCs + 1} \tag{5.11}$$
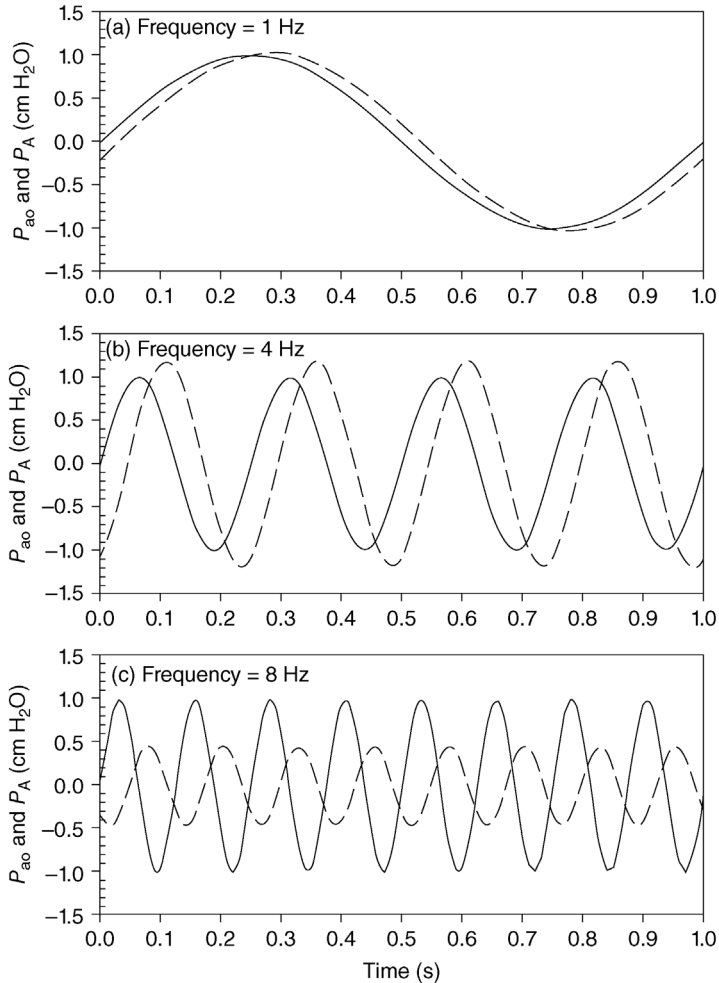
FIGURE 5.1   Steady-state responses (dashed tracings) at P_A of linearized lung mechanics model to sinusoidal excitation (solid tracings) at P_{ao}. Frequencies of sine waves are at (a) 1 Hz, (b) 3 Hz, and (c) 8 Hz.

An alternative (and shorter) approach to deriving the frequency response function $H_0(\omega)$ is by evaluating $H_0(s)$ along the imaginary axis on the s-plane, that is, *by setting* $s = j\omega$. Substituting $j\omega$ for s in Equation 5.11, we obtain

$$H_0(\omega) = \frac{1}{LC(j\omega)^2 + RCj\omega + 1} \tag{5.12}$$

Since $j^2 = -1$, rearranging terms in Equation 5.12 leads to Equation 5.6.
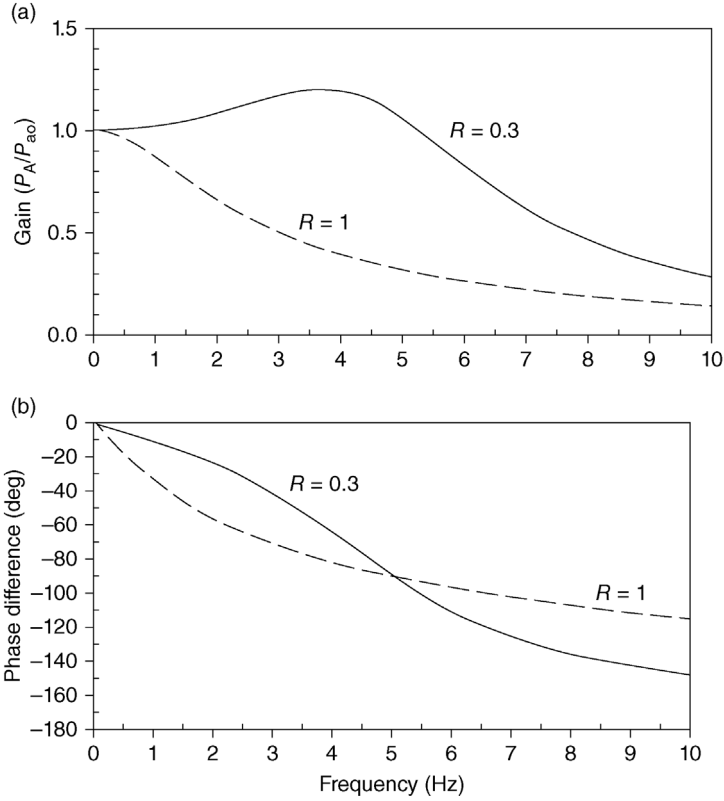
FIGURE 5.2 Frequency responses of the linearized lung mechanics model in under-damped (solid curves) and overdamped (dashed curves) conditions. The values shown for R are in cm $H_2O\,s\,L^{-1}$. Other parameter values are $C = 0.1\,L\,cm\,H_2O^{-1}$ and $L = 0.01\,cm$ $H_2O\,s^2\,L^{-1}$.

5.1.2 Closed-Loop Frequency Response

Now, consider the closed-loop situation where there is proportional feedback of $P_A(t)$. The corresponding transfer function would be as given in Equation 4.5b. Evaluating the frequency response, we obtain

$$H_c(\omega) = \frac{1}{(1 + k - LC\omega^2) + jRC\omega} \tag{5.13}$$

Thus, for the closed-loop case, the magnitude and phase of the frequency response are given by

$$|H_c(\omega)| = \frac{1}{\sqrt{(1 + k - LC\omega^2)^2 + R^2C^2\omega^2}} \tag{5.14}$$

and

$$\phi_c(\omega) = -\tan^{-1}\left(\frac{RC\omega}{1 + k - LC\omega^2}\right) \tag{5.15}$$

The closed-loop frequency responses are shown together with the open-loop responses in Figure 5.3. Here, the feedback gain k has been assumed to be unity. Closing the loop leads to a reduction of the steady-state gain from 1 to 0.5, that is, $|H_c(\omega=0)|=0.5$ compared to $|H_0(\omega=0)|=1$. This is consistent with the results that were presented in Chapters 2 and 4. Closing the loop also shifts the location of the resonance peak to a substantially higher frequency ($\sim$6.5 versus $\sim$4 Hz). This is also consistent with the impulse and step responses of the underdamped system in
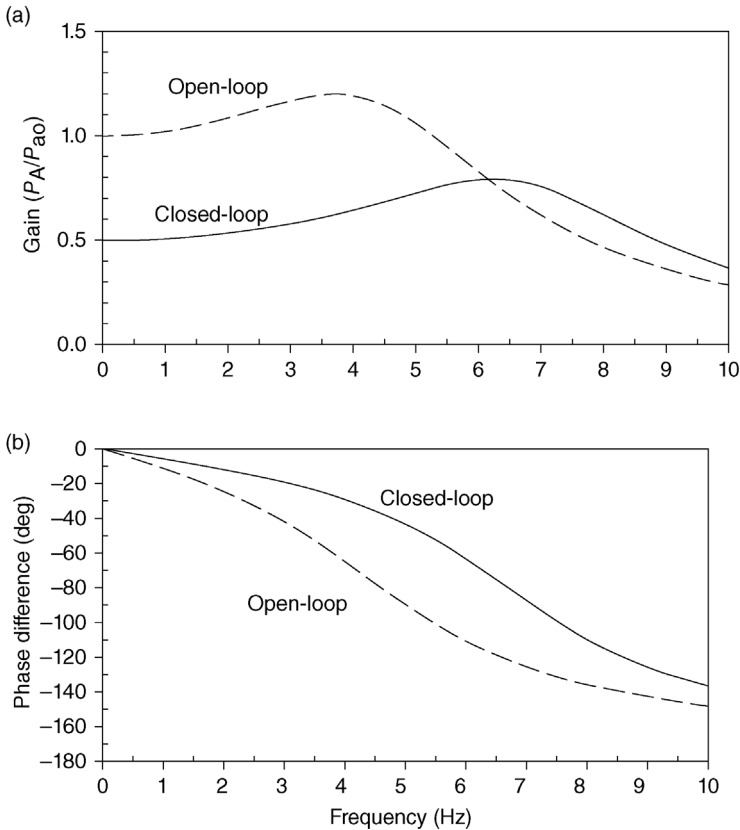
FIGURE 5.3 Frequency responses of the linearized lung mechanics model in closed-loop (solid curves) and open-loop modes. The values of the parameters used are $R=0.3$ cm $H_2O\,L^{-1}$, $C=0.1$ L cm H_2O^{-1}, $L=0.01$ cm $H_2O\,s^2\,L^{-1}$ and $k=1$ (or equivalently, $\lambda=2$).

Chapter 4 where we found an increase in frequency but decrease in amplitude of the transient oscillations. The phase portion of the frequency response shows a general decrease in the phase lag introduced by the closed-loop system vis-à-vis the open-loop system. This is equivalent to our earlier finding in Chapter 4 that closing the loop produces an increase in the speed of response of the system.

5.1.3   Relationship between Transient and Frequency Responses

Since the transfer function of any linear system is the Laplace transform of its impulse response and the frequency response can be deduced by replacing s in the Laplace transform by $j\omega$, it follows that the frequency response can be derived by taking the Fourier transform of the impulse response. This implies that one should be able to deduce various features of the transient response from the corresponding frequency response. Consider an idealized linear system with the following frequency response:

$$H(\omega) = e^{-j\omega\tau}, \quad -\omega_c \le \omega \le \omega_c = 0, \quad |\omega| > \omega_c \tag{5.16}$$

that is, this system has a gain of unity at angular frequencies between $-\omega_c$ and ω_c and zero gain at all frequencies outside this range. The *bandwidth* of a linear system is defined as the range of frequencies over which the system gain exceeds $1/\sqrt{2}$ or 0.7071 (see sections to follow). Thus, in this case, ω_c represents the system bandwidth. The phase of this system is linear with frequency within the bandwidth; the slope of the phase curve (line) is $-\tau$, that is, the output is delayed by τ relative to the input at all frequencies. This frequency response is displayed in Figure 5.4a and b.

To deduce the corresponding impulse response $h(t)$, we take the inverse Fourier transform of Equation 5.16:

$$\begin{aligned}
h(t) &= \frac{1}{2\pi} \int_{-\infty}^{\infty} H(\omega)e^{j\omega t}d\omega \\
&= \frac{1}{2\pi} \int_{-\omega_c}^{\omega_c} e^{-j\omega\tau}\, e^{j\omega t}\, d\omega \\
&= \frac{\omega_c}{\pi} \operatorname{sinc}(\omega_c(t-\tau))
\end{aligned} \tag{5.17}$$

where the function sinc(x) represents sin(x)/x. Figure 5.4c shows the form of $h(t)$. Note three important features in Equation 5.17. First, the impulse response peaks τ units of time after the input impulse has occurred, due to the delay inherent in H (ω). Second, the maximum value of the impulse response or the *peak amplitude* (see Section 4.4.2.1) is proportional to the bandwidth ω_c. This makes sense since a larger
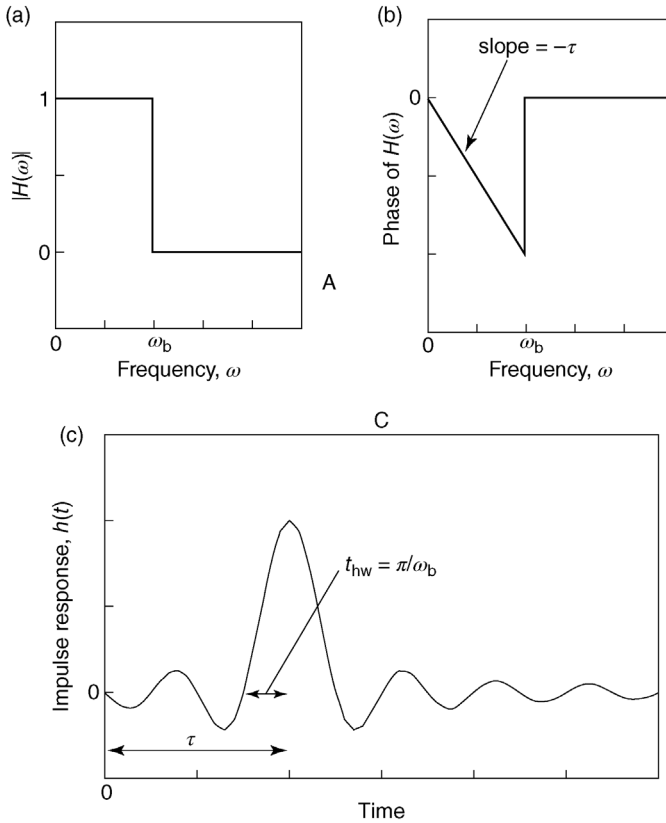
FIGURE 5.4 Relationship between frequency response (a and b) and the impulse response (c) of a linear system.

bandwidth allows the impulse response to be composed of a broader range of frequencies: In particular, the higher frequencies add sharpness or abruptness to the impulse response. In the limit, as bandwidth becomes infinite, the peak amplitude also becomes infinite as the impulse response approaches a delta function, that is, the same form as the input itself. The third important feature relates to the *half-width*, t_{hw}, of the impulse response, that is, the time taken for the main response to the impulse to fully develop. From Equation 5.17, t_{hw} can be deduced from the interval between the impulse response peak and the preceding zero-crossing (see Figure 5.4c):

$$t_{hw} = \frac{\pi}{\omega_c} \tag{5.18}$$

Since the step response is simply the integral of the impulse response, it also can be shown that the rise time T_r of the step response is proportional to t_{hw} and therefore

inversely proportional to ω_c. In fact, for a second-order system such as our linearized lung mechanics model, the following approximate relationship holds:

$$T_r \approx \frac{2}{\omega_c} \tag{5.19}$$

In Section 4.4.1, we showed that the *natural frequency* ω_n and *damping ratio* ζ are the two key parameters that characterize the impulse and step responses of the generalized second-order linear system. These parameters can be easily derived from the frequency response of the same system. Consider the linearized lung mechanics model in open-loop mode, the frequency response of which is given by Equations 5.8 and 5.9. Note from Equation 4.49 that $\omega_n = 1/(LC)^2$. Thus, $1 - LC\omega_n^2 = 0$, so that $\varphi_0(\omega_n) = -\tan^{-1}(\infty) = -90°$. Therefore, by locating the frequency at which the phase plot attains a phase lag of 90°, we can deduce ω_n. Evaluation of the frequency response magnitude at ω_n allows us to deduce ζ, since

$$|H_0(\omega_n)| = \frac{1}{RC\omega_n} = R\sqrt{\frac{C}{L}} = \frac{1}{2\zeta} \tag{5.20}$$

where the last equality in Equation 5.20 is based on Equation 4.50.

For the (open-loop) case where $L = 0.01$, $C = 0.1$, and $R = 0.3$, note that $\omega_n = 31.62 \, \text{rad s}^{-1}$, corresponding to a frequency of approximately 5 Hz. At this frequency, $|H_0(\omega_n)| \sim 1.05$. It is important to note that this does not correspond to the *resonant frequency*, which is located at $\sim$4 Hz ($\omega_r \sim 25$). Moreover, the peak value of $|H_0(\omega)|$, which occurs at the resonant frequency, is $\sim$1.2. In fact, resonance occurs at the natural frequency only when there is no damping in the system ($R = 0$).

5.2   GRAPHICAL REPRESENTATIONS OF FREQUENCY RESPONSE

5.2.1   Bode Plot Representation

In the field of control engineering, Bode plots represent one of the most popular classical methods for displaying the frequency response of a linear system. These plots are similar to but differ from the graphs presented in Figures 5.2 and 5.3 in that the gain (or magnitude) and frequency scales are presented in logarithmic form, while the phase remains on a linear scale. In addition, the frequency scale is generally displayed in terms of the angular frequency ω, and therefore in units of radians per second.

The gain of the frequency response $H(\omega)$ is expressed in units of *decibels* (dB), defined as follows:

$$\begin{aligned} G_{dB}(\omega) &= 10 \log_{10}|H(\omega)|^2 \\ &= 20 \log_{10}|H(\omega)| \end{aligned} \tag{5.21}$$

The first equality in Equation 5.21 is shown to emphasize the fact that for a linear system, $|H(\omega)|^2$ represents the ratio between the *power* of the output and the power of the input signal. The value of ω at which only half of the input signal power is transmitted is known as the *corner* or *cutoff* frequency ω_c. Thus, $|H(\omega_c)|^2 = 0.5$. Since $|H(\omega_c)| = 1/\sqrt{2}$, it turns out that the range of frequencies between 0 and ω_c is also the system bandwidth, as discussed previously. Using Equation 5.21, we find that $G_{dB}(\omega_c) = -3$ dB. This definition assumes only the case where the system takes the form of a low-pass filter, where the high frequencies are attenuated. For systems that are high-pass in nature, the corner frequency is defined as the frequency at which the input signal power is amplified and doubled at the output. In this case, $|H(\omega_c)|^2 = 2$, and therefore, $G_{dB}(\omega_c) = 3$ dB. Thus, in general, the corner frequency is the frequency at which the gain of the linear system is changed by 3 dB.

The logarithmic nature of G_{dB} represents one of the major strengths of the Bode plot. In general, it is possible to factorize the numerator and denominator of any given frequency response function into a cascade of first-order systems. For instance, consider the following generalized frequency response function:

$$H(\omega) = \frac{G_{SS}\prod_{m=1}^{M}(1 + j\omega\tau_m)}{(j\omega)^{N}\prod_{i=1}^{P}(1 + j\omega\tau_i)} \tag{5.22}$$

Note that in Equation 5.22, G_{SS} is the steady-state gain since $|H(\omega = 0)| = G_{SS}$. If we express the magnitude of $H(\omega)$ in terms of logarithmic gain, then

$$G_{dB}(\omega) = 20\log_{10}\left(\frac{G_{SS}\prod_{m=1}^{M}\sqrt{1 + \omega^2\tau_m^2}}{\omega^N\prod_{i=1}^{P}\sqrt{1 + \omega^2\tau_i^2}}\right) \tag{5.23a}$$

Evaluating the logarithm of the expression in curly brackets in Equation 5.23a, we obtain

$$G_{dB}(\omega) = 20\log_{10}G_{SS} + \sum_{m=1}^{M}20\log_{10}\left(1 + \omega^2\tau_m^2\right)^{1/2} + 20\log_{10}\omega^{-N}$$
$$+ \sum_{i=1}^{P}20\log_{10}\left(1 + \omega^2\tau_i^2\right)^{-1/2} \tag{5.23b}$$

This representation converts the logarithm of the products of several factors into equivalent sums of the logarithms of these factors. As a result, the contribution of each term is additive, which makes it easy to determine how the individual factors contribute to the overall gain. The overall phase of $H(\omega)$ can also be decomposed into the sum of all its individual components:

$$\phi(\omega) = \sum_{m=1}^{M}\tan^{-1}(\omega\tau_m) - \frac{N\pi}{2} - \sum_{i=1}^{P}\tan^{-1}(\omega\tau_i) \tag{5.24}$$

From the above example, one can see that it is useful to consider the magnitude and phase contributions of the basic components of $H(\omega)$, which take the form of either $(j\omega)^{-1}$ or $(1+j\omega\tau)^{\pm 1}$. Let us first consider the term $(j\omega)^{-1}$. Note that since $j\omega$ is purely imaginary and has no real part, its phase contribution is $\pi/2$ rad. Consequently, the phase contribution of $(j\omega)^{-N}$ is $-N\pi/2$ rad, the negative sign implying that the phase shift is a lag. The magnitude contribution of $(j\omega)^{-N}$ is infinite at zero frequency, but zero at infinite frequency.

Now, consider the term $(1+j\omega\tau)$. This factor adds a phase shift of $+\tan^{-1}(\omega\tau)$, that is, a phase lead, to $H(\omega)$. At very low frequencies ($\omega \ll 1/\tau$), this phase lead would be close to zero. At very high frequencies, that is, $\omega \gg 1/\tau$, this phase lead would approach $\pi/2$ rad or 90°. What about the gain contributions of this factor? At very low frequencies ($\omega \ll 1/\tau$), the gain would be $\sim 20\log_{10}(1)=0$ dB, that is, this would appear as a straight line on the zero-decibel axis. At very high frequencies ($\omega \gg 1/\tau$), the gain would approximate $20\log_{10}(\omega\tau)$ dB, and thus behave like a straight line (on the Bode plot) with a slope of 20 dB per decade. These two straight lines bound the actual gain plot and are known as the *low-frequency* and *high-frequency asymptotes*, respectively. Conversely, each $(1+j\omega\tau)^{-1}$ factor would contribute a phase shift of $-\tan^{-1}(\omega\tau)$, that is, a phase lag, to $H(\omega)$. As in the previous case, at very low frequencies, the phase lag would approach zero, while the low-frequency asymptote for the gain plot would coincide with the zero-decibel axis. At very high frequencies, the phase lag would approach 90°, and the high-frequency asymptote for the gain plot would be a straight line with slope -20 dB per decade. The Bode plots for these two basic functions are displayed in Figure 5.5, together with the low-frequency and high-frequency asymptotes.

The Bode plots of the frequency response of the linearized lung mechanics model discussed earlier are presented in Figure 5.6. It should be noted that these plots contain the same information as the linear frequency response plots displayed in Figure 5.3. One difference is that the frequency scale is expressed in terms of ω, in units of radians per second. Another important feature is that the logarithmic scaling enhances the appearance of the resonance peaks in both the open- and closed-loop systems. In both cases, the bandwidth of the system can be readily determined as the frequency range over which the gain lies above -3 dB.

5.2.2 Nichols Charts

Instead of presenting gain and phase in separate plots, an alternative approach is to plot the logarithmic magnitude in decibels versus phase for a range of frequencies. These plots are known as *Nichols charts*. The log-magnitude versus phase curves for the frequency responses $(1+j\omega\tau)$ and $(1+j\omega\tau)^{-1}$ are displayed in Figure 5.7. The closed circles on these plots correspond to the frequencies listed. In this case, the individual values of the product $\omega\tau$ are shown, so that the same plots would apply irrespective of the specific value of τ being employed. Figure 5.8 shows the Nichols charts for the linearized lung mechanics model in open- and closed-loop modes. The values placed next to the closed circles represent the corresponding
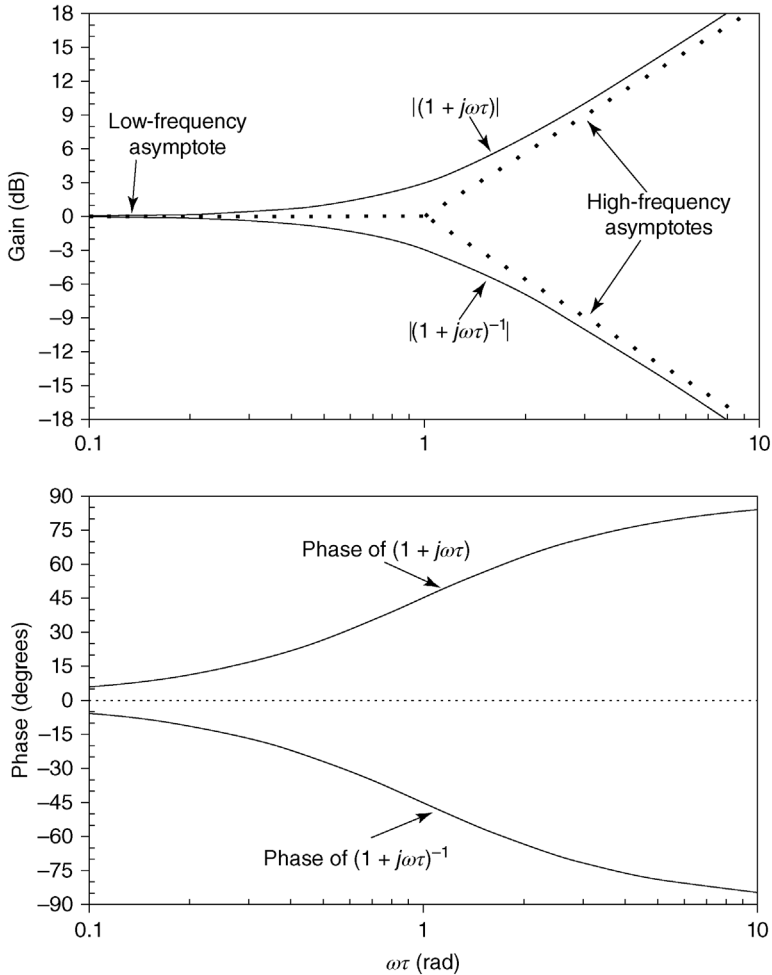
FIGURE 5.5   Bode plots of the first-order frequency response functions $(1 + j\omega\tau)$ and $(1 + j\omega\tau)^{-1}$. Note, in this case, that the "frequency scale" has been normalized and presented in terms of the product $\omega\tau$.

angular frequencies ω, in radians per second. These curves convey the same information that was contained in the Bode plots of Figure 5.6 and the linear frequency response plots of Figure 5.3. However, as we will see later, the shapes of these curves at the points where gain approaches 0 dB and the phase approaches 180° can yield useful information about system stability.

5.2.3   Nyquist Plots

Nyquist plots are sometimes also called *polar plots*. Here, the frequency response $H(\omega)$ is plotted on a plane in which the horizontal axis reflects the magnitude of the real
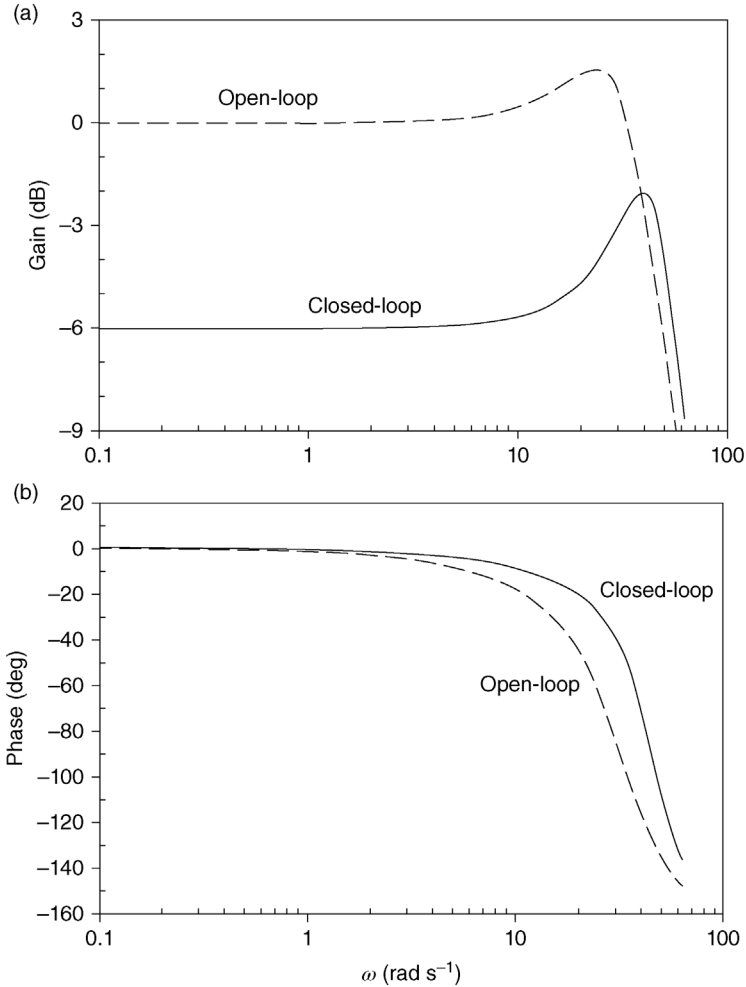
FIGURE 5.6   Bode plots of the frequency response of the linearized lung mechanics model in open- and closed-loop modes.

part of $H(\omega)$, while the vertical axis reflects the imaginary part. Thus, at any frequency ω, $H(\omega)$ is represented by a vector linking the origin to the point in question, and the length of the vector represents the magnitude of $H(\omega)$. As illustrated in the inset in Figure 5.9, the angle subtended by this line and the positive real axis represents φ, the phase of $H(\omega)$. The sign convention generally adopted is that *anticlockwise* rotations of the vector $H(\omega)$ from the positive real axis yield *positive values for* φ.

Nyquist plots corresponding to the basic frequency response functions $1/j\omega\tau$, $(1 + j\omega\tau)$ and $(1 + j\omega\tau)^{-1}$ are shown in Figure 5.9. The plot for $1/j\omega\tau$ coincides with the negative portion of the imaginary axis: When $\omega = 0$, $1/j\omega\tau$ is at $-j\infty$, but as ω becomes large, $1/j\omega\tau$ approaches the origin along the imaginary axis. The locus of $(1 + j\omega\tau)$
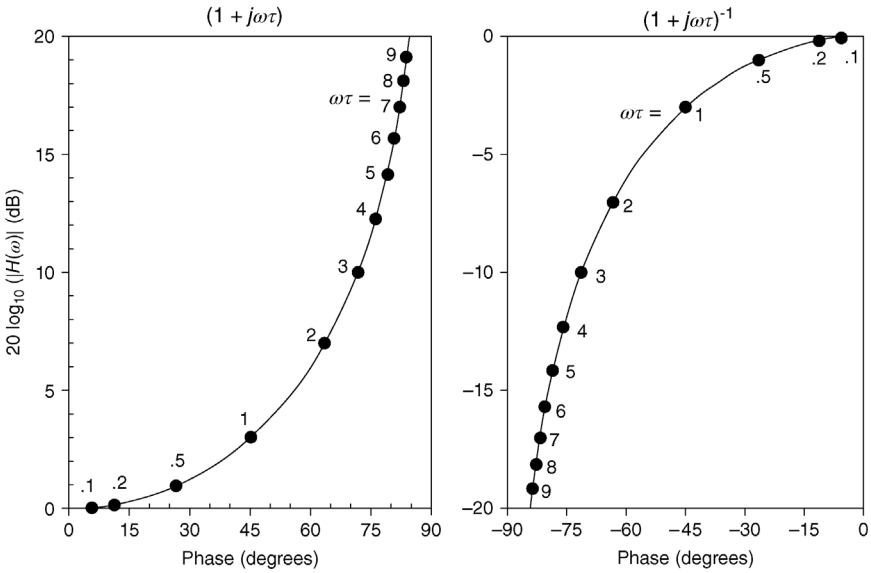
FIGURE 5.7 Nichols charts for the frequency response functions $(1 + j\omega\tau)$ and $(1 + j\omega\tau)^{-1}$.
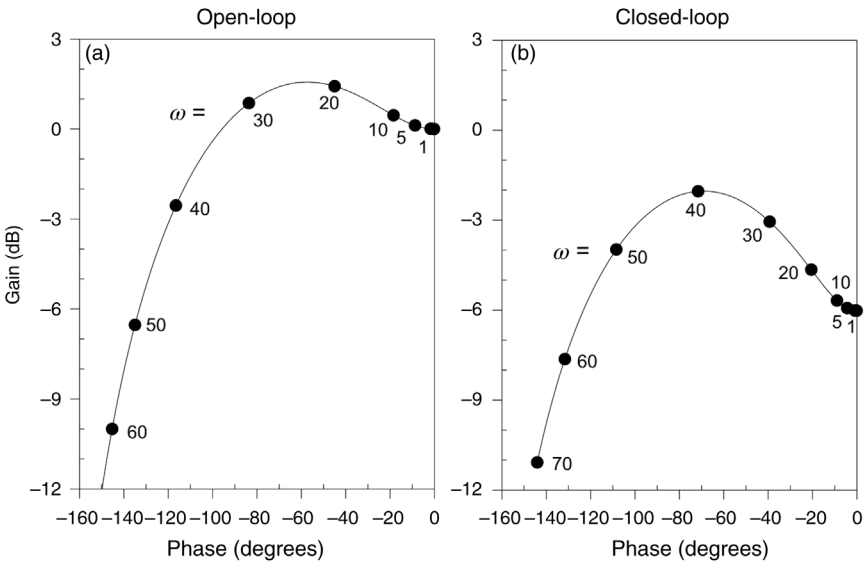
FIGURE 5.8 Nichols charts for the linearized lung mechanics model in open-loop (part (a)) and closed-loop (part (b)) modes. The parameter values assumed are the same as the values employed in Figures 5.3 and 5.5.
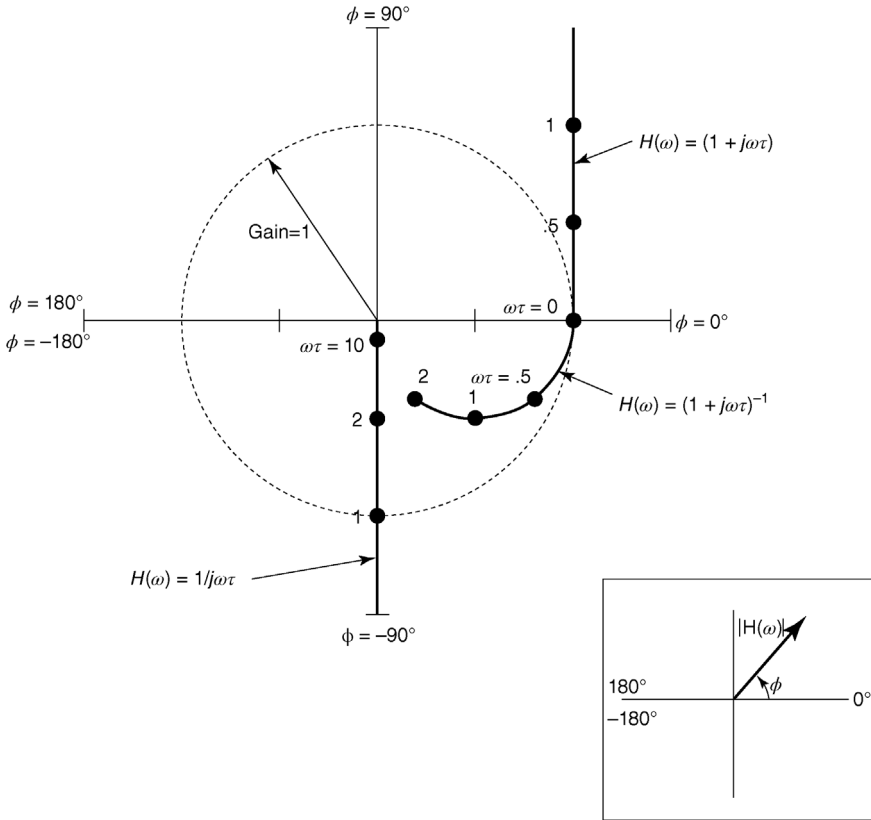
FIGURE 5.9 Nyquist plots for the basic frequency response functions $1/j\omega\tau$, $(1 + j\omega\tau)$, and $(1 + j\omega\tau)^{-1}$. Selected values of $\omega\tau$ are shown as closed circles. Dotted circle represents locus of points where gain equals unity. Inset shows definitions of $|H(\omega)|$ and ϕ – anticlockwise rotations of vector $H(\omega)$ from the positive real axis yield positive values of ϕ, and vice versa.

begins with a gain of unity on the real axis. As ω increases, this frequency response function moves vertically upward, tracing a path that is parallel with the positive imaginary axis. By contrast, the locus traced by $(1 + j\omega\tau)^{-1}$ is a semicircular arc that begins at 1 on the real axis when $\omega = 0$ and ends at the origin when $\omega = 4$.

The loop-like forms of the Nyquist plots presented in Figure 5.10 are more representative of the frequency responses of physiological systems. The particular plots shown characterize the frequency responses of the linearized lung mechanics model in both open- and closed-loop modes. These frequency responses represented here are exactly the same as those shown in Figure 5.3 where magnitude and phase were separately plotted against frequency. However, it is clear from the Nyquist plots that the points at which these curves intersect the imaginary axis (i.e., when $\varphi = -90°$ and $\omega = \omega_n$) do not correspond to the points of resonance at which the gain is maximum (and $\omega = \omega_r$).
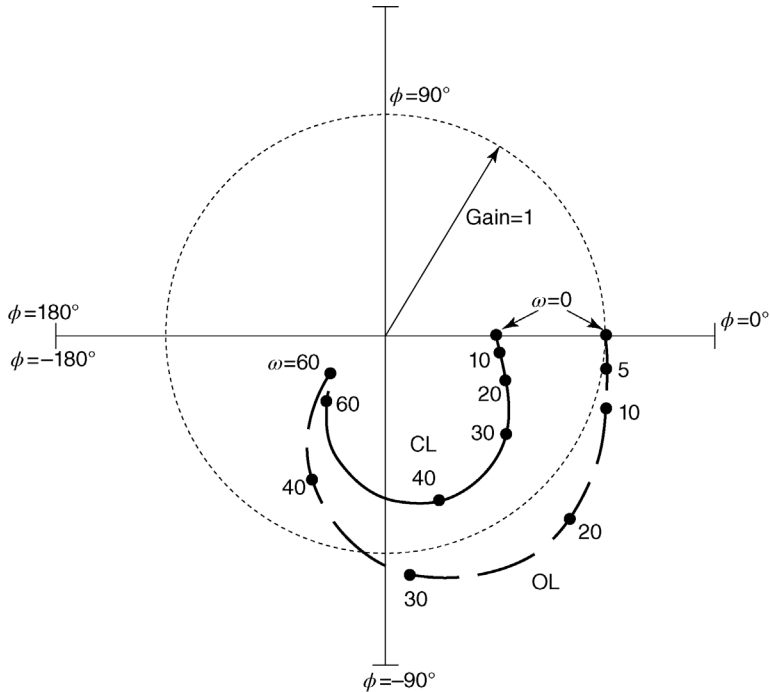
FIGURE 5.10 Nyquist plots for the frequency responses of the linearized lung mechanics model in open-loop (>OL=, broken curve) and closed-loop (>CL=, solid curve) modes. Dotted circle represents the locus of points at which the gain equals 1.

5.3 FREQUENCY-DOMAIN ANALYSIS USING MATLAB AND SIMULINK

5.3.1 Using MATLAB

To demonstrate the utility of employing MATLAB to examine the frequency response of a known system, we turn again to our linearized model of lung mechanics. We will use the closed-loop transfer function expression given in Equation 4.5b, since this can be converted to the open-loop expression by simply setting k equal to zero. Assuming that the values of L, R, C, and k have been preassigned, the following command lines set up the transfer function, Hs, of the model and produces the frequency vector w that contains the range of frequencies (in rad s^{-1}) to be examined:

```
>> num = [1];
>> den = [L*C   R*C   (1+k)];
>> Hs = tf(num, den);
>> f = 0:0.1:10;
>> w = 2*pi*f;
```

The MATLAB Control System Toolbox function `freqresp` is used next to compute the frequency response, `Hw`, of `Hs` over the frequency range of 0–10 Hz. Since `Hw` is a complex mutlidimensional array, the `squeeze` function is used to collapse it into a complex vector, which is subsequently decomposed into magnitude and phase components using the `abs` and `angle` functions, respectively. Finally, the magnitude and phase components are plotted against `w`.

```
>> Hw = freqresp(Hs, w);
>> Hwmag= abs(squeeze(Hw))';
>> Hwpha= 180*angle(squeeze(Hw))/pi;
>> subplot(2,1,1); plot(w,Hwmag);
>> ylabel('Freq Resp Magnitude');grid on;
>> subplot(2,1,2); plot(w,Hwpha);
>> xlabel('Frequency (rad/s)');
>> ylabel('Freq Resp Phase (deg)'); grid on;
```

The above command lines will produce linearly scaled frequency response plots of the type shown in Figures 5.2 and 5.3.

To produce Bode plots, the following commands can be used:

```
>> bode(Hs,w);
>> [Hwmag,Hwpha] = bode(Hs,w);
```

The first command line will lead to the automatic generation of Bode gain and phase diagrams. The second command line will not produce the plots, but will save the results in the variables `Hwmag` and `Hwpha`.

In similar fashion, the Nichols chart can be generated using the following lines:

```
>> nichols(Hs,w);
>> [Hwmag,Hwpha] = nichols(Hs,w);
```

Again, the second line will only compute the results but will not produce the plots.

Finally, the Nyquist plot can be produced as follows:

```
>> nyquist(Hs,w);
>> [Hwreal,Hwimag] = nyquist(Hs,w);
```

In this case, however, the second command line yields the real and imaginary parts of the frequency response, `Hw`, and not the magnitude and phase.

All of the above command lines are contained in a script file called `fda_11m.m`, which has been included in the library of MATLAB and SIMULINK files that accompany this book.
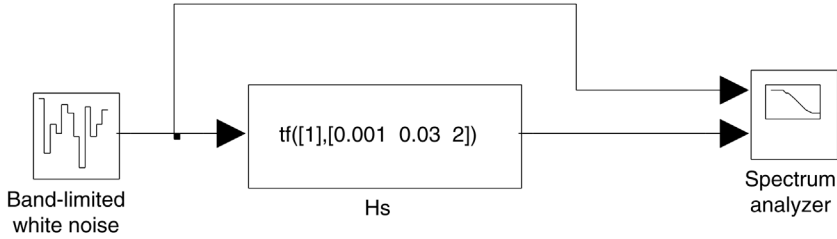
FIGURE 5.11   SIMULINK model used for determining the frequency response of the linearized lung mechanics (closed-loop) transfer function.

5.3.2   Using SIMULINK

The MATLAB functions described in the previous section are extremely useful when the exact transfer function of the system being analyzed is known. However, with more complicated models where there may exist several subsystems connected through forward and feedback loops, deriving a closed form for the overall frequency response can be very laborious. In such situations, an alternative approach would be to perturb the model with a known input, monitor the resulting output, and use both input and output to deduce the frequency response of the system. This is a basic *system identification* technique; systems identification and parameter estimation will be discussed in greater detail in Chapter 7.

Figure 5.11 provides an illustration of how the frequency response of our linearized lung mechanics model can be "measured." Here, we assume the following values for the model parameters (see Equation 4.5b): $L = 0.01$ cm $H_2O \, s^2 \, L^{-1}$, $R = 0.3$ cm $H_2O \, s \, L^{-1}$, $C = 0.1$ L cm H_2O^{-1}, and $k = 1$ (i.e., closed-loop mode), so that the particular transfer function employed here is given by

$$H(s) = \frac{1}{0.001s^2 + 0.03s + 2} \tag{5.25}$$

This model is represented in Figure 5.11 by the `LTI System` block, labeled `Hs`, found in the SIMULINK *Control System Toolbox* library. Although `Hs` is represented by the simple form shown in this example, in general it could be composed of several interconnected subsystems. The point of relevance here is that one has to identify the input and output that relate to the overall transfer function of the model. White noise is fed into the input of Hs, and both input and output are fed into a block known as `Spectrum Analyzer`, found in the *Sink* sub-library of the *DSP System Toolbox*. The `Spectrum Analyzer` produces the graphical results shown in Figure 5.12: Here Figure 5.12a displays the input and output time-courses, and Figure 5.12b and c show the frequency response magnitude and phase. Note that since the results displayed are computed from data sets of finite duration, it is inevitable that "noise" will appear in the estimated frequency response plots. The model shown in Figure 5.11 has been saved in the SIMULINK model file `fdallm.slx`, which may be found in the library of MATLAB/SIMULINK files that accompany this book.
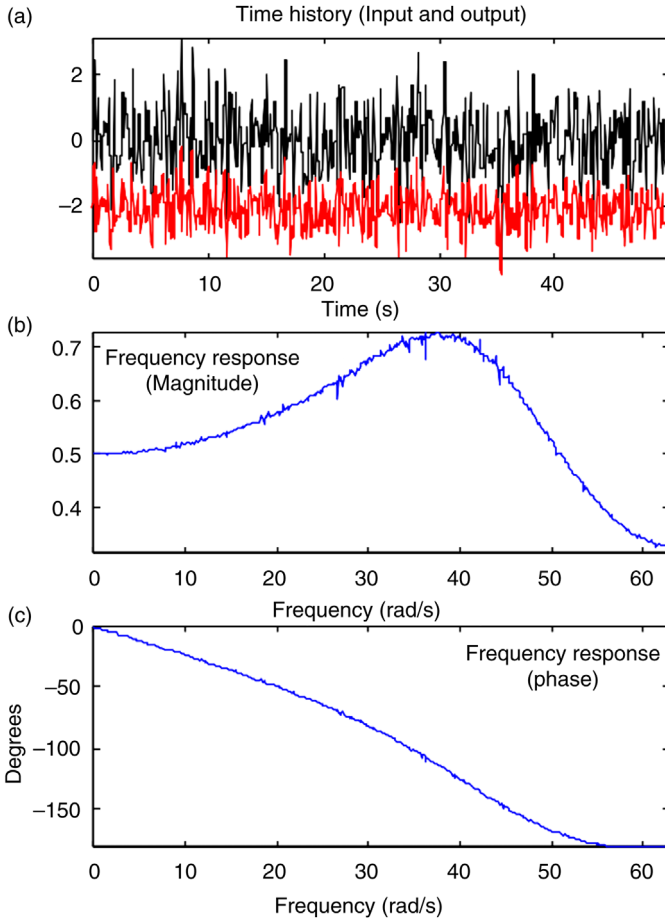
FIGURE 5.12 Determination of frequency response of linearized lung mechanics model using the SIMULINK program shown in Figure 5.11. Part (a) shows time-courses of the input to and output from the model. Note that the actual mean level of the output signal (lower tracing) was zero: The displayed time-course was shifted vertically to enhance clarity of presentation.

An important detail that the user should note is that, when setting up `Spectrum Analyzer`, a value is required for the `Sample time`. This value allows the block to assign a timescale to the input and output data. The `Sample time` also determines the frequency range over which the estimated frequency response is plotted. In our example, we were concerned with the frequency range of 0–10 Hz, corresponding to a range in ω of about 0–63 rad s^{-1}. Thus, we chose a minimum sampling frequency of 20 Hz, which translated to a sampling interval of 0.05 s.

5.4 ESTIMATION OF FREQUENCY RESPONSE FROM INPUT–OUTPUT DATA

5.4.1 Underlying Principles

The algorithm embedded in `Spectrum Analyzer` is based on the following principle that is valid for all linear systems: The frequency response of the system under study can be derived by dividing the cross-spectrum of the input and output by the spectrum of the input. This principle is derived from the basic linear properties of convolution and superposition. Assuming $x(t)$ and $y(t)$ to represent the input and output of the linear system with impulse response $h(t)$, we begin by recalling the convolution equation displayed in Equation 2.45:

$$y(t) = \int_0^\infty h(t')x(t - t')dt' \qquad (5.26)$$

Multiplying both sides of Equation 5.26 by $x(t - \tau)$ and then taking expectations, we obtain

$$E[x(t - \tau)y(t)] = \int_0^\infty h(t')E[x(t - \tau)x(t - t')]dt' \qquad (5.27a)$$

where the *expectations* operator $E[\cdot]$ is defined by

$$E[z] = \int_{-\infty}^\infty zp(z)dz \qquad (5.28)$$

and $p(z)$ is the probability distribution function of the variable z. However, by definition, the left-hand side of Equation 5.27a yields the cross-correlation function between x and y, $R_{xy}(\tau)$, while the expectation term on the right-hand side is equal to the autocorrelation function of x, $R_{xx}(\tau)$. Thus, we replace Equation 5.27a with

$$R_{xy}(\tau) = \int_0^\infty h(t')R_{xx}(\tau - t')dt' \qquad (5.27b)$$

It can be shown that *Fourier transformation* of R_{xy} and R_{xx} yields the cross-spectrum S_{xy} and autospectrum S_{xx}, respectively; this equivalence principle is also known as the *Wiener−Khinchine* theorem. And since the frequency response $H(\omega)$ is obtained by Fourier transforming $h(t)$, the time-convolution on the right-hand side of Equation 5.27b can be converted into a product in the frequency domain:

$$S_{xy}(\omega) = H(\omega)S_{xx}(\omega) \qquad (5.29a)$$

where

$$S_{xy}(\omega) = \int_{-\infty}^{\infty} R_{xy}(\tau)e^{-j\omega\tau}d\tau \tag{5.30a}$$

$$S_{xx}(\omega) = \int_{-\infty}^{\infty} R_{xx}(\tau)e^{-j\omega\tau}d\tau \tag{5.30b}$$

$$H(\omega) = \int_{-\infty}^{\infty} h(\tau)e^{-j\omega\tau}d\tau \tag{5.30c}$$

Thus, the frequency response can be estimated from

$$H(\omega) = \frac{S_{xy}(\omega)}{S_{xx}(\omega)} \tag{5.29b}$$

Note that for good estimates of $H(\omega)$ to be derived from Equation 5.29b, it is important for S_{xx} to be positive over the range of interest for ω. Ideally, the input or stimulus signal should have a spectrum that is relatively flat over a wide bandwidth, that is, the input should be *broadband*. In fact, the optimal type of input, from the viewpoint of estimation, is white noise (see Chapter 8).

5.4.2 Physiological Application: Forced Oscillation Technique in Respiratory Mechanics

The "forced oscillation technique" has been used to noninvasively assess lung mechanical function in humans. Figure 5.13 shows how this is implemented in the clinical setting. The subject breathes through a mouthpiece or face mask that is connected to a breathing circuit with a loudspeaker at the other end. The loudspeaker is used to generate fluctuations in airflow to perturb the subject's respiratory system. At the same time, the circuit is flushed by a bias flow that brings fresh air into the system and removes exhaled CO_2. The tubing and inlets associated with the bias flow present high impedance to the high-frequency oscillations but low resistance to spontaneous breathing, so that the subject is able to breathe normally during the test. The oscillatory airflow delivered into the subject's mouth is measured using a pneumotachograph. The airway opening pressure is measured near the mouthpiece. Thus, in this example, the airflow generated by the loudspeaker is considered the "input," whereas the resulting fluctuations in pressure are considered the "output" of the respiratory system in response to the random flow perturbations. The transfer function relating the input airflow perturbations to the output pressure fluctuations is called the "respiratory system impedance," Z_{rs}. If we assume the lungs act as a single
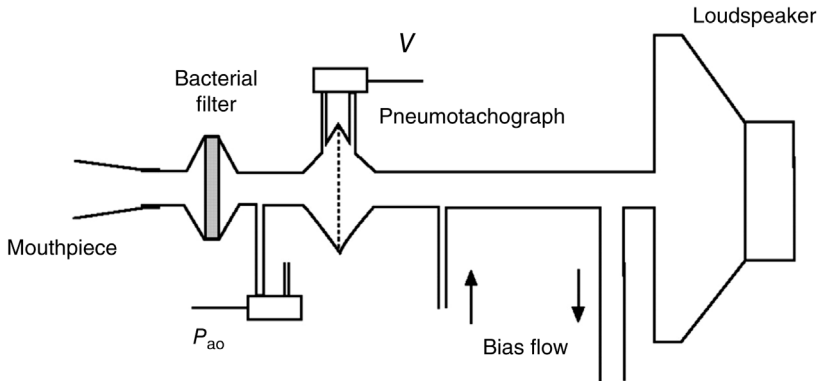
FIGURE 5.13 Schematic illustration of the equipment setup for the forced oscillation technique. (Reproduced with permission from Oostveen et al. (2003).)

compartment, the $R-L-C$ electrical analog (see Figure 4.1) would provide an approximate model of respiratory system mechanics:

$$P_{ao} = L\frac{dQ}{dt} + RQ + \frac{1}{C}\int Qdt \qquad (5.31)$$

where Q is the airflow. Taking Laplace transforms of both sides, we get

$$P_{ao}(s) = LsQ(s) + RQ(s) + \frac{Q(s)}{Cs} \qquad (5.32)$$

Thus,

$$Z_{rs}(s) \equiv \frac{P_{ao}(s)}{Q(s)} = Ls + R + \frac{1}{Cs} \qquad (5.33)$$

By setting $s = j\omega$, we obtain an expression for the frequency dependence of respiratory system impedance, that is, the frequency response of this system:

$$Z_{rs}(\omega) = R + j\left(L - \frac{1}{C\omega}\right) \qquad (5.34)$$

Note that the real part of this frequency response is simply the resistance and independent of frequency. The imaginary part is often called the "reactance" and contains contributions from inertance and compliance that are of opposite signs. When frequency is low, the compliance contribution to reactance is large and highly negative. When frequency is high, the inertance contribution dominates and reactance becomes highly positive. At the resonant frequency, the two contributions cancel each other out and reactance becomes zero; at this point, the impedance consists solely of the resistance contribution.

5.5   FREQUENCY RESPONSE OF A MODEL OF CIRCULATORY CONTROL

The regulation of heart rate and systemic blood pressure is achieved in short term primarily through the feedback control via the arterial baroreflexes. However, both cardiovascular variables are continually perturbed by respiration. Breathing can affect heart rate and arterial blood pressure through a number of mechanisms. First, respiratory-induced intrathoracic pressure changes exert a direct effect on arterial pressure, which, in turn, affects heart rate through the baroreflexes. Second, the present evidence suggests a direct coupling between the respiratory pattern generator in the medulla and the autonomic centers that influence heart rate. Third, vagal feedback from the pulmonary stretch receptors during breathing has been shown to reflexively affect heart rate. Finally, changes in heart rate can lead to changes in cardiac output, which, in turn, produce arterial blood pressure fluctuations that alter heart rate through the baroreflexes. The overall effect of respiration on heart rate, commonly referred to as the *respiratory sinus arrhythmia*, can be quantified in terms of a frequency response function. Changes in phase and/or magnitude of this frequency response function would suggest changes in one of the factors that influence autonomic control of heart rate.

5.5.1   The Model

The model of circulatory control that we will examine was developed by Saul et al. (1991) from the Harvard Medical School and Massachusetts Institute of Technology. The SIMULINK implementation of this model (filename: `rsa.slx`) is shown in Figure 5.14. Respiration, measured in the form of lung volume change V is
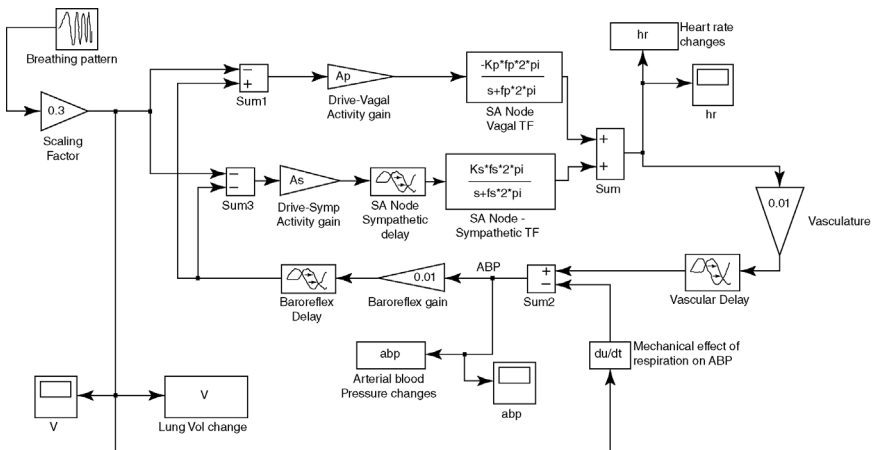
FIGURE 5.14   SIMULINK model of circulatory control that accounts for the effect of respiration on heart rate and arterial blood pressure.

assumed to directly affect the autonomic inputs to the sino-atrial node: Inspiration leads to decreases in both vagal and sympathetic efferent activity (note signs in summing blocks). The model does not distinguish between respiratory inputs from the pulmonary stretch receptors from the central drive that originates in the medullary centers. Feedback from the baroreceptors also directly influences the autonomic inputs to the heart: A rise in arterial blood pressure, `abp`, produces a decrease in sympathetic activity and an increase in parasympathetic activity. During inspiration, the decrease in vagal efferent activity acts on the sino-atrial node to increase heart rate `hr`. The transfer function that models the dynamics of this relationship is a simple low-pass filter with a cutoff frequency (f_p) that is on the order of 0.2 Hz and a negative gain $-K_p$. In contrast, the response of the sino-atrial node to sympathetic stimulation is considerably slower. In addition to a latency of 1–2 s, the transfer function that characterizes the dynamics of sympathetic activity to heart rate conversion has a cutoff frequency f_s of 0.015 Hz. In this case, the gain is positive.

Changes in heart rate are assumed to affect arterial blood pressure after a delay of 0.42 s. For simplicity, the transfer function representing the properties of the arterial vasculature is assumed static with a gain of 0.01 mmHg min bt^{-1}. In addition, since the transduction of `abp` into baroreceptor output occurs with very rapid dynamics, we assume that the baroreflex can be adequately represented by a static gain (equal to 0.01) in series with a fixed delay of 0.3 s. Finally, the direct mechanical effects of respiration on `abp` are modeled as a negative differentiator, that is, inspiration tends to decrease `abp`, while expiration tends to increase it. Thus, the model simulates respiratory sinus arrhythmia by allowing the direct autonomic stimulation of heart rate. In addition, the resulting changes in heart rate and the direct mechanical effects of respiration produce fluctuations in `abp`, which subsequently affect `hr` via the baroreflexes.

5.5.2   Simulations with the Model

To determine the frequency response of the circulatory control model, we employ a source block that produces a `chirp` signal. This is a sine wave, the frequency of which increases linearly with time. In our case, we set the parameters of the chirp block such that we start off with a frequency of 0.005 Hz and end with a frequency of 0.5 Hz after a duration of 300 s (simulation time). Since the amplitude of the chirp signal is not adjustable, a gain block of 0.3 is included between the source block and the rest of the model. This limits the peak-to-peak amplitude of the "respiration signal" to 0.6 L. Before starting the simulation, the m-file `rsa_var.m` has to be executed in order to assign values to the adjustable parameters of the model. The following nominal parameter values represent the normal subject in supine posture: SA node vagal transfer function gain, $K_p = 6$; SA node sympathetic transfer function gain, $K_s = 18$; SA node vagal transfer function cutoff frequency, $f_p = 0.2$ Hz; SA node sympathetic transfer function cutoff frequency, $f_s = 0.015$ Hz. The relative weight factors for the conversion of respiratory drive or baroreflex drive to

efferent neural activity are A_p (for the vagal branch) $= 2.5$ and A_s (for the sympathetic branch) $= 0.4$.

Figure 5.15 displays the results obtained from one simulation run; for the sake of clarity, only 100 s of the simulated "data" are shown. Figure 5.15a shows the chirp signal (respiratory input) used to stimulate the model. The corresponding changes in heart rate predicted by the model are displayed in Figure 5.15b. Note that at low frequencies, heart rate fluctuates almost in synchrony with lung volume change; however, at the higher frequencies, it tends to lag respiration. Also, the amplitude of
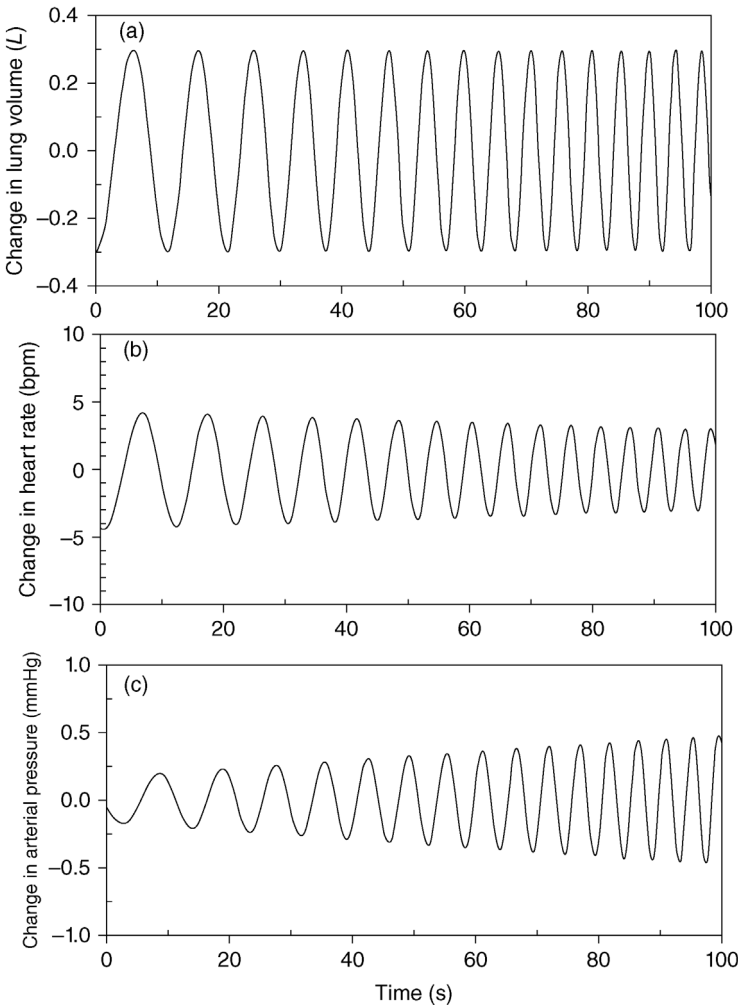
FIGURE 5.15   Responses in heart rate and arterial blood pressure to a controlled breathing pattern (slow-to-high frequency), as predicted by the SIMULINK model of circulatory control ("normal" conditions).

the heart rate signal decreases with increasing frequency, underscoring the low-pass nature of the overall frequency response. The predicted behavior of arterial blood pressure is somewhat different: As frequency increases, the respiratory-induced changes in abp become larger. This results from the growing influence of the direct mechanical effects of breathing on blood pressure as frequency increases.
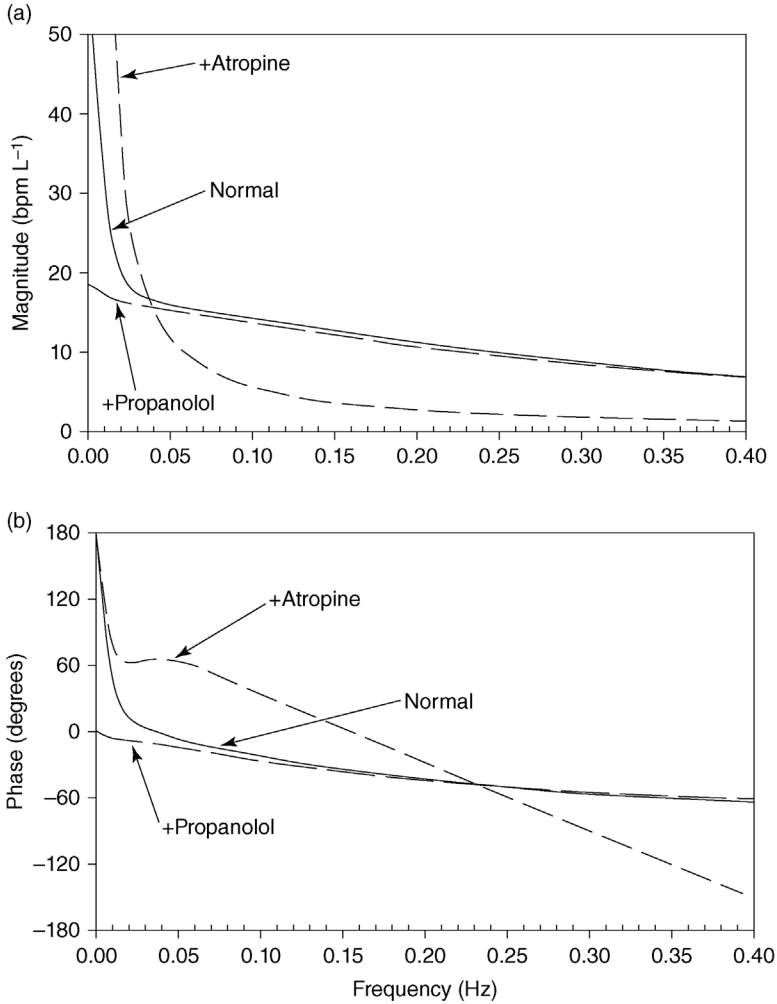
5.5.3 Frequency Response of the Model

Using the method described in Section 5.4.1, the frequency response of the model can be deduced from the input and simulated output. Instead of inserting the Spectrum Analyzer block, the reader can also save the input (V) and output (hr or abp) variables to the Workspace, and use the following MATLAB code (saved as rsa_tf.m) to deduce the frequency response:

```
% We assume the sampling interval is 0.1 s so that N=3000
% for a total simulation time of 300 s
>> freq = [0:1/300:5]';
% compute Power spectrum of V and Cross-spectrum between
% V and hr
>> Pv = psd(V,N,10);
>> Pvhr = csd(V,hr,N,10);
% compute Frequency Response magnitude and phase
>> Hvhr = Pvhr./Pv;
>> Hvhrmag = abs(Hvhr);
>> Hvhrpha = angle(Hvhr)*180/pi;
```

The chirp signal is useful as an input waveform since it produces a reasonably broad spectrum over the frequency range of interest: 0–0.4 Hz. Figure 5.16 displays the magnitude (part (a)) and phase (part (b)) components of the frequency response between respiration and heart rate estimated for the simulated supine normal subject (solid curves). The low-pass nature of the magnitude response is clearly evident; however, the frequency response values toward the low (0 Hz) and high (0.4 Hz) ends of the range displayed cannot be regarded as accurate since most of the spectral power of the chirp input is contained in the frequencies in the middle of this range.

The results of two other simulation cases are also presented in Figure 5.16. The first simulates how the frequency response of the respiratory sinus arrhythmia would change if the "subject" were given a dose of atropine (+atropine, dashed curves) that produces complete *parasympathetic blockade*. In addition, the model parameters are also modified to simulate the subject in a standing posture, when the sympathetic influence on heart rate is enhanced. Under such conditions, heart rate control would be modulated predominantly by the sympathetic nervous system. Not surprisingly, the resulting frequency response magnitude curve shows a substantial increase at frequencies below 0.03 Hz and a large decrease at frequencies higher than 0.1 Hz. The phase curve shows a much steeper slope, indicating an increase in the lags inherent in the system. The values of the model parameters employed here are $A_p = 0.1$, $K_p = 1$, $f_p = 0.07$ Hz, $A_s = 4.0$, $K_s = 9$, and $f_s = 0.015$ Hz.

FIGURE 5.16 Frequency responses of the circulatory control model under conditions that simulate normal heart rate control, complete β-adrenergic blockade (+propanolol), and complete parasympathetic blockade (+atropine).

In the other simulation case, the "subject" is given a dose of propranolol, which produces *β-adrenergic blockade*. Furthermore, we assume a supine posture, thus making vagal modulation the predominant mode of control. The frequency response curves corresponding to this condition are labeled +propranolol. Compared to the control case, there is little change in the frequency response above 0.05 Hz. However, loss of sympathetic modulation leads to a significant decrease in frequency response magnitude and phase at the very low frequencies. Under this "purely vagal" state, the phase difference between respiration and heart rate is relatively small over the 0–0.4 Hz range, indicating that the respiratory-induced

changes in heart rate occur rapidly. The parameter values used to represent this state are $A_p = 2.5$, $K_p = 6$, $f_p = 0.2$ Hz, $A_s = 0.1$, $K_s = 1$, and $f_s = 0.015$ Hz.

PROBLEMS

P5.1. Consider the simplified model of eye-movement control displayed in Figure P4.3. Assuming that $G/J = 14{,}400 \, \text{rad}^2 \, \text{s}^{-2}$, $B/J = 24 \, \text{rad s}^{-1}$, and $K_v = 0.01$, compute the frequency response for this model. Display the magnitude and phase components of the frequency response in the form of

   (a) linear-scale frequency response plots (e.g., Figures 5.2 and 5.3);

   (b) Bode plots;

   (c) Nichols charts;

   (d) Nyquist plots.

P5.2. **(a)** Derive the expressions for the Bode plots (magnitude and phase) of the transfer function $H(s) = e^{-sT}$, where T is a constant. Sketch the magnitude and phase curves as a function of frequency ω.

   (b) Derive the expressions for the Bode plots (magnitude and phase) of the transfer function:

$$H(s) = e^{\left[-sT + (sT)^2/2\right]}$$

where T is a constant.

Sketch the magnitude and phase curves as a function of frequency ω.

P5.3. Recall the simplified linear model of the baroreflex regulation of heart rate and arterial blood pressure introduced in Problem P4.1. ΔHR and ΔABP represent changes (from the mean) of heart rate and arterial blood pressure, respectively. The "blocks" represent the various physiological components of this reflex. G, k_b, and k_c represent constant gains (all with positive values), while τ represents time constant (in seconds) for the response of the sino-atrial (SA) node (in the heart) to changes in efferent vagal traffic (ΔVg). The expression shown in the box representing the SA node is its transfer function, with s being the Laplace variable. The external stimulus (labeled ΔZ) to this closed-loop system represents the effect of administering a vasoconstrictor drug (e.g., phenylephrine) that acts to raise blood pressure. Assuming that $\tau = 1$, $G = 1$, $k_b = 1$ and $k_c = 1$, and ΔZ is a sine wave with unit amplitude and frequency of 1 Hz, calculate the amplitude and phase of ΔABP relative to ΔZ.

P5.4. Determine the frequency responses of the ventilatory control model shown in Figure P4.5, assuming (a) $\alpha = 0$ (no rate sensitivity), (b) $\alpha = 1/2$ (lag-lead feedback), and (c) $\alpha = 2$ (lead-lag feedback).

P5.5. Derive a closed-form expression for the frequency response of the circulatory control model shown in Figure 5.13, with respiration as the input and heart rate as the output. Using the parameter values given in the SIMULINK implementation of the model (`rsa.slx`) and Section 5.4.3, deduce and plot the frequency responses of this model for the three cases shown in Figure 5.15: (a) normal supine subject, (b) following atropine administration, and (c) following propranolol infusion. Compare these plots with those presented in Figure 5.15.

P5.6. The model of Figure 5.13 can be used to investigate the dynamics of blood pressure regulation by the baroreflexes in the following way. In the SIMULINK model, `rsa.slx`, remove the respiratory input from the model and add an external source that imposes a random excitation directly on arterial blood pressure. This can be achieved experimentally in approximate fashion by imposing positive and negative pressure changes on the neck, thereby changing carotid sinus pressure. Assuming the applied pressure time-course to be the input and the resulting heart rate changes to be the output, use the method outlined in Section 5.3.2 to deduce the frequency response of the closed-loop baroreflex control system.

P5.7. Derive an expression for the closed-loop frequency response of the neuromuscular reflex model displayed in Figure 4.11, assuming the external moment M_x to be the input and angular displacement of the forearm θ to be the output. Using the parameter values given in Section 4.6.2, display the magnitude and phase plots of the frequency response.

P5.8. Using the SIMULINK implementation of the neuromuscular reflex model (`nmreflex.slx`), displayed in Figure 4.12, estimate the frequency response of the closed-loop system by using the method discussed in Section 5.3.2. Employ a random noise source as the driving input, M_x. Check your results against the magnitude and phase plots of frequency response deduced analytically in Problem P5.7.

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6

STABILITY ANALYSIS: LINEAR APPROACHES

6.1 STABILITY AND TRANSIENT RESPONSE

The concept of stability is best explained by referring to the illustration in Figure 6.1. The four plots in this figure represent the impulse responses from a linear system under different conditions. In case (a), the response is similar to that exhibited by an overdamped or critically damped system. In case (b), the system exhibits an underdamped response, that is, there is some oscillation in the response but it is eventually damped out. In both cases, the system is said to be *stable*. However, in case (d), the impulsive stimulus produces a response that is oscillatory with growing amplitude; as a consequence, the system output never returns to its original operating point prior to stimulation. This is the hallmark of an *unstable* system. With these examples in mind, we introduce the following definition of stability: *A stable dynamic system is one that will respond to a bounded input with a bounded response*. Apart from the clear-cut cases of "stable" and "unstable" systems, there are *conditionally or marginally stable* systems that exhibit undamped oscillations. Here, the response is bounded but never returns to the steady operating level prior to perturbation; however, such systems can exhibit unbounded responses if stimulated with certain bounded inputs, such as sinusoidal waves with frequencies that match the characteristic frequencies of the system.

We examined to considerable detail in Chapter 4 how first- and second-order systems, operating in either open- or closed-loop modes, respond to impulsive or
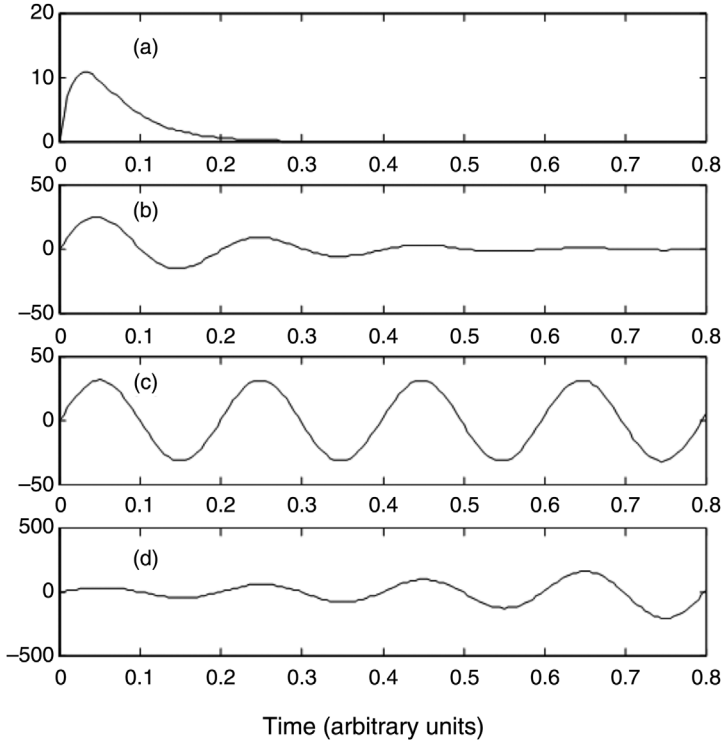
FIGURE 6.1 Responses of a system that is (a) stable and overdamped; (b) stable and underdamped; (c) marginally stable or oscillatory; and (d) unstable.

step inputs. Recall that the dynamics of the system responses, that is, the "transient responses," were determined by the roots of the denominator, or the *poles*, of the model transfer function. For the closed-loop $(k > 0)$ generalized second-order system, this was shown in Equation 4.51 to be

$$\frac{Y(s)}{X(s)} = \frac{G_{SS}\,\omega_n^2}{s^2 + 2\zeta\omega_n s + (1 + kG_{SS})\omega_n^2} \tag{6.1}$$

so that the system poles are given by the roots (α_1 and α_2) of

$$s^2 + 2\zeta\omega_n s + (1 + kG_{SS})\omega_n^2 = 0 \tag{6.2}$$

that is,

$$\alpha_{1,2} = -\zeta\omega_n \pm \omega_n\sqrt{\zeta^2 - (1 + kG_{SS})} \tag{6.3}$$

The poles can be real, imaginary, or complex, depending on the size of ζ^2 relative to the term $1 + kG_{SS}$, as shown by Equation 6.3. Thus, the impulse response

corresponding to Equation 6.1 is given by

$$h(t) = \frac{G_{SS}\omega_n^2}{\alpha_2 - \alpha_1}(e^{\alpha_2 t} - e^{\alpha_1 t}) \tag{6.4}$$

For positive values of ζ, ω_n, k, and G_{SS}, Equation 6.4 shows that the transient response will take on one of the forms represented in cases (a)–(c) in Figure 6.1, but not case (d). This arises from the fact that in cases (a)–(c), the real parts of the roots are either negative or zero, so that the terms within the curly parentheses in Equation 6.4 represent exponential decaying, exponentially damped, or simply sinusoidal dynamics. The exponentially growing behavior of the *unstable* system would only occur if the *real parts of α_1 and/or α_2 were positive*. This could be so if ζ or ω_n were negative; however, this would not be physically feasible. The only possible way in which the closed-loop model represented by Equation 6.1 could realistically be made unstable would be by making k negative: If k were to be negative and to assume a magnitude larger than $1/G_{SS}$, one of the roots in Equation 6.3 would start to become positive real, and the resulting impulse response (Equation 6.4) would increase exponentially with time.

We can summarize the conclusions from the above discussion on stability by extending the results to more generalized systems of the forms shown in Figure 6.2. In Figure 6.2a, the dynamics of the forward and feedback components are characterized by transfer functions $P(s)$ and $Q(s)$, respectively. The gain of the
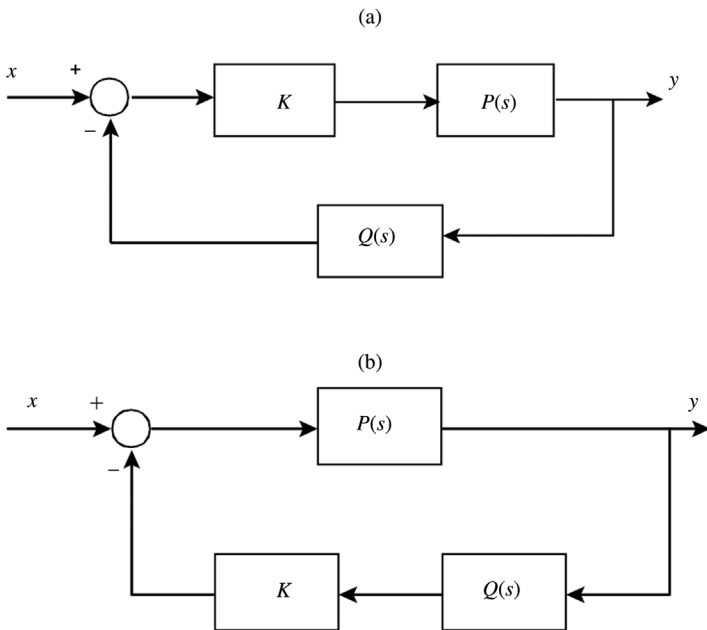
FIGURE 6.2   Closed-loop systems with variable forward (a) and feedback (b) gains.

forward path is controlled by the static factor K, which can be varied between zero and infinity. Figure 6.2b shows a similar configuration except that, here, the feedback gain can be varied by varying K between zero and infinity. For the kind of system represented in Figure 6.2a, the closed-loop transfer function is

$$H_A(s) = \frac{Y(s)}{X(s)} = \frac{KP(s)}{1 + KP(s)Q(s)} \tag{6.5}$$

For the type of system shown in Figure 6.2b, the closed-loop transfer function is

$$H_B(s) = \frac{Y(s)}{X(s)} = \frac{P(s)}{1 + KP(s)Q(s)} \tag{6.6}$$

Note that in both cases, the denominator of the closed-loop transfer function is the same. Thus, for given forms of $P(s)$ and $Q(s)$ and gain K, both types of systems will have the same transient response. As in the example of the lung mechanics model, the transient response and stability of both these systems are determined by the poles of their closed-loop transfer functions, that is, the roots of the following characteristic equation:

$$1 + KP(s)Q(s) = 0 \tag{6.7}$$

Note that the product $KP(s)Q(s)$ yields the *loop transfer function* {LG(s)} of both closed-loop systems. (Recall from Section 3.2 that the magnitude of LG(s) is the *loop gain* of the feedback control system.) Thus, Equation 6.7 can be generalized further to

$$1 + \mathrm{LG}(s) = 0 \tag{6.8}$$

Extending the result arrived at earlier, we can conclude that in each of the closed-loop systems shown in Figure 6.2, the transient response will become unstable if the *real part of any root* of its characteristic equation (Equation 6.7) is *positive*.

6.2   ROOT LOCUS PLOTS

The root locus method is a classical procedure used to determine how the poles of the closed-loop transfer function would change as a function of a system parameter (generally, some gain constant), given the location of the open-loop poles and zeros. In fact, the "root locus" is the path on the complex s-plane traced by the closed-loop poles when the system parameter in question varies over a range of values. To illustrate how this method is applied, we turn once again to our simple lung mechanics model. Referring back to Figure 4.2b, we find that the lung mechanics model is merely a special case of the closed-loop

form shown in Figure 6.2, where

$$P(s) = \frac{1}{LCs^2 + RCs + 1} \qquad (6.9)$$

$$Q(s) = 1 \qquad (6.10)$$

and

$$K = k \qquad (6.11)$$

For given values of the lung mechanics parameters (L, C, and R), the root locus will show us how the dynamic behavior of the model changes as the feedback gain takes on a range of values. Applying Equation 6.7, the task then is to solve for the roots of the following characteristic equation as k varies:

$$1 + \frac{k}{LCs^2 + RCs + 1} = 0 \qquad (6.12)$$

This is equivalent to solving

$$LCs^2 + RCs + 1 + k = 0 \qquad (6.13)$$

The general solution for the roots of Equation 6.13 is

$$\alpha_{1,2} = \frac{-R \pm \sqrt{R^2 - 4L(1 + k)/C}}{2L} \qquad (6.14)$$

To solve Equation 6.14, we will assume, as in Section 4.3, that $L = 0.01$ cm $H_2O\,s^2\,L^{-1}$, $C = 0.1\,L$ cm H_2O^{-1}, and $R = 1$ cm $H_2O\,s\,L^{-1}$. It is generally useful to determine the locations of the closed-loop poles when k assumes its two most extreme values. First, note that when $k = 0$, solving Equation 6.13 becomes equivalent to determining the locations of the poles of the open-loop system, $P(s)$. Substituting the above values of L, C, R, and k into Equation 6.14, we obtain the following solutions: $\alpha_1 = -88.73$ and $\alpha_2 = -11.27$. For these parameter values, both poles are real and negative. At the other extreme, when k becomes infinitely large, Equation 6.14 yields the solutions: $\alpha_1 = -50 + j4$ and $\alpha_2 = -50 - j4$. Thus, for very large k, the poles become complex. Finally, the value of k at which the two real poles become complex can be found by solving for the value of k where the expression inside the square-root operation in Equation 6.14 goes to zero, that is,

$$k = \frac{R^2C}{4L} - 1 \qquad (6.15)$$

Substituting in values for L, R, and C, we obtain $k = 1.5$.

The complete root locus plot of the system in question can be obtained easily with the use of the MATLAB Control System Toolbox function `rlocus`. This

function assumes that the product $P(s)Q(s)$ yields a transfer function H that takes the form of the ratio of two polynomial functions of s:

$$P(s)Q(s) = H(s) = \frac{N(s)}{D(s)} \tag{6.16}$$

Then, Equation 6.7 can be recast into the following form:

$$D(s) + KN(s) = 0 \tag{6.17}$$

`rlocus` finds the solution to Equation 6.17 for all values of K between 0 and infinity. For our specific example, the following MATLAB command lines can be used to plot the corresponding root locus:

```
>> Ns = [1];
>> Ds = [L*C R*C 1];
>> Hs = tf(Ns,Ds);
>> rlocus(Hs);
```

The resulting root locus plot is displayed in Figure 6.3. Note that the locations of the closed-loop poles when k equals zero, 1.5, and infinity are exactly as we had deduced earlier. Also, since the poles always lie on the left-hand side of the s-plane (i.e., real parts of poles are always negative), the closed-loop system is stable for all positive values of feedback gain, k. Note that the root locus gives us a good global picture of the transient response characteristics of the system, but tells us little about its frequency response.

As a further example, we consider the linear lung mechanics model when integral feedback control is employed instead of proportional feedback, that is, in this case, the fluctuations in alveolar pressure are integrated before being fed back to the comparator. This system is displayed in Figure 6.4. Referring to Figure 6.2b, the forward transfer function $P(s)$ remains the same, while the feedback transfer function $Q(s)$ is now given by
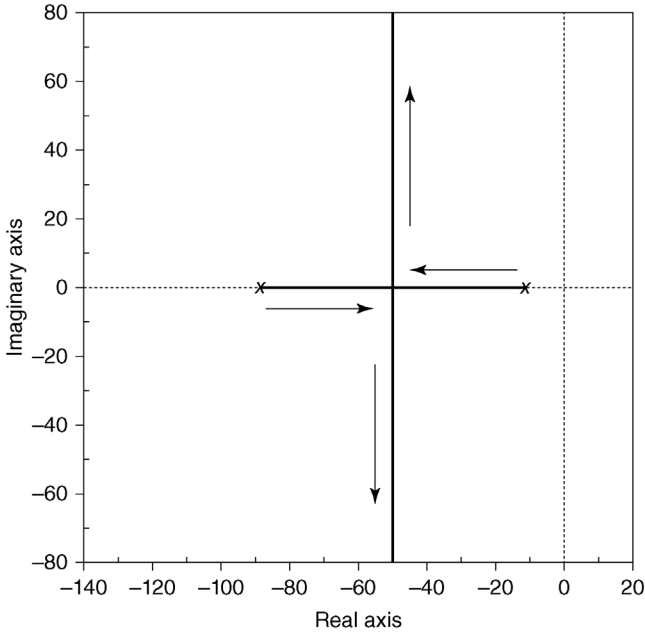
$$Q(s) = \frac{1}{s} \tag{6.18}$$

The characteristic equation (Equation 6.17) now assumes the specific form:

$$LCs^3 + RCs^2 + s + k = 0 \tag{6.19}$$

The above equation can be solved easily using MATLAB by simply inserting an extra term into the row vector that represents $D(s)$:

```
>> Ns = [1];
>> Ds = [L*C R*C 1 0];
>> Hs = tf(Ns,Ds);
>> rlocus(Hs);
```

FIGURE 6.3   Root locus plot for the lung mechanics model with proportional feedback. Locations marked x indicate positions of poles when $k=0$. Arrows indicate direction in which the poles move as k is increased from zero to infinity. Dotted horizontal and vertical lines represent the real and imaginary axes, respectively.

The corresponding root locus plot, displayed in Figure 6.5, shows a form that differs significantly from the plot in Figure 6.3. Because the characteristic equation is now third order, there are three poles instead of two. When $k=0$, all three poles are located on the real axis, one of which is situated at the origin ($s=0$). As k increases, the most negative pole becomes progressively more negative while remaining real. However, when k increases beyond 2.64, the other two poles become complex, that is, the transient response becomes a damped oscillation that becomes less and less damped, and the frequency of which
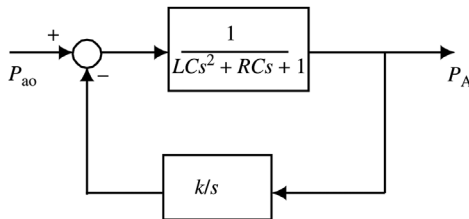
FIGURE 6.4   Linear lung mechanics model with integral feedback.
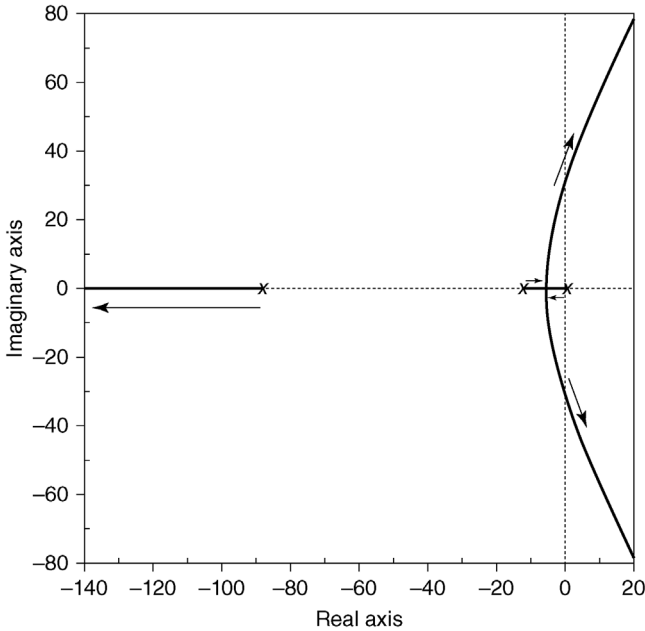
FIGURE 6.5   Root locus plot for the lung mechanics model with integral feedback. Poles at $k = 0$ are marked x. Arrows indicate direction in which the three poles move as k is increased toward infinity.

progressively increases, as k continues to increase. Finally, when k exceeds 100, the system becomes unstable as these two poles move into the right-hand side of the s-plane, that is, the impulse response assumes the form of a growing oscillation. These results demonstrate that the integral feedback system depicted here exhibits dynamics that are less stable than the proportional feedback case, although it gives a faster response.

6.3   ROUTH–HURWITZ STABILITY CRITERION

The root locus method requires the evaluation of all roots of the characteristic equation in order to determine whether a given system is stable or unstable. The Routh–Hurwitz technique is a classical stability test that enables such a determination without the need to actually evaluate the roots. With the computational tools that are available nowadays, this test has become somewhat obsolete. Nonetheless, we will describe it here for the sake of completeness.

We assume the following general form for the characteristic equation of the closed-loop system in question:

$$a_n s^n + a_{n-1} s^{n-1} + \cdots + a_1 s + a_0 = 0 \qquad (6.20)$$

The Routh–Hurwitz technique requires the computation of values based on the coefficients of the characteristic equation and the arrangement of these values into an array of the following construction (with each row corresponding to a power of s, as indicated in the margin to the left of the array):

$$
\begin{array}{c|cccc}
s^n & a_n & a_{n-2} & a_{n-4} & \vdots \\
s^{n-1} & a_{n-1} & a_{n-3} & a_{n-5} & \vdots \\
s^{n-2} & b_1 & b_2 & b_3 & \vdots \\
s^{n-3} & c_1 & c_2 & c_3 & \vdots \\
\cdots & \cdots & \cdots & \cdots & \vdots \\
s^1 & p_1 & & & \\
s^0 & q_1 & & &
\end{array}
\tag{6.21}
$$

where

$$
b_1 = \frac{a_{n-1}a_{n-2} - a_n a_{n-3}}{a_{n-1}}, \quad b_2 = \frac{a_{n-1}a_{n-4} - a_n a_{n-5}}{a_{n-1}}, \quad c_1 = \frac{b_1 a_{n-3} - b_2 a_{n-1}}{b_1}
\tag{6.22}
$$

and so on. The Routh–Hurwitz criterion states that the number of closed-loop poles located on the right-hand side of the s-plane is given by the number of changes in sign in the first column of the constructed array in Equation 6.21. Thus, for a *stable* system, there should be *no changes in sign in the first column of the array*.

To illustrate the use of the Routh–Hurwitz test, we will apply it to the two examples discussed in Section 6.2. Consider first the linear lung mechanics model with proportional feedback. The characteristic equation here is given by Equation 6.13. Using Equation 6.21, the Routh array is

$$
\begin{array}{c|cc}
s^2 & LC & 1+k \\
s & RC & 0 \\
s^0 & 1+k & 0
\end{array}
\tag{6.23}
$$

For positive values of L, C, and R and for $k \geq 0$, there are no changes in sign in the terms of the *first column* of the array. Therefore, by the Routh–Hurwitz criterion, the system defined by Equation 6.13 will always be stable.

In the lung mechanics model with integral feedback, the characteristic equation is given by Equation 6.19. In this case, the first two columns of the Routh array are

$$
\begin{array}{c|cc}
s^3 & LC & 1 \\
s^2 & RC & k \\
s & 1 - \dfrac{Lk}{R} & 0 \\
s^0 & k & 0
\end{array}
\tag{6.24}
$$

For positive values of L, C, R, and k, note that all terms in the first column of Equation 6.24 will be positive, except for the third term that can become negative if

$$k > \frac{R}{L} \tag{6.25}$$

For $R = 1$ cm H_2O s L^{-1} and $L = 0.01$ cm H_2O s^2 L^{-1}, Equation 6.25 predicts that there will be two changes of sign in Equation 6.24 when k exceeds the value of 100: There is one change of sign from positive to negative (sign of RC to sign of $1 - Lk/R$), and a second change of sign from negative to positive (sign of $1 - Lk/R$ to sign of k). This implies that there are two closed-loop poles located on the right-hand side of the s-plane – exactly the result obtained in Section 6.2 when we employed the root locus method. Therefore, this closed-loop system can become unstable when $k > R/L$.

6.4 NYQUIST CRITERION FOR STABILITY

One primary disadvantage of the Routh–Hurwitz method is that the test becomes difficult to apply when the characteristic equation cannot be simply expressed as a polynomial function of s. Pure time delays are abundantly found in physiological systems. Although these can be approximated by rational polynomial expressions (see Problem P4.3), the resulting characteristic equation can become extremely unwieldy. For these kinds of problems, it is generally more convenient to employ the Nyquist stability test.

The formal mathematical development of the Nyquist stability criterion will not be presented here, as it involves a fair bit of complex variable theory; the interested reader can find this in most engineering texts on control systems. Instead, we will employ a more intuitive approach by illustrating the basic notions underlying this criterion. Consider the very simple negative feedback system shown in Figure 6.6. The input u represents a disturbance to the system that is nonzero for only a brief period of time. The system output x is fed back through a static gain $-K$ and added to the input before being fed forward through the time-delay (T_D) block. Note that the
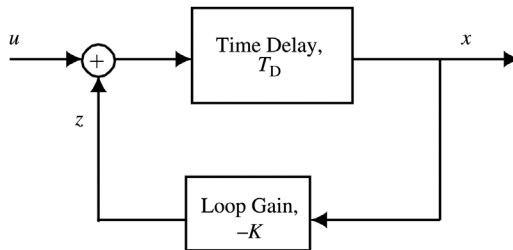
FIGURE 6.6 Simple negative feedback system with delay. Note that, in this example, the negativity in feedback is embedded in the "loop gain" block.

negativity in feedback control is implemented through the assignment of a negative value to the feedback gain and the addition (not subtraction) of the resulting feedback signal (z) at the level of the summing junction.

Figure 6.7 shows three of the many possibilities with which the system responds following the imposition of a transient disturbance u. In all these cases, we have assumed that u takes the form of a half-sine wave. Consider case (a) in which we assume $K = 0.5$ and T_D is half the duration of u. Here, the initial passage of u through the forward and feedback blocks produces an inverted and attenuated (by 50%) half-sine wave at z, labeled 1 in Figure 6.7a (upper panel). Propagation of this signal around the loop a second time would produce a response at z of the form labeled as 2. Similarly, the third and fourth traversals around the entire loop would produce "3"
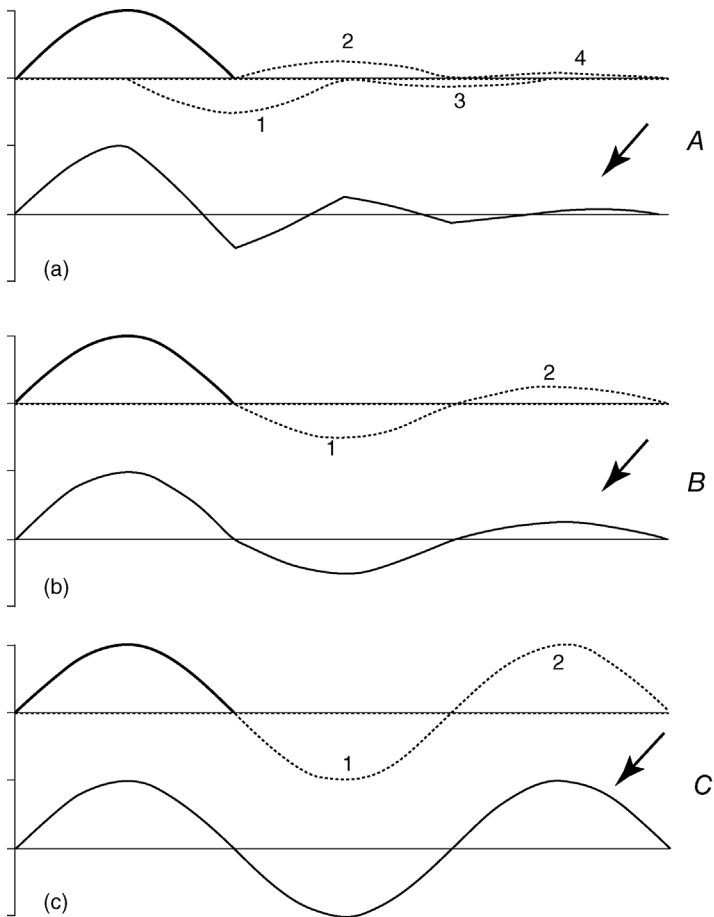
FIGURE 6.7 Response of the negative feedback system with delay to an initial disturbance when (a) loop gain magnitude <1 and phase lag <180°, (b) loop gain magnitude <1 and phase lag = 180°, and (c) loop gain magnitude = 1 and phase lag = 180°.

and "4", respectively. This process would continue until the "echoes" of the initial u are finally damped out totally. Each time there is a complete traversal around the loop, there is a change in sign, reflecting the negative nature of the feedback. Superimposition of all these individual effects on the initial disturbance leads to the complete response at x shown in the lower panel of Figure 6.7a. Thus, the system response to u is a rapidly damped oscillation, that is, the system in this case is stable. In Figure 6.7b, K is assigned the value of 0.5 again, but T_D is now increased to a value equal to the duration of u. This produces the responses in z labeled as "1" after the first traversal around the loop and "2" after the second traversal (Figure 6.7b, upper panel). The net result again is a damped oscillation (Figure 6.7b, lower panel). However, in this case, the oscillation appears to be more slowly damped and the frequency of the oscillation is also lower. Note, in this case, that the corrective actions ("1" in response to u and "2" in response to "1") have been delayed so much that they occur out of phase (i.e., at 180°) with the preceding fluctuations. Finally, in case (c), the time delay is kept the same as that for case (b), but the loop gain magnitude is increased to 1. As in case (b), due to the increased delay, the feedback signal tends to reinforce the effect of the initial disturbance rather than to cancel it out. However, in this case, since there is no attenuation in the feedback loop, the net result is a sustained oscillation or period equal to twice the length of T_D. It is easy to see that if the loop gain were to be increased further to a value exceeding 1, the system response would be an oscillation with growing amplitude.

From the examples shown in Figure 6.7, it is clear that a closed-loop system can become unstable if the total phase lag imposed by all system components around the loop equals 180° and the loop gain magnitude is at least unity. This is the basic notion underlying the Nyquist criterion. Thus, in order to determine whether a specific closed-loop system is stable or unstable, we would first deduce the *loop transfer function* (i.e., the product of all component transfer functions around the closed loop). The Nyquist plot of the loop transfer function is generated. Note that a loop gain of unit magnitude and phase lag of 180° is represented by the point $(-1 + j0)$ on the complex plane. *The system is stable if the $(-1 + j0)$ point lies to the left of the Nyquist plot as the locus is traversed in the direction of increasing frequency*. Another way of stating this criterion is that in the stable system, the Nyquist plot *will not encircle* the $(-1 + j0)$ point. In order to make this determination, it is generally necessary to evaluate the loop transfer function from zero frequency to infinity, or at least, over a wide band of frequencies. This criterion, as stated above, is valid as long as the loop transfer function does not contain any poles with positive real parts. If this condition does not apply, one has to employ a different version of the Nyquist criterion; more details of the method under such circumstances can be found elsewhere (Dorf and Bishop, 2011).

The reader may recall that the Nyquist representation was previously discussed in Section 5.2.3. However, one should be cautioned that, in Chapter 5, the examples shown were those in which we characterized the frequency responses corresponding to the open- and closed-loop transfer functions of the systems in question. For a determination of stability, we need to evaluate the *loop transfer function*, which

will yield a Nyquist plot quite different from the Nyquist plots that correspond to the open- or closed-loop transfer functions of the same system.

To illustrate the application of the Nyquist stability criterion, we turn once again to the two examples that have been discussed earlier. For the linear lung mechanics model with proportional feedback, the loop transfer function, $H_{L1}(s)$, is given by

$$H_{L1}(s) = \frac{k}{LCs^2 + RCs + 1} \tag{6.26}$$

The frequency response corresponding to Equation 6.26 is obtained by substituting $j\omega$ for s:

$$H_{L1}(\omega) = \frac{k}{(1 - LC\omega^2) + jRC\omega} \tag{6.27}$$

The Nyquist plots corresponding to Equation 6.27 are shown in Figure 6.8 for three values of feedback gain ($k = 1$, 10, and 100). Note that at $\omega = 0$, $H_{L1}(\omega) = k$. However, the zero frequency values for $H_{L1}(\omega)$ when $k = 10$ and $k = 100$ lie outside the range displayed. In Equation 6.27, also notice that when $\omega \rightarrow \infty$, $H_{L1}(\omega) \rightarrow 0$. Thus, for each plot, the Nyquist locus begins at the point $(k + j0)$ at zero frequency and ends at the origin at infinite frequency. The direction of traversal of each locus with increasing values of frequency is indicated by the arrows (Figure 6.8). Except for the hypothetical case when k becomes infinite, it can be seen that none of the Nyquist loci touch or encircle the $(-1 + j0)$ point (represented as the filled circle in Figure 6.8). Thus, this system is stable for all finite values of feedback gain.

The loop transfer function for the linear lung mechanics model with integral feedback, $H_{L2}(s)$ is given by

$$H_{L2}(s) = \frac{k}{LCs^3 + RCs^2 + s} \tag{6.28}$$

The frequency response corresponding to the above transfer function is

$$H_{L2}(\omega) = \frac{k}{-RC\omega^2 + j\omega(1 - LC\omega^2)} \tag{6.29}$$

In this case, note that when $\omega \rightarrow 0$, $H_{L2}(\omega) \rightarrow j\infty$. When ω is very large, the term in ω^3 will become much more important than the other terms in the denominator in Equation 6.29. Thus, when $\omega \rightarrow \infty$, $H_{L2}(\omega) \rightarrow 0$, with the Nyquist locus at high values of ω tending to approach the origin along the positive imaginary axis. Figure 6.9 shows the behavior of Nyquist loci at four different values of feedback gain ($k = 1$, 10, 100, and 1000). All loci start off from $-j\infty$ and curve in toward the origin as frequency increases toward infinity. For values of k below 100, the Nyquist loci remain to the right of the $(-1 + j0)$ point, so the system remains stable. When $k = 100$, the Nyquist locus passes through the $(-1 + j0)$ point, indicating that the system becomes conditionally stable in this condition. Then,
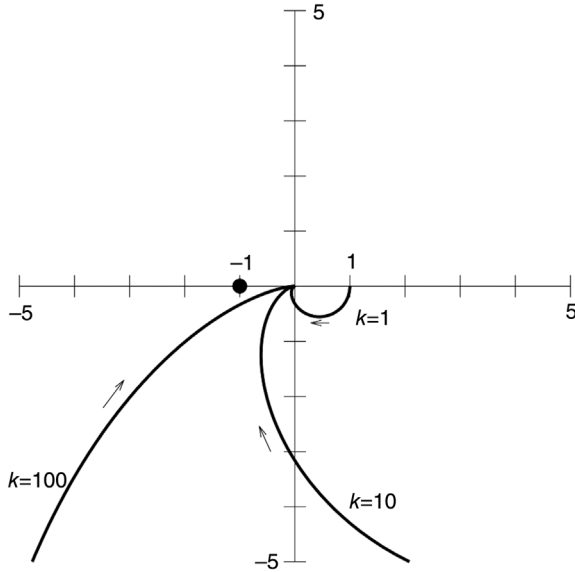
FIGURE 6.8 Nyquist plots for the linear lung mechanics model with proportional feedback. Feedback gains shown are $k = 1$, 10, and 100. For the latter two cases, only portions of the Nyquist plots lie outside of the scale shown. Arrows indicate direction of Nyquist trajectories as frequency increases from 0 to infinity. Filled circle represents location of the $-1 + j0$ point.
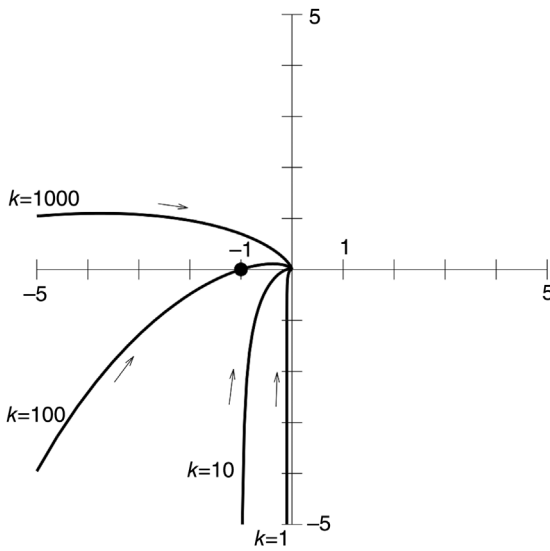
FIGURE 6.9 Nyquist plots for the linear lung mechanics model with integral feedback. In all cases ($k = 1$, 10, 100, 1000), portions of the Nyquist plot lie outside of scale shown. Arrows indicate direction of Nyquist trajectories as frequency increases from 0 to infinity. Filled circle represents location of the $-1 + j0$ point.

when $k > 100$, the system becomes unstable with the Nyquist plot encircling the $(-1 + j0)$ point.

6.5 RELATIVE STABILITY

While it is useful to be able to determine under what conditions a closed-loop system would become marginally stable or unstable, it is equally useful to have a means of assessing how "far" the point of instability is from the current state of system stability. Consider two stable systems. Suppose the effect of an impulsive distur-bance elicits a rapidly damped, nonoscillatory response from the first system, while the same disturbance produces an underdamped oscillatory response from the second system. Although both systems are "stable," it would be reasonable to conclude that the first may be considered "more stable" than the second. This is the basis underlying the notion of *relative stability*.

The relative stability of a given system can be quantified in terms of either the *gain margin* or the *phase margin*. Both provide measures of the $(-1 + j0)$ point from specific points on the locus of the loop transfer function. The *gain margin* refers to the *factor by which the loop gain corresponding to a phase of* $-180°$ *has to be increased before it attains the value of unity*. The frequency at which the phase of the loop transfer function becomes $-180°$ is known as the *phase crossover frequency*. To illustrate this point, consider a special case of the linear lung mechanics model with integral feedback that has the following specific loop transfer function:

$$H_{L2}(s) = \frac{1}{s^3 + 3s^2 + 2s} \tag{6.30}$$

The Bode magnitude and phase plots corresponding to Equation 6.30 are displayed in Figure 6.10. Note that at the phase crossover frequency ω_{pc}, when the phase of $H_{L2}(\omega_{pc})$ becomes equal to $-180°$, the gain of $H_{L2}(\omega_{pc})$ remains less than unity, implying that this system is stable. By definition, the gain margin (GM) in this case is given by

$$\left| H_{L2}(\omega_{pc}) \right| \times GM = 1 \tag{6.31}$$

Therefore, the gain margin, expressed in decibels, is

$$GM_{dB} = 20 \log_{10} \left(\frac{1}{\left| H_{L2}(\omega_{pc}) \right|} \right) = -20 \log_{10} \left| H_{L2}(\omega_{pc}) \right| \tag{6.32}$$

As shown in the top panel of Figure 6.10, GM_{dB} is given by the vertical distance between the 0 dB axis and the point on the gain plot at which the phase crossover frequency is attained.
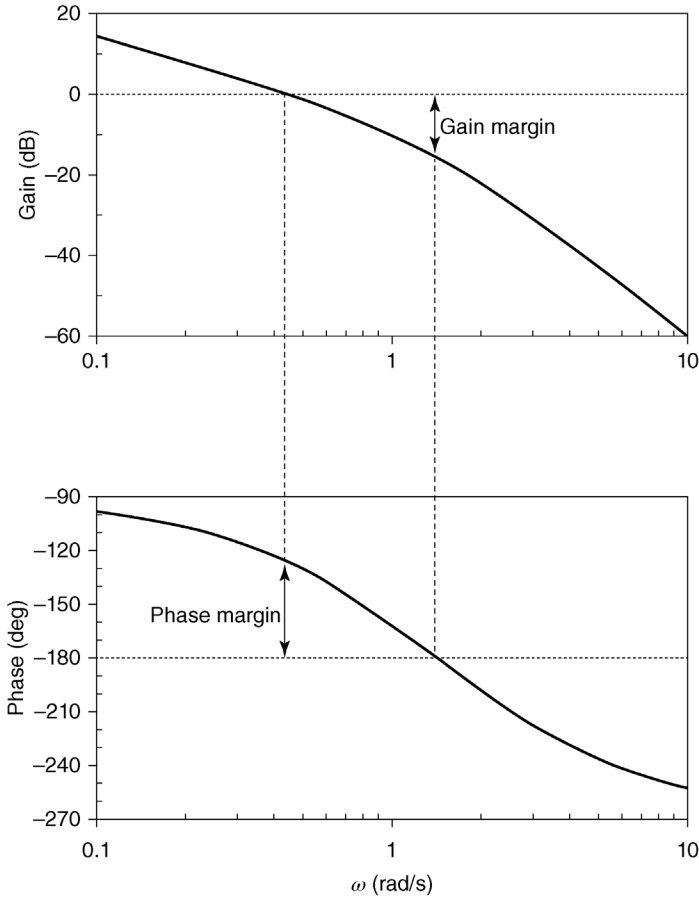
FIGURE 6.10 Derivation of gain and phase margins from the Bode plots of the loop transfer function.

The other measure of relative stability, the *phase margin*, is defined as *the shift in phase at unit loop gain necessary to produce a phase lag of* 180°. The frequency at which the loop gain becomes equal to unity is known as the *gain crossover frequency* (ω_{gc}). Referring to the lower panel of Figure 6.10, note that the phase margin is given by the vertical distance between the −180° line and the point on the Bode phase plot at which the gain crossover frequency is attained.

The gain and phase margins can also be readily deduced from the Nichols chart (Figure 6.11) and the Nyquist plot (Figure 6.12). In the case of the Nyquist plot, the intersection between the Nyquist locus and the negative horizontal axis yields $|H_{L2}(\omega_{pc})|$, which, by Equation 6.31, gives the reciprocal of GM. Note, however, that here GM is expressed as a ratio and not in terms of decibels. It can be appreciated from Figures 6.10 through 6.12 that larger positive values for the gain
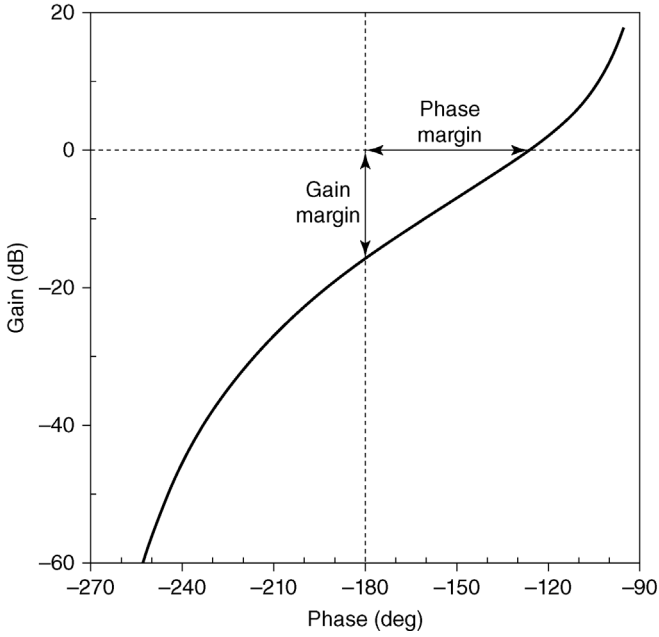
FIGURE 6.11   Derivation of gain and phase margins from the Nichols chart of the loop transfer function.
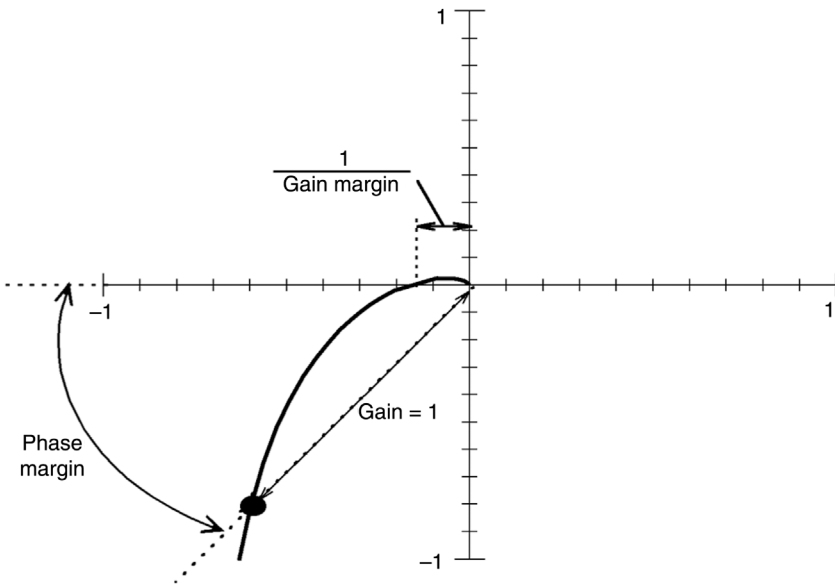
FIGURE 6.12   Derivation of gain and phase margins from the Nyquist plot of the loop transfer function.

or phase margins imply greater relative stability. On the other hand, negative values for either gain or phase margins would imply that the system in question is already unstable.

Numerical evaluation of the gain and phase margins can be performed easily using the function margin in the MATLAB Control System Toolbox. For instance, to evaluate the gain and phase margins for the loop transfer function defined by Equation 6.30, the following MATLAB command lines (contained in script file gpmargin.m) may be applied:

```
>>%First construct loop transfer function and vector of
frequencies
>> num = [1];
>> den = [1 3 2 0];
>> Hs = tf(num, den);
>> f = 0.01:0.01:10;
>> w = 2*pi*f;

>>%Compute magnitude (mag) and phase (pha)
of loop transfer function
>> [mag,pha] = bode(Hs,w)

>>%Compute gain margin (GM), phase margin (PM), gain
crossover
>>%frequency (wcG) and phase crossover frequency (wcP)
>> [GM,PM,wcG,wcP] = margin(mag,pha,w);

>>%Plot Bode diagram showing gain and phase margins
>> margin(mag,pha,w);
```

6.6   STABILITY ANALYSIS OF THE PUPILLARY LIGHT REFLEX

The pupillary light reflex has been studied extensively using control system analysis, beginning with bioengineering pioneers Lawrence Stark (1959) and Manfred Clynes (1960). The purpose of this reflex is to regulate the total light flux reaching the retina, although the same pupil control system is also used to alter the effective lens aperture so as to reduce optical aberrations and increase depth of focus. The reflex follows the basic scheme shown in Figure 6.13a. An increase in the intensity (*I*) of ambient light elevates the total light flux (*L*) received by the retina, which converts the light into neural signals. The afferent neural information is sent via the optic nerve to the lateral geniculate body and then to the pretectal nucleus. Subsequently, the Edinger–Westphal nucleus sends efferent neural signals back toward the periphery to the iris sphincter and dilator muscles, which, respectively, contract and relax to reduce the pupil area (*A*).

Not surprisingly, quantitative investigations into this feedback control scheme have revealed significant nonlinearities in each of the system components. However, Stark came up with a linear characterization that provides a reasonably good
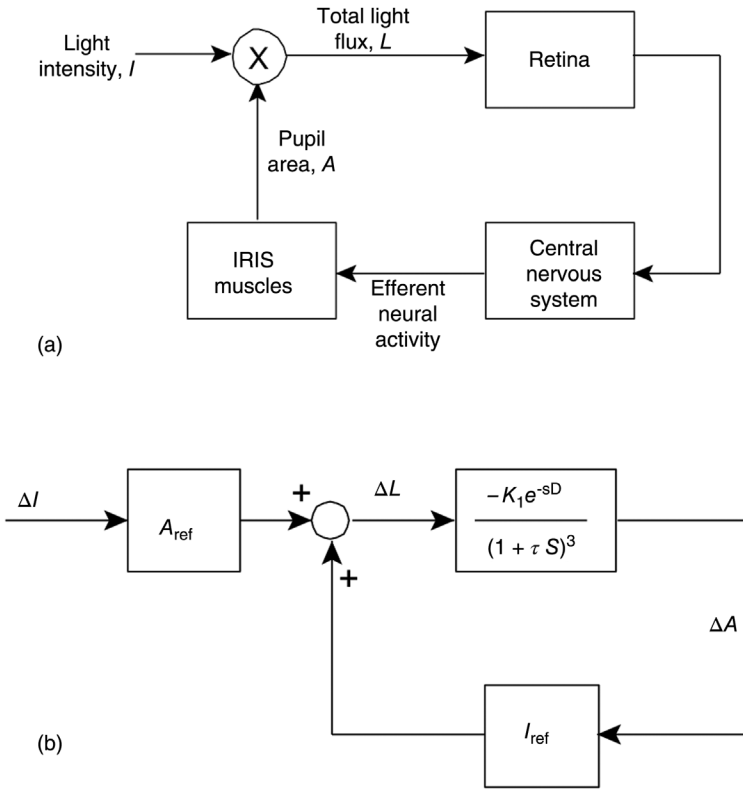
FIGURE 6.13 (a) Functional scheme for the pupillary control system. Note that the total light flux L is given by the product of light intensity I and pupil area A; (b) linearized (small-signal) model of the pupillary light reflex.

approximation of the underlying dynamics when the changes involved are relatively small. This linearized model is schematized in Figure 6.13b. Using an ingenious experimental design (see Figure 7.13), he was able to functionally "open the loop" of this reflex and measure the dynamics of this system. (This technique is discussed further in Section 7.4.5.) He found that the dynamics could be modeled by a third-order transfer function with time constant τ, in series with a pure time delay D. In Figure 6.13b, ΔI represents a small change in light intensity from the reference intensity level, I_{ref}, while ΔA represents the corresponding change in pupil area from the reference value A_{ref} and ΔL is the change in total light flux reaching the retina. Based on the model, the closed-loop transfer function of the pupillary reflex can be deduced as

$$\frac{\Delta A}{\Delta I} = \frac{-A_{\text{ref}} K_1 e^{-sD}/(1+\tau s)^3}{1 + \left[I_{\text{ref}} K_1 e^{-sD}/(1+\tau s)^3 \right]} \tag{6.33}$$

By inspection of Figure 6.13b and Equation 6.33, one can readily infer that the *loop transfer function* of this model is given by

$$H_L(s) = \frac{Ke^{-sD}}{(1 + \tau s)^3} \tag{6.34}$$

where $K = I_{\text{ref}} K_1$. Therefore, the characteristic equation for the closed-loop model is

$$1 + \frac{Ke^{-sD}}{(1 + \tau s)^3} = 0 \tag{6.35}$$

From his measurements on normal humans, Stark found the following values for the model parameters: $K = 0.16$, $D = 0.18$ s, and $\tau = 0.1$ s. In the next two sections that follow, we will assume the above values of D and τ in our analyses and determine the critical value of K above which the model becomes unstable.

6.6.1   Routh–Hurwitz Analysis

In Section 6.3, when the Routh–Hurwitz stability criterion was first discussed, application of the test was simple since the examples considered had characteristic equations that could be expressed as polynomials in s. In Equation 6.35 however, the presence of the time delay complicates matters a little. Llaurado and Sun (1964) suggested that this problem can be circumvented by expanding e^{-sD} as a power series in s:

$$e^{-sD} \approx 1 - Ds + \frac{D^2 s^2}{2} - \frac{D^3 s^3}{6} \tag{6.36}$$

Substituting the above approximation for e^{-sD} into Equation 6.35 and collecting terms for each power of s, we obtain the following third-order polynomial expression in s:

$$\left(\tau^3 - \frac{KD^3}{6}\right)s^3 + \left(3\tau^2 + \frac{KD^2}{2}\right)s^2 + (3\tau - KD)s + (1 + K) = 0 \tag{6.37a}$$

Inserting the numerical values for D and τ into Equation 6.37a, we obtain

$$(0.001 - 0.000972K)s^3 + (0.03 + 0.0162K)s^2 + (0.3 - 0.18K)s + (1 + K) = 0 \tag{6.37b}$$

The Routh array corresponding to Equation 6.37b is

$$
\begin{array}{c|ccc}
s^3 & 0.001 - 0.000972K & 0.3 - 0.18K & 0 \\
s^2 & 0.03 + 0.162K & 1 + K & 0 \\
s & \dfrac{890 - 431.72K - 19.44K^2}{300 + 162K} & 0 & 0 \\
s^0 & 1 + K & 0 & 0
\end{array} \tag{6.38}
$$

The requirement for stability is that all terms in the first column of the above array should have the same sign. Since the term corresponding to s^0 is $1 + K$, and K must be positive, then for the system to be stable, all terms in the first column of the Routh array must also be positive. Thus, the following inequalities have to be satisfied simultaneously:

$$0.001 - 0.000972K > 0 \tag{6.39a}$$

$$0.03 + 0.0162K > 0 \tag{6.39b}$$

$$\frac{890 - 431.72K - 19.44K^2}{300 + 162K} > 0 \tag{6.39c}$$

$$1 + K > 0 \tag{6.39d}$$

From the first inequality (Equation 6.39a), we find that $K < 1.029$. The second and fourth inequalities are satisfied for all values of K that are greater than zero. In the third inequality, the quadratic expression in the numerator of the left-hand side has to be factorized first. From this, it can be deduced that the inequality is satisfied if $-24.196 < K < 1.996$. Thus, combining this result with that from the first inequality, we conclude that for the closed-loop system to be stable, K must be less than 1.029. Since the average value of K measured by Stark was 0.16, we can conclude from Routh–Hurwitz analysis that the normal pupillary reflex is a highly stable negative feedback system.

6.6.2 Nyquist Analysis

The frequency response $H_L(\omega)$ corresponding to the loop transfer function can be obtained by substituting $j\omega$ for s in Equation 6.34:

$$H_L(\omega) = \frac{Ke^{-j\omega D}}{(1 + j\omega\tau)^3} = \frac{Ke^{-j\omega D}}{(1 - 3\omega^2\tau^2) + j\omega\tau(3 - \omega^2\tau^2)} \tag{6.40}$$

The problem of evaluating the transfer function with time delay in Equation 6.40a can be approached in a number of ways. One way is to apply the power series expansion approach employed in the previous section. Another possibility is to employ a Padé approximation to the delay, as illustrated in Problem P4.2. The first of these methods converts the time-delay transfer function into a polynomial function of s, while the second approximates it with a transfer function that consists of the ratio of two polynomials in s. However, a third approach is to express $H_L(\omega)$ in polar form and recognize that the delay will only affect the phase component of the transfer function. MATLAB offers a convenient means of performing this computation over a given range of

frequencies. An illustration of this procedure is given in the MATLAB command lines below:

```
>>% Construct undelayed transfer function & evaluate
frequency response
>> num = [K];  den = [tau^3  3*tau^2  3*tau  1];
>> Hs = tf(num, den);
>> [R,I] = nyquist(Hs);  I = squeeze(I);  R = squeeze(R);
>>
>>% Add delay to results
>> Rdel = real((R+j*I).*exp(-j*w*D));
>> Idel = imag((R+j*I).*exp(-j*w*D));
>>
>>% Plot final Nyquist diagram
>> plot(Rdel,Idel);
```

The complete script file (named `pupil.m`) for evaluating Equation 6.40 and plotting the Nyquist diagrams is included in the library of MATLAB/SIMULINK files accompanying this book.

Figure 6.14a displays the Nyquist plot for the normal pupil control system with $K = 0.16$, generated using the above MATLAB code and assuming the parameter values, $D = 0.18\,\text{s}$ and $\tau = 0.1\,\text{s}$. The Nyquist plots corresponding to increased values of K: 1.6 (dotted curve) and 2 (solid curve) are shown in Figure 6.14b. Note that the scale of the axes in Figure 6.14b has been increased to cover a substantially larger range of values. It is clear that the pupillary reflex model is unstable when $K = 2$ but stable when $K = 1.6$. Further computations show that the critical value of K for the development of self-sustained oscillations is 1.85. This is larger than the 1.029 value deduced in Section 6.6.1. However, the critical value arrived at here is the more accurate prediction, since in the previous section, an approximation had to be assumed in order to represent the pure-delay transfer function as a power series.

Examination of all the Nyquist plots in Figure 6.14 shows that the intersection of each locus with the negative real axis (i.e., phase $= -180°$) occurs at the same critical frequency ω_c, regardless of the value of K. This can be confirmed by analysis of the phase of $H_L(\omega)$. At the critical frequency, since the phase of $H_L(\omega)$ must equal $-\pi\,\text{rad}$ (or $-180°$), we have

$$-\pi = -\omega_c D - \tan^{-1}\left(\frac{\omega_c \tau\left(1 - \omega_c^2 \tau^2\right)}{\left(1 - 3\omega_c^2 \tau^2\right)}\right) \tag{6.41}$$

Note that ω_c can be deduced by solving Equation 6.41. However, in Equation 6.41, since none of the terms depend on K, ω_c must also be independent of K. ω_c can be found from solution of Equation 6.41 or from inspection of the Nyquist plots to be equal to 7.1 rad s^{-1}, which corresponds to an absolute frequency of 1.1 Hz. This predicted frequency is close to the frequency of continuous oscillations of the
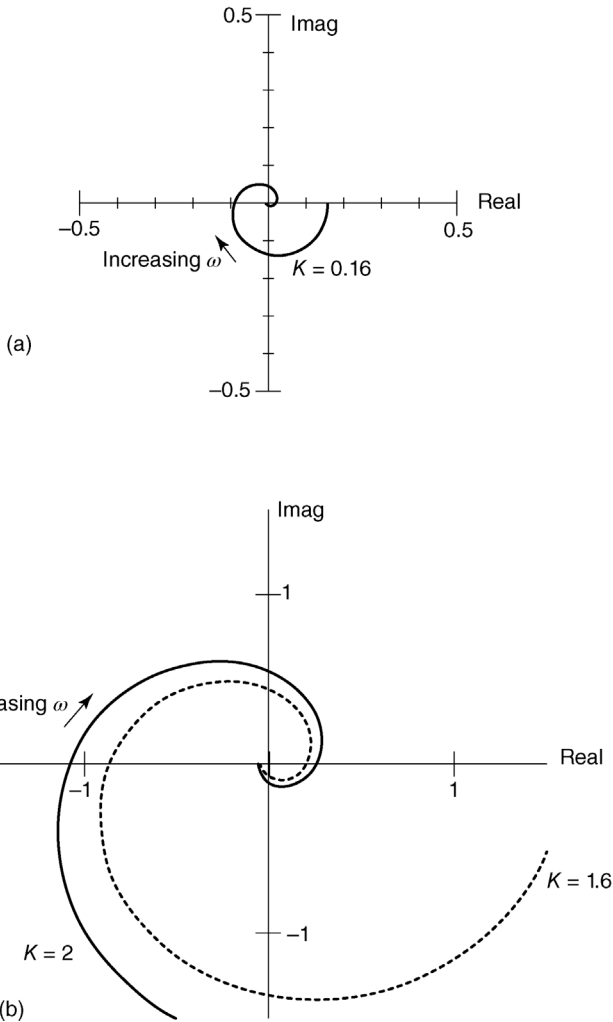
FIGURE 6.14   Nyquist plots of the linearized pupillary reflex model at (a) normal loop gain factor ($K = 0.16$); (b) elevated loop gain factors ($K = 1.6$ and $K = 2$).

pupil, known as "hippos," which have been observed under certain pathological conditions. Stark was able to artificially induce hippus in normal subjects using a clever experimental design. He focused a thin beam of light at the edge of the pupil. This stimulation of the reflex led to constriction of the pupil, which produced a large decrease in retinal illumination (since much of it was now blocked by the iris). This, in turn, acted through the reflex to dilate the pupil, restoring the effect of the applied retinal illumination. This experimental design was tantamount to elevating the loop gain of the closed-loop system tremendously, thereby setting the

stage for a self-sustained oscillation to occur. The frequency of this oscillation was found to be close to that predicted by the model. The good agreement between model prediction and experimental observation supports the approximate validity of the linear assumption in this case.

6.7 MODEL OF CHEYNE–STOKES BREATHING

The term *periodic breathing* refers to the cyclic modulation of respiration that occurs over the timescale of several breaths. The resulting ventilatory pattern may or may not include periods of apnea, in which breathing ceases altogether. Periodic breathing does not commonly occur in normals during wakefulness; however, its frequency of incidence increases dramatically during ascent to altitude as well as during sleep onset. An exaggerated form of periodic breathing, known as *Cheyne–Stokes breathing*, is frequently observed in patients with congestive heart failure. A large body of evidence suggests that periodic breathing results from an instability in the feedback control system that regulates ventilation and arterial blood gases. In this section, we will demonstrate that this is a reasonable hypothesis by applying stability analysis to a linearized model of chemoreflex regulation of ventilation. Recall that a steady-state nonlinear model for arterial CO_2 and O_2 regulation was discussed previously in Section 3.7. In the present model, we assume that the system is operating under normoxic conditions, so that the chemoreflex response to hypoxia can be ignored. However, the various components of the model are assigned dynamic properties. Since the response of the "central" chemoreceptors located in the ventral medulla is much more sluggish than that of the "peripheral" (carotid body) chemoreceptors, it is convenient to assume that there are functionally two feedback loops in this system: one representing the central chemoreflex and the other representing the peripheral chemoreflex. We also incorporate into the model the delays taken to transport blood from the lungs and the chemoreceptors. A simplified schematic diagram of this dynamic model is shown in Figure 6.15.

6.7.1 CO$_2$ Exchange in the Lungs

The dynamic equivalent of the gas exchange equation given in Equation 3.48 is

$$V_{lung} \frac{dP_{ACO_2}}{dt} = \left(\dot{V}_E - \dot{V}_D \right)\left(P_{ICO_2} - P_{ACO_2} \right) + 863Q\left(C_{vCO_2} - C_{aCO2} \right) \quad (6.42)$$

where Q represents pulmonary blood flow, V_{lung} is the effective CO_2 storage capacity of the lungs, and C_{aCO2} and C_{vCO2} are the CO_2 concentrations in arterial and mixed venous blood, respectively. Other symbols are as defined previously in Section 3.7. It should also be noted that, in the steady state, the last term in Equation 6.42 would equal 863 $\dot{V}_{CO2}$, where $\dot{V}_{CO2}$ is the metabolic production rate of CO_2.
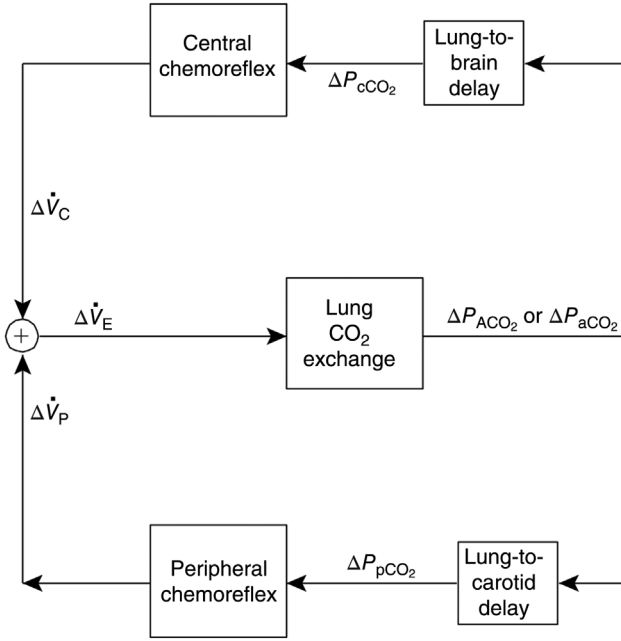
FIGURE 6.15 Linearized dynamic model of the chemoreflex control of ventilation.

Suppose that small perturbations are imposed on $\dot{V}_E$ ($\Delta\dot{V}_E$) and that these lead to small perturbations in P_{ACO2} (ΔP_{ACO2}) and C_{aCO2} (ΔC_{aCO2}). If we ignore the effect of the arterial blood gas fluctuations on mixed venous CO_2 concentration (since the body tissues represent a very large buffer of CO_2 changes), assume that dead space ventilation remains constant and ignore terms involving the product $\Delta\dot{V}_E\Delta P_{ACO2}$, we can derive the following small-signal expression from Equation 6.42:

$$V_{lung}\frac{d\left(\Delta P_{ACO_2}\right)}{dt} = -\left(\dot{V}_E - \dot{V}_D\right)\Delta P_{ACO_2} + \left(P_{ICO_2} - P_{ACO_2}\right)\Delta\dot{V}_E - 863Q\Delta C_{aCO_2}$$

$$(6.43a)$$

If we approximate the blood CO_2 dissociation curve with a straight line relating C_{aCO2} to P_{aCO2} with slope K_{CO2}, and impose the assumption of alveolar–arterial P_{CO2} equality, we obtain from Equation 6.43a the following result:

$$V_{lung}\frac{d\left(\Delta P_{aCO_2}\right)}{dt} + \left(\dot{V}_E - \dot{V}_D + 863QK_{CO_2}\right)\Delta P_{aCO_2} = \left(P_{ICO_2} - P_{aCO_2}\right)\Delta\dot{V}_E$$

$$(6.43b)$$

Note that in Equations 6.43a and 6.43b, $\dot{V}_E$ and P_{aCO2} represent the steady-state operating levels of minute ventilation and arterial CO_2 tension, respectively. If we

take the Laplace transform of Equation 6.43b and rearrange terms, we can obtain the following expression for the transfer function, $H_{lung}(s)$, of the lungs:

$$H_{lung}(s) \equiv \frac{\Delta P_{aCO_2}}{\Delta \dot{V}_E} = \frac{-G_{lung}}{\tau_{lung}s + 1} \tag{6.44}$$

where

$$\tau_{lung} = \frac{V_{lung}}{\dot{V}_E - \dot{V}_D + 863QK_{CO_2}} \tag{6.45}$$

and

$$G_{lung} = \frac{P_{aCO_2} - P_{ICO_2}}{\dot{V}_E - \dot{V}_D + 863QK_{CO_2}} \tag{6.46}$$

Equations 6.44 through 6.46 indicate that, under small-signal conditions, the dynamics of CO_2 exchange in the lungs may be modeled approximately as a simple first-order system with time constant τ_{lung} and gain G_{lung}. Note, however, that τ_{lung} and G_{lung} will vary, depending on the steady-state operating levels of $\dot{V}_E$ and P_{aCO2}. This reflects the fundamentally *nonlinear* nature of the gas exchange process. Another important detail is that the negative value for $H_{lung}(s)$ in Equation 6.44 merely implies that the negative feedback in this closed-loop system is embedded in the CO_2 exchange process (i.e., when ventilation increases, P_{aCO2} decreases). G_{lung} will always be positive since P_{aCO2} must be greater than P_{ICO2} (for positive metabolic CO_2 production rates).

6.7.2   Transport Delays

We assume that pulmonary end-capillary blood returning to the heart will take some time (T_p) to arrive at the peripheral chemoreceptors (carotid bodies) and a slightly longer time ($T_c > T_p$) to first appear at the site of the central (medullary) chemoreceptors. Thus,

$$\Delta P_{pCO_2}(t) = \Delta P_{aCO_2}\left(t - T_p\right) \tag{6.47a}$$

$$\Delta P_{cCO_2}(t) = \Delta P_{aCO_2}(t - T_c) \tag{6.47b}$$

All mixing effects in the vasculature during the convective process are ignored. The Laplace transforms corresponding to Equations 6.47a and 6.47b are

$$\Delta P_{pCO_2}(s) = e^{-sT_p}\Delta P_{aCO_2}(s) \tag{6.48a}$$

$$\Delta P_{cCO_2}(s) = e^{-sT_c}\Delta P_{aCO_2}(s) \tag{6.48b}$$

6.7.3 Controller Responses

Following Bellville et al. (1979), we assume the following dynamic relations for the peripheral and central chemoreflex responses:

$$\tau_p \frac{d\dot{V}_p}{dt} + \dot{V}_p = G_p \left[P_{pCO_2} - I_p \right] \tag{6.49a}$$

$$\tau_c \frac{d\dot{V}_c}{dt} + \dot{V}_c = G_c \left[P_{cCO_2} - I_c \right] \tag{6.49b}$$

$$\dot{V}_c + \dot{V}_p = \dot{V}_E \tag{6.49c}$$

In the above controller equations, τ_p and τ_c represent the characteristic response times of the peripheral and central chemoreflexes, respectively; it is assumed that $\tau_c \gg \tau_p$. G_p and G_c represent the steady-state gains for the peripheral and central controllers, respectively. The brackets on the right-hand side of Equations 6.49a and 6.49b are used to imply a thresholding operation: That is, these terms will be set equal to zero if the quantities within the parentheses become negative. Thus, I_p and I_c represent the corresponding apneic thresholds for the peripheral and central chemoreceptors, respectively.

Assuming that the "set point" of operation is nowhere in the vicinity of the apneic thresholds, Equations 6.49a and 6.49b can be linearized using small-signal analysis. The result of this analysis following Laplace transformation yields

$$\Delta \dot{V}_p(s) = \frac{G_p}{\tau_p s + 1} \Delta P_{pCO_2}(s) \tag{6.50a}$$

$$\Delta \dot{V}_c(s) = \frac{G_c}{\tau_c s + 1} \Delta P_{cCO_2}(s) \tag{6.50b}$$

6.7.4 Loop Transfer Functions

Corresponding to the two feedback loops in this model are two loop transfer functions: one for the peripheral chemoreflex loop ($H_{Lp}(s)$) and the other for the central chemoreflex loop ($H_{Lc}(s)$). These are derived by combining Equations 6.44, 6.48a, and 6.50a:

$$H_{Lp}(s) \equiv \frac{\Delta \dot{V}_p(s)}{\Delta \dot{V}_E(s)} = \frac{G_{lung} G_p e^{-sT_p}}{(\tau_{lung} s + 1)(\tau_p s + 1)} \tag{6.51a}$$

$$H_{Lc}(s) \equiv \frac{\Delta \dot{V}_c(s)}{\Delta \dot{V}_E(s)} = \frac{G_{lung} G_c e^{-sT_c}}{(\tau_{lung} s + 1)(\tau_c s + 1)} \tag{6.51b}$$

The overall frequency response of the loop transfer function is defined as

$$H_L(\omega) \equiv \frac{\Delta \dot{V}_p(\omega) + \Delta \dot{V}_c(\omega)}{\Delta \dot{V}_E(\omega)} \tag{6.52}$$

If we combine Equations 6.51a and 6.51b and introduce the substitution $s = j\omega$, we obtain the following expression for the overall frequency response of the two loops:

$$H_L(\omega) = \frac{G_{\text{lung}}}{(1 + j\omega\tau_{\text{lung}})} \left(\frac{G_p e^{-j\omega\tau_p}}{(1 + j\omega\tau_p)} + \frac{G_c e^{-j\omega\tau_c}}{(1 + j\omega\tau_c)} \right) \tag{6.53}$$

The stability of the respiratory control model for a given set of parameters (V_{lung}, Q, K_{CO2}, G_p, τ_p, G_c, τ_c) and conditions ($\dot{V}_E$, P_{aCO2}, P_{ICO2}) can be tested by applying the Nyquist criterion to Equation 6.53.

6.7.5   Nyquist Stability Analysis Using MATLAB

The generation of the Nyquist diagram from Equation 6.53 can be carried out relatively easily using MATLAB. The following shows sample lines of MATLAB code that can be used for this purpose:

```
>> % Construct loop transfer functions
>> num1 = [Glung*Gp]; den1 = [taulung*taup  (taulung+taup)  1];
>> Hs1 = tf(num1, den1);
>> num2 = [Glung*Gc]; den2 = [taulung*tauc  (taulung+tauc)  1];
>> Hs2 = tf(num2, den2);

>> % Compute Nyquist results, excluding effect of pure delays
>> [R1,I1] = nyquist(Hs1,w);  R1 = squeeze
(R1);  I1 = squeeze(I1);
>> [R2,I2] = nyquist(Hs2,w);  R2 = squeeze(R2);  I2 =
squeeze(I2);

>> % Add delay to results
>> R1del = real((R1 + j*I1).*exp(-j*w*Tp));
>> I1del = imag((R1 + j*I1).*exp(-j*w*Tp));
>> R2del = real((R2 + j*I2).*exp(-j*w*Tc));
>> I2del = imag((R2 + j*I2).*exp(-j*w*Tc));
>> Rdel = R1del + R2del;
>> Idel = I1del + I2del;

>> % Plot Nyquist diagram of overall frequency response
>> axis square; plot(Rdel,Idel); grid;
```

Figure 6.16 shows Nyquist plots representing the overall frequency responses in the case of the typical normal subject (N) and the patient with congestive heart
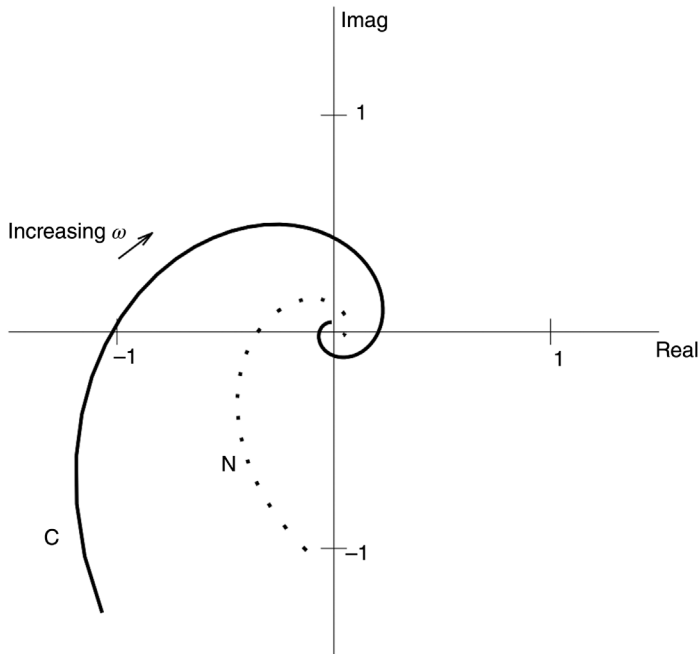
FIGURE 6.16 Nyquist plots representing the frequency responses of the linearized ventilatory control model in the normal subject ("N", dotted curve) and patient with congestive heart failure ("C", continuous curve). Frequencies represented in plots range from 0.01 to 0.1 Hz.

failure (C). We have assumed the following parameter values to represent *both* types of subjects: $V_{lung} = 2.5\,L$, $K_{CO2} = 0.0065\,\text{mmHg}^{-1}$, $G_p = 0.02\,L\,s^{-1}\,\text{mmHg}^{-1}$, $G_c = 0.04\,L\,s^{-1}\,\text{mmHg}^{-1}$, $\tau_p = 20\,s$, $\tau_c = 120\,s$, $\dot{V}_E = 0.12\,L\,s^{-1}$, $\dot{V}_D = 0.03\,L\,s^{-1}$, $P_{ICO2} = 0$, and $P_{aCO2} = P_{ACO2} = 40\,\text{mmHg}$. In the normal subject, the following circulatory parameter values were assigned: $T_p = 6.1\,s$ and $T_c = 7.1\,s$ and $Q = 0.1\,L\,s^{-1}$. In the patient with congestive heart failure, we assumed a halving of cardiac output and a doubling of the circulatory delays: $Q = 0.05\,L\,s^{-1}$, $T_p = 12.2\,s$, and $T_c = 14.2\,s$. The MATLAB script file (nyq_resp.m) used to generate the Nyquist plots shown is included with the set of MATLAB/SIMULINK files that accompany this book.

The Nyquist plots in Figure 6.16 represent a bandwidth of frequencies that range from 0.01 to 0.1 Hz. These correspond to interbreath periodicities of cycle durations 10–100 s. In the normal subject (N), the Nyquist plot shows a stable system with a critical loop gain (i.e., at −180°) of 0.34. This critical point occurs at $f = 0.0295\,\text{Hz}$ or the equivalent of a periodicity of 34 s. Thus, when transient oscillations in ventilation appear in the normal subject, we would expect these oscillations to have a cycle duration of 34 s. On the other hand, in the subject with congestive heart failure, the halving of Q and doubling of circulatory delays lead to a rotation and stretching of the Nyquist locus. Since the locus encircles the

$(-1+j0)$ point, we can conclude that under the assigned conditions, this system is unstable. The loop gain at $-180°$ is now 1.02, with the critical frequency occurring at $f = 0.0165$ Hz, which is equivalent to a periodicity of approximately 61 s. These cycle durations are consistent with the oscillation periods that have observed in normals and heart failure subjects who exhibit Cheyne–Stokes respiration. For more complicated models and analyses, the reader is referred to journal reports such as those published by Khoo et al. (1982), Carley and Shannon (1988), and Nugent and Finley (1987).

PROBLEMS

P6.1. Figure P6.1 shows a simple negative feedback control system with a variable gain K in the feedback element. Determine the smallest value of K that would render this closed-loop system unstable, using (a) the Routh–Hurwitz test and (b) the Nyquist stability criterion.
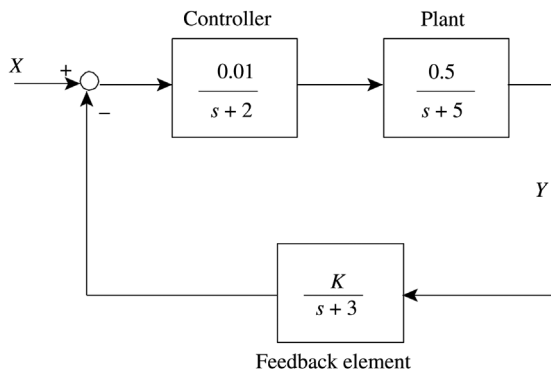
FIGURE P6.1   Simple control system with variable feedback gain.

P6.2. Consider the simple model of eye-movement control shown schematically in Figure P4.1. Assume that $G/J = 14{,}400 \, \text{rad}^2 \, \text{s}^{-2}$, $B/J = 24 \, \text{rad s}^{-1}$.

(a) If $K_v = 0.01$, deduce the gain and phase margins of this closed-loop system.
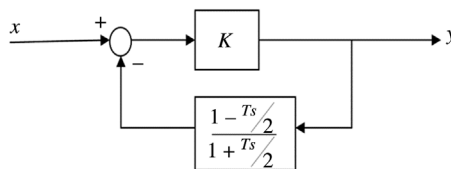
FIGURE P6.2   Negative feedback system with the first-order Padé approximation of delay T as the feedback block.

(b) Using the Routh–Hurwitz and Nyquist stability tests, determine the value of K_v at which you might expect the model to exhibit self-sustained oscillations in θ.

P6.3. In the model of ventilatory control with feedback from the intrapulmonary CO_2 receptors shown in Figure P4.3, determine how rate sensitivity is expected to affect relative stability. Compute the gain and phase margins of this system, and display the corresponding Bode diagrams, when the rate sensitivity factor α assumes the following values: (a) $\alpha = 0$; (b) $\alpha = 1/2$; and (c) $\alpha = 2$.

P6.4. In the analysis of the pupillary light reflex model discussed in Section 6.6, the transfer function representing the pure delay D was approximated as a power series in the Laplace variable s. Repeat the Routh–Hurwitz and Nyquist stability analyses, assuming a first-order Padé approximation to the delay, that is,

$$e^{-sD} = \frac{1 - (Ds/2)}{1 + (Ds/2)}$$

In each case, find the value of the steady-state loop gain K that would lead to the production of self-sustained oscillations in pupil diameter.

P6.5. Consider the neuromuscular reflex model of Figure 4.12. Develop a MATLAB program that would enable you to assess the relative stability of this model as the feedback gain β is changed. Assume the following values for the rest of the model parameters: $J = 0.1 \text{ kg m}^2$, $k = 50 \text{ N m}$, $B = 2 \text{ N m s}$, $\tau = 1/300 \text{ s}$, $T_d = 0.02 \text{ s}$, and $\eta = 5$. Determine if your prediction of the critical value of β for instability to occur is compatible with simulation results using the SIMULINK program `nmreflex.slx`.

P6.6. It is known that hyperoxia, induced by breathing a gas mixture with high O_2 content, can substantially attenuate the CO_2 sensitivity of the peripheral chemoreceptors. As a first approximation, we can assume that this sets the parameter G_p in the chemoreflex model of Section 6.7 equal to zero. This effectively reduces the model to only one feedback loop – that involving the central chemoreflex. Employing Routh–Hurwitz and Nyquist stability analyses, show that administration of inhaled O_2 would eliminate Cheyne–Stokes breathing in the patient with congestive heart failure. Use the parameter values given in Section 6.7.5.

P6.7. Develop a SIMULINK representation of the chemoreflex model described in Section 6.7, using the parameter values pertinent to the normal subject. Investigate the stability of the SIMULINK model by introducing impulsive perturbations into the closed-loop system. Determine how changes in the following model parameters may promote or inhibit the occurrence of periodic breathing: (a) V_{lung}; (b) P_{ICO2}; (c) G_c; and (d) G_p. Verify your

conclusions using the Nyquist analysis technique illustrated in Section 6.7.

P6.8. Consider the closed-loop system in Figure P6.2, where the transfer function of the feedback block is the first-order approximation to the delay T. Derive the characteristic equation of this system and determine, by evaluating the roots of the characteristic equation, whether the system is stable or unstable when $K = 2$. Assume $T = 1$.

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7

DIGITAL SIMULATION OF CONTINUOUS-TIME SYSTEMS

7.1 PRELIMINARY CONSIDERATIONS: SAMPLING AND THE Z-TRANSFORM

Up to this point, the models and their associated signals have been assumed to be characterized in continuous time: there is an infinitesimal difference in time between one time point and the next time point. Ordinary or partial differential equations have been the mathematical representations that we have employed to characterize the dynamics of models of physiological systems, and these operate in the continuous-time domain. Analytical solutions exist for a large class of continuous-time differential equations, but for the rest the only viable path is to employ a numerical method of solution. However, to do so requires a conversion of the problem from one in continuous time to the equivalent problem in discrete time. The way this is achieved in practice is to sample the continuous-time signal (commonly referred to as the *analog* signal) on a periodic basis. Continuous-time *systems* can also be converted to discrete-time *systems*, and in this chapter, we will demonstrate that this can be accomplished using different methods, each with different ramifications.

In Figure 7.1, we consider from a theoretical perspective what exactly occurs in the transformation of a continuous-time signal $x(t)$ into a discrete-time signal $x_D(n)$. The first part of the transformation involves the multiplication of $x(t)$ by a train $p(t)$ of unit impulses uniformly spaced T time units apart. Thus, $p(t)$ is
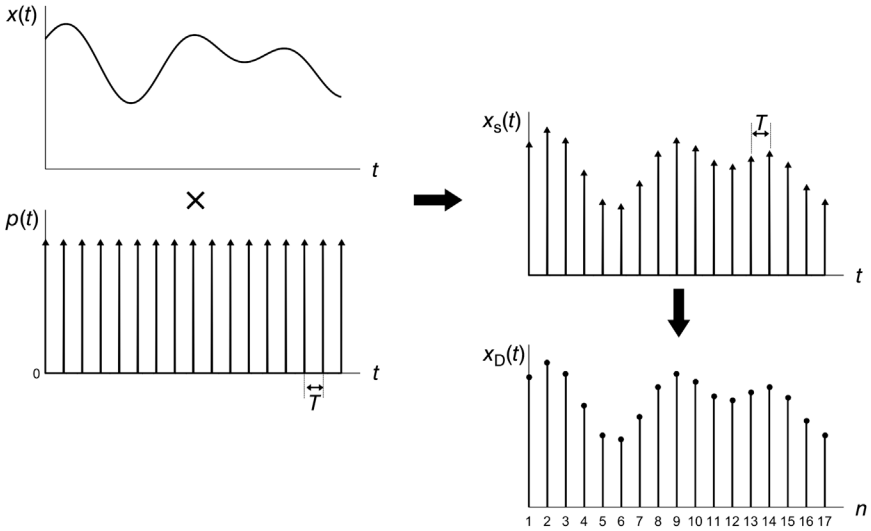
FIGURE 7.1   Schematic illustration of the process of sampling a continuous-time signal and converting it to discrete-time signal.

defined as

$$p(t) = \sum_{n=-\infty}^{\infty} \delta(t - nT) \tag{7.1}$$

where $n = 0, \pm1, \pm2, \pm3, \ldots, \pm\infty$, and $p(t) = 0$ when $t \neq nT$.

The resulting product is

$$x_s(t) = x(t) \cdot p(t) = x(t) \cdot \sum_{n=-\infty}^{\infty} \delta(t - nT) \tag{7.2}$$

Thus,

$$x_s(t) = \sum_{n=-\infty}^{\infty} x(nT) \cdot \delta(t - nT) \tag{7.3}$$

Since

$$\delta(t - nT) = 0, \quad \text{when} \quad t \neq nT \tag{7.4}$$

at the time points at which $x(t)$ is sampled, we have

$$x_D(n) = x(nT) \tag{7.5}$$

Note that, as a discrete-time signal, $x_D(n)$ is not defined between consecutive values of n.

Now, take the two-sided Laplace transform of Equation 7.3:

$$X_s(s) = \int_{-\infty}^{\infty} e^{-st} x_s(t) dt = \int_{-\infty}^{\infty} e^{-st} \sum_{n=-\infty}^{\infty} x(nT)\delta(t - nT) dt \qquad (7.6a)$$

$$= \sum_{n=-\infty}^{\infty} x(nT) \int_{-\infty}^{\infty} e^{-st} \delta(t - nT) \, dt \qquad (7.6b)$$

$$= \sum_{n=-\infty}^{\infty} x(nT) e^{-snT} \qquad (7.6c)$$

If we define the following equality

$$z = e^{sT} \qquad (7.7)$$

and use Equation 7.5, the right-hand side of Equation 7.6c can be converted into a form that contains the variable z. We can then define the following function of z:

$$X_D(z) = \sum_{n=-\infty}^{\infty} x_D(n) z^{-n} \qquad (7.8)$$

Equation 7.8 yields a mapping between the discrete-time signal $x(n)$ and the corresponding transformed quantity in the complex z-domain, $X_D(z)$. This "mapping" is called the *z-transform*.

The similarity in form between Equations 7.6c and 7.8 indicates that there is a one-to-one mapping between Laplace transform of $x_s(t)$ and the z-transform of $x_D(n)$. It can also be demonstrated that as T goes to zero, $X_D(z)$ converges to $X(s)$.

The utility of the z-transform for solving difference equations in discrete-time systems parallels that of the Laplace transform for solving differential equations in continuous time. A very simple result that is useful to keep in mind when employing the z-transform is the "delay theorem":

$$\sum_{n=-\infty}^{\infty} x_D(n - m) z^{-n} = z^{-m} \sum_{n=-\infty}^{\infty} x_D(n) z^{-n} = z^{-m} X_D(z) \qquad (7.9)$$

We will employ this result frequently in the following sections when we convert difference equations into transfer functions.

7.2 METHODS FOR CONTINUOUS-TIME TO DISCRETE-TIME CONVERSION

In this section, we examine four ways of converting a continuous-time linear system to a discrete-time linear system. In order to keep our focus on the conceptual aspects of these four methods, we will base our considerations on a highly simplified linear system: the lung mechanics model displayed in Figure 4.1. We will assume further that the fluid inertance effects are negligible, and thus the inductance element L will be equal to zero. Let $P_A = y$ and $P_{ao} = x$. Then, from Equation 4.3, we have

$$\tau \frac{dy}{dt} + y = x \tag{7.10}$$

where $\tau = RC$. We showed in Equation 4.7 that the transfer function with $x(P_{ao})$ as input and $y(P_A)$ as output (for the open-loop configuration of the model) is given by

$$H(s) = \frac{Y(s)}{X(s)} = \frac{1}{\tau s + 1} \tag{7.11}$$

The corresponding impulse response for this model is

$$h(t) = \frac{1}{\tau} e^{-t/\tau} \tag{7.12}$$

7.2.1 Impulse Invariance

In the impulse invariance method, the impulse response is sampled at a uniform interval of T time units, and thus the resulting sampled discrete-time impulse response is

$$h_D(n) = \frac{1}{\tau} e^{-nT/\tau}, \quad n \geq 0 \tag{7.13a}$$

$$\text{and} \quad h_D(n) = 0, \quad n < 0 \tag{7.13b}$$

Note that Equation 7.13b holds because $h_D(n)$ is causal.

Taking the z-transform of $h_D(n)$, we obtain

$$H_D(z) = \sum_{n=-\infty}^{n=\infty} h_D(n) z^{-n} \tag{7.14a}$$

But because of Equation 7.13b,

$$H_D(z) = \sum_{n=0}^{n=\infty} h_D(n) z^{-n} \tag{7.14b}$$

Substituting for $h_D(n)$ in Equation 7.14b, we get

$$H_D(z) = \frac{1}{\tau} \sum_{n=0}^{n=\infty} e^{-nT/\tau} z^{-n} = \frac{1}{\tau} \sum_{n=0}^{n=\infty} \left(e^{-T/\tau} z^{-1} \right)^n \tag{7.15}$$

However, note the equality:

$$\sum_{n=0}^{\infty} \propto^n = \frac{1}{1 - \propto} \tag{7.16}$$

Using Equation 7.16 in Equation 7.15 yields

$$H_D(z) = \frac{1/\tau}{1 - e^{-T/\tau} z^{-1}} \tag{7.17}$$

We can use Equation 7.17 to derive the equivalent difference equation relating the output $y(n)$ to the input $x(n)$ by recognizing that $H_D(z)$ is by definition $Y(z)/X(z)$. Then, substituting into Equation 7.17 and rearranging terms, we get

$$\left(1 - e^{-T/\tau} z^{-1} \right) Y(z) = \frac{X(z)}{\tau} \tag{7.18}$$

Taking the inverse z-transform of both sides of Equation 7.18, we obtain

$$y(n) = e^{-T/\tau} y(n-1) + \frac{x(n)}{\tau} \tag{7.19}$$

Equation 7.19 gives the solution to the equivalent discrete-time equivalent of Equation 7.10 for any type of input. A hallmark of this solution is that it is recursive in nature: At each time point, y depends on its own past value, as well as the values of the input $x(n)$. Thus, note that if the input is a unit impulse, that is,

$$x(0) = 1 \quad \text{and} \quad x(n) = 0, \quad \text{for} \quad n > 0 \tag{7.20}$$

then Equation 7.19 yields the impulse response:

$$y(n) = h_D(n) = e^{-T/\tau} h_D(n-1) = e^{-2T/\tau} h_D(n-2) = \cdots = \frac{1}{\tau} e^{-nT/\tau} \tag{7.21}$$

which is the expression for the impulse response that we started with in Equation 7.13a.

7.2.2 Forward Difference

In this method, we make use of the following numerical approximation for dy/dt that becomes more and more exact as T tends toward zero:

$$\frac{dy}{dt} = \frac{y(n+1) - y(n)}{T} \tag{7.22}$$

Thus, substituting this into Equation 7.10 yields

$$\tau \frac{y(n+1) - y(n)}{T} + y(n) = x(n) \tag{7.23}$$

Rearranging terms in Equation 7.23, we obtain

$$y(n+1) = \left(1 - \frac{T}{\tau}\right)y(n) + \left(\frac{T}{\tau}\right)x(n) \tag{7.24a}$$

or equivalently,

$$y(n) = \left(1 - \frac{T}{\tau}\right)y(n-1) + \left(\frac{T}{\tau}\right)x(n-1) \tag{7.24b}$$

Equation 7.24b provides a somewhat different solution in discrete time to Equation 7.10 than we had found using the impulse invariance method (Equation 7.19).

Using Equation 7.20 in Equation 7.24b allows us to derive the discrete-time impulse response of this system:

$$h_D(n) = 0, \quad n = 0 \tag{7.25a}$$

$$h_D(n) = \left(1 - \frac{T}{\tau}\right)^{n-1} \frac{T}{\tau}, \quad n > 0 \tag{7.25b}$$

The corresponding discrete-time transfer function can be derived by taking the z-transform of Equation 7.24b:

$$Y(z) = \left(1 - \frac{T}{\tau}\right)z^{-1}Y(z) + \left(\frac{T}{\tau}\right)z^{-1}X(z) \tag{7.26a}$$

Rearranging terms, Equation 7.26a yields the z-transform for the discrete-time system in question:

$$H_D(z) \equiv \frac{Y(z)}{X(z)} = \frac{(T/\tau)\, z^{-1}}{1 - (1 - (T/\tau))z^{-1}} \tag{7.26b}$$

7.2.3 Backward Difference

Here, the following numerical approximation for dy/dt, based on the difference between the current time point and the previous time point, is used:

$$\frac{dy}{dt} = \frac{y(n) - y(n-1)}{T} \tag{7.27}$$

Substituting into Equation 7.10, we have

$$\tau \frac{y(n) - y(n-1)}{T} + y(n) = x(n) \tag{7.28}$$

Rearranging terms, we obtain

$$y(n) = \frac{1}{1 + (T/\tau)} y(n-1) + \frac{T/\tau}{1 + (T/\tau)} x(n) \tag{7.29}$$

As in the previous cases, we obtain the impulse response by setting $x(n)$ to be equal to the unit impulse (Equation 7.20):

$$h_{\mathrm{D}}(n) = \left(1 + \frac{T}{\tau}\right)^{-n} \left(\frac{T/\tau}{1 + (T/\tau)}\right), \quad n \geq 0 \tag{7.30}$$

The transfer function for the backward difference system can be derived from Equation 7.29 by taking the z-transform of Equation 7.29:

$$Y(z) = \frac{1}{1 + (T/\tau)} z^{-1} Y(z) + \frac{T/\tau}{1 + (T/\tau)} X(z) \tag{7.31a}$$

Rearranging terms in Equation 7.31a yields the transfer function:

$$H_{\mathrm{D}}(z) \equiv \frac{Y(z)}{X(z)} = \left(\frac{T/\tau}{1 + (T/\tau)}\right) \left(1 - \frac{1}{1 + (T/\tau)} z^{-1}\right)^{-1} \tag{7.31b}$$

7.2.4   Bilinear Transformation

The bilinear transformation is best known through the following mapping between the s terms in the continuous-time transfer function $H(s)$ and the z terms in the discrete-time transfer function $H(z)$:

$$s = \frac{2}{T} \frac{1 - z^{-1}}{1 + z^{-1}} \tag{7.32a}$$

The following expression characterizes the same mapping, but expressing z as a function of s:

$$z = \frac{1 + (T/2)s}{1 - (T/2)s} \tag{7.32b}$$

One disadvantage of the bilinear transformation is that we need to have the expression representing the transfer function $H(s)$ before the conversion to discrete time can be performed.

For a more intuitive interpretation of this transformation, consider the inverse of Equation 7.32a:

$$\frac{1}{s} = \frac{T}{2}\frac{1 + z^{-1}}{1 - z^{-1}} \tag{7.32c}$$

Denoting the right-hand side of Equation 7.32c as $H(z)$, and remembering that it is by definition equal to $Y(z)/X(z)$:

$$\frac{Y(z)}{X(z)} = \frac{T}{2}\frac{1 + z^{-1}}{1 - z^{-1}} \tag{7.33a}$$

Rearranging terms in Equation 7.33a, we obtain

$$Y(z) - z^{-1}Y(z) = \frac{T}{2}\left(X(z) + z^{-1}X(z)\right) \tag{7.33b}$$

Taking the inverse z-transform of Equation 7.33b yields

$$y(n) = y(n - 1) + \frac{T}{2}(x(n) + x(n - 1)) \tag{7.34}$$

Equation 7.34 provides a useful, practical interpretation of Equation 7.32c. What it says is that the bilinear transformation converts the integration operation in continuous time (represented by $1/s$) into the equivalent operation of numerical integration in discrete time by employing the "trapezoidal rule."

We now turn back to deriving the discrete-time equivalent of the first-order continuous-time transfer function given in Equation 7.11. Starting with Equation 7.11 and applying the transformation defined in Equation 7.32a, we obtain

$$H_{\mathrm{D}}(z) = \frac{1}{1 + (2\tau/T)\left[(1 - z^{-1})/(1 + z^{-1})\right]} \tag{7.35}$$

From Equation 7.35, it is easy to show that the corresponding finite difference equation is

$$y(n) = \frac{1 - (T/2\tau)}{1 + (T/2\tau)}y(n - 1) + \frac{T/2\tau}{1 + (T/2\tau)}(x(n) + x(n - 1)) \tag{7.36}$$

The corresponding discrete-time impulse response can be derived from Equation 7.36 by setting $x(n)$ to be equal to the unit impulse (Equation 7.20):

$$h_{\mathrm{D}}(n) = \frac{T/2\tau}{1 + (T/2\tau)}, \quad n = 0 \tag{7.37a}$$

$$h_{\mathrm{D}}(n) = \frac{T/\tau}{(1 + (T/2\tau))^2}, \quad n = 1 \tag{7.37b}$$

$$h_D(n) = \left(\frac{1 - (T/2\tau)}{1 + (T/2\tau)}\right)^{n-1} \left(\frac{T/\tau}{1 + (T/2\tau)^2}\right), \quad n > 1 \qquad (7.37c)$$

7.3   SAMPLING

In the previous section, we showed that a given continuous-time system can be converted into more than one equivalent discrete-time systems, depending on the method employed to perform the analog-to-digital transformation. Another important parameter in this process is the rate at which the sampling of the continuous-time signal is carried out. It should be quite intuitive that, with a very slow sampling rate, one could miss much of the dynamics of a particular signal. In the example discussed above, different values of the ratio T/τ could lead to discrete-time equivalents with very different system dynamics. In addition, there is another fundamental phenomenon that poses its own challenges, if certain constraints are not kept – and this is the problem of *aliasing* that arises from employing sampling rates that are too low, relative to the dynamics of the continuous-time system.

Consider a continuous-time impulse response $h(t)$. The frequency response of this system is given by the Fourier transform of $h(t)$:

$$H(\omega) = \int_{-\infty}^{\infty} h(t)e^{-j\omega t} dt \qquad (7.38)$$

Conversely, the inverse Fourier transform of $H(\omega)$ yields the impulse response:

$$h(t) = \frac{1}{2\pi} \int_{-\infty}^{\infty} H(\omega)e^{j\omega t} d\omega \qquad (7.39)$$

If $h(t)$ is sampled at uniform intervals of T, the values of $h(t)$ at those points would be

$$h(nT) = \frac{1}{2\pi} \int_{-\infty}^{\infty} H(\omega)e^{jn\omega T} d\omega \qquad (7.40)$$

Equation 7.40 can be reformulated in a somewhat different way for use later. We make the following change in variables:

$$\phi = \omega T \qquad (7.41a)$$

From Equation 7.41a, we obtain

$$d\omega = \frac{1}{T} d\phi \qquad (7.41b)$$

Then, Equation 7.40 can be rewritten as

$$
\begin{aligned}
h(nT) &= \frac{1}{2\pi T} \int_{-\infty}^{\infty} H(\phi) e^{jn\phi} d\phi \\
&= \frac{1}{2\pi T} \sum_{m=-\infty}^{\infty} \int_{(2m-1)\pi}^{(2m+1)\pi} H(\phi) e^{jn\phi} d\phi
\end{aligned}
\tag{7.42a}
$$

Define

$$
\Omega \equiv \phi - 2\pi m \tag{7.43}
$$

Substituting into Equation 7.42a yields

$$
\begin{aligned}
h(nT) &= \frac{1}{2\pi T} \sum_{m=-\infty}^{\infty} \int_{-\pi}^{\pi} H(\Omega + 2\pi m) e^{jn\phi} d\Omega \\
&= \frac{1}{2\pi} \int_{-\pi}^{\pi} \sum_{m=-\infty}^{\infty} \frac{H(\Omega + 2\pi m)}{T} e^{jn\phi} d\Omega
\end{aligned}
\tag{7.42b}
$$

To better appreciate how aliasing affects the sampling process, we first return to the definition of the z-transform in Equation 7.8. Recall that the mapping between the s-domain and the z-domain is given by Equation 7.7. If we are interested in determining how the frequency response of a continuous-time system translates to the frequency response of its equivalent discrete-time system, what we would do is to evaluate the z-transform of the discrete-time system along the contour of the unit circle in the z-domain, that is,

$$
z = e^{-j\Omega n} \tag{7.44}
$$

Thus, the z-transform of the discrete-time impulse response becomes

$$
H_d(\Omega) = \sum_{n=-\infty}^{\infty} h_D(nT) e^{-j\Omega n} \tag{7.45}
$$

Note that Equation 7.45 says that the frequency response of the discrete-time system is given by the *discrete-time Fourier transform* of its impulse response. Conversely, we can represent the discrete-time impulse response as the inverse of the Fourier transform of its frequency response:

$$
h_d(nT) = \frac{1}{2\pi} \int_{-\pi}^{\pi} H_D(\Omega) e^{j\Omega n} d\Omega \tag{7.46}
$$

Since $h(nT)$ in Equation 7.42b is equal to $h_D(nT)$ in Equation 7.46, we can equate the term within the integral on the right-hand-side of Equation 7.42b to the corresponding term in Equation 7.46:

$$H_d(\Omega) = \sum_{m=-\infty}^{\infty} \frac{H(\Omega + 2\pi m)}{T} \tag{7.47}$$

From Equations 7.41a and 7.43, we see that

$$\omega = \frac{\Omega + 2\pi m}{T} \tag{7.48}$$

Substituting Equation 7.48 into Equation 7.47, we obtain the following result:

$$H_d(\Omega) = \sum_{m=-\infty}^{\infty} \frac{H(\omega - (2\pi m/T))}{T} \tag{7.49}$$

Equation 7.49 is highly significant in that it shows the fundamental relationship between the frequency response of a continuous-time system and the corresponding frequency response of its discrete-time equivalent derived by sampling the impulse response of the continuous-time system. This is best understood by presenting the concept in graphical form, as displayed in Figure 7.2. Figure 7.2a shows the frequency response (magnitude) plot of the continuous-time system, with the abscissa representing angular frequency ω. Since the (absolute) sampling frequency is $1/T$, where T is the sampling interval, the *angular* sampling frequency ω_s is related to T in the following way:

$$\omega_s = \frac{2\pi}{T} \tag{7.50}$$

Figure 7.2b displays the frequency response (magnitude) of the corresponding discrete-time system. Equation 7.49 shows that this frequency response is a reduced-amplitude version of the original frequency response. In addition, the frequency response of the continuous-time system is duplicated an infinite number of times and centered around multiples of $2\pi m$, where $m = 0, \pm 1, \pm 2$, and so on. Figure 7.2 highlights another important detail that appears in Equation 7.49: That the magnitude of the frequency response of the continuous-time system $H(\omega)$ (Figure 7.2a) is scaled by the factor T in the frequency response of the corresponding discrete-time system $H_D(\Omega)$ (Figure 7.2b). The scaled duplicates of the original frequency response are known as *aliases*. The relationship between $H(\omega)$ and $H_D(\Omega)$ is best understood if we recall from Equation 7.7 that the sampling process is equivalent to mapping the imaginary axis $(j\omega)$ of the s-plane into the unit circle of the z-plane, that is,

$$e^{j\Omega} = e^{j\omega T} \tag{7.51}$$

(a)

(b)
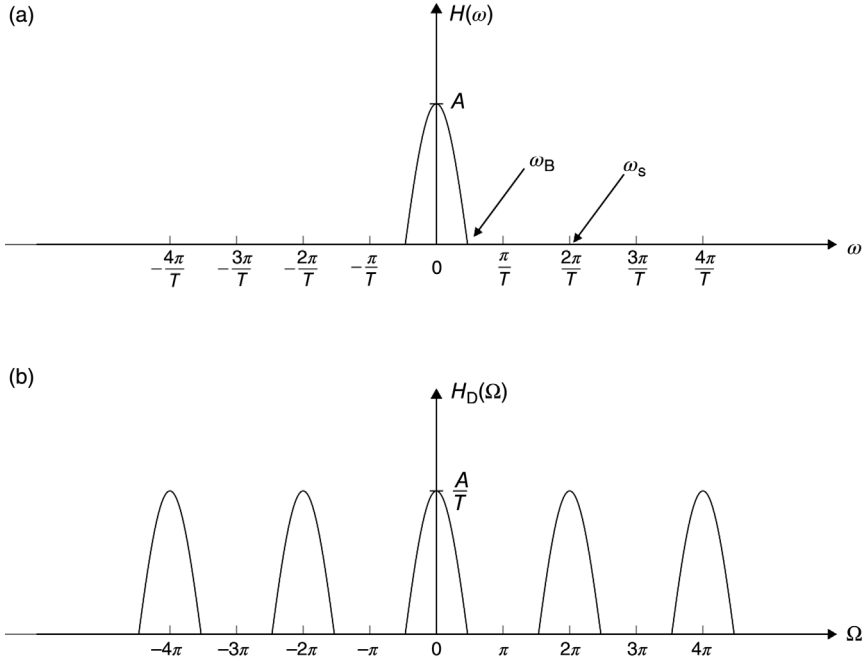
FIGURE 7.2 (a) Frequency response of a continuous-time system. (b) Frequency response of the discrete-time equivalent of the above continuous-time system derived using impulse invariance. $\omega_B =$ highest frequency associated with dynamics of the continuous-time system; $\omega_s =$ sampling frequency.

As illustrated in Figure 7.3, the segment of the $j\omega$ axis from $\omega = -\pi/T$ to $\omega = \pi/T$ gets mapped to the unit circle of the z-plane from $\Omega = -\pi$ to $\Omega = \pi$ in the anticlockwise direction. What about the segment of $j\omega$ from $\omega = \pi/T$ to $\omega = 3\pi/T$ in the s-plane? One can surmise from Figures 7.2 and 7.3 that this next segment is mapped into the z-plane as another anticlockwise wrap around the unit circle, from $\Omega = -\pi$ to $\Omega = \pi$. Similarly, each "strip" of length $2\pi/T$ of the $j\omega$ axis gets wrapped around the unit circle in the z-plane.

Based on the considerations illustrated in Figures 7.2 and 7.3, it is not too difficult to understand why $H_D(\Omega)$ contains multiple aliases of $H(\omega)$. However, it is important to note that each alias of $H_D(\Omega)$ takes the exact form of $H(\omega)$ under certain constraints. In Figure 7.2, $H(\omega)$ is shown to have a bandwidth (highest frequency) of ω_B, and ω_B is less than π/T. Since the (angular) sampling frequency ω_s is equal to $2\pi/T$,

$$\omega_B < \frac{\omega_s}{2} \tag{7.52}$$
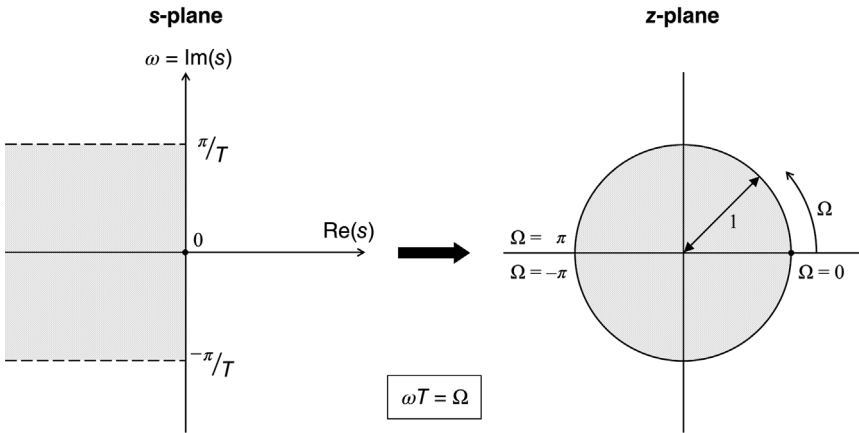
FIGURE 7.3 Schematic illustration of how the *s*-plane (associated with continuous-time system) maps to the *z*-plane (associated with discrete-time system), according to the relationship $z = e^{sT}$.

Figure 7.4 shows an example when the condition specified in Equation 7.52 does not hold. Here, because $\omega_B > \omega_s/2$, the ends of the frequency responses of the main transfer function and its aliases run into one another, causing distortion in the regions where there is overlap. When this happens, parts of the system response with frequencies higher than $\omega_s/2$ appear as components in frequencies lower than $\omega_s/2$. This phenomenon is known as *aliasing*. $\omega_s/2$ is also known as the *Nyquist frequency* or *folding frequency*. Equation 7.52 represents the concise version of the Nyquist–Shannon sampling theorem, namely, that a continuous-time signal can be fully reconstructed from its discrete-time equivalent only when the sampling frequency is greater than or equal to twice the highest frequency component of original signal. A corollary of this theorem is that if one samples a periodic phenomenon at the primary frequency of the process, then the aliasing effect will make the dynamic phenomenon appear static. This is the principle by which the stroboscopic effect works.

7.4 DIGITAL SIMULATION: STABILITY AND PERFORMANCE CONSIDERATIONS

In this section, we compare how well the various methods of converting continuous-time systems to discrete-time systems work when they are implemented in MATLAB. We use the example of the first-order system discussed in Section 7.3. Assuming the associated time constant τ to be equal to 1 s, we will consider how the dynamics of the equivalent discrete-time systems derived from the four methods of discrete-to-continuous-time conversion compare with the dynamics of the original continuous-time system. The MATLAB program
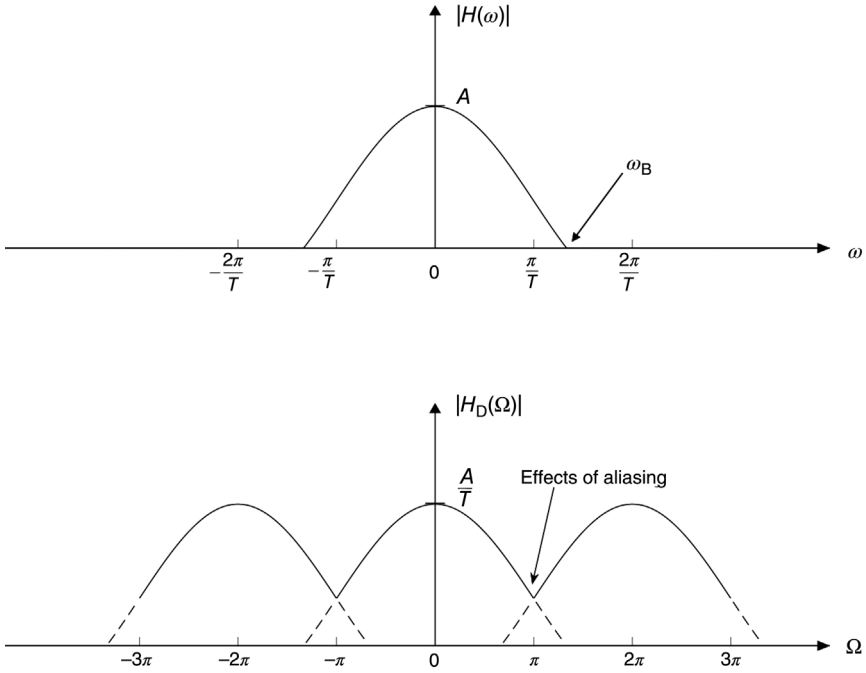
FIGURE 7.4 Aliasing occurs when the bandwidth or the highest frequency of the continuous-time system is greater than half the sampling frequency, that is, $\omega_B > \omega_s/2$. This is equivalent to $\omega_s < 2\omega_B$, thus violating the sampling theorem.

`CT2DTsys_impresp.m` implements the four methods of conversion discussed in the last section:

```
%% Continuous time
h = 1/tau * exp(-tc/tau);

%% Discrete time
x = zeros(size(t));
x(t==0) = 1; %input = unit impulse x(n=0) = 1
n0 = find(t==0); %n=0

%Method 1: impulse invariance
yii = nan(size(t));
yii(t<0) = 0; %y(n<0)
for nn=n0:length(t)
    yii(nn) = exp(-T/tau) * yii(nn-1) + x(nn)/tau; %y(n>0)
end

%Method 2: forward difference
yfd = nan(size(t));
yfd(t<0) = 0; %y(n<0)
```

```
for nn=n0:length(t)
    yfd(nn) = (1 - T/tau)*yfd(nn-1) + (T/tau)*x(nn-1);
end
yfd = yfd/T;   % scaling factor


%Method 3: backward difference
ybd = nan(size(t));
ybd(t<0) = 0; %y(n<0)
Ky = 1/(1 + T/tau); %scaling factor of y
Kx = (T/tau)/(1 + T/tau); %scaling factor of x
for nn=n0:length(t)
    ybd(nn) = Ky*ybd(nn-1) + Kx*x(nn);
end
ybd = ybd/T;   % scaling factor


%Method 4: bilinear transformation
ybt = nan(size(t));
ybt(t<0) = 0; %y(n<0)
Ky = (1 - T/(2*tau)) / (1 + T/(2*tau)); %scaling factor of y
Kx = (T/(2*tau)) / (1 + T/(2*tau)); %scaling factor of x
for nn=n0:length(t)
    ybt(nn) = Ky*ybt(nn-1) + Kx*(x(nn) + x(nn-1));
end
ybt = ybt/T;   % scaling factor
```

As displayed in Figure 7.5a and b, the thick black curve represents the impulse response of the continuous-time system, that is, $h(t)$ in Equation. 7.12. The impulse invariance method consists of simply sampling $h(t)$ at uniform time intervals of T; the mathematical representation is given in Equation 7.13a. Figure 7.5 displays the sampled points as closed black circles that lie on the trajectory of $h(t)$ at two different sampling intervals: $T = 0.1$ s (Figure 7.5a) and $T = 1$ s (Figure 7.5b). This is clearly the reason why this method is known as "impulse invariance." With the forward difference method, the impulse response of the discrete-time equivalent is described by Equations 7.25a and 7.25b. As shown in Figure 7.5 (upright triangles), the peak of the impulse response is delayed by one point. In the case for $T = 0.1$, since the impulse response begins with magnitude zero at time zero, the discrepancy between the discrete-time and continuous-time impulse responses is largest before $t = 0.5$ s, but both responses converge subsequently. However, when $T = 1$, the impulse response of the discrete-time system generated using forward difference oscillates between values of -2 and 2. Thus, clearly, with relatively large T (with respect to τ), the stable continuous-time system gets converted into an unstable discrete-time equivalent. On the other hand, with the backward difference method (inverted triangles, Figure 7.5), the impulse response of the discrete-time system converges toward the trajectory of $h(t)$, regardless of whether T is 0.1 or 1 s, following an initial period of discrepancy. Similarly, the impulse response of the discrete-time equivalent obtained by using the bilinear transformation (open squares, Figure 7.5) converges toward $h(t)$ after the second
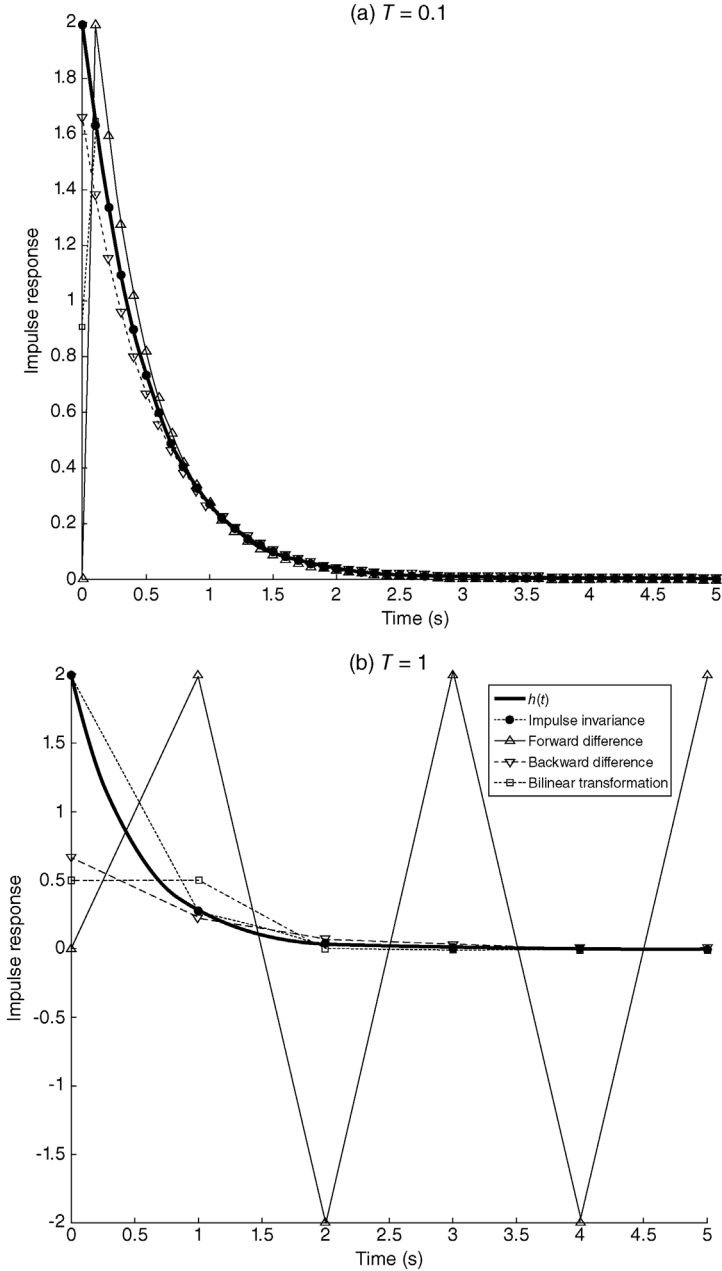
FIGURE 7.5   Impulse responses of discrete-time systems derived from a first-order low-pass continuous-time system using various methods of CT–DT conversion: impulse invariance (closed circles), forward difference (upright triangles), backward difference (inverted triangles), and bilinear transformation (squares), using time step (T) of (A) 0.1 s and (B) 1 s.

point. Note that, in the code for `CT2DTsys_impresp.m` (displayed above), a scaling factor of $1/T$ is applied to the solutions corresponding to the forward difference, backward difference, and bilinear transformation systems in order to make the magnitudes of the discrete-time impulse responses comparable to that of the continuous-time system. This scaling factor stems from the fundamental difference between continuous-time systems and discrete-time systems. In continuous time, the "impulse" takes the form of an infinitely high and infinitely thin "spike," but the total area under the spike is one. In discrete time, the "impulse" simply takes on the value of 1 at time zero. While the discrete-time system generated using impulse invariance has an impulse response function whose values fall on the impulse response of the continuous-time system, its step response needs to be scaled appropriately to match the step response of the continuous-time system.

The relationship between τ, which reflects the dynamics of the continuous-time system, and the sampling interval T used in developing the discrete-time equivalent is simply another manifestation of the relationship between the system bandwidth ω_B and the sampling frequency ω_s, as we had discussed in Section 7.3. When the sampling frequency is less than twice the bandwidth of the continuous-time system in question, aliasing occurs. Equivalently, when the ratio of T to τ becomes too large, aliasing introduces "distortion" into the dynamics of the discrete-time system vis-à-vis the original continuous-time system. This is the reason why the impulse responses of the discrete-time systems become progressively more different from that of the parent continuous-time system as T increases. On the other hand, the unstable behavior of the discrete-time system generated via the forward difference method with large T is derived from a different source. Recall, from Chapter 6, that the poles of a stable continuous-time system are always located on the left-hand side of the s-plane, that is, the real parts of the poles must be negative. Now, consider Figure 7.3 that shows how the s-plane gets mapped into the z-plane. Notice that the left-hand side of the s-plane maps into the area within the unit circle in the z-plane (shaded regions in Figure 7.3). Thus, discrete-time equivalent of a continuous-time system will be stable as long as the poles of the discrete-time system fall within the unit circle. Now, consider Equation. 7.26b, the transfer function corresponding to the discrete-time system derived using the forward difference method. This transfer function can be rewritten as

$$H_D(z) = \frac{T/\tau}{z - (1 - (T/\tau))} \qquad (7.26c)$$

Thus, the pole at $s = -1/\tau$ in the continuous-time system gets mapped into a pole at $z = 1 - T/\tau$ in the discrete-time equivalent derived using forward difference method. Note that, when T ranges between zero and 2τ, the pole of the discrete-time system falls within the unit circle (i.e., $-1 < z < 1$). But when $T > 2\tau$, this pole will lie outside the unit circle – This is when the discrete-time equivalent of the stable continuous-time system becomes unstable. Similar considerations can be applied to the backward difference and bilinear transformation methods – But in these cases, the corresponding discrete-time system always remains stable.

7.5 PHYSIOLOGICAL APPLICATION: THE INTEGRAL PULSE FREQUENCY MODULATION MODEL

A major motivation for converting a model containing continuous-time systems and signals into a representation in which these systems and signals are now expressed in discrete time is that this allows for more convenient estimation of the model parameters, especially when the measurements employed for estimation are collected on a sample-by-sample basis. This is essentially what happens anyway since analog signals have to be digitized before being acquired on any computer. However, when sampling frequency is very high relative to the dynamics of the system under study, we can still employ continuous-time models (e.g., in the form of differential equations) but use a wide plethora of numerical integration techniques to solve these equations. But there are many instances in which the physiological variables under study occur naturally on a sample-by-sample basis. The obvious examples are cardiac variables, such as heart period and stroke volume, both of which can be quantified on a beat-to-beat basis. Respiratory variables can be expressed on a per-breath basis. Physiological oscillations are so ubiquitous that it is not unusual to quantify the underlying time base in units of "cycles." Since arterial blood pressure fluctuates between systolic and diastolic levels within each cardiac cycle, one can define new descriptors such as the cycle-averaged blood pressure, systolic pressure, and diastolic pressure on a beat-by-beat basis.

Neural signals are another excellent example of the kind of model where continuous-time inputs can yield outputs that may be approximated as discrete "spikes." In this case, the underlying "drive" may be continuous, but the output is in the form of a train of neural impulses. Generally, when the "drive" is high, the neural system would depolarize more rapidly and generate an action potential more quickly – As such, a high drive would produce a high rate of neuronal firing. The integral pulse frequency modulation (IPFM) model, introduced by Bayly (1968), has been employed in many theoretical studies of neuronal dynamics. Figure 7.6 displays a schematic diagram that highlights how the IPFM model works. The following equations specify the operations of each of the modules in
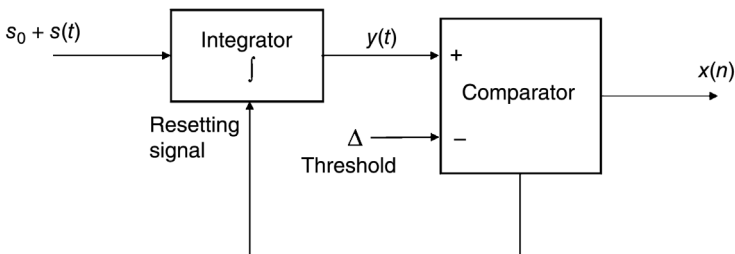
FIGURE 7.6 Schematic diagram of the integral pulse frequency modulation (IPFM) model. (Adapted from Chiu and Kao (2001).)

the IPFM model:

$$y(t) = \int_{t_n}^{t} [(s_0 + s(t))dt] \tag{7.53}$$

where $t_n \leq t \leq t_{n+1}$, and

$$y(t_{n+1}) - y(t_n) = \Delta \tag{7.54}$$

Note, in Equation 7.53, that s_0 represents the intrinsic drive, while $s(t)$ represents the modulated component of the drive. The square brackets [. . .] operate by disallowing any negative values to occur; if the argument becomes negative, the square brackets will function as a thresholding operation, setting everything to zero if the argument within the square brackets goes negative. Δ represents a threshold that determines the intrinsic frequency of the generated pulses when $s(t)$ is equal to zero. The output of the integral in Equation 7.53 is constantly compared to the selected threshold Δ, and once the difference between y at time t_{n+1} and y at time t_n equals Δ, a spike is generated at the output of the comparator module. At that same instant, a signal is sent to the integrator to reset and start integrating the input again.

A SIMULINK implementation of the IPFM model (`IPFM.slx`) is displayed in Figure 7.7. Figure 7.7a shows the IPFM model as a subsystem that receives the neural drive input and outputs the corresponding response in the form of a spike train. Figure 7.7b shows the internal workings of the IPFM. The "neural drive" takes the form of a continuous-time signal with mean value s_0 and fluctuating component $s(t)$. It has units of impulses (or cycles) per second – Hence, it represents the instantaneous neural firing frequency. This continuous-time signal is first integrated and compared with the threshold Δ. When the integral has risen to the point at which it attains the value of the threshold, the model generates a "spike." This "process" may be thought of as being analogous to the depolarization of the nerve cell membrane prior to the point at which an action potential is generated. The SIMULINK implementation shown here assumes a threshold value of 1, and we can consider this example as a model of how the totality of autonomic input to the heart generates the surge of electrical activity that triggers ventricular contraction (observable via the electrocardiogram as the "R-wave"). The "neural drive" in this case would be the instantaneous heart rate. In the SIMULINK implementation, the instantaneous heart rate (in cycles per second) is integrated continuously until the integral attains the value of 1. At this point, the "hit crossing" block generates a unit impulse (spike), the integrator is reset to zero, and integration of the input (instantaneous heart rate) resumes, starting from zero. The total duration over which each cycle of integration takes place equals the heart period for that beat. In the SIMULINK implementation, the tracking of the "R-to-R interval" is taken care of through the use of the second integrator, which integrates a constant
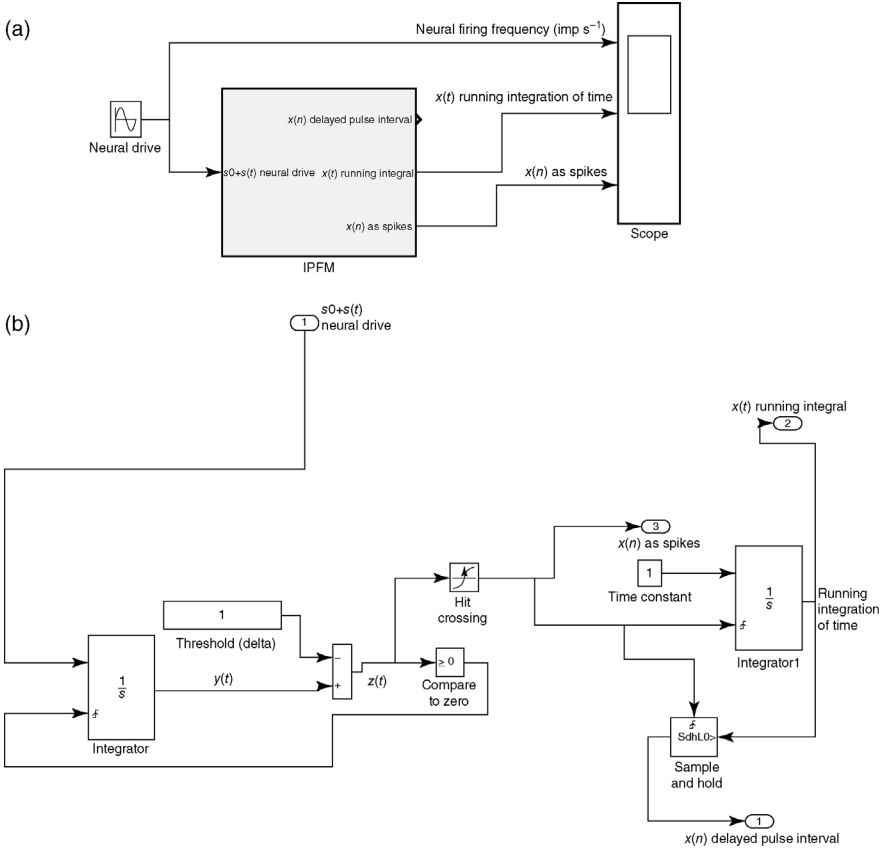
FIGURE 7.7 SIMULINK implementation of IPFM model. (a) Overall model showing IPFM subsystem with input (neural drive or neural firing frequency) and outputs. (b) SIMULINK structure of IPFM mechanism (see text for explanation).

input value of 1 until it is reset to zero by the next "spike" issued by the "hit crossing" block:

$$x(n) = \int\limits_{t_n}^{t_{n+1}} 1 \, dt \qquad (7.55)$$

Figure 7.8 displays the results of running IPFM.slx with a constant level of cardiac autonomic input equivalent to a heart rate of 0.5 beat s^{-1} (or 30 beats min^{-1}) for the first 30 s and a different constant level of 1 beat s^{-1} (or 60 beats min^{-1}) for the following 30 s (part (a)). Figure 7.8b shows, for each beat, the running time count (output of the second integrator) that occurs in parallel with the integration of the cardiac autonomic input signal (accomplished through the first integrator in
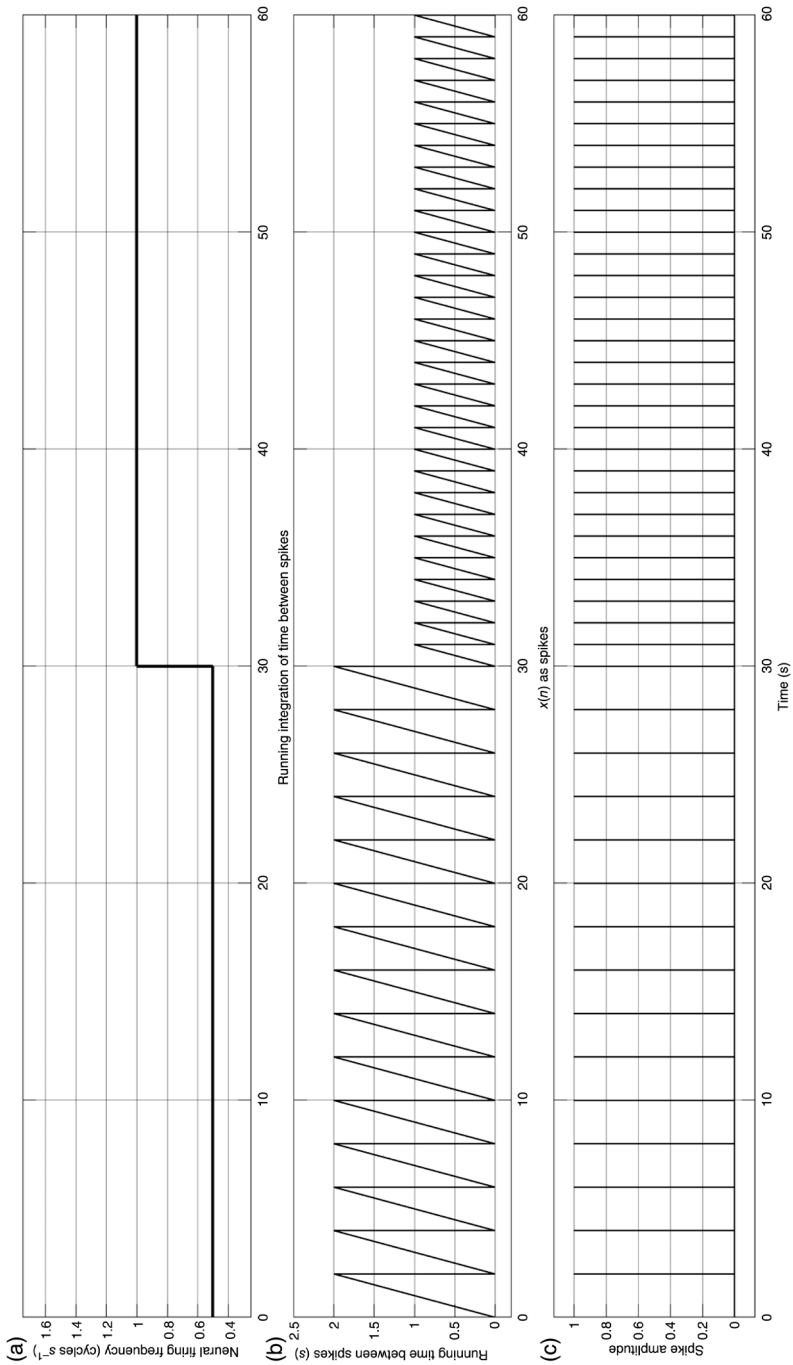
FIGURE 7.8 Simulation of IPFM model with fixed neural drives.

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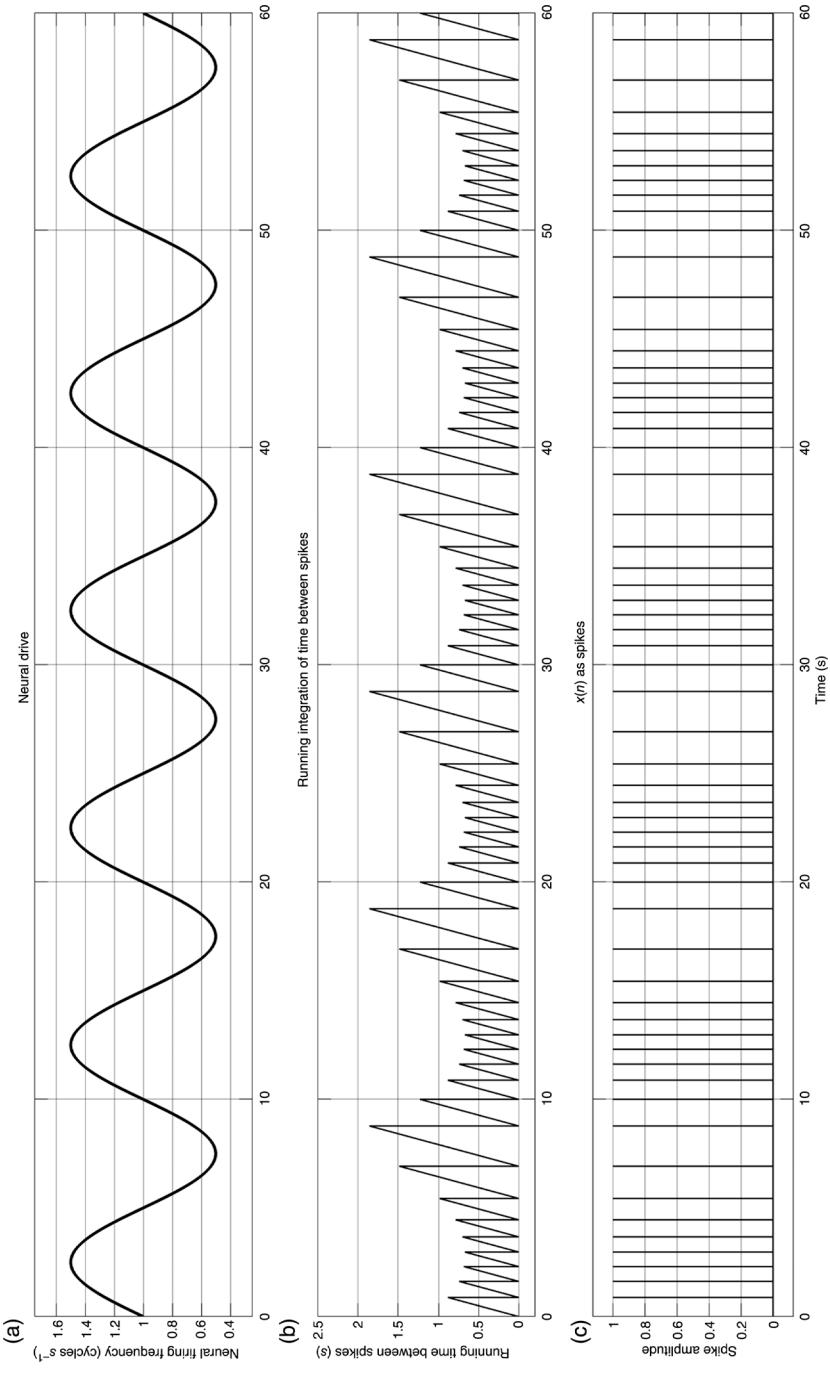
FIGURE 7.9 Simulation of IPFM model with sinusoidally modulated neural drive.

Figure 7.7). Recall that the first integrator resets to zero once the integral attains the value of 1. The second integrator resets to zero simultaneously, but in this case, the highest value of the integral achieved before it is reset yields the duration of time elapsed since the previous beat. In this example, this time interval equals 2 s in the first half of the simulation and 1 s in the second half. Figure 7.8c displays the main output of the IPFM model, that is, the "spikes" of unit amplitude that are generated with periodicities consistent with the input neural drive. In this first half of this simulation, the heart rate is 0.5 beat s^{-1}, equivalent to a heart period of 2 s, whereas in the second half, the heart rate of 1 beat s^{-1} yields a heart period of 1 s. Figure 7.9 shows another simulation, but this time, the autonomic input to the heart fluctuates sinusoidally with an amplitude of 0.5 beat s^{-1} around a mean level of 1 beat s^{-1}. This input represents an oscillatory drive that should make instantaneous heart rate vary between 0.5 and 1.5 beats s^{-1}. When it is sent through the IPFM, the output is a train of spikes (beats) that varies in interbeat interval between 0.6 and 1.8 s (Figure 7.9b and c).

PROBLEMS

P7.1. Consider a saline-filled catheter that has been inserted into the brachial artery of a patient so that the proximal tip of the catheter is exposed to blood flowing through the artery at pressure P_a. The distal tip of the catheter is connected to a pressure transducer. The transducer works by means of an internal thin diaphragm that deflects by an amount proportional to the difference between the pressure in the transducer chamber (P_m) and the ambient pressure (which we will consider to be equal to zero). This arrangement is displayed in Figure P7.1. Under static conditions, P_m should be exactly equal to P_a. However, this will not be true if P_a varies dynamically. How much P_m differs
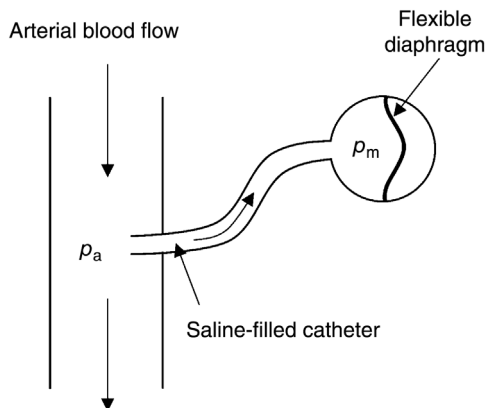
FIGURE P7.1   Schematic illustration of catheter–transducer system for measuring arterial pressure.

from P_a at any given time would depend on the response characteristics of the catheter–transducer system. If the mechanical properties of the transducer diaphragm and the dynamics of fluid motion in the catheter are known (based on prior testing), it is possible to employ a simple model to determine how the true arterial pressure signal is likely to be distorted dynamically by the measurement process.

(a) Derive the simplest linear lumped parameter model of the catheter–transducer arrangement that relates P_m to P_a. Include in the model the effects of (i) resistance R to fluid motion along the catheter, (ii) inertance L due to fluid acceleration along the catheter, and (iii) compliance C of the transducer diaphragm. We will consider the saline inside the catheter to be incompressible and the catheter wall to be nondistensible.

(b) Use the forward difference (Euler) method for converting the continuous-time model above into a discrete-time model. With the resulting difference equation, compute the responses in P_m of the discrete-time model to a unit step in P_a when the time step (sampling interval) $T = 0.1$ s and when $T = 2$ s. Assume the following values for the model parameters: $R = 0.05$, $L = 0.1$, and $C = 10$.

(c) Using SIMULINK, determine the response of the *continuous-time system* to a unit step, and display this alongside the two responses obtained in part (b).

(d) Determine an expression for the transfer function of the discrete-time system ($P_m(z)/P_a(z)$). By examining the locations of the poles of this system on the z-plane, explain why the stability properties of the two discrete-time representations ($T = 0.1$ versus $T = 2$) are different.

P7.2. In an experiment on humans, the ventilatory response to a single-breath challenge of CO_2 was measured, that is, during one breath, the inhaled CO_2 concentration was changed abruptly from 0 to 10% against a background mixture of air. Subsequently, the same subjects were exposed to the same CO_2 challenge, except that this was performed against a background mixture of hypoxic gas. In all subjects, the following model was found to provide an adequate fit to the data:

$$y(n) = ay(n-1) + bx(n-M)$$

where x and y represent changes from the mean levels of inhaled CO_2 concentration and ventilation, respectively. n represents the current breath, and M represents the delay (in number of breaths) between exposure to CO_2 and the change in ventilation that follows. Note that the measurements were made on a breath-by-breath basis, and therefore as a first approximation, they may be considered samples of an underlying continuous-time process that were acquired with a sampling interval equal to the subject's average breath period (T).

(a) Derive an expression for the transfer function (i.e., $H(s) = Y(s)/X(s)$) of the equivalent continuous-time model, assuming impulse invariance. Show clearly how a, b, M, and T are related to the parameters of $H(s)$.

(b) In one subject, suppose the following values were estimated from the data:

Normoxia: $a = 0.684$, $b = 0.059$, $M = 3$, $T = 3.3$ s
Hypoxia: $a = 0.624$, $b = 0.257$, $M = 2$, $T = 2.6$ s

How has hypoxia affected the steady-state gain and time constant of the underlying continuous-time model for the CO_2 ventilatory response for this particular subject?

P7.3. Consider the following discrete-time linear system with transfer function (z-domain) as given below:

$$H(z) = \frac{z}{z - 0.5}$$

(a) Derive the corresponding finite difference equation for this system that will enable you to determine how the output $y(n)$ would respond to the input $x(n)$ in the (discrete) time domain.

(b) Sketch as accurately as possible the response of the above system to a unit impulse.

(c) The values tabulated below represent the output of the system to an unknown input signal. Assume both output and input were sampled at 1 Hz. Determine the corresponding values of the input signal at the times displayed in the table below:

Time, t (s)	0	1	2	3	4	5	6	7
$y(t)$	−4	3	8	−2	6	−7	−5	1

P7.4. Combine the IPFM model, implemented in SIMULINK as `IPFM.slx`, with the model of circulatory control introduced in Section 5.5.1 (`rsa.slx`), so that the extended "RSA" model will generate simulated "R-waves," similar to the ECG spikes that accompany each heart beat. Then, using the successive intervals between adjacent R-waves, produce plots of heart period variability similar to those displayed in Figure 5.15. Generate such plots for the "normal," "+atropine," and "+propranolol" conditions. By resampling these R-to-R interval time series with a uniform sampling interval of 0.5 s and applying `rsa_tf.m` to the resulting time series, determine if the corresponding frequency responses are similar to the plots displayed in Figure 5.16.

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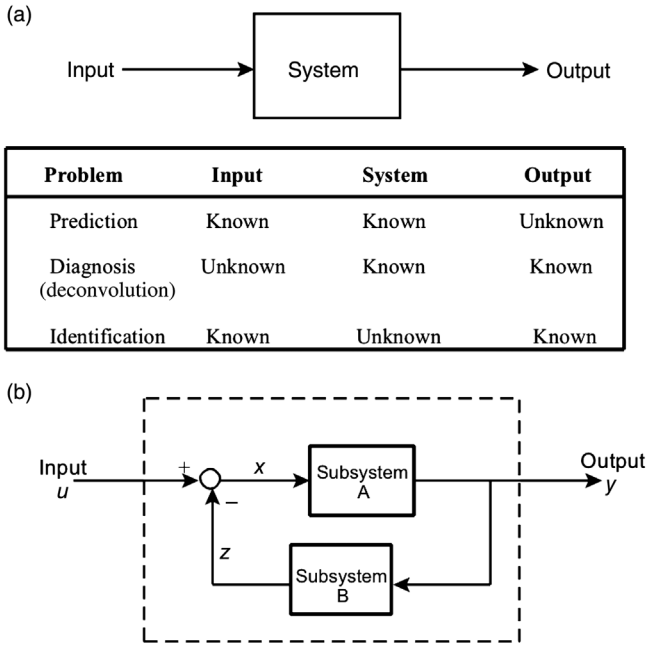Oppenheim, A.V., and A.S. Willsky. *Signals and Systems*, 2nd edition, Pearson, New York, 1996.

8

MODEL IDENTIFICATION AND PARAMETER ESTIMATION

8.1 BASIC PROBLEMS IN PHYSIOLOGICAL SYSTEM ANALYSIS

In the past several chapters, we have examined a variety of techniques for analyzing the steady-state and dynamic characteristics of feedback control systems. A common thread among all these different methods has been the use of the *systems approach*: The physiological process under study is decomposed into a number of interconnected "systems" or "subsystems," under the assumption that each of these components can be characterized functionally by a set of differential equations or their Laplace equivalents. This is displayed schematically in Figure 8.1a. Up to this point, we have always assumed that the equations in each of the "boxes" are known or can be derived by applying physical principles and physiological insight to the process in question. Thus, knowing the form of the independent variable (the "input"), the equations (representing the "system") can be solved to deduce the form of the dependent variable (the "output"). This type of analysis is known as the *forward problem* or *prediction problem*. Predictions allow us to determine if the model postulated provides an accurate characterization of the process under study. A somewhat greater challenge is posed by the *inverse problem*. Here, a model of the process in question is available and the output is measured; however, the input is not observable and therefore has to be deduced. This is known as the *diagnosis problem* and often involves the need for *deconvolution* of the model impulse response with the output in order to deduce the input.

(a)

Problem	Input	System	Output
Prediction	Known	Known	Unknown
Diagnosis (deconvolution)	Unknown	Known	Known
Identification	Known	Unknown	Known

(b)

FIGURE 8.1   (a) The three fundamental problems in system analysis. (b) Identification of a closed-loop system.

The third type of problem, that of *system identification*, is the most pervasive in physiological system analysis. There are two basic approaches to system identification. It is often the case, when dealing with physiological processes, that we have only very limited insight into the underlying mechanisms or that the complexity of the processes involved is just too overwhelming. Under such conditions, it would be difficult to begin with physical principles to derive the differential equations that appropriately characterize the system under study. At this level of knowledge (or lack of it), it would probably be more useful to probe the system in question with known stimuli and to record the system's response to these inputs. This is the *black-box* or *nonparametric* approach to system identification, where little is assumed about the system except, perhaps, whether we expect it to be linear or nonlinear. Ideally, we would deduce from the measured input and output the system impulse response, if it is linear, or kernels if it is nonlinear (see Chapter 10), that subsequently are used to catalog the behavior of the unknown process. The result would be a purely *empirical* model of the system under study.

In the case of systems for which *some* knowledge regarding mechanisms is available, it is generally possible to put this knowledge to use by coming up with a mathematical description (which could consist of a set of differential or difference equations or their frequency-domain equivalents). This characterizes the second

approach to system identification, in which a *structural* or *gray box* model is constructed. While the "structure" of such a model is derived from what we know about the physiological process being studied, there remain unknown model coefficients or *parameters* that have to be determined. Thus, the next stage in system identification in such cases, following model building, is the problem of *parameter estimation*. For this reason, structural models fall into the category referred to as *parametric models*. In the case of the linear lung mechanics model that we analyzed in Chapters 4 through 6, the unknown parameters (R, L, and C) each bear a one-to-one correspondence to a physiological entity – airway resistance, fluid inertance, and lung compliance. But this is not a requirement of all parametric models. *Functional* or *minimal models* are models that contain only parameters that can be estimated from input–output data. Frequently, some of these parameters may be related to the underlying physiological entities, but a one-to-one correspondence may not exist. In many models of pharmacokinetics, for example, there are often assumed "compartments" that may be used to account for effects arising from many different sources but not one single definable physiological entity. In some other functional models, a negative delay may have to be postulated; such a parameter clearly has no physiological meaning but may be needed in order to fully characterize the observed system behavior.

The control engineering literature is replete with countless methods of system identification, particularly for linear systems. In this chapter, we will discuss the few basic techniques that have been most commonly applied in physiology system analysis. While there is a large body of literature on the theory of system identification in simple single-input single-output systems, there has been relatively much less work published on the identification of closed-loop systems. The fact that most physiological systems are closed-loop systems can introduce some complications into the process of system identification or parameter estimation. Referring to the example illustrated in Figure 8.1b, if we could only measure the input (u) and the output (y) of the overall closed-loop system, but the internal variables x and z were unobservable, it would be impossible for us to know (just based on the measurements of u and y) that this is in fact a closed-loop system. On the other hand, if we could measure x in addition to u and y, then in principle, we would be able to identify subsystem A; then, having identified the overall closed-loop system, we would be able to determine subsystem B. Similarly, if we could measure the feedback variable z, it would be possible (at least in theory) to determine subsystem B; then, from knowledge of the overall closed-loop model, we would be able to deduce subsystem A. This is the fundamental basis of *closed-loop estimation*. In some cases, it may be possible to "open the loop" through surgical, physiological, or pharmacological interventions. Indeed, some of the biggest advances in physiology have resulted from clever experimental designs that allowed the researchers to "open the loop" in one or more of these ways. We will review some examples of these in the sections to follow.

8.2 NONPARAMETRIC AND PARAMETRIC IDENTIFICATION METHODS

We begin by reviewing some of the basic computational techniques commonly employed in the identification of single-input, single-output, open-loop systems. While most physiological models have been developed assuming a continuous-time base, in practice physiological measurements are generally obtained as *discrete-time* samples of the signals under study. Some of these time series contain measurements obtained at a fixed sampling rate, for example, arterial blood pressure and the electroencephalograph. There are also measurements that involve pulsatile or cyclic quantities, for example, beat-to-beat heart period and breath-by-breath ventilation, which provide information on a per cycle basis. These measurements have an integer time base – in beats, cycles, or breaths, for example; but they do not give information that is uniformly sampled in absolute time, because the cycle periods vary with each cycle. Since the process of system identification requires the use of real data, the vast majority of identification techniques that have been developed assume a discrete-time base. Thus, for the most part, system identification problems are solved by numerical methods and do not have closed-form analytical solutions. An important assumption that we will make, however, is that the sampling interval has been selected to be small enough so that the time series obtained adequately capture the fastest dynamics present in the observed signals. As demonstrated in Chapter 7, sampling the input and output signals at rates that are lower than one-half of the highest frequency present in the signals can lead to the problem of aliasing, in which the sampled data may appear to contain dynamic components that were really not contained in the original signals. Use of these aliased input and output time series would definitely lead to erroneous estimates of the system impulse responses or transfer functions.

8.2.1 Numerical Deconvolution

The most direct nonparametric techniques for linear system identification have been discussed earlier in Chapter 4. The response to the step input has been one of the most commonly used methods for characterizing physiological system dynamics, provided the stimulus can indeed be made to follow a time-course that closely approximates a step. Having found the step response, the impulse response can be deduced by differentiating the former with respect to time. In general, impulsive inputs cannot be easily implemented in physiological applications. If the step input is also not a convenient option, one might resort to stimulating the system under study with a bolus type of input. Then, in order to estimate the impulse response from the bolus response and the input, one can employ the method of numerical deconvolution.

Assuming that the data samples are obtained at uniformly spaced time intervals of T s, the convolution integral for continuous-time linear time-invariant systems

(Equation 2.56) can be converted into its equivalent in discrete-time as follows:

$$y(n) = \sum_{k=0}^{n} h(n-k)u(k)T \tag{8.1}$$

where $u(n)$ represents the input, $y(n)$ the output, and $h(n)$ the (discrete-time) impulse response at time $t = nT$. *In order to make the magnitude of $h(n)$ consistent with $h(t)$, the sampling interval T is introduced to scale the convolution sum accordingly on the right-hand side of Equation 8.1.* In the special cases where $n = 0$, 1, and 2, Equation 8.1 becomes

$$y(0) = h(0)u(0)T \tag{8.1a}$$

$$y(1) = [h(1)u(0) + h(0)u(1)]T \tag{8.1b}$$

$$y(2) = [h(2)u(0) + h(1)u(1) + h(0)u(2)]T \tag{8.1c}$$

Thus, from Equation 8.1a, assuming $u(0) \neq 0$, we find that

$$h(0) = \frac{y(0)/T}{u(0)} \tag{8.2a}$$

Similarly, rearranging Equation 8.1b, we obtain

$$h(1) = \frac{([y(1)/T] - h(0)u(1))}{u0} \tag{8.2b}$$

so that once $h(0)$ has been deduced from Equation 8.2a, the next point in the impulse response function, $h(1)$, can be determined from Equation 8.2b. Subsequently, from Equation 8.1c, we get

$$h(2) = \frac{([y(2)/T] - h(1)u(1) - h(0)u(2))}{u0} \tag{8.2c}$$

where $h(2)$ can be determined, since $h(0)$ and $h(1)$ are now known. This estimation procedure is continued for all subsequent values of $h(n)$. Thus, the general deconvolution formula is

$$h(n) = \frac{\left([y(n)/T] - \sum_{k=1}^{n} h(n-k)u(k)\right)}{u0} \tag{8.2d}$$

While Equation 8.2d is valid in principle, in practice it is hardly used. The reason is that small values for $u(0)$ can amplify errors enormously, and errors made in each sequential estimate of h tend to accumulate.

8.2.2   Least-Squares Estimation

A key problem of numerical deconvolution is that the estimated impulse response function is "forced" to satisfy Equation 8.1 even when it is clear that the output measurements y will contain noise. The effect of this noise accumulates with each step in the deconvolution process. One way to get around this problem is to build some averaging into the estimation procedure.

To do this, we restate the problem in the following way: Given N pairs of input–output measurements, estimate the impulse response function (consisting of p points, where $p \ll N$) that would allow Equation 8.1 to be satisfied *on average*. To develop this mathematically, we recast the relationship between input and output measurements in the following form:

$$y(n) = \sum_{k=0}^{p-1} h(k)u(n-k)T + e(n), \quad n = 0, 1, \ldots, N-1 \tag{8.3}$$

where $e(n)$ represents the error between the measured (noisy) value and the "best estimate" of the response at time $t = nT$. The "best estimate" of the response is obtained by selecting the impulse response function $\{h(k), k = 0, 1, \ldots, p-1\}$ that would minimize the sum of the squares of all the errors, $\{e(n), n = 0, 1, \ldots, N-1\}$. This method is analogous to the fitting of a straight line to a given set of data points, except that the "line" in this case is a multidimensional surface.

To find the least-squares estimate of $h(n)$, we proceed by defining the following matrix and vector quantities:

$$\underline{y} = \begin{bmatrix} y(0) & y(1) & \cdots & y(N-1) \end{bmatrix}' \tag{8.4}$$

$$\underline{h} = \begin{bmatrix} h(0) & h(1) & \cdots & h(p-1) \end{bmatrix}' \tag{8.5}$$

$$\underline{e} = \begin{bmatrix} e(0) & e(1) & \cdots & e(N-1) \end{bmatrix}' \tag{8.6}$$

and

$$U = \begin{bmatrix} u(0) & 0 & \cdots & 0 \\ u(1) & u(0) & \cdots & 0 \\ \vdots & \vdots & \vdots & \vdots \\ \vdots & \vdots & \vdots & \vdots \\ u(N-1) & u(N-2) & \cdots & u(N-p) \end{bmatrix} T \tag{8.7}$$

In Equations 8.4 through 8.6, $\underline{y}$, $\underline{h}$, and $\underline{e}$ are column vectors (and the superscript $'$ represents the transpose operation). Then, the system of equations represented by Equation 8.3 can be compactly rewritten in matrix notation as

$$\underline{y} = U\underline{h} + \underline{e} \tag{8.8}$$

Let J represent the sum of squares of the errors. Then,

$$J = \sum_{n=0}^{N-1} e(n)^2 = \underline{e}'\underline{e} \tag{8.9}$$

Combining Equations 8.8 and 8.9, we get

$$J = \left(\underline{y} - \mathbf{U}\underline{h}\right)'\left(\underline{y} - \mathbf{U}\underline{h}\right) \tag{8.10}$$

To find the minimum J, we differentiate Equation 8.10 with respect to the vector $\underline{h}$, and equate all elements in the resulting vector to zero:

$$\frac{\partial J}{\partial \underline{h}} = -2\mathbf{U}'\underline{y} + 2\mathbf{U}'\mathbf{U}\underline{h} = \underline{0} \tag{8.11}$$

Rearranging Equation 8.11, we find that the least-squares solution for the impulse response function is

$$\underline{h} = (\mathbf{U}'\mathbf{U})^{-1}\mathbf{U}'\underline{y} \tag{8.12}$$

It can be shown further that a lower bound to the estimate of the variance associated with the estimated elements of $\underline{h}$ is given by

$$\mathrm{var}(\underline{h}) = \mathrm{diag}\left((\mathbf{U}'\mathbf{U})^{-1}\right)\sigma_{\mathrm{e}}^2 \tag{8.13}$$

where σ_{e}^2 is the variance of the residual errors $\{e(n), n = 0, 1, \ldots, N-1\}$, that is,

$$\sigma_{\mathrm{e}}^2 = \frac{1}{N-1} \sum_{n=0}^{N-1} e(n)^2 \tag{8.14}$$

This method produces much better results for $h(n)$ than numerical deconvolution, since we are using N pieces of information to deduce estimates of p unknowns, where p should be substantially smaller than N. How small the ratio p/N should be depends on the relative magnitude of noise in the data. As a rough rule of thumb, p/N should be smaller than 1/3. Another requirement for obtaining good estimates of $h(n)$ is that the matrix $\mathbf{U}'\mathbf{U}$ must not be ill-conditioned since it has to be inverted: As one can see from Equation 8.13, the variance of $h(n)$ becomes infinite if $\mathbf{U}'\mathbf{U}$ is singular. Since $\mathbf{U}$ consists of all the input measurements, the conditioning of $\mathbf{U}'\mathbf{U}$ depends on the time-course of the stimulus sequence. This problem will be discussed further in Section 8.3.

The practical implementation of this method is relatively straightforward in MATLAB. An example of the MATLAB code that can be used to estimate the

elements of vector **h** and their associated standard errors (in column vector hse) is given in the script file sysid_ls.m. The main portion of this code is displayed below:

```
>> % Construct observation matrix UU
>> UU = zeros(N,p);
>> for i=1:p,
>>    if i==1
>>       UU(:,1)=u;
>>    else
>>       UU(:,i)= [zeros(i-1,1)' u(1:N-i+1)']';
>>    end
>> end;
>> UU = T*UU;
>>
>> % Construct autocorrelation matrix
>> AA = UU'*UU;
>> b = UU'*y;
>>
>> % Compute estimate of h
>> h = AA\b;
>>
>> % Compute estimated standard errors of h, hse
>> e = y - U*h;
>> sigma = std(e);
>> AAinv = inv(AA);
>> hse = zeros(size(h));
>> for i=1:p,
>>    hse(i) = sqrt(AAinv(i,i))*sigma;
>> end
```

The above code assumes that the input and output data are contained in the N-element column vectors u and y, respectively. As an illustration of how one can apply the estimation algorithm, we use sample input and output "data" generated by the linear lung mechanics model discussed in Chapters 4 and 5. The following parameter values are assumed: $L = 0.01$ cm H_2O $s^2 L^{-1}$, $R = 1$ cm H_2O s L^{-1}, and $C = 0.1$ L cm H_2O^{-1}. The input in this case is a unit step in P_{ao}, beginning at time $t = 0$. The output is the model response in P_A. Gaussian white noise is added to the output to simulate the effects of measurement noise. The simulated input and output measurements are displayed in Figure 8.2a; these time series are also contained in the file labeled data_llm.mat. The estimated impulse response is shown in Figure 8.2b, along with upper and lower bounds that reflect the estimates plus and minus 1 standard error. Superimposed on the estimates is the "true" impulse response, which appears as the smooth curve. The fluctuations in the estimated impulse response illustrates how sensitive it is to measurement noise, since in this case the p/N ratio of $\sim 1/2$ was large. Note that, as in Section 8.2.1, multiplying the
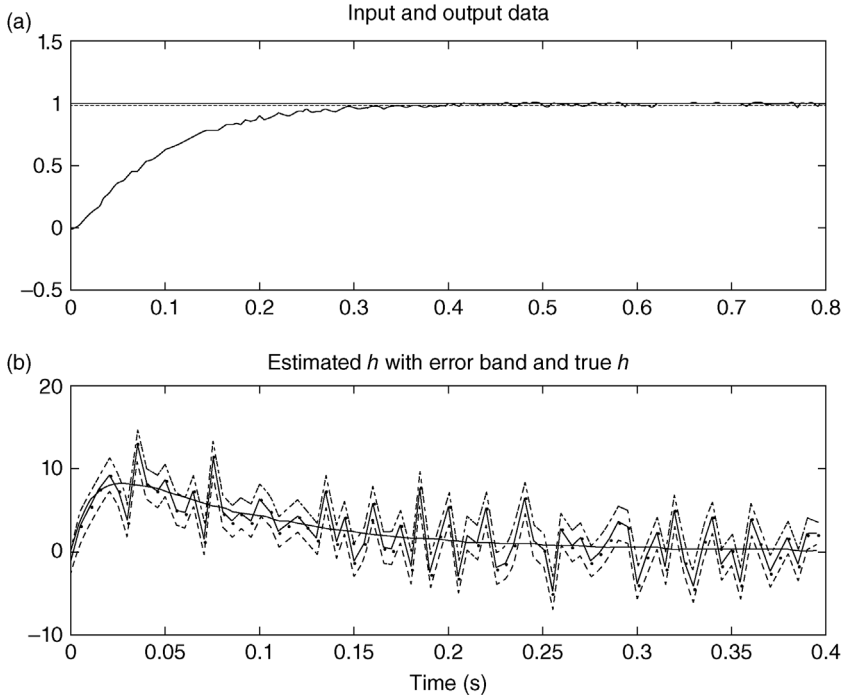
FIGURE 8.2 (a) Step change in P_{ao} and resulting response in P_{A} (with noise added). (b) Estimated impulse response with error bounds superimposed on "true" impulse response (smooth curve). Horizontal axes represent time in seconds.

input time series by the sampling interval T (see Equations 8.3 and 8.7) introduces the correct scaling so that $h(n)$ can be compared directly with $h(t)$.

Although we have confined the application of this identification method to linear systems here, in Chapter 10 we show that Equation 8.3 can be readily extended to take the form of a *Volterra* series, which also contains *nonlinear* dependencies of $y(n)$ on $u(n)$. This formulation allows us to estimate the parameters that characterize the dynamics of a certain class of open-loop nonlinear systems.

8.2.3 Estimation Using Correlation Functions

Starting with Equation 8.3, for any $m \geq 0$, if we multiply both sides of the equation by $u(n-m)$, sum up all $N-m$ nonzero terms, and then divide through by $N-m$, we will obtain

$$\hat{R}_{uy}(m) = \sum_{k=0}^{p-1} h(k)\hat{R}_{uu}(m-k)T + \hat{R}_{ue}(m) \tag{8.15a}$$

where

$$\hat{R}_{uy}(m) = \frac{1}{N-m} \sum_{n=m}^{N-1} u(n-m)y(n) \tag{8.16}$$

$$\hat{R}_{uu}(m-k) = \frac{1}{N-m} \sum_{n=m}^{N-1} u(n-m)u(n-k) \tag{8.17}$$

and

$$\hat{R}_{ue}(m) = \frac{1}{N-m} \sum_{n=m}^{N-1} u(n-m)e(n) \tag{8.18}$$

Note that $\hat{R}_{uy}$, $\hat{R}_{uu}$, and $\hat{R}_{ue}$ in Equations 8.16 through 8.18 represent ensemble averages of the lagged products between u and y, u and itself, and u and e. Assuming that the stochastic processes that underlie u and y are *ergodic*, these ensemble averages provide reasonably good estimates of the cross-correlation between u and y, the autocorrelation in u, and the cross-correlation between u and e, respectively (see Section 5.3.2). We select that solution of h such that $\hat{R}_{ue}(m)$ becomes zero for all values of m – thus, Equation 8.15a becomes

$$\hat{R}_{uy}(m) = \sum_{k=0}^{p-1} h(k)\hat{R}_{uu}(m-k)T, \quad m = 0, 1, \ldots, p-1 \tag{8.15b}$$

Equation 8.15b may be considered the discrete-time version of Equation 5.27b, and may be solved by applying a little matrix algebra, as in Section 8.2.2. We define the following vector and matrix quantities:

$$\underline{\hat{\mathbf{R}}}_{uy} = \begin{bmatrix} \hat{R}_{uy}(0) & \hat{R}_{uy}(1) & \cdots & \hat{R}_{uy}(p-1) \end{bmatrix}' \tag{8.19}$$

and

$$\underline{\hat{\mathbf{R}}}_{uu} = \begin{bmatrix} \hat{R}_{uu}(0) & \hat{R}_{uu}(1) & \cdots & \hat{R}_{uu}(p-1) \\ \hat{R}_{uu}(1) & \hat{R}_{uu}(0) & \cdots & \hat{R}_{uu}(p-2) \\ \vdots & \vdots & \vdots & \vdots \\ \vdots & \vdots & \vdots & \vdots \\ \hat{R}_{uu}(p-1) & \hat{R}_{uu}(p-2) & \cdots & \hat{R}_{uu}(0) \end{bmatrix} \tag{8.20}$$

Note from Equation 8.17 that $\hat{R}_{uu}(m-k) = \hat{R}_{uu}(k-m)$, and therefore the matrix $\underline{\hat{\mathbf{R}}}_{uu}$ in Equation 8.20 is symmetric. Equation 8.15b can be written in

matrix form as

$$\underline{\hat{\mathbf{R}}}_{uy} = T\hat{\mathbf{R}}_{uu}\underline{\mathbf{h}} \tag{8.21}$$

where $\underline{\mathbf{h}}$ is defined by Equation 8.5. Since all elements of $\hat{\mathbf{R}}_{uy}$ and $\hat{\mathbf{R}}_{uu}$ can be computed from the input and output data by applying Equations 8.16 and 8.17, the unknown impulse response function is determined through the solution of Equation 8.21 by inverting $\hat{\mathbf{R}}_{uu}$:

$$\underline{\mathbf{h}} = \frac{1}{T}\hat{\mathbf{R}}_{uu}^{-1}\underline{\hat{\mathbf{R}}}_{uy} \tag{8.22}$$

It can be shown that, aside from possible differences in the details of computing the autocorrelation and cross-correlation functions, Equation 8.22 is essentially equivalent to Equation 8.12. As before, the feasibility of applying this approach depends on the invertibility of the autocorrelation matrix $\hat{\mathbf{R}}_{uu}$.

8.2.4   Estimation in the Frequency Domain

Since the Laplace transform of the impulse response is the system transfer function, carrying out the system identification process in the frequency domain should, in principle, yield the same results as any of the time-domain methods discussed earlier. The problem of transfer function identification is actually the same as that of estimating the frequency response. The underlying idea is very simple and is illustrated in Figure 8.3. At each frequency in the range of interest, apply a sinusoidal input of known amplitude and phase to the system under study; then, measure the resulting output. If the system is linear, the measured output also will be a sinusoid of the same frequency. The ratio between the magnitude of the output and the magnitude of the input ($= A_{\text{out}}/A_{\text{in}}$ in Figure 8.3) would yield the system *gain* at that frequency. The phase difference between the output and input waveforms ($= 2\pi T_{\text{p}}/T_{\text{c}}$) would be the system *phase* at that frequency. By repeating this measurement over all frequencies of interest, one would be able to arrive at the frequency response of the system and therefore obtain an estimate of the transfer function.

Although the above method can provide very good estimates of the system transfer function at the frequencies investigated, one major drawback is that the entire identification procedure can be extremely time-consuming and therefore impractical for application in human or animal studies. An alternative would be to employ the spectral analysis technique presented in Section 5.4.1; the basic idea here is that the frequency response is estimated from the ratio between the input–output cross-spectrum (S_{uy}) and the input power spectrum (S_{uu}), that is,

$$H(\omega_k) = \frac{S_{uy}(\omega_k)}{S_{uu}(\omega_k)} \tag{8.23}$$
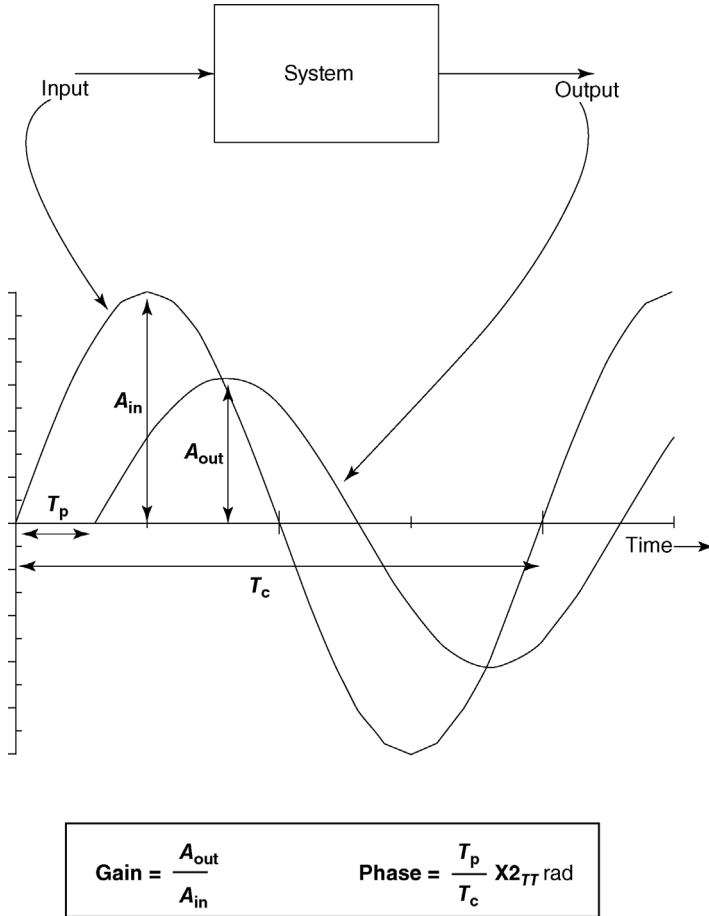
FIGURE 8.3 Illustration of transfer function identification using sinusoidal inputs. System gain and phase at given frequency $\omega = 2\pi/T_c$ are as defined above.

where $\omega_k = 2\pi k/pT$ $\{k = 0, 1, \ldots, p-1\}$, and the spectral quantities are defined in the following way:

$$S_{uy}(\omega_k) = \sum_{m=0}^{p-1} \hat{R}_{uy}(m)e^{-j\omega_k mT}, \quad k = 0, 1, \ldots, p-1 \tag{8.24}$$

$$S_{uu}(\omega_k) = \sum_{m=0}^{p-1} \hat{R}_{uu}(m)e^{-j\omega_k mT}, \quad k = 0, 1, \ldots, p-1 \tag{8.25}$$

As shown in Equations 8.24 and 8.25, $S_{uy}(\omega_k)$ and $S_{uu}(\omega_k)$ are computed by applying the discrete Fourier transform to $\hat{R}_{uy}$ and $\hat{R}_{uu}$, respectively. Equations 8.23

through 8.25 are the discrete frequency equivalents of Equations 5.29 and 5.30, which were applied to correlation quantities based on continuous time. As mentioned in Section 5.3.2, all values of S_{uu} must be positive in order for meaningful estimates of the transfer function to be obtained. Whether this condition is attained depends on the form of the input sequence, as we will see in Section 8.3.

Once $H(\omega)$ has been estimated, it is possible to interpret the results in the context of physiologically meaningful entities if a parametric model is available. As an example of how this can be done, consider the linear lung mechanics model that was discussed in Chapters 4 and 5. From Equation 5.6, the frequency response predicted from this model takes the following form:

$$H_{\text{model}}(\omega) = \frac{1}{(1 - LC\omega^2) + jRC\omega} \tag{8.26}$$

The values of the lung mechanical parameters (R, L, and C) that most closely correspond to the measured frequency response $H_{\text{meas}}(\omega)$ can be estimated by first defining a "criterion function" J, which represents the "distance" between H_{model} and H_{meas} and, second, by searching for the parameter values that minimizes this distance. Since H_{model} and H_{meas} are complex-valued functions of ω, a suitable criterion function could be

$$J = \sum_{k=0}^{p-1} \left((\text{Re}\,[H_{\text{meas}}(\omega_k)] - \text{Re}\,[H_{\text{model}}(\omega_k)])^2 + (\text{Im}\,[H_{\text{meas}}(\omega_k)] - \text{Im}\,[H_{\text{model}}(\omega_k)])^2 \right)$$

$$\tag{8.27}$$

The above expression assumes that frequency response measurements are available at the frequencies ω_k, where $k = 0, 1, \ldots, p - 1$. The methodology for minimizing J is described in the next section.

8.2.5   Optimization Techniques

As we had mentioned at the beginning of this chapter, the identification of "gray box" or "parametric" models consists of two stages. First, the model structure has to be developed, consistent with prior knowledge about the physiological system in question. Frequently, this takes the form of a set of differential equations. Once the model has been formulated, the next task is to estimate the unknown model parameters by minimizing (or maximizing) some criterion that reflects the goodness of fit between the model predictions and the observed output measurements. When dealing with models represented by differential equations of high order, we mentioned in Section 2.8 that it is generally better, from the viewpoint of numerical stability, to employ a *state-space* framework. Another advantage of employing a state-space model is that the analysis can readily be extended to *nonlinear* systems. We will now illustrate how this system identification technique works by considering

our favorite example of the linear lung mechanics model. A more advanced example, involving the analysis of a nonlinear model, is given in Section 8.5.1.

8.2.5.1 *State-Space Model Formulation*   The differential equation characterizing the lung mechanics model was derived in Section 4.1 and is given by

$$LC\frac{d^2 P_A}{dt^2} + RC\frac{dP_A}{dt} + P_A = P_{ao} \tag{8.28}$$

Since P_{ao} is the input and P_A is the output of this system, we make the new variable assignments:

$$y_1 = P_A \quad \text{and} \quad u = P_{ao} \tag{8.29a,b}$$

Also, assume

$$y_2 = \frac{dP_A}{dt} = \frac{dy_1}{dt} \tag{8.30}$$

Then, we can rewrite Equation 8.28 as

$$LC\frac{dy_2}{dt} + RCy_2 + y_1 = u \tag{8.31}$$

Then, using Equations 8.30 and 8.31, rearranging terms, and writing the two equations in matrix form, we obtain

$$\frac{d}{dt}\begin{bmatrix} y_1 \\ y_2 \end{bmatrix} = \begin{bmatrix} 0 & 1 \\ -\dfrac{1}{LC} & -\dfrac{R}{L} \end{bmatrix} \begin{bmatrix} y_1 \\ y_2 \end{bmatrix} + \begin{bmatrix} 0 \\ \dfrac{1}{LC} \end{bmatrix} u \tag{8.32a}$$

If we define

$$\underline{\mathbf{y}} \equiv \begin{bmatrix} y_1 \\ y_2 \end{bmatrix} \tag{8.33}$$

$$\mathbf{A} = \begin{bmatrix} 0 & 1 \\ -\dfrac{1}{LC} & -\dfrac{R}{L} \end{bmatrix} \tag{8.34}$$

and

$$\mathbf{B} = \begin{bmatrix} 0 \\ \dfrac{1}{LC} \end{bmatrix} \tag{8.35}$$

Equation 8.32a becomes

$$\frac{d}{dt}\underline{y} = \mathbf{A}\underline{y} + \mathbf{B}u \tag{8.32b}$$

Thus, we have converted the second-order scalar differential equation (Equation 8.31) into the equivalent first-order matrix state equation. This type of equation can be conveniently solved by numerical integration using one of the MATLAB ordinary differential equation solver functions: `ode45`, `ode23`, `ode113`, `ode15s`, and `ode23s`. An even easier way is to construct a state-space representation of the model within MATLAB using the `ss` function, and then use the `lsim` function (see Section 4.6) to generate the model response to a given input waveform $u(t)$. The MATLAB script file (provided as the file `sss_llm.m`) that performs these tasks is displayed below:

```
>> A = [0   1; -1/L/C  -R/L];
>> B = [0   1/L/C]';
>> t = [0:0.005:0.8]';
>> u = ones(size(t));

% Construct the system using state-space formulation
>> Hs = ss(A,B,[1 0],0);

% Solve state space equation using lsim and plot results
>> y = lsim(Hs,u,t);
>> plot(t,u,t,y);
```

In the above MATLAB script file, note that we have assumed the following companion "observation equation":

$$\hat{y} = \mathbf{D}\underline{y} + Eu \tag{8.36}$$

where $\hat{y}$ represents the measured output of this system, that is, P_A. In this case, we are able to measure the state variable y_1 ($=P_A$) directly. Thus, here, we have

$$\mathbf{D} = \begin{bmatrix} 1 & 0 \end{bmatrix} \quad \text{and} \quad E = 0 \tag{8.37a,b}$$

This accounts for the last two items in the argument list of the function `ss` in the above MATLAB script. One other detail is that the matrices $\mathbf{A}$ and $\mathbf{B}$ do not need to be evaluated directly. If the transfer function of the model is available, then the MATLAB function `tf2ss` can be used to convert the system representation from transfer function format to state-space format:

```
>> [A,B,D,E]  = tf2ss(num,den);
```

8.2.5.2 Optimization Algorithm   Having constructed the model, the next step is to select the means by which the response of the model to a given input sequence can

be compared with the response of the physiological system to the same input. The comparison is made through the use of a *criterion function* that provides a measure of the *goodness of fit* between the two time series. There are many possible candidates for the criterion function, but the one most commonly employed is the sum of squares of the differences between the measured and predicted outputs:

$$J = \sum_{n=0}^{N-1} \left(y(n) - y_{\text{pred}}(n) \right)^2 = \sum_{n=0}^{N-1} e(n)^2 \qquad (8.38)$$

This is the same expression as that presented in Equation 8.9.

With the criterion function having been defined, the problem of parameter estimation becomes transformed into a problem of optimization, where the objective is to find the combination of parameter values that minimizes the criterion function. The entire scheme of parameter estimation is illustrated in the schematic block diagram shown in Figure 8.4. It should be noted that if the model selected provides an accurate representation of the dynamics of the real system, then the residual errors $\{e(n), n = 0, 1, \ldots, N-1\}$ should closely reflect the measurement noise affecting the output. On the other hand, if the selected model is largely "wrong" and does not provide an adequate description of the output dynamics, there will be a significant contribution from *structural errors* as well. This is one of the major drawbacks of opting to employ a "structural" model: Erroneous information about the dynamics of the underlying physiology can translate into large errors in
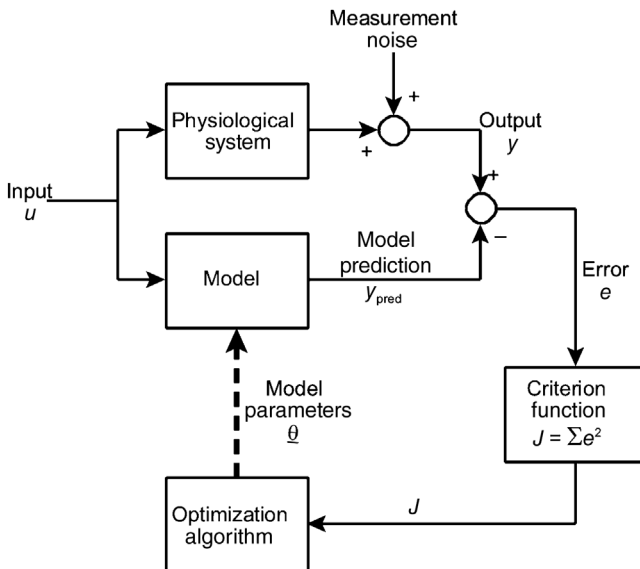
FIGURE 8.4   Schematic diagram of the optimization approach to parameter estimation.

the parameter estimates and/or an inability of the model predictions to "fit" the data well.

The choice of the algorithm employed to perform the minimization of the criterion function is also important. There is a large array of algorithms available, but it is not within the scope of this chapter to examine all or even most of these. The most commonly applied methods employ the *gradient descent* approach. These methods are best explained by considering a problem in which two parameters need to be estimated. We will refer to these two parameters as θ_1 and θ_2. Since J is a function of θ_1 and θ_2, evaluating J over selected ranges of θ_1 and θ_2 would yield a surface in a three-dimensional space with the Cartesian axes formed by θ_1, θ_2, and J. Suppose the surface looks like the contour map shown in Figure 8.5a, where each
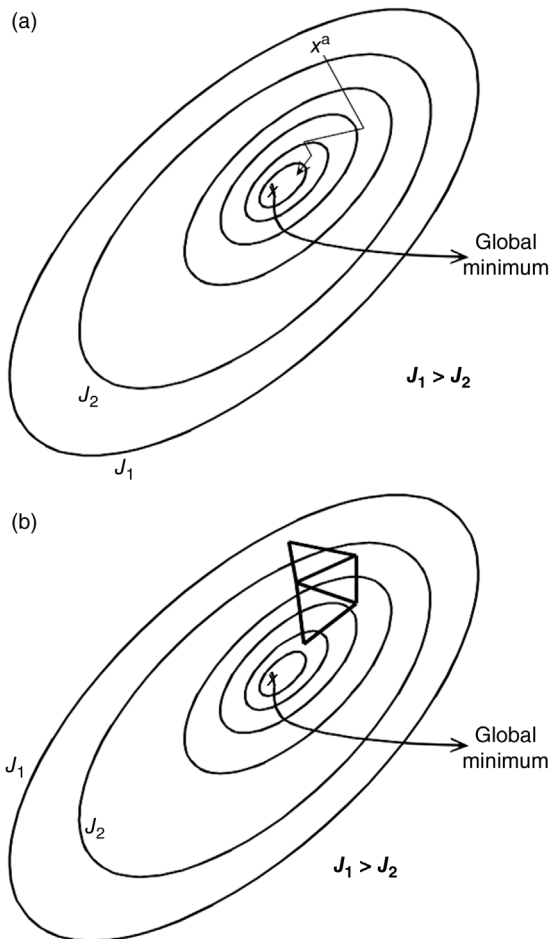
FIGURE 8.5 Methods for finding the minimum of the criterion function surface. (a) Steepest gradient method. (b) Simplex method.

contour corresponds to a uniform value of *J*. In the gradient descent approach, we start off with an initial guess of the parameters, represented as the point *a*. Then, information about the slope of the local terrain is obtained, and based on this information, we move down the slope along the direction of "steepest descent." The size of the step taken in this direction differs with the different gradient descent methods, with some methods using information about the curvature (i.e., second derivatives) of the surface as well.

One drawback of the gradient approach is the need to evaluate the derivatives of *J* with respect to all the parameters being estimated at every iteration step. Furthermore, these methods are generally quite susceptible to producing solutions that correspond to false local minima, if the *J*-surface is highly irregular, as would be the case when the signal-to-noise ratio is low. A popular alternative, which does not require any derivative computations at all, is the Nelder–Mead simplex algorithm. For a three-parameter problem, the *simplex* takes the form of a tetrahedron, while for the two-parameter problem, it is a triangle. Figure 8.5b shows the same criterion function surface discussed above, together with the simplex (triangle) and how the shape and position of the triangle moves over the course of a few iterations. The vertices of the triangle represent the three points on the *J*-surface that are known at any given iteration of the algorithm. Starting at the initial three points, the triangle is reflected over the two vertices with the lowest *J*-values and the height of the triangle is expanded or contracted so that the remaining vertex is located at the point of lowest *J*-value. Next, the triangle is reflected over the two of the three vertices that have the lowest *J*-values, and the new third vertex is found by stretching or shrinking the height of the triangle. This process is repeated until some tolerance for convergence toward the minimum is met.

The MATLAB function `fminsearch` employs the Nelder–Mead simplex algorithm to determine the minimum point of a given multidimensional function. The MATLAB script (contained in the m-file named `popt_llm.m`) presented below shows an example of how `fminsearch` can be used to estimate the values of the two unknown parameters in the state-space formulation of the linear lung mechanics model (Section 8.2.5.1).

```
>> global u y
>> theta_init(1)=input
('  Enter initial value of 1st parameter >>');
>> theta_init(2)=input
('  Enter initial value of 2nd parameter >>');

% Perform optimization to minimize the objective function J
% defined by the function "fn_llm"
>> [theta,options] = fminsearch('fn_llm',theta_init);
```

Two items are required as inputs to `fminsearch`. The first is a user-defined function that defines the model being employed and returns to `fminsearch` the value of the criterion function at each iteration in the optimization process. In our

particular example, we have named this function `fn_llm`. The second input is a column vector (`theta_init`) containing an initial guess of the parameters to be estimated. `fminsearch` produces two sets of outputs. The first set contains the estimated parameter vector (`theta`). The second set of outputs (`options`) contains information about the minimization process. For instance, the 10th element of `options` contains the total number of iterations or function calls in the optimization run. Besides, in this example, the data file `data_llm.mat` has to be loaded prior to running `popt_llm` so that the input and output data are present in the workspace as vectors `u` and `y`, respectively, for `fminsearch` to work on. Since the function `fn_llm` must also use these data, the `global` declaration is included in both `popt_llm` and `fn_llm` to make `u` and `y` universally accessible. The relevant portion of the MATLAB code for `fn_llm` is given below, with the complete listing given in the m-file `fn_llm.m`.

```
>> function J = fn_llm(theta)
>> global u y
>> A = [0  1; -theta(1)  -theta(2)];
>> B = [0  theta(1)]';
>> Hs = ss(A,B,[1 0],0);
>> ypred = lsim(Hs,u,t);
>> e = y - ypred;
>> J = sum(e.^2);
```

It should be noted from the script for `fn_llm` that the two parameters being estimated, θ_1 and θ_2, correspond to the lung mechanical parameters $1/LC$ and R/L, respectively (see Equations 8.34 and 8.35). Application of this algorithm to the simulated data given in `data_llm.mat` produces the estimated parameter values of 1006.3 and 100.8, which agree closely with the "true" values for $1/LC$ and R/L of 1000 and 100, respectively:

```
Final Parameter Values:
  1.0e+003 *
    1.0063
    0.1008
Total Number of Iterations:
  198
```

8.3 PROBLEMS IN PARAMETER ESTIMATION: IDENTIFIABILITY AND INPUT DESIGN

8.3.1 Structural Identifiability

The problem of structural identifiability is intimately coupled to the problem of model-building. In theory, if knowledge about the underlying physiology of the

system in question is available, it should be possible for us to translate this knowledge into a parametric model by applying the basic laws of physics and chemistry. The more we know about the system, the more details we would be able to add to the model. In general, a more detailed and complex model would be expected to account for a greater range of observations under a larger variety of conditions. However, the price that one has to pay for the increased model complexity is the emergence of more model *parameters*, the values of which have to be assumed or estimated. In the several models that we have discussed in previous chapters, we assumed the parameter values to be known. For example, in the linear lung mechanics model, we assumed values for R, L, and C that were considered "representative" of the population of subjects with normal lungs. This assumption ignores the fact that there is a considerable degree of variability in these lung mechanical parameters across subjects that one can consider "normal." On the other hand, we could choose to estimate the parameters in each individual subject. The problem of structural identifiability arises when the information that is required for the parameter estimation process is incomplete. This could be due to the inaccessibility of certain signals or the lack of dynamic content in the stimulus.

As an example, consider the differential equation (Equation 8.28) that character- izes the linear lung mechanics model. Here, there are three unknown parameters: R, L, and C. However, the mathematical structure of this model turns out to be such that the parameters only appear as paired combinations of one another: LC and RC. As a consequence, the dynamics of the model are determined by only two parameters (LC and RC) and not by the original three (R, L, and C). This fact again becomes evident when one looks at the state-space formulation of the model in Equation 8.32a. Here, only two independent parameters determine the solution (i.e., dynamics) for the vector $\underline{y}$, and these are $1/LC$ and R/L. Thus, it is clear that, using only measurements of P_A and P_{ao}, the linear lung mechanics model is not completely identifiable in terms of all three parameters – R, L, and C. We should stress that this assertion on identifiability (or rather, the lack of it) holds true regardless of whether noise is present or absent in the measurements. On the other hand, this model can become fully identifiable if an additional channel of mea- surement, such as airflow, were to become available. For instance, one could estimate C separately from static changes, for example, from the ratio of the change in lung volume resulting from an applied change in pressure. Then, by combining this additional piece of information with the two parameters that can be estimated from the step-response in P_A, we would be able to identify all three original model parameters.

8.3.2 Sensitivity Analysis

A model that has been found to be structurally identifiable may still turn out to be unidentifiable in practice, if the parameter estimation process is sufficiently degraded by the presence of measurement noise. Therefore, having arrived at a structurally identifiable model, the next test that we should subject the model to is

the determination of whether the parameters that need to be estimated are resolvable in the presence of noise. Since the parameter estimation process requires us to find the lowest point on the multidimensional surface of the criterion function, it follows that parameter identifiability depends heavily on the quality of the J-surface. Figure 8.6 illustrates this statement with the help of two hypothetical examples that assume the case involving only a single parameter (θ_1). In both cases, we also assume that the presence of measurement noise limits the resolvability of changes in J to a value ΔJ. In Figure 8.6a, there is a deep minimum. The error in the parameter estimate $(\Delta \theta_1)$ made in arriving at a solution that is located at a criterion function value ΔJ above the global minimum is small. On the other hand, in Figure 8.6b, the J-surface contains a very shallow minimum. In this case, the effect of the same amount of measurement noise is a much larger error in the parameter estimate. What distinguishes part (a) from part (b) in this example is the fact that in part (a), a given
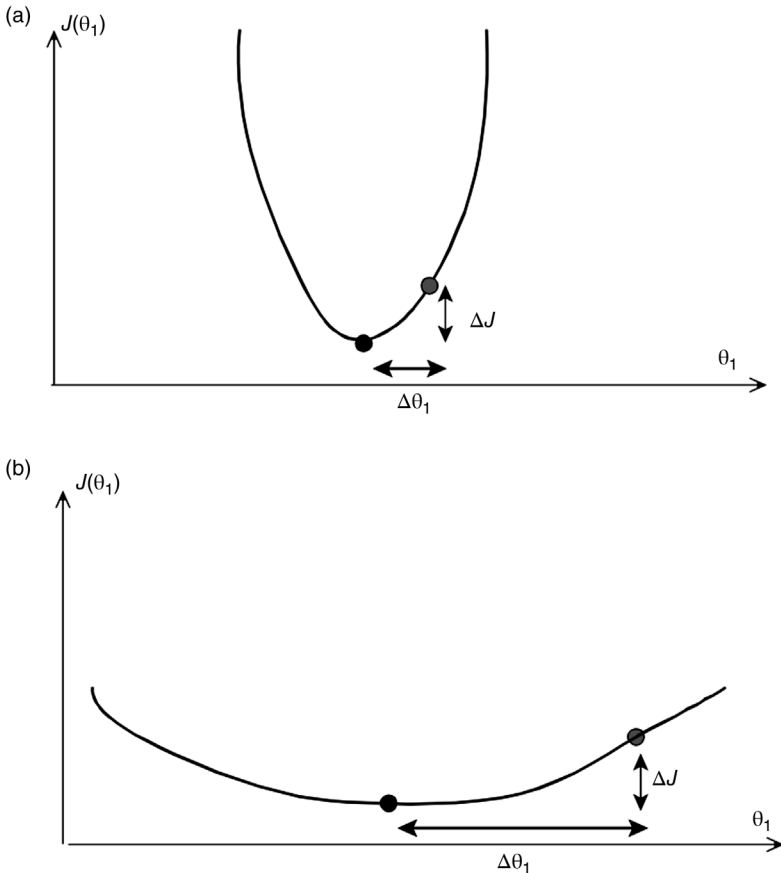
FIGURE 8.6   Relationship between sensitivity to parameter variations and parameter estimation error. (a) High sensitivity. (b) Low sensitivity.

change in the parameter value leads to a large (and, therefore, highly observable) change in the model output or J. Thus, in part (a), the model possesses *high sensitivity* to parameter variations, whereas in part (b) sensitivity is low.

The inverse relationship between sensitivity and parameter estimation error can be demonstrated analytically. We begin by recalling the definition of the criterion function (see Equation 8.38) but rewriting it in vector form:

$$\mathbf{J} = -\left(\underline{\mathbf{y}} - \underline{\mathbf{y}}_{\text{pred}}\right)' \left(\underline{\mathbf{y}} - \underline{\mathbf{y}}_{\text{pred}}\right) \tag{8.39}$$

Differentiating $\mathbf{J}$ with respect to the parameter vector $\underline{\boldsymbol{\theta}}$, we obtain

$$\frac{d\mathbf{J}}{d\underline{\theta}} = -\left(\underline{\mathbf{y}} - \underline{\mathbf{y}}_{\text{pred}}\right)' \frac{d\underline{\mathbf{y}}_{\text{pred}}}{d\underline{\theta}} \tag{8.40}$$

Note that the derivative on the right-hand side of Equation 8.40 is a $N \times p$ matrix, the elements of which represents the effect of a small change in each parameter on the model output. Thus, we can refer to this entity as the *sensitivity matrix* $\mathbf{S}$:

$$\mathbf{S} = \frac{d\underline{\mathbf{y}}_{\text{pred}}}{d\underline{\theta}} \tag{8.41}$$

Suppose $\underline{\boldsymbol{\theta}}^*$ represents the parameter vector at the global minimum point on the J-surface. Then, by applying a Taylor's series expansion and keeping *only first-order terms*, the model output in the vicinity of the minimum point can be expressed as

$$\underline{\mathbf{y}}_{\text{pred}}\left(\underline{\boldsymbol{\theta}}\right) = \underline{\mathbf{y}}_{\text{pred}}\left(\underline{\boldsymbol{\theta}}^*\right) + \frac{d\underline{\mathbf{y}}_{\text{pred}}}{d\underline{\theta}}\bigg|_{\underline{\boldsymbol{\theta}}^*} \left(\underline{\boldsymbol{\theta}} - \underline{\boldsymbol{\theta}}^*\right) = \underline{\mathbf{y}}_{\text{pred}}\left(\underline{\boldsymbol{\theta}}^*\right) + \mathbf{S}_{\underline{\boldsymbol{\theta}}^*}\left(\underline{\boldsymbol{\theta}} - \underline{\boldsymbol{\theta}}^*\right) \tag{8.42}$$

where $\mathbf{S}_{\underline{\boldsymbol{\theta}}^*}$ denotes the matrix $\mathbf{S}$ evaluated at the minimum point. At the minimum point, $\partial \mathbf{J}/\partial \underline{\boldsymbol{\theta}}$ in Equation 8.40 becomes a null vector. Thus, substituting Equation 8.42 into Equation 8.40, we obtain

$$-\frac{d\mathbf{J}}{d\underline{\boldsymbol{\theta}}} = 0 = \left(\underline{\mathbf{y}} - \underline{\mathbf{y}}_{\text{pred}}\left(\underline{\boldsymbol{\theta}}^*\right)\right)' \mathbf{S}_{\underline{\boldsymbol{\theta}}^*} - \left(\underline{\boldsymbol{\theta}} - \underline{\boldsymbol{\theta}}^*\right)' \mathbf{S}'_{\underline{\boldsymbol{\theta}}^*}\mathbf{S}_{\underline{\boldsymbol{\theta}}^*} \tag{8.43}$$

which can be rearranged to yield the following expression for the parameter estimate error:

$$\left(\underline{\boldsymbol{\theta}} - \underline{\boldsymbol{\theta}}^*\right)' = \underline{\mathbf{e}}^{*'} \mathbf{S}_{\underline{\boldsymbol{\theta}}^*} \left(\mathbf{S}_{\underline{\boldsymbol{\theta}}^*}' \mathbf{S}_{\underline{\boldsymbol{\theta}}^*}\right)^{-1} \tag{8.44}$$

where $\underline{e}^*$ represents the vector that contains the residual errors between the measurements and the predicted output values. Left-multiplying both sides of Equation 8.44 with $(\underline{\theta} - \underline{\theta}^*)$ and applying the expectation operator (see Equation 5.28) to both sides of the resulting equation, we get

$$\mathbf{P_{\underline{\theta}}} \equiv E\left[(\underline{\theta} - \underline{\theta}^*)(\underline{\theta} - \underline{\theta}^*)'\right] = \left(\mathbf{S}_{\underline{\theta}^*}{'}\mathbf{S}_{\underline{\theta}^*}\right)^{-1}\mathbf{S}_{\underline{\theta}^*}'\,E\left[\underline{e}^*\underline{e}^{*'}\right]\mathbf{S}_{\underline{\theta}^*}\left(\mathbf{S}_{\underline{\theta}^*}{'}\mathbf{S}_{\underline{\theta}^*}\right)^{-1} \quad (8.45)$$

where $\mathbf{P_{\underline{\theta}}}$ is also known as the *parameter error covariance matrix*. The diagonal elements of $\mathbf{P_{\underline{\theta}}}$ contain the variances of all p parameters in $\underline{\theta}$, whereas the off-diagonal elements represent the cross-covariances between the different paired combinations of the parameters. If we assume the sequence of residual errors to be white, that is, the present error is uncorrelated with past or future errors, then the matrix $\mathbf{E}[\underline{e}^*\underline{e}^{*'}]$ reduces to the identity matrix scaled by a factor equal to the variance σ^2 of the residual errors. Thus, Equation 8.45 simplifies to

$$\mathbf{P_{\underline{\theta}}} = \sigma^2\left(\mathbf{S}_{\underline{\theta}^*}{'}\mathbf{S}_{\underline{\theta}^*}\right)^{-1} \quad (8.46)$$

From Equation 8.46, it is important to note that each element of the symmetric matrix $\mathbf{S}'_{\underline{\theta}^*}\mathbf{S}_{\underline{\theta}^*}$ reflects the change in model output resulting from small changes in all possible pairings of the parameters. If changes in one or more of the parameters have no effect on the model output (zero sensitivity), then one or more columns and rows of $\mathbf{S}'_{\underline{\theta}^*}\mathbf{S}_{\underline{\theta}^*}$ will be zero; as a result, $\mathbf{S}'_{\underline{\theta}^*}\mathbf{S}_{\underline{\theta}^*}$ will be singular and the parameter errors will be infinite. This occurs when the model is not structurally identifiable. In structurally identifiable models, $\mathbf{S}'_{\underline{\theta}^*}\mathbf{S}_{\underline{\theta}^*}$ can still become close to singular if there are strong interdependencies between some of the parameters; in this case, there will be strong correlations between columns or rows of matrix $\mathbf{S}'_{\underline{\theta}^*}\mathbf{S}_{\underline{\theta}^*}$. Inversion of this close-to-singular matrix will yield variance and covariance values in $\mathbf{P_{\underline{\theta}}}$ that are unacceptably large. However, it is important to bear in mind from Equation 8.46 that, even under circumstances where model sensitivity is high, it is still possible for the parameters to be poorly estimated if the variance of the measurement noise (σ^2) is very large.

Equation 8.46 provides lower bound estimates of the variances and cross-covariances associated with the model parameters when these are estimated from noisy measurements. However, computation of the $\mathbf{P_{\underline{\theta}}}$ matrix is based on local changes in the vicinity of the optimal set of parameter values. A common alternative method of assessing model sensitivity is to base the calculations over a larger range of parameter value changes. In this approach, the criterion function J is evaluated over a selected span of values (say, $\pm 50\%$) for each parameter in turn, while holding the rest at their nominal values. Ideally, the "nominal" or "reference" values selected should correspond to the optimal set $\underline{\theta}^*$. The form of the criterion function J_θ is the same as that given in Equation 8.39, except that in this case the vector of observations $\underline{y}$ is replaced by $\underline{y}_{\text{ref}}$, where the latter represents the model predictions
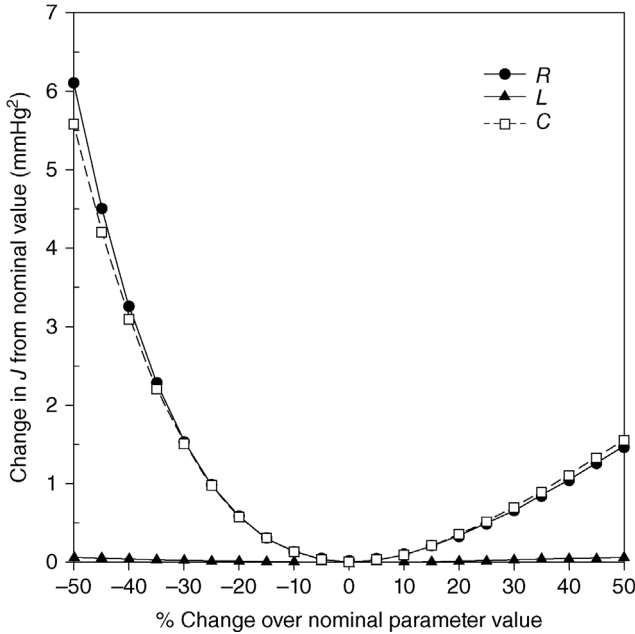
FIGURE 8.7 Sensitivity of the linear lung mechanics model to variations in the model parameters about their nominal values ($R = 1$, $L = 0.01$, $C = 0.1$). Flat curve for L (i.e., very low sensitivity) suggests identifiability problems for this parameter.

when the parameters are at their nominal values. Here, $\underline{y}_{pred}$ corresponds to the vector of model predictions at any of the parameter combinations ($\neq \underline{\theta}^*$) being evaluated. An example of this type of sensitivity analysis is shown in Figure 8.7 for the linear lung mechanics model. The nominal parameter set in this case is $R = 1$ cm H_2O s L^{-1}, $L = 0.01$ cm H_2O s L^{-2}, and $C = 0.1$ L cm H_2O^{-1}. The model is assumed to be perturbed by a unit step in P_{ao}. The plot for R, for instance, shows changes in J that would result if R were to be varied over the range 0.5–1.5 cm H_2O s L^{-1}, while L and C are kept at their nominal values. The model output is reasonably sensitive to changes in R and C, but virtually insensitive to changes in L. This kind of "flatness" in the sensitivity curve provides a good indication that at least one of the parameters will not be identifiable. This conclusion is consistent with our analysis of structural identifiability of this model in Section 8.3.1. The sensitivity results in Figure 8.7 were generated by the MATLAB script file `sensanl.m` (which also calls the function `fn_rlc.m`).

8.3.3   Input Design

The result represented by Equation 8.46 is valid for any general parametric model. Comparison of this result with Equation 8.13 shows a striking similarity between

the two equations. This similarity is by no means coincidental. In fact, Equation 8.13 represents a special case of Equation 8.46 when the assumed "model" is simply the impulse response of the system under study. To demonstrate this, note that the vector $\underline{\mathbf{h}}$ containing the sampled impulse response is the unknown parameter vector $\underline{\theta}$ that we would like to estimate. Thus, the model predictions are given by

$$y_{\text{pred}}(n) = \sum_{k=0}^{p-1} \theta_k u(n-k)T, \quad n = 0, 1, \ldots, N-1 \tag{8.47}$$

where

$$\theta_k = h(k) \tag{8.48}$$

The sensitivity of the nth output value to changes in the kth parameter is

$$S_{nk} \equiv \frac{\partial y_{\text{pred}}(n)}{\partial \theta_k} = u(n-k)T, \quad n = 0, 1, \ldots, N-1; \quad k = 0, 1, \ldots, p-1 \tag{8.49}$$

Looking back at Equation 8.7, it can be easily seen that the matrix $\mathbf{U}$ is simply a special case of the sensitivity matrix $\mathbf{S}$. Consequently, we have the following equality:

$$\mathbf{S}'_{\underline{\theta}^*}\mathbf{S}_{\underline{\theta}^*} = \mathbf{U}'\mathbf{U} \tag{8.50}$$

While it was not specifically mentioned in the previous section, it is clear from Equation 8.50 that the $p \times p$ matrix $\mathbf{S}'_{\underline{\theta}^*}\mathbf{S}_{\underline{\theta}^*}$ is a function of the input time-course. This implies that if the researcher has control over the type of stimulus that can be administered to the system in question, it should be possible to design the input waveform in such a way as to best "condition" the matrix $\mathbf{S}'_{\underline{\theta}^*}\mathbf{S}_{\underline{\theta}^*}$ so that the elements of its inverse can be minimized. From linear algebra, we know that a matrix is singular if there is linear dependence between any two or more of its columns (or rows). The best-case scenario for matrix inversion occurs when the matrix to be inverted is diagonal and all the diagonal elements are nonzero. For the matrix $\mathbf{U}'\mathbf{U}$ to become diagonal, it would be necessary to choose an input time-course in which any sample in the waveform is uncorrelated with all other samples. Another way of saying this is that the input waveform should have zero auto-correlation over all lags, except at the zeroth lag (which simply measures how correlated the signal is with itself). One type of input waveform that has this kind of autocorrelation function is white noise. This is one of the reasons for the popularity

of white noise as a test input. Another reason relates to the fact that the white noise time series also has a power spectrum that is essentially flat over a broad range of frequencies. This *persistently exciting* kind of stimulation allows the system to be probed over a larger range of dynamic modes.

Although white noise has been employed as a test input in many studies investigating various neural systems, it has not been used as much for identifying other physiological systems. A major reason for this is the practical difficulty of implementing this kind of input forcing. The *pseudo-random binary sequence* (PRBS) offers an attractive alternative that is very easy to implement and can lead to good estimation results in many applications. The PRBS is so named because the time series produced is actually periodic with a cycle duration of $N+1$ samples, if N is the total number of points in the sequence. However, within one period of this series, each sample is virtually uncorrelated with other samples. One of the most commonly used methods for generating the PRBS employs binary shift-registers with feedback. Figure 8.8a displays a four-stage shift-register. The process begins with all stages assigned a value of 1. Then, at the end of each time-step (T), the value contained in each stage is moved to the right by one stage. The value in the rightmost stage of the shift-register ($=1$) becomes the first value of the PRBS. At the same time, this value is fed back toward the first stage and is exclusive "OR"ed ("XOR"ed) with the value originally in the first stage. In this case, applying Boolean arithmetic, we get $1 \oplus 1 = 0$. Thus, at the end of the first time-step, the values in the shift-register are 0111. During the next time-step, the value 1 in the rightmost stage is moved to the right and becomes the second value in the PRBS. At the same time, this value (1) is fed back to the first stage and XORed with its current value (0); thus, $1 + 0 = 1$. This new result is assigned to the first stage. At the end of the second time-step, the values in the shift-register are 1011. This process continues until the values in the shift-register revert to 1111, which was what it had started with. It can be easily shown that the four-stage shift register assumes the 1111 value at the end of the 16th time-step and the whole sequence repeats itself. The output of this process is a 15-point sequence with random-like properties, as depicted in Figure 8.8b. The autocorrelation function of this kind of sequence approximates that of white noise up to a maximum lag number of 14, as is shown in Figure 8.8c. However, beyond this range, it is clear that the sequence is periodic. The maximum autocorrelation value for a PRBS signal of amplitude A is A^2, and the minimum value is $-A^2/N$. Thus, in our example, the maximum and minimum values turn out to be 0.25 and -0.01667, respectively. Although the PRBS example shown here is based on a four-stage shift-register, the latter may be extended to more stages. For an m-stage shift-register, the total output sequence will consist of $2^m - 1$ "random" values. The PRBS signal displayed in Figure 8.8b was generated by executing the MATLAB script file `prbs.m`.

Using the PRBS as an input can lead to a dramatic simplification of the correlation method of system identification (see Section 8.2.3). If an N-sample
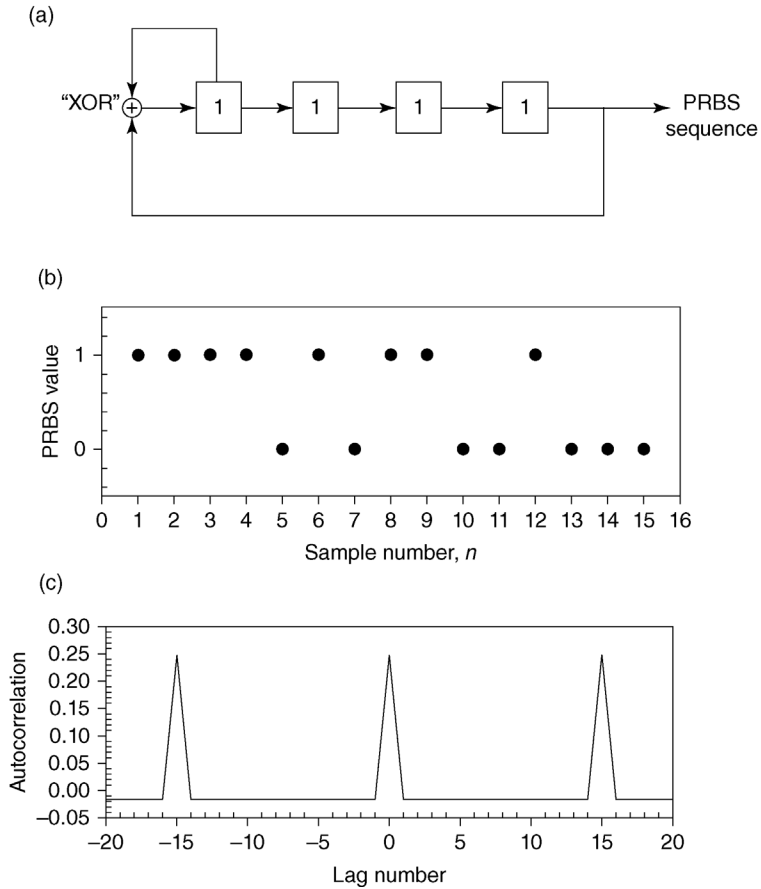
FIGURE 8.8 (a) Shift register method for generating a 15-point pseudorandom binary sequence. (b) The 15-point PRBS signal generated from (a). (c) Theoretical autocorrelation function of the 15-point PRBS signal.

PRBS input of amplitude A is employed, the autocorrelation matrix $\hat{\mathbf{R}}_{uu}$ (of size $N \times N$) becomes

$$
\hat{\mathbf{R}}_{uu} = \frac{A^2}{N}
\begin{bmatrix}
N & -1 & \cdots & -1 \\
-1 & N & \cdots & -1 \\
\vdots & \vdots & \vdots & \vdots \\
\cdots & \cdots & \cdots & \cdots \\
-1 & -1 & \cdots & N
\end{bmatrix}
\tag{8.51}
$$

This matrix can be easily inverted, taking the following form:

$$\hat{\mathbf{R}}_{uu}^{-1} = \frac{N}{A^2(N+1)} \begin{bmatrix} 2 & 1 & \cdots & 1 \\ 1 & 2 & \cdots & 1 \\ \vdots & \vdots & \vdots & \vdots \\ \vdots & \vdots & \vdots & \vdots \\ 1 & 1 & \cdots & 2 \end{bmatrix} \tag{8.52}$$

One can verify that the right-hand side of Equation 8.52 is the inverse of $\hat{\mathbf{R}}_{uu}$ by multiplying this by the right-hand side of Equation 8.51 and showing that the result yields the identity matrix.

Then, applying Equation 8.22, we obtain the impulse response vector:

$$\mathbf{h} = \frac{N}{A^2(N+1)\,T} \begin{bmatrix} 2 & 1 & \cdots & 1 \\ 1 & 2 & \cdots & 1 \\ \vdots & \vdots & \vdots & \vdots \\ \vdots & \vdots & \vdots & \vdots \\ 1 & 1 & \cdots & 2 \end{bmatrix} \begin{bmatrix} \hat{R}_{uy}(0) \\ \hat{R}_{uy}(0) \\ \vdots \\ \vdots \\ \hat{R}_{uy}((N-1)T) \end{bmatrix} \tag{8.53}$$

By evaluating the right-hand side of Equation 8.53, we can decompose the above matrix equation into the following set of equations:

$$h(k) = \frac{N}{A^2(N+1)\,T} \left(\hat{R}_{uy}(k) + \sum_{i=0}^{N-1} \hat{R}_{uy}(i) \right), \quad k = 0, 1, \ldots, N-1 \tag{8.54}$$

The expression for $h(k)$ in Equation 8.54 makes it necessary only to compute the cross-correlation between the input and output sequences. Explicit matrix inversion is thereby averted. However, a serious practical limitation of Equation 8.54 is that the errors associated with the estimates of the impulse response can be unacceptably large if the input and output measurements are very noisy, since N values of $h(k)$ have to be estimated from N pairs of input–output data (see Section 8.2.2). A good example of the application of this technique to the identification of a physiological system is given in Sohrab and Yamashiro (1980).

8.4   IDENTIFICATION OF CLOSED-LOOP SYSTEMS: "OPENING THE LOOP"

The system identification methods discussed in Section 8.3 were based implicitly on the assumption of an open-loop system: The stimulus (input) to the system was assumed to be unaffected by the response (output). However, since most

physiological control processes operate under closed-loop conditions, researchers have applied a variety of techniques to "open the loop" by isolating the components of interest from other components that comprise the entire system. In some cases, "opening the loop" has meant literally that the subsystem of interest was surgically separated from the rest of the system. Denervation, ablation of certain focal areas, and the redirection of blood flow have become standard techniques in physiological investigations. Another group of methods have been less invasive, involving the use of pharmacological agents to minimize or eliminate potentially confounding influences while the component of interest is studied. A third class of techniques apply clever, noninvasive experimental manipulations to the intact physiological system in order to open the loop functionally rather than physically or pharmacologically. In this chapter, we will review several classic examples that represent the wide spectrum of these methods.

8.4.1   The Starling Heart–Lung Preparation

In Section 3.5, we discussed a simple closed-loop model of cardiac output regulation, consisting of essentially two major subsystems, one comprising of the heart and pulmonary circulation and the other representing the systemic circulation. The now legendary experiments by Patterson et al. (1914) provided the first systematic characterization of the former subsystem, thereby enabling the measurement of the intrinsic response of the heart to changes in venous return and arterial blood pressure. As illustrated schematically in Figure 8.9, the heart and lungs were surgically isolated from the rest of the systemic circulation. By connecting the right atrium to a reservoir of blood placed above it and controlling the flow of blood from the reservoir to the heart, the researchers were able to artificially vary the right atrial pressure. Blood ejected from the left ventricle was led to an arterial capacitance and then through an adjustable resistance (Starling resistance) back to the venous reservoir after being heated to body temperature. Adjustment of the Starling resistor or the vertical position of the arterial capacitance allowed the researchers to control arterial (or aortic) pressure. In this way, systematic changes in right atrial pressure and arterial pressure were related to the corresponding cardiac output. These data formed the basis of Guyton's cardiac function curves (see Section 3.5.1).

8.4.2   Kao's Cross-Circulation Experiments

Kao and Ray (1954) performed experiments on anesthetized dogs to determine whether the increase in cardiac output observed during exercise was due to neural or humoral (blood-borne) factors. In order to separate the neural from humoral effects, their experiments were designed in the following way. In each experiment, two anesthetized dogs were used. The hind limbs of the "neural dog" were stimulated electrically so that muscular work was induced. However, arterial blood perfusing
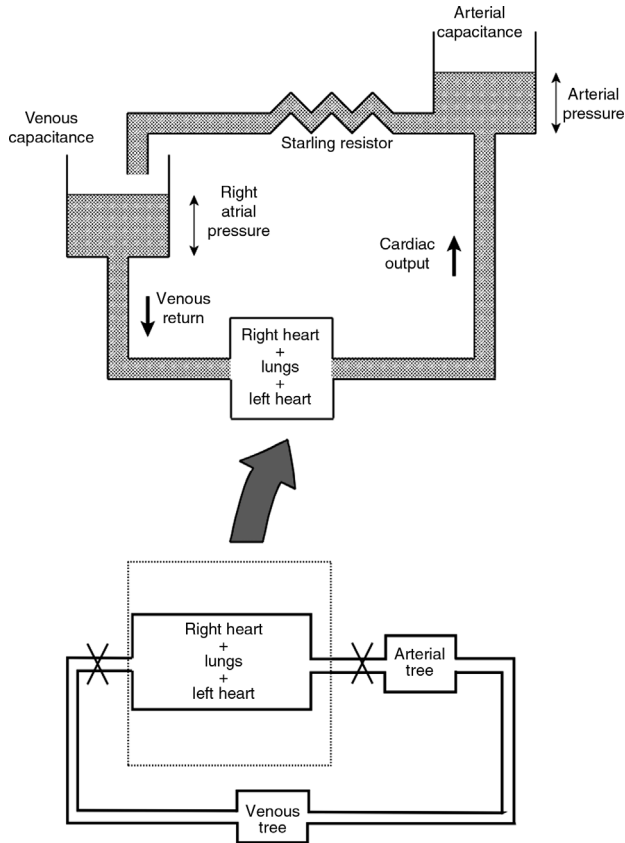
FIGURE 8.9   Schematic illustration of the way in which Starling and coworkers "opened the loop" to study the control of cardiac output.

the hind limbs of this dog came from the second dog, and venous blood leaving the limbs were directed back to the "humoral dog." The basic experimental design is displayed in Figure 8.10. The authors hypothesized that (i) if the exercise-induced cardiac output increase was solely due to neural feedback from the exercising limbs, the "neural dog" would continue to show this increase, while the "humoral dog" should not respond at all; and (ii) if the exercise-induced cardiac output increase was solely due to humoral factors, the "humoral dog" should show this increase, while there should be no response in the "neural dog." Based on the results of nine pairs of these animals, it was found that cardiac output increased significantly in both "neural" and "humoral" dogs. This led the authors to conclude that both neural and humoral factors are involved in the regulation of cardiac output during muscular activity.
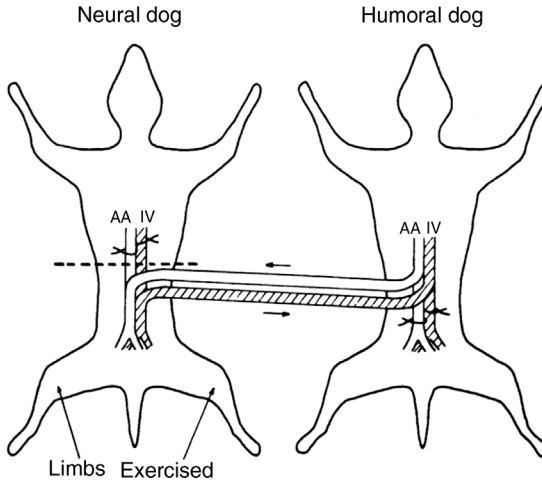
FIGURE 8.10 Kao's experimental design for separating neuromuscular feedback from humoral effects on exercise-induced hyperpnea. (Reproduced with permission from Kao and Ray (1954).)

8.4.3 Artificial Brain Perfusion for Partitioning Central and Peripheral Chemoreflexes

In Section 6.7, we examined the stability properties of a model of the chemoreflex regulation of ventilation. The analysis showed that the gains and time constants associated with the central and peripheral chemoreflexes are important determinants of respiratory stability. The question of being able to measure the dynamics of these two chemoreflexes in isolation from one another was addressed by Berkenbosch et al., 1979 in a series of experiments that employed the clever technique of artificial brain perfusion. This method is illustrated schematically in Figure 8.11. In anesthetized cats, the researchers directed blood from one of the femoral arteries through an extracorporeal circuit in which the blood was equilibrated in a foamer with a gas mixture of known composition, defoamed, and then returned to the cat through a cannulated vertebral artery. The other vertebral artery was clamped, so that the brain was perfused only by the blood leaving the extracorporeal circuit. This allowed the P_{CO2}, P_{O2}, and pH of the blood perfusing the medullary chemosensitive regions to be maintained at constant levels set by the researchers. This effectively "opened" the central chemoreflex loop. Consequently, the effects of dynamic changes in arterial P_{CO2} or P_{O2} (produced by inhalation of hypercapnic or hypoxic gas mixtures) on the peripheral chemoreflex contribution to ventilation could be measured in isolation from the central contribution.
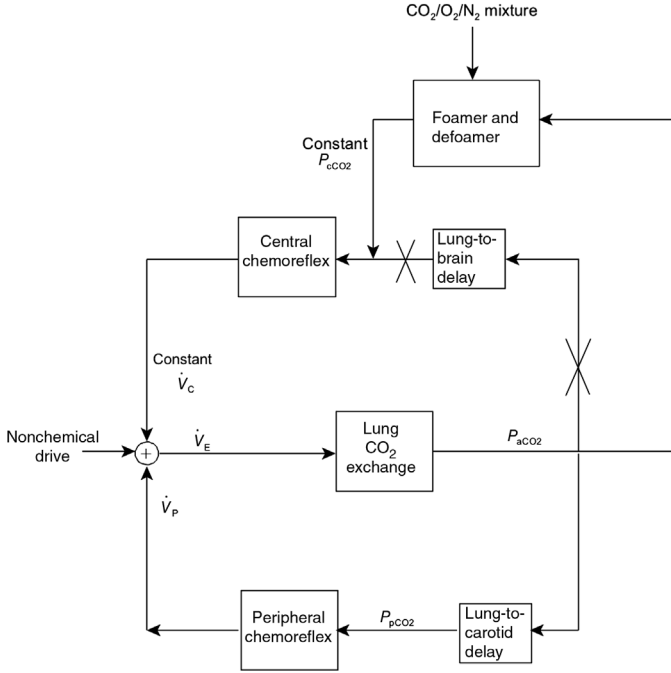
FIGURE 8.11 Schematic representation of the artificial brain perfusion setup for separating central and peripheral chemoreflex drives.

8.4.4 The Voltage Clamp

The basic mechanism for the neuronal action potential is a classic example of a physiological process in which both negative and positive feedback occur. Consider the Hodgkin–Huxley model shown in circuit form in Figure 8.12a and block diagram form in Figure 8.12b; the model has been simplified here to exclude the leakage channel due to the chloride ions. Normally, potassium ions (K^+) tend to leak out of the nerve cell because of the much larger K^+ concentration in the axosplasm relative to the extracellular fluid. The opposite occurs with the sodium ions (Na^+). When the membrane is depolarized by a presynaptic stimulus, the variable Na^+ conductance increases rapidly and considerably, allowing a large influx of Na^+ ions from the extracellular fluid, which depolarizes the cell membrane even further. Thus, the positive feedback dominates this initial phase of the action potential. Fortunately, the increase in Na^+ conductance is short-lived and the influx of Na^+ ions slows after a fraction of a millisecond. At the same time, the K^+ conductance starts to increase, following a short delay. This allows K^+ ions to flow out of the axoplasm, acting to reverse the depolarization of the membrane. The repolarization speeds up the decline in Na^+ conductance, which, in turn, promotes the
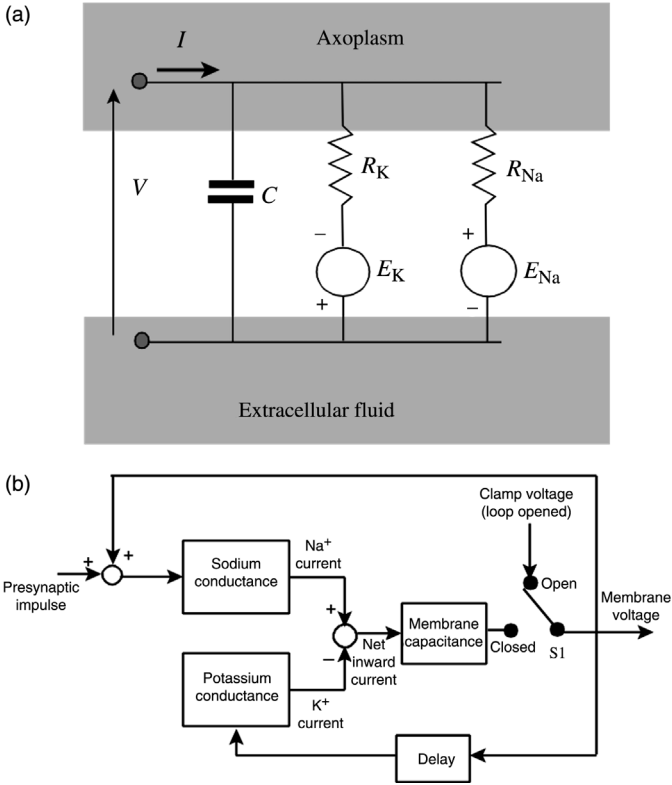
FIGURE 8.12   (a) Simplified schematic of the Hodgkin–Huxley model. (b) "Opening the loop" via application of the voltage clamp technique.

repolarization process (Figure 8.12b). Through the insertion of an electrode into the axoplasm, it is possible to precisely control the voltage inside the nerve cell. By applying a step depolarization through this electrode and keeping the applied voltage constant, one is effectively "opening" both the positive and negative feedback loops. Then, by measuring the current flowing across the membrane and by altering the composition of the extracellular fluid to isolate the Na^+ from K^+ effects, it is possible to deduce the time-courses of the Na^+ and K^+ conductances to the step depolarization. This was the basic methodology employed by Hodgkin et al. (1952), as well as researchers after them, to study the mechanisms underlying the generation of the action potential.

8.4.5   Opening the Pupillary Reflex Loop

The model of the pupillary light reflex that we employed in Section 6.6 to demonstrate stability analysis was based largely on Stark's ingenious experiments in which he developed techniques to functionally open the reflex loop. The two

basic means by which this was done are illustrated in Figure 8.13. In the normal closed-loop state, an increase in total light flux impinging on the retina results in a reduction in pupil area that, assuming the light intensity remains constant, would decrease total light flux to offset the initial increase. In Figure 8.13a, Stark used a pupillometer to measure the size of the pupil and an adjustable light source that delivered light at intensities that were inversely proportional to the pupil area. By introducing these devices into the feedback loop, he was able to offset the effect of changing pupil area on total light flux by raising the light intensity. In this way, the total light flux could be controlled quite precisely, enabling him to deduce the loop transfer function characteristics of the reflex. Another technique that he used to effectively "open the loop" is illustrated in Figure 8.13b. Here, he applied a very narrow beam of light through the pupil. By restricting the cross-sectional area of the beam to a size that was smaller than the residual area of the pupil, total light flux was rendered completely independent of pupil area, since the area of the light beam
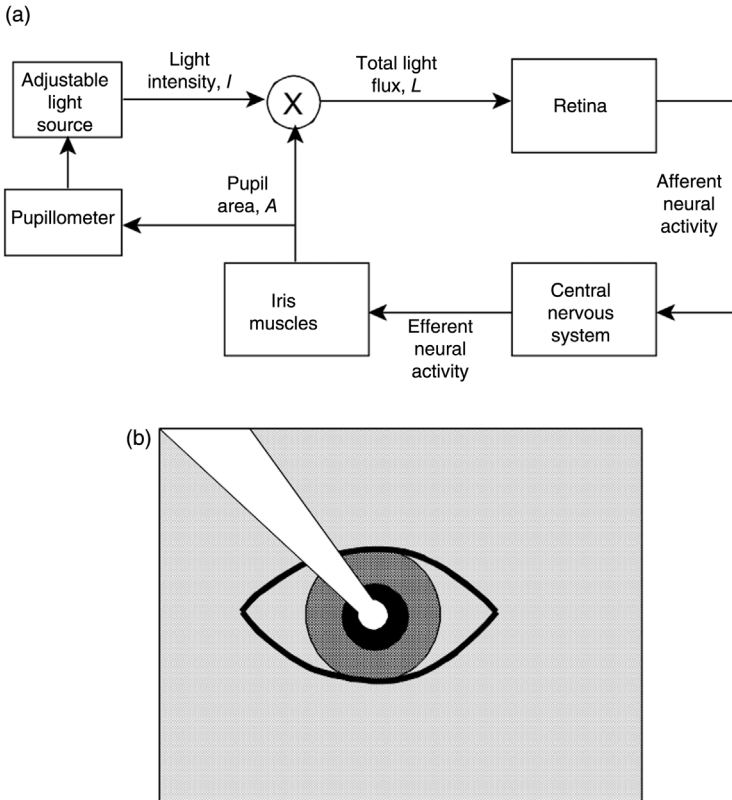
FIGURE 8.13   Two methods of functionally opening the pupillary reflex loop. (a) Modulation of applied light intensity using measurements of pupil area. (b) Application of narrow light beam to residual area of pupil.

impinging on the retina was not affected by changes in pupil size. Under these conditions, it was possible to completely control the time-course of the input (total light flux) and measure the corresponding response (pupil area) of the "opened" reflex loop.

8.4.6 Read Rebreathing Technique

Under normal operating circumstances, ventilation ($\dot{V}_E$) and arterial P_{CO_2} (P_{aCO_2}) are tightly coupled through the powerful negative feedback loops of the chemoreflexes: Any increase in P_{aCO_2} leads rapidly to increase in $\dot{V}_E$, which act to offset the initial rise in P_{aCO_2}. However, Read (1967) found a simple experimental technique of functionally breaking this closed-loop relationship. The subject breathes into and out of a small (4–6 L) rebreathing bag that is filled with an initial gas mixture containing 7% CO_2 in oxygen. After an initial transient phase, an equilibrium is established between arterial blood, oxygenated mixed venous blood, and gas in the lungs and rebreathing bag. Thereafter, the P_{CO_2} in both blood and gas phases increases linearly with time, and $\dot{V}_E$ also increases proportionally, without reversing the rise in P_{CO_2} as one would expect in the closed-loop situation. The way in which the technique works is best appreciated from a modeling perspective. The following differential equation provides the simplest dynamic characterization of CO_2 exchange at the level of the body tissues:

$$\frac{V_t}{863}\frac{dP_{vCO_2}}{dt} = QK_{CO_2}\left(P_{aCO_2} - P_{vCO_2}\right) + \dot{V}_{CO_2} \tag{8.55}$$

Equation 8.55 assumes the capacitance effect of the body tissues to be lumped into the volume V_t. $\dot{V}_{CO_2}$ is the metabolic production rate of CO_2. Following the establishment of the equilibrium between the arterial and mixed venous blood and gas in the lungs and bag, it can be seen that in Equation 8.55 the arteriovenous gradient disappears and the derivative becomes a constant proportional to the CO_2 metabolic production rate. Integrating Equation 8.55 results in P_{vCO_2} assuming a linear dependence on time. Since P_{aCO_2}, alveolar P_{CO_2} (P_{ACO_2}), and the bag P_{CO_2}, are equilibrated with P_{vCO_2}, these variables also increase linearly with time during the rest of the rebreathing process. With the linearly rising arterial and tissue P_{CO_2}, brain tissue P_{CO_2} will also increase in linear fashion, driving $\dot{V}_E$ along a similar time-course. The increasing $\dot{V}_E$ is mediated almost completely by the central chemoreceptors, since the high oxygenation levels suppress peripheral chemoreception. However, because of the equilibration between the bag (inspired) P_{CO_2} and P_{ACO_2} (see Equation 6.42), the increasing $\dot{V}_E$ is prevented from influencing P_{ACO_2}, hence breaking the negative feedback in this closed-loop system. A schematic block diagram of the rebreathing model is shown in Figure 8.14.
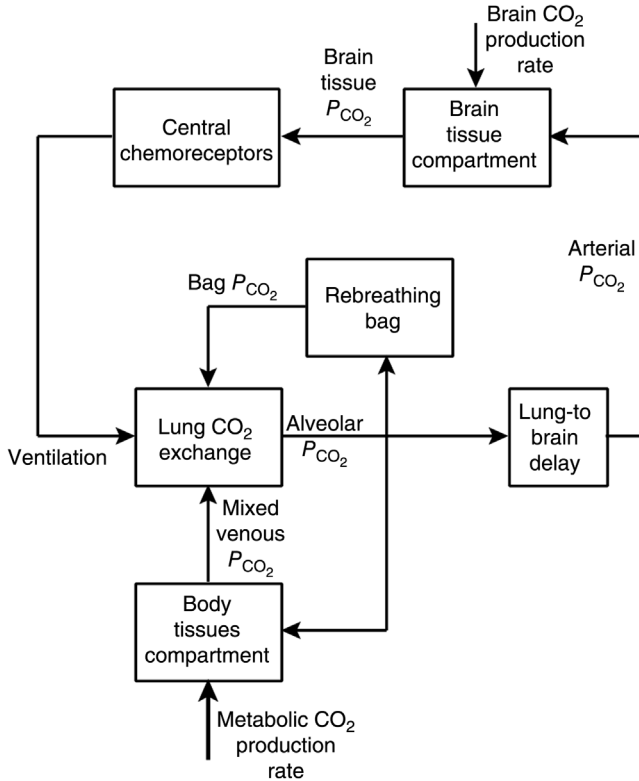
FIGURE 8.14   Schematic block diagram of respiratory control during rebreathing.

8.5   IDENTIFICATION UNDER CLOSED-LOOP CONDITIONS: CASE STUDIES

Although we have seen a wide range of physiological examples in which exper-
imental interventions were employed to "open the loop," such techniques are not
always applicable. Moreover, a major criticism leveled against this kind of
approach is that the system under study is placed under nonphysiological conditions
and subjected to nonphysiological inputs when these interventions are applied.
Ideally, we would like to identify the physiological system under "normal operating
conditions" when its feedback loops are functionally intact. However, consider the
problem involved with identifying the impulse response $h(t)$ of the closed-loop
system component shown in Figure 8.15. The unknown (and unobservable)
disturbance $u(t)$ that enters the closed loop represents both a "measurement"
and "process" noise input. It is considered "measurement noise" since it corrupts
the measurements $x(t)$, which otherwise would be related solely to $y(t)$, the input to
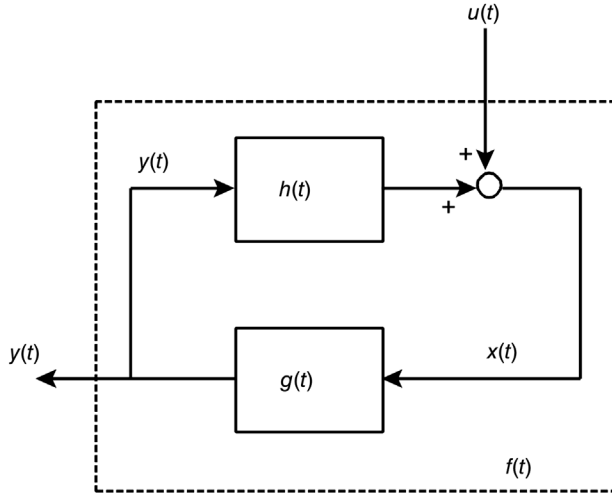
FIGURE 8.15   Correlation of the process noise input $u(t)$ with $y(t)$ complicates the identification $h(t)$ from closed-loop measurements $y(t)$ and $x(t)$. $f(t)$ is the impulse response of the closed-loop system.

the system component. This is clear from the mathematical expression relating $x(t)$ to $y(t)$ and $u(t)$:

$$x(t) = \int_0^\infty h(\tau)y(t-\tau)dt + u(t) \tag{8.56}$$

$u(t)$ also takes the form of "process noise" since it enters the closed-loop system and becomes correlated with $y(t)$. If we consider $u(t)$ the input and $y(t)$ the output of the overall system (as defined by the dashed rectangle in Figure 8.15), we obtain

$$y(t) = \int_0^\infty f(\tau)u(t-\tau)d\tau \tag{8.57}$$

In order to obtain an unbiased estimate of $h(t)$ from Equation 8.56 by the least-squares approach using $y(t)$ as input and $x(t)$ as output, the final solution must be such that $u(t)$ becomes uncorrelated with (or orthogonal to) $y(t)$. However, as Equation 8.57 clearly shows, $y(t)$ is correlated with $u(t)$. Thus, the direct application of open-loop system identification methods to this problem will not yield accurate estimates of $h(t)$. A couple of approaches for circumventing this problem are described in the following two sections. One other approach is to impose constraints on the orthogonality condition, but this falls outside the scope of the present

discussion. For further information about this last method, the interested reader is referred to a study by Khoo (1989a, 1989b).

8.5.1   Minimal Model of Blood Glucose Regulation

One effective way of partitioning the effects of the feedforward and feedback components of a closed-loop system from one another is to assume a model structure for at least one of these components. Then, if the effects of all other extraneous influences (process noise) entering the closed-loop system are small relative to the magnitude of the system responses, the parameters of the assumed model can be estimated. The "minimal model" of blood glucose regulation, developed by Bergman et al. (1979), represents a good example of this kind of approach. Referring to Figure 8.15, suppose that $x(t)$ and $y(t)$ correspond to the plasma glucose and insulin concentrations at time t, respectively. Then, the impulse response function $h(t)$ would represent glucose regulation kinetics, while $g(t)$ would reflect the dynamics of insulin production and utilization. The closed-loop system is perturbed by an impulsive input $u(t)$, consisting of an intravenous injection $(300\,\mathrm{mg\,kg^{-1}}$ in dogs) of glucose. By using the resulting time-courses in $y(t)$ *and x(t) as input and output*, respectively, the model of *glucose dynamics* can be identified. Subsequently, by using $x(t)$ as input and $y(t)$ as output, the parameters of the model of *insulin dynamics* can be estimated. Bergman and coworkers have referred to this methodology as *partition analysis*, since both halves of the closed-loop system are identified as if they were in the open-loop state. It should be emphasized that the key assumptions that make this kind of closed-loop estimation possible are (a) the imposition of structure and causality on the dynamics characterizing glucose and insulin production and utilization and (b) relatively large signal-to-noise ratios in the measurements.

In this section, we will discuss only the estimation of the minimal model of glucose regulation, that is, how insulin affects glucose. The estimation of the converse model in which glucose affects insulin will not be considered. Thus, the input here is the measured plasma insulin concentration $y(t)$ following the intravenous glucose injection, while the output is the corresponding measured blood glucose concentration $x(t)$. The model employed by Bergman contains the features incorporated in the glucose kinetics model proposed by Stolwijk and Hardy (see Sections 3.6 and 4.8), but is more realistic in that it allows for the delayed effect of insulin on glucose disappearance, a feature that has been observed. The insulin concentration $y(t)$ does not affect glucose dynamics directly. Instead, it acts through a "remote compartment," so the *effective* insulin concentration $y_{\mathrm{eff}}(t)$ is given by

$$\frac{dy_{\mathrm{eff}}}{dt} = k_2 y(t) - k_3 y_{\mathrm{eff}}(t) \qquad\qquad (8.58a)$$

where k_2 and k_3 represent the fractional rate parameters for insulin transport into and elimination from the remote compartment. This compartment is "remote" in that y_{eff}

is not directly measurable. It should also be noted that the volume of the remote compartment has been factored into the rate constants k_2 and k_3. The rate of change of glucose in the blood plasma is given by

$$\frac{dx}{dt} = \text{net rate of glucose production by the liver}$$

$$-\text{rate of glucose utilization by other tissues} \qquad (8.59a)$$

where

$$\text{Net rate of glucose production by the liver} = B_0 - k_5 x(t) - k_6 y_{\text{eff}}(t)x(t) \quad (8.60)$$

and

$$\text{Rate of glucose utilization by other tissues} = R_{d0} + k_1 x(t) + k_4 y_{\text{eff}}(t)x(t) \quad (8.61)$$

In Equation 8.60, B_0 represents the rate of glucose production by the liver. The rate of glucose uptake by the liver is assumed to be proportional to an insulin-independent component (through rate constant k_5) and an insulin-dependent component (through rate constant k_6). Similarly, in Equation 8.61, the rate of glucose utilization by nonhepatic tissues is assumed to have a constant component, a component proportional to glucose concentration and a component sensitive to both glucose and effective insulin concentration. Substituting Equations 8.60 and 8.61 into Equation 8.59a and rearranging terms, we obtain the result:

$$\frac{dx}{dt} = [B_0 - R_{d0}] - [k_5 + k_1]x(t) - [k_6 + k_4]y_{\text{eff}}(t)x(t) \qquad (8.59b)$$

As in Equation 8.58a, the effective plasma glucose capacitance is factored into the parameters on the right-hand side of Equation 8.59b.

Equations 8.58a and 8.59b provide a complete characterization of glucose kinetics. However, it is obvious that there are too many redundant parameters. For instance, in Equation 8.59b, it would not be possible to estimate B_0 and R_{d0} separately; only the combined term $[B_0 - R_{d0}]$ can be identified. The same is true for $[k_5 + k_1]$ and $[k_6 + k_4]$. In addition, since $y_{\text{eff}}(t)$ is not measurable, a further reduction in parametrization can be achieved by defining the new variable $z(t)$ that is proportional to $y_{\text{eff}}(t)$:

$$z(t) \equiv [k_6 + k_4]y_{\text{eff}}(t) \qquad (8.62)$$

Substituting Equation 8.62 into Equations 8.58a and 8.59b, we obtain

$$\frac{dz}{dt} = -p_2 z(t) + p_3 y(t) \qquad (8.58b)$$

and

$$\frac{dx}{dt} = p_4 - p_1 x(t) - z(t)x(t) \tag{8.59c}$$

where $p_1 = k_1 + k_5$, $p_2 = k_3$, $p_3 = k_2(k_4 + k_6)$, and $p_4 = B_0 - R_{d0}$. Equations 8.58b and 8.59c provide the same dynamic characterization of glucose regulation for the minimum number of unknown parameters that have to be estimated from the input–output data. For this reason, it is referred to as a *minimal model*.

The way in which the unknown parameters p_1, p_2, p_3, and p_4 are estimated is as follows. First, we begin with initial guesses for the unknown parameters. Using the measured input time-course $y(t)$ and the initial parameter values, Equation 8.58b is first solved to obtain the value of z at the current time-step. Using this value of z in Equation 8.59c and integrating this equation, the glucose concentration at the next time-step can be computed. This process is repeated until predictions for $x(t)$ have been made for the entire duration of the experiment. The predictions are compared with the actual blood glucose measurements, and the value of the criterion function (sum of squares of the differences between measured and predicted glucose values) is computed. An optimization algorithm is used to search for another combination of the four unknown parameters that would produce a lower value of the criterion function. Using the new combination of parameter values, $z(t)$ and $x(t)$ are again solved using Equations 8.58b and 8.59c, and the whole process is repeated until the incremental reduction in criterion function is considered insignificant.

An example of the results achieved with minimal model estimation is displayed in Figure 8.16. "Data" required for the estimation were generated using a SIMULINK implementation (named `gmm_sim.slx`) of Bergman's models of both glucose and insulin subsystems (Bergman et al., 1979, 1985; Toffolo et al., 1980). These were combined and made to operate in closed-loop mode. Random perturbations were added to the glucose concentration $x(t)$ predicted by the model to simulate "measurement noise" in the glucose observations. The SIMU-LINK model, shown in Figure 8.17, produced samples of $x(t)$ and the plasma insulin concentration $y(t)$ at intervals of 1 min to mimic the blood sampling conducted in the real experiments. These "measurements" are shown as the closed circles in Figure 8.16a and b. This particular set of "measurements" have also been saved in the MATLAB data file: `data_gmm.mat`. Parameter estimation is performed using the Nelder–Mead simplex algorithm, which is implemented in MATLAB with the function `fminsearch`. The primary command lines of the MATLAB script file, labeled `gmm_est.m`, are as follows:

```
>> options(1) = 1;
>> [p,options] = fminsearch('fn_gmm',p_init,options,[]);
```

where the function `fn_gmm` (in MATLAB file `fn_gmm.m`) is used by `gmm_est.m` to produce values of the criterion function *J* with each iteration of the algorithm. For
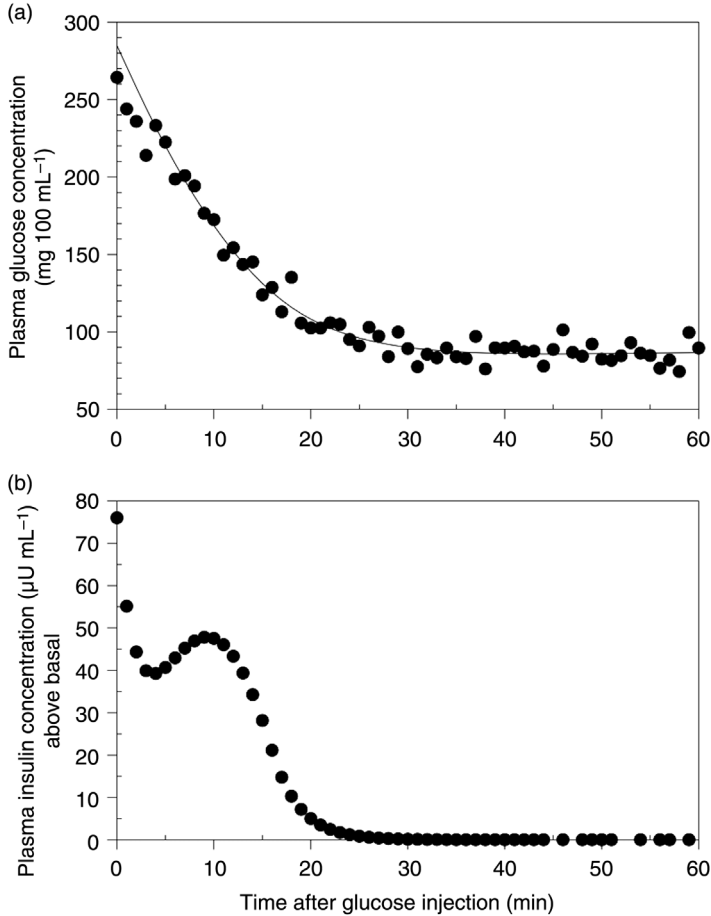
FIGURE 8.16 "Measurements" of blood glucose and insulin levels (closed circles) following intravenous bolus infusion of $300\,\mathrm{mg\,kg^{-1}}$ of glucose. The best-fit prediction for glucose is shown as the solid curve.

each new set of parameter values, `fn_gmm` solves the model equations given in Equations 8.58b and 8.59c using the Euler method of integration (with time-steps of 0.01 min) and computes the sum of squares of the differences between the "observed" glucose concentration samples and the values predicted by model solution. The parameters to be estimated are the four unknown coefficients in Equations 8.58b and 8.59c: p_1, p_2, p_3, and p_4; in addition, the "true" glucose concentration at time zero, $x(0)$, is treated as the fifth unknown parameter. In the above MATLAB command lines, note that the array `options` is used as both an input and output argument. This is done (specifically, the first element of `options` is set equal to 1) to allow the algorithm to display the value of J, along with the

(a)

(b)

FIGURE 8.17 SIMULINK model `gmm_sim.slx` of combined glucose–insulin kinetics based on Bergman's minimal models. (a) Model input and outputs. (b) Details of the SIMULINK implementation.

parameter values associated with the simplex vertices, at each stage of the computations. The final estimated set of parameter values are $p_1 = 0.068$, $p_2 = 0.091$, $p_3 = 6.72\,\text{H}\,10^{-5}, p_4 = 6.03$, and $x(0) = 284.9\,\text{mg}\,100\,\text{ml}^{-1}$. These may be compared with the "true" parameter values used in the SIMULINK program: $p_1 = 0.049$, $p_2 = 0.091$, $p_3 = 8.96\,\text{H}\,10^{-5}$, and $p_4 = 4.42$. These latter values were selected from the results obtained by Bergman and coworkers from their experiments on dogs. The "best-fit" model prediction is shown as the solid curve in Figure 8.16.

The minimal model has been employed successfully in many clinical studies to quantitate insulin sensitivity and glucose effectiveness in various populations at risk for diabetes. More details on the relationship between these indices and the model parameters may be found in the original papers by Bergman and coworkers listed in the Bibliography section of this chapter.

8.5.2 Closed-Loop Identification of the Respiratory Control System

In this section, we illustrate a somewhat different approach to closed-loop identification. In the previous example, an optimization technique was employed for parameter estimation. As we had mentioned earlier, one disadvantage of this kind of iterative method is the possibility of convergence to a local minimum instead of the global solution. Here, we take the alternative approach of least-squares estimation, where the optimal solution is arrived at in one computational step. Another difference that we will highlight here is the use of a persistently exciting input to stimulate the closed-loop system, instead of the brief but potent impulsive disturbance employed in the minimal model of glucose regulation. Practical considerations dictate the use of the former type of input in the case of the respiratory control system. A potent impulsive disturbance in this case would take the form of an inhaled breath of gas with very high CO_2 content. Such a potent stimulus would be certain to evoke a behavioral response in addition to the chemoreflex-mediated changes, thereby allowing the measurement process to affect the system under observation. In this case, the stimulus takes the form of a pseudorandom binary sequence (PRBS) in the inhaled P_{CO2}. This allows the system to be excited with relatively low CO_2 concentrations over a broad range of frequencies within the limited experimental duration. An example of the practical implementation of this kind of PRBS time-course in inhaled P_{CO2} (P_{ICO2}) and the resulting effects on alveolar P_{CO2} (P_{ACO2}) and ventilation ($\dot{V}_E$) in a normal human subject is displayed in Figure 8.18. As in the previous section, partition analysis is employed in the identification procedure. The first stage of the analysis involves the estimation of the parameters of the plant (i.e., gas exchange in the lungs), using measurements of $\dot{V}_E$ and P_{ICO2} as inputs and P_{ACO2} as output. The second stage consists of the estimation of the controller and lung-to-chemoreceptor delay using P_{ACO2} as the input and $\dot{V}_E$ as the output (Figure 8.19).

8.5.2.1 Identification of the Plant The model employed to represent the CO_2 exchange in the lungs is the small-signal expression derived in Equation 6.43b.
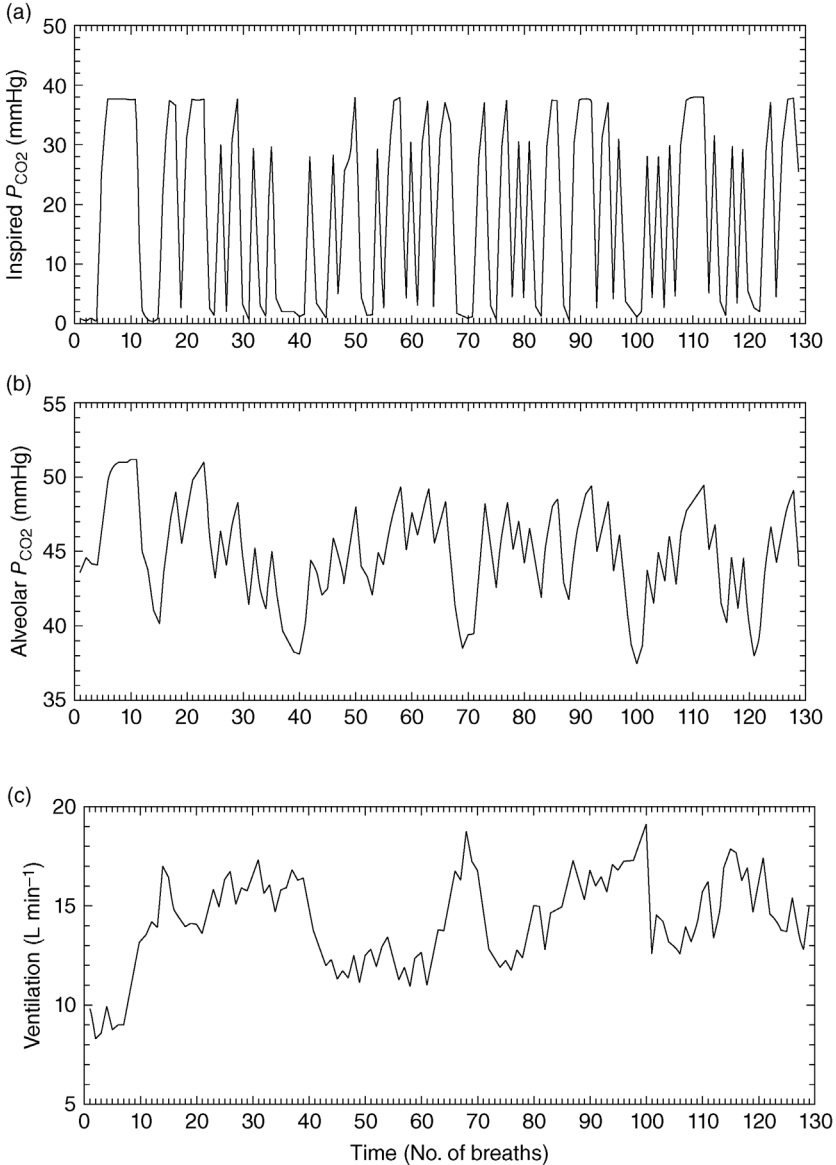
FIGURE 8.18 Responses in P_{ACO2} (part (b)) and ventilation (part (c)) produced in a normal subject during inhalation of 6% CO_2 in air, administered on a pseudorandom binary basis (P_{ICO2}, part (a)). (Reproduced from Ghazanshahi and Khoo (1997).)
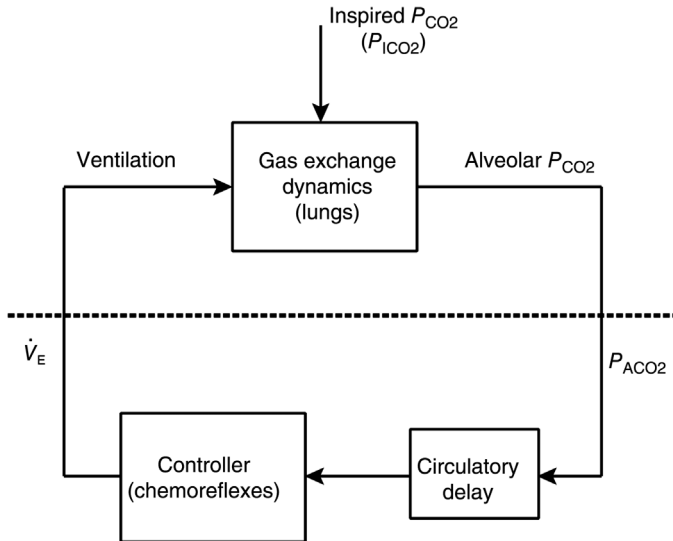
FIGURE 8.19   Application of partition analysis for identification of the plant (above dashed line) and controller (below dashed line) portions of the closed loop.

However, in Equation 8.63, we have also allowed for perturbations in P_{ICO2} (which, in Equation 6.43b, was kept constant):

$$\tau_L \frac{d\left(\Delta P_{ACO_2}\right)}{dt} + \Delta P_{ACO_2} = G_1 \Delta P_{ICO_2} - G_2 \Delta \dot{V}_E \tag{8.63}$$

where ΔP_{ACO2}, $\Delta \dot{V}_E$, and ΔP_{ICO2} represent small changes in P_{ACO2}, $\dot{V}_E$, and P_{ICO2} about their equilibrium values, and

$$G_1 = \frac{\dot{V}_E - \dot{V}_D}{\dot{V}_E - \dot{V}_D + 863 Q K_{CO_2}} \tag{8.64}$$

$$G_2 = \frac{P_{ACO_2} - P_{ICO_2}}{\dot{V}_E - \dot{V}_D + 863 Q K_{CO_2}} \tag{8.65}$$

Since, in this analysis, our attention is focused on the characterization of how small *changes* in P_{ICO2} elicit *changes* in P_{ACO2} and $\dot{V}_E$, we assume, to a first approximation, that the operating values of P_{ACO2}, $\dot{V}_E$, $\dot{V}_D$, and P_{ICO2} are constant. Hence, we regard the two factors G_1 and G_2 to be constant-valued parameters, which have to be estimated from the measurements, as we demonstrate below.

Since the measurements of P_{ACO2}, P_{ICO2}, and $\dot{V}_E$ are not made continuously in time but are obtained on a breath-by-breath basis, for purposes of parameter estimation, it is more useful to express the plant equation in the form of a difference equation with a discrete time base (with "breaths" as the unit of time). Furthermore,

P_{ACO2} cannot be directly sampled; instead, we assume that end-tidal P_{CO2} (the highest value of P_{CO2} measured in the exhaled stream during expiration) reliably reflects P_{ACO2}. By integrating Equation 8.63 from the end of the previous breath to the end of the current breath, the differential equation can be converted into a difference equation of the following form:

$$\Delta P_{ACO_2}(n) + \alpha\Delta P_{ACO_2}(n-1) = \beta_1 \Delta P_{ICO_2}(n) - \beta_2 \Delta \dot{V}_E(n) + e(n) \qquad (8.66)$$

where n represents the current breath number and $0 \leq n \leq N-1$, N being the total number of breaths used for data analysis. Recalling the material covered in Chapter 7, this type of continuous-time to discrete-time conversion is an example of the *impulse invariance* method (Jackson, 1995).

It can be further shown that α, β_1, and β_2 in Equation 8.63 are related to the parameters G_1, G_2, and τ_L of Equation 8.60 through the following relations:

$$\beta_1 = \frac{G_1}{\tau_L} \qquad (8.67)$$

$$\beta_2 = \frac{G_2}{\tau_L} \qquad (8.68)$$

$$\alpha = -e^{-T/\tau_L} \qquad (8.69)$$

where T is the "sampling interval" that, in this case, would be the breath duration. Strictly speaking, T would vary from breath to breath, since the breathing frequency is somewhat variable. However, previous studies in this field have demonstrated that assuming T to be constant and equal to the *average breath duration* simplifies matters considerably without affecting the outcome of the analysis significantly in most experimental situations.

In Equation 8.66, the last term $e(n)$ is added to account for the residual error between the measured ΔP_{ACO2} and model-predicted ΔP_{ACO2}. Equation 8.66 is a special case of the general class of models known as ARX (autoregressive with exogenous input) models (Ljung, 1987). In this special case, α, β_1, and β_2 are the unknown parameters to be estimated using ΔP_{ACO2} as the output measurement and ΔP_{ICO2} and $\Delta \dot{V}_E$ as the inputs. Estimation of these parameters can be easily achieved using least-squares minimization, as the following equations illustrate. Rewriting Equation 8.66 for all values of n in vector form, we have

$$\begin{bmatrix} \Delta P_{ACO_2}(0) \\ \Delta P_{ACO_2}(1) \\ \vdots \\ \vdots \\ \Delta P_{ACO_2}(N-1) \end{bmatrix} = \begin{bmatrix} 0 & \Delta P_{ICO_2}(0) & -\Delta \dot{V}_E(0) \\ -\Delta P_{ACO_2}(0) & \Delta P_{ICO_2}(1) & -\Delta \dot{V}_E(1) \\ \vdots & \vdots & \vdots \\ \vdots & \vdots & \vdots \\ -\Delta P_{ACO_2}(N-2) & \Delta P_{ICO_2}(N-1) & -\Delta \dot{V}_E(N-1) \end{bmatrix} \begin{bmatrix} \alpha \\ \beta_1 \\ \beta_2 \end{bmatrix} + \begin{bmatrix} e(0) \\ e(1) \\ \vdots \\ \vdots \\ e(N-1) \end{bmatrix}$$
$$(8.70)$$

It can be seen that Equation 8.70 is of the form displayed in Equation 8.8, that is,

$$\mathbf{y} = \mathbf{U}\underline{\boldsymbol{\theta}} + \underline{\mathbf{e}} \tag{8.71}$$

where $\mathbf{y}$ represents the column vector on the left-hand-side of Equation 8.67, $\mathbf{U}$ is the $N \times 3$ matrix on the other side of the equation, and $\underline{\boldsymbol{\theta}}$ is the vector containing the unknown parameters. Thus, $\underline{\boldsymbol{\theta}}$ can be estimated using Equation 8.12, which we rewrite as Equation 8.72:

$$\underline{\boldsymbol{\theta}} = (\mathbf{U}'\mathbf{U})^{-1}\mathbf{U}'\mathbf{y} \tag{8.72}$$

8.5.2.2 *Identification of the Controller and Circulatory Delay*   To model the controller and circulatory delay, we assume the form proposed by Bellville et al. (1979) (see Section 6.8.3). Using the Laplace transform version of this model, we have (from Equations 6.50a and 6.50b)

$$\Delta \dot{V}_E(s) = \left(\frac{G_c}{\tau_c s + 1} + \frac{G_p}{\tau_p s + 1} \right) e^{-sT_d} \Delta P_{ACO_2}(s) \tag{8.73a}$$

Here, we have made the simplifying assumption of using only one common circulatory delay, T_d, in place of the separate central (T_c) and peripheral (T_p) delays assumed in the chemoreflex model of Section 6.8.3. Employing a common denominator for both terms in the summation of Equation 8.73a, we can rewrite the equation in the following form:

$$\Delta \dot{V}_E(s) = \left(\frac{(G_c \tau_p + G_p \tau_c)s + (G_c + G_p)}{\tau_c \tau_p s^2 + (\tau_c + \tau_p)s + 1} \right) e^{-sT_d} \, \Delta P_{ACO_2}(s) \tag{8.73b}$$

When inverse-Laplace transformed back into the time domain, Equation 8.73b takes the following differential equation form:

$$\tau_c \tau_p \frac{d^2(\Delta \dot{V}_E)}{dt^2} + (\tau_c + \tau_p)\frac{d(\Delta \dot{V}_E)}{dt} + \Delta \dot{V}_E = (G_c \tau_p + G_p \tau_c)\frac{d(\Delta P_{ACO_2}(t - T_d))}{dt}$$
$$+ (G_c + G_p)\Delta P_{ACO_2}(t - T_d) \quad (8.74)$$

As in Section 8.5.2.1, since the measurements are made on a breath-by-breath basis, it is more convenient to assume a discrete-time base and recast the model in finite difference form, as in Equation 8.66 for the plant. In this case, the corresponding finite difference equation is

$$\Delta \dot{V}_E(n) + a_1 \Delta \dot{V}_E(n-1) + a_2 \Delta \dot{V}_E(n-2) = b_0 \Delta P_{ACO_2}(n - N_d)$$
$$+ b_1 \Delta P_{ACO_2}(n - 1 - N_d) + \varepsilon(n) \quad (8.75)$$

where, as in Equation 8.66, n represents the current breath number and $0 \le n \le N$ -1. N_d represents the circulatory delay in number of breaths, that is, $N_d = T_d/T$. In this case, $\varepsilon(n)$ is added to account for the discrepancy between the model-predicted $\Delta \dot{V}_E$ and the measured $\Delta \dot{V}_E$. Equation 8.75 can also be cast in the following form:

$$\Delta \dot{V}_E(n) = \begin{bmatrix} -\Delta \dot{V}_E(n-1) & -\Delta \dot{V}_E(n-2) & \Delta P_{ACO_2}(n-N_d) \end{bmatrix}$$

$$\Delta P_{ACO_2}(n-N_d-1) \begin{bmatrix} a_1 \\ a_2 \\ b_0 \\ b_1 \end{bmatrix} + \varepsilon(n) \qquad (8.76)$$

By applying Equation 8.76 to all N sets of data points, we can again construct a matrix equation of the form displayed in Equation 8.70, and thus estimate the unknown parameters a_1, a_2, b_0, and b_1 using least-squares minimization. However, in order to solve for the unknown parameters, it is necessary to know what N_d is. Determination of N_d is done in the following way. We first select a range of physiologically feasible values for N_d. The lung-to-ear delay in most normals is generally in the range of 6–12 s. Thus, a reasonable range for N_d might be 1–4 for breath durations that range from 3 to 6 s. For each of these values of N_d, we solve the least-squares minimization problem and estimate a_1, a_2, b_0, and b_1. For each case, we compute J, the residual sum of squares of the differences between the measured and predicted $\Delta \dot{V}_E$ (as given in Equation 8.35). The "best" estimate of N_d is that value that yields the lowest value of J.

Having estimated the unknown parameters, a_1, a_2, b_0, and b_1, it is possible in principle to relate them to the gains (G_c and G_p) and time constants (τ_c and τ_p) that characterize the corresponding differential equation (Equation 8.74) in a way similar to Equations 8.67 through 8.69. However, in this case, the relations will be nonlinear and the latter group of parameters would generally tend to be sensitive to errors in the estimates of a_1, a_2, b_0, and b_1. A more robust alternative approach is to characterize the controller dynamics in terms of its corresponding unit impulse response. This can be achieved quite easily by setting $\Delta P_{ACO2}(0)$ to 1 and $\Delta P_{ACO2}(n)$ to 0 for all $n > 0$, and computing $\Delta \dot{V}_E$ recursively from Equation 8.75. The error terms $\varepsilon(n)$ are set equal to zero during this computation. Note that this would produce results similar to the PRBS technique described in Section 8.3.3. However, one can expect much less noisy estimates of the impulse response function from the present method, since $h(n)$ in this case would be derived from only four parameters estimated from a large number (128 or higher) of data points. Further details on the application of this approach to human respiratory data and the results obtained may be found in Khoo et al. (1995) and Ghazanshahi and Khoo (1997). In the problems given at the end of this chapter, the reader will be able to explore this technique in greater detail by applying the accompanying MATLAB script file `rcs_est.m` to a number of data sets (`prbs1.mat`, `prbs2.mat`, `prbs3.mat`, `prbs4.mat`) obtained from human experiments.

8.5.3   Closed-Loop Identification of Autonomic Control Using Multivariate ARX Models

Another example of the ARX modeling approach employed in Section 8.5.2 is presented here for application to a different physiological system in which there are two inputs and one output. It is well accepted that rhythm of the heart and other pulsatile cardiovascular variables is not regular but contains a significant degree of variability. Variability of the heart period, or the inverse of heart rate, arises from two major sources: (i) respiration affects heart period through the mechanical effect of pleural pressure transmission, as well as autonomically through neural coupling between the medullary centers and the heart; (ii) fluctuations in blood pressure also affect heart period through the arterial baroreflexes, mediated by the carotid and aortic baroreceptors. Heart period is measured directly from the interval between successive R-waves in the electrocardiogram; hence, we denote changes in heart period as ΔRRI. Jo et al. (2003) postulated the following ARX model for relating changes in heart period to respiration (V) and changes in systolic blood pressure (ΔSBP):

$$\Delta\mathrm{RRI}(n) = -\sum_{i=1}^{n_a} a_i \Delta\mathrm{RRI}(n-i) + \sum_{j=0}^{n_b} b_j V(n - N_{\mathrm{RCC}} - j)$$

$$+ \sum_{k=0}^{n_c} c_k \Delta\mathrm{SBP}(n - N_{\mathrm{ABR}} - k) + e(n) \qquad (8.77)$$

Since the model includes measurements of both respiration and beat-to-beat values of heart period and systolic blood pressure, the data are "resampled" in units of time less than the minimum heart period; in the study of Jo et al. (2003), the "resampling interval" was 0.5 s. Thus, in Equation 8.77, the time base is given in terms of number of samples (n). N_{RCC} and N_{ABR} are the latencies (in units of samples) associated with the respiratory–cardiac coupling (RCC) and baroreflex (ABR) mechanisms, respectively. $e(n)$ represents the variability in ΔRRI not explained by the RCC and ABR mechanisms. The above model is similar in form to models that have been employed in other studies of heart rate variability (Barbieri et al., 2001; Baselli et al., 1988; Mullen et al., 1997).

The unknown ARX model coefficients (a_i, b_j, and c_k) in Equation 8.77 are estimated using least-squares minimization in the same way as described in Section 8.5.2.2. The least-squares minimization procedure is repeated over a range of values for the latencies (N_{RCC} and N_{ABR}) and model orders (n_a, n_b, and n_c). Thus, in a sense, Equation 8.77 represents not just a single model, but a substantial number of model candidates, each with a unique combination of n_a, n_b, n_c, N_{RCC}, and N_{ABR}. As such, we have to select the "best" among all the model candidates. For this purpose, a metric that quantifies the quality of fit between model prediction and data has to be employed. However, a model candidate that has a larger number of parameters than another that has fewer parameters is more likely to produce a better fit to the data. On

the other hand, the former model candidate with the larger number of parameters to be estimated is also likely to have larger parameter variances and cross-covariances. Rissanen (1978) introduced the minimum description length (MDL) as a measure of the quality of fit that also penalizes for increasing model complexity:

$$\text{MDL} = \log \left(\frac{\sigma_{\text{e}}^2}{\sigma_{\Delta\text{RRI}}^2} \right) + NC_{\text{tot}} \times \frac{\log (N)}{N} \tag{8.78}$$

In Equation 8.78, σ_{e}^2 is the variance of the residual errors between the measurements and the predicted ΔRRI, $\sigma_{\Delta\text{RRI}}^2$ is the variance of the measured ΔRRI, and N is the total number of data points. NC_{tot} is the total number of ARX model coefficients:

$$NC_{\text{tot}} = n_{\text{a}} + (n_{\text{b}} + 1) + (n_{\text{c}} + 1) \tag{8.79}$$

Note that MDL decreases as the fit between model predictions and data improves, but increases as the number of parameters in the model increases (i.e., as model complexity increases).

   Selection of the "optimal" candidate model is based on a global search (over all model orders and latencies) for the minimum MDL. Thus, unlike the example in Section 8.5.2, in which a fixed "structure" of the model was assumed, in this application the model is allowed to be more flexible and "data driven" by searching for that combination of number of terms (n_{a}, n_{b}, n_{c}, N_{RCC}, N_{ABR}) in Equation 8.77 that produces the lowest global value of MDL. Once the ARX coefficients for the optimal candidate model are estimated, the impulse response functions for the RCC ($h_{\text{RCC}}(n)$) and baroreflex ($h_{\text{ABR}}(n)$) components of the model can be computed from the recursive equations below, by setting the initial values of the inputs V and ΔSBP to unity and all other values to zero:

$$h_{\text{RCC}}(n) = -\sum_{i=1}^{p} a_i h_{\text{RCC}}(n - i) + \sum_{j=0}^{q} b_j V(n - N_{\text{RCC}} - j) \tag{8.80}$$

$$h_{\text{ABR}}(n) = -\sum_{i=1}^{p} a_i h_{\text{ABR}}(n - i) + \sum_{k=0}^{m} c_k \Delta\text{SBP}(n - N_{\text{ABR}} - k) \tag{8.81}$$

where $0 \leq n \leq p - 1$.

   In Equations 8.80 and 8.81, the values of $h_{\text{RCC}}(n)$ and $h_{\text{ABR}}(n)$ for all $n < 0$ are set equal to zero. It should be kept in mind also that $h_{\text{RCC}}(n)$ and $h_{\text{ABR}}(n)$ are estimated from input–output data assuming a discrete time base, and therefore any comparison to the continuous-time versions of these impulse responses, $h_{\text{RCC}}(t)$ and $h_{\text{ABR}}(t)$, will need to include a scaling factor based on the sampling interval T (see Sections 8.2.1 and 8.2.2).

   An example of the kind of data collected in the study of Jo et al. (2003) is shown in Figure 8.20. The goal of this study was to derive estimates of $h_{\text{RCC}}(n)$ and $h_{\text{ABR}}(n)$
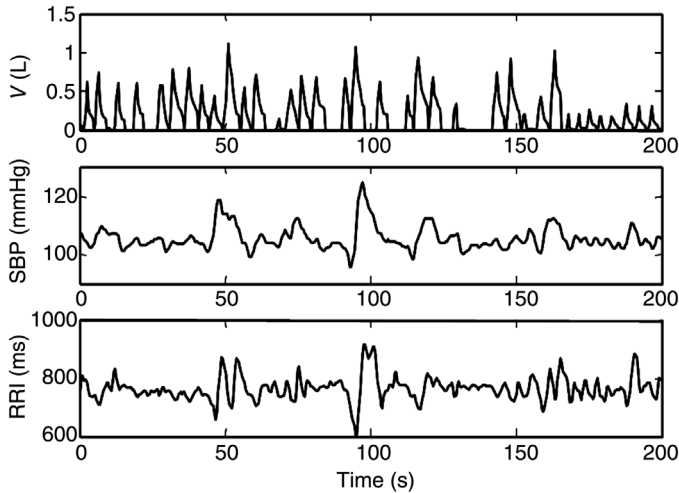
FIGURE 8.20 Waveforms of respiration (V), beat-to-beat systolic blood pressure (SBP), and heart period (RRI) measured in a sleeping subject. The subject was placed on a bilevel positive pressure ventilator that provided 5 cm H_2O pressure boost during inspiration on random breaths; this was intended to increase the variability of ventilation.

from sleeping subjects using noninvasive measurements of respiration (obtained from nasal airflow), continuous blood pressure monitoring, and electrocardiogram. In order to enhance the accuracy of parameter estimation, the pattern of breath-to-breath ventilation was made more "broadband" by connecting the subject's mask to a bilevel positive pressure ventilation that, on a random basis, would increase inspiratory pressure by 5 cm H_2O over the baseline level for one or more breaths. This intervention increased the variability of the breathing pattern, resulting in a wide range of large and small and longer and shorter breaths (Figure 8.20). This consequently induced larger variations in SBP and RRI. By applying the methodology discussed above, the impulse responses in the RCC and ABR components of heart period variability can be estimated. Examples of the estimated $h_{RCC}(n)$ and $h_{ABR}(n)$ from one of the subjects participating in the study are displayed in Figure 8.21. One can interpret $h_{RCC}(n)$ to represent the time-course in ΔRRI following a very rapid inspiration and expiration of 1 L of air. The large negative dip in $h_{RCC}(n)$ is consistent with the well-accepted observation that, in respiratory sinus arrhythmia, inspiration leads to an acceleration of heart rate (or equivalently, a reduction in RRI). $h_{ABR}(n)$ represents the time-course of ΔRRI resulting from an abrupt increase in SBP of 1 mmHg. In this case, there is an initial large positive overshoot and a subsequent smaller negative undershoot, consistent with the notion that the net baroreflex response to an increase in blood pressure is a slowing of heart rate (or equivalently, an increase in RRI). The interested reader is referred to the study of Jo et al. (2003) for further details in the modeling approach and results.
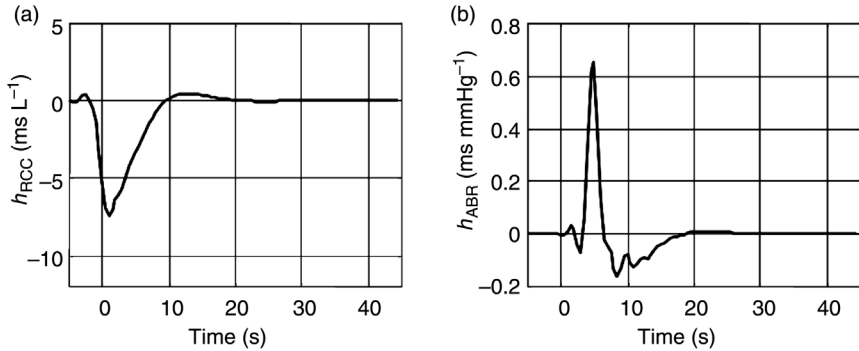
FIGURE 8.21 Estimated impulse responses of the respiratory–cardiac coupling (RCC, part (a)) and baroreflex (ABR, part (b)) components of the model of heart rate variability derived from one subject's data.

8.6   IDENTIFICATION OF PHYSIOLOGICAL SYSTEMS USING BASIS FUNCTIONS

8.6.1   Reducing Variance in the Parameter Estimates

In this chapter, we have introduced a number of techniques for system identification of parametric and nonparametric models, but these fall into either one of two general approaches. Optimization of a selected criterion function is the underlying principle for both approaches, but what distinguishes one approach from the other is the way in which the global minimum value of the criterion function is arrived at. Most parametric models are identified using multistep, iterative optimization techniques – gradient descent and simplex are examples of two classes of these methods. Nonparametric models and some parametric models can be identified using least-squares minimization or related techniques. This is possible because in such cases, the model is "linear in the parameters," and as a result, the parameter estimation process is analogous to multiple linear regression. The least-squares estimation and correlation methods introduced in Sections 8.2.2 and 8.2.3 are prime examples of this approach when the impulse response function is directly estimated from the input and output data sets. The key assumption here is that the impulse response lasts over a finite duration. Depending on the "persistence" of the true impulse response and the length of the data sets that one has to work with, the "truncation" of the impulse response can definitely bias the estimates of the impulse response coefficients. On the other hand, assuming a longer impulse response for a given total number of data samples would lead to greater variance in the estimates of the impulse response coefficients – the well-known problem of overparametrization. Employing an ARX model structure, as in Sections 8.5.2 and 8.5.3, reduces the number of model parameters that have to be estimated. On the other hand, the autoregressive structure of ARX makes it such that the disturbances are coupled to

the system dynamics. Thus, if the extraneous "noise" is large compared to the deterministic part of the observed output, the parameter estimates can be unreliable and biased. Aside from the case in which the model is used as a "one-step ahead predictor," predictions that employ only past inputs and past *predicted* outputs can diverge from the actual data.

8.6.2   Use of Basis Functions

One approach that can dramatically reduce the impact of the aforementioned problems assumes that the impulse response can be represented as a weighted sum of orthonormal basis functions. The advantage of this technique is that it greatly reduces the number of parameters to be estimated and also constrains the behavior of the impulse response, thereby reducing variance in the parameter estimates. With the improved estimation robustness, this technique can be applied on relatively short and noise-contaminated data where the inputs do not have to be strictly broadband and Gaussian. The Laguerre set has been employed in many studies of physiological systems since these functions have a built-in exponential decay that enables the constructed impulse response to mimic long-tailed distributions (Marmarelis, 1993; Chon et al., 1996; Belozeroff et al., 2002).

The impulse response for a single-input single-output linear system is expanded into the weighted sum of $q + 1$ basis functions as follows:

$$h(n) = \sum_{i=0}^{q} c(i)B_i(n), \quad 0 \le n \le p - 1 \tag{8.82}$$

If the set of basis functions are Laguerre, then

$$B_i(n) = L_i(n), \quad 0 \le i \le q \tag{8.83}$$

where

$$L_i(n) = a^{(n-i)/2}(1 - \alpha)^{1/2} \sum_{k=0}^{i} (-1)^k \binom{n}{k} \binom{i}{k} a^{i-k}(1 - \alpha)^k, \quad n \ge 0 \tag{8.84}$$

For a more efficient implementation, $L_i(n)$ can be computed recursively as follows:

$$L_0(n) = \sqrt{\alpha^n(1 - \alpha)} \tag{8.85a}$$

$$L_i(n) = \sqrt{\alpha}L_i(n - 1) + \sqrt{\alpha}L_{i-1}(n) - L_{i-1}(n - 1), \quad i > 0 \tag{8.85b}$$

The parameter $(0 < \alpha < 1)$ determines the rate of exponential decay of the Laguerre functions. α is selected based on the "memory" or effective duration $(=p)$ of the impulse response and the number of Laguerre functions used for the expansion, so
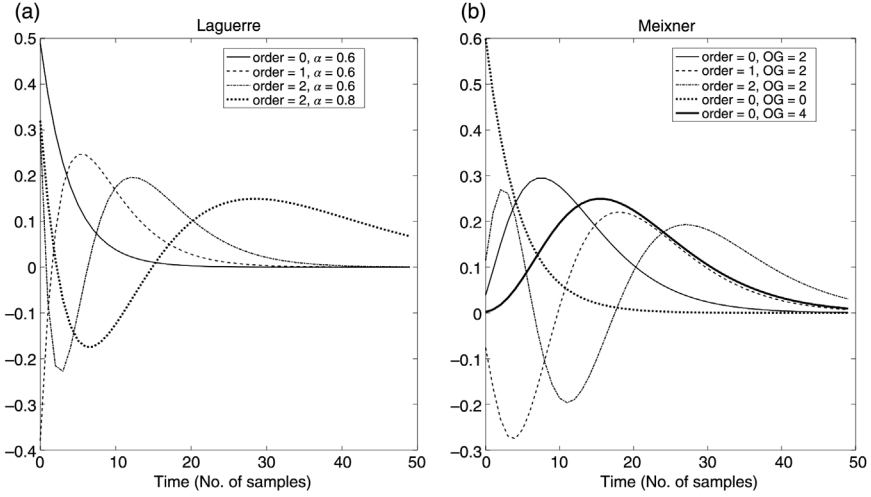
FIGURE 8.22 Basis functions used for kernel expansion in the model: Laguerre (a) and Meixner (b). See text for more details.

that all these functions would converge sufficiently close to zero toward the end of the system memory. The MATLAB function `genalphacoef` generates the α value that is consistent with the system "memory" and the maximum order of Laguerre function to be used in constructing the impulse response function. The function `laguer` generates the family of $q + 1$ Laguerre functions for a given values of α and p. Both of these functions appear in the larger MATLAB script file `laguerest.m`. Figure 8.22a displays the zeroth-, first-, second-, and third-order Laguerre functions for $\alpha = 0.6$ and a system memory of 25 samples.

Substituting Equations 8.82 and 8.83 into Equation 8.3 yields

$$y(n) = \sum_{k=0}^{p-1} h(k)u(n-k)T + e(n) = \sum_{k=0}^{p-1} \sum_{i=0}^{q} c(i)L_i(k)u(n-k)T + e(n) \quad (8.86a)$$

In Equation 8.86, if we reverse the order of the summations for indices k and i, we will get

$$y(n) = \sum_{i=0}^{q} c(i)v_i(n) + e(n) \quad (8.86b)$$

where

$$v_i(n) = \sum_{k=0}^{p-1} L_i(k)u(n-k)T \quad (8.87)$$

The problem of estimating p unknown coefficients of the impulse response function in Equation 8.86a has now been converted into a problem of estimating q ($\ll p$) unknown weights of the basis functions used in this method, as displayed in Equation 8.87. This feature is what accounts for the advantage of employing the basis function expansion method. Note that the time series $v_i(n)$, where $i = 0$, $1, \ldots, q$, represents filtered versions of the input signal, since the input is convolved with the set of basis functions that range from order 0 through q. Thus, the coefficients of the basis functions can be estimated through least-squares minimization. Once the coefficient estimates have been calculated, the impulse response function of the system to be identified can be deduced using Equation 8.82.

A generalized form of Laguerre functions, known as Meixner-like basis functions, can also be employed in the aforementioned estimation procedure (Asyali and Juusola, 2005). Like the Laguerre function, the Meixner function has an exponential decaying form, making it suitable for modeling biological impulse responses, since the impulse response of a stable system always decays to zero. A feature of the Meixner function not found in the Laguerre function is a parameter known as the "order of generalization"; this parameter allows us to have control over the rise time of the function, making it suitable for modeling impulse responses that have pure delays and exhibit slow initial dynamics. The higher the order of generalization, the slower the initial rise time. A Meixner function can be generated from the corresponding Laguerre function simply by an orthogonal transformation (Brinker, 1995). Figure 8.22b displays a number of Meixner functions of different model orders and orders of generalization.

As an example of the application of the Laguerre expansion technique, we use the same input and output "data" that were analyzed using the least-squares correlation method (implemented through `sysid_ls.m`) in Section 8.2.2. However, in this case, we apply the aforementioned Laguerre expansion technique implemented through the program `laguerest.m` accompanying this volume. As in the case of the ARX model discussed in Section 8.5.3, the "optimal" q (number of basis functions) is arrived at by determining which of the candidate models produces the minimum MDL. The results of this exercise are displayed in Figure 8.23. In Figure 8.23a, the input step in P_{ao} and the resulting response in P_A (noisy tracing) are displayed along with the best-fit model prediction (smooth tracing). Figure 8.23b shows the true impulse response (thick tracing) along with the estimated impulse response (thin tracing). Comparison with the results of using the correlation method (Figure 8.2) clearly suggests that the application of the Laguerre expansion technique has drastically reduced the variance of the parameter estimates, although it has introduced a certain degree of bias into the estimation process.

8.6.3 Baroreflex and Respiratory Modulation of Heart Rate Variability

Methods employing basis function expansions, as described in Section 8.6.2, have been applied to characterize the dynamics of the two main autonomic mechanisms that contribute to heart rate variability. Belozeroff et al. (2002, 2003) used the
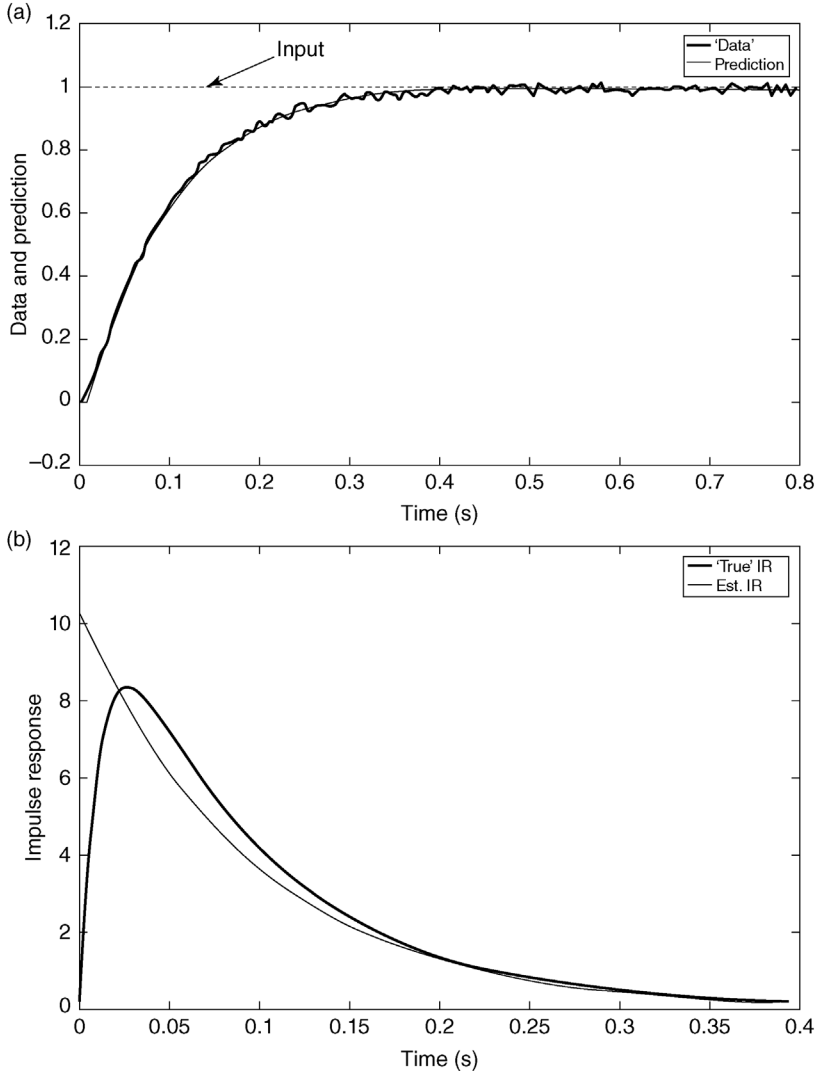
FIGURE 8.23 Results obtained from applying the Laguerre expansion technique to simulated "data" from lung mechanics model used in Figure 8.2. Part (a) shows input P_{ao} (step), the output "data," and the best fit to the data. Part (b) shows the estimated impulse response plotted along with the "true" impulse response of the model.

Laguerre expansion technique, while Chaicharn et al. (2009) employed Meixner basis functions. As in Section 8.5.3, the starting point in these studies was the following model relating changes in heart period to respiration and change in systolic blood pressure:

$$\Delta\text{RRI}(n) = \sum_{i=0}^{p-1} h_{\text{RCC}}(i)V(n - N_{\text{RCC}} - i) + \sum_{i=0}^{p-1} h_{\text{ABR}}(i)\Delta\text{SBP}(n - N_{\text{ABR}} - i) + e(n)$$

(8.88)

where $0 \leq n \leq N - 1$.

In both studies, the first step was to express each of the impulse responses as the weighted sum of the set of basis functions:

$$h_{\text{RCC}}(n) = \sum_{i=0}^{q} c_{\text{RCC}}(i)B_i(n)$$

(8.89)

$$h_{\text{ABR}}(n) = \sum_{i=0}^{q} c_{\text{ABR}}(i)B_i(n)$$

(8.90)

The system "memory" (p) assumed for the above impulse responses was 50 sampling intervals, each sampling interval being 0.5 s. Since the measurements were made noninvasively under closed-loop conditions between SBP and RRI, it was necessary to impose causality constraints on the baroreflex impulse response in an explicit fashion during the parameter estimation procedure. In the studies by Belozeroff and Chaicharn cited earlier, a minimum value of 1 sample (0.5 s) was assumed for N_{ABR}, the latency associated with the baroreflex.

Using Equations 8.89 and 8.90 in Equation 8.88 reduced the latter to the following form:

$$\Delta\text{RRI}(n) = \sum_{i=0}^{q} c_{\text{RCC}}(i)v_{\text{RCC}}(i, n) + \sum_{i=0}^{q} c_{\text{ABR}}(i)v_{\text{ABR}}(i, n) + e(n)$$

(8.91)

where

$$v_{\text{RCC}}(i, n) = \sum_{k=0}^{p-1} B_i(k)V(n - k - N_{\text{RCC}})$$

(8.92)

and

$$v_{\text{ABR}}(i, n) = \sum_{k=0}^{p-1} B_i(k)\Delta\text{SBP}(n - k - N_{\text{ABR}})$$

(8.93)

The basis function coefficients were solved using least-squares estimation for a wide range of combinations of model order (q), delays (N_{ABR} and N_{RCC}), and, in the case for Meixner functions, various orders of generalization. The "optimal" model was selected based on a search for the minimum MDL, similar to the procedure described for the multi-input ARX model in Section 8.5.3.

Figure 8.24 displays the results obtained from the study of Belozeroff et al. (2003) for estimates of the respiratory–cardiac coupling (RCC, part (a)) and baroreflex (ABR, part (b)) impulse responses estimated from 11 control (normal) subjects and 11 subjects with obstructive sleep apnea (OSA); measurements were obtained from all subjects during wakefulness. The symbols (open circles = normals, closed circles = OSA) represent the means of each subject group, with standard errors of the mean shown as error bars. The basic forms of the estimated RCC and ABR impulse responses agree well with the corresponding estimates displayed in Figure 8.21 derived from a different study (Jo et al., 2003) and collected from a subject during sleep. What is interesting is that both RCC and ABR impulse responses estimated from the OSA group were notably smaller than the corresponding impulse responses derived from the normal controls, indicating impaired autonomic control of heart rate in the OSA subjects. It should be pointed out that the individual signal characteristics of RRI and SBP by themselves did not reveal any apparent differences between the subjects of the two groups, whereas comparison of the estimated impulse responses of the RCC and ABR mechanisms showed clear differences. This result is a prime example of the class of problems where examining the dynamics of the "system" linking input and output signals can yield greater dividends than examining the waveforms of the individual signals themselves.
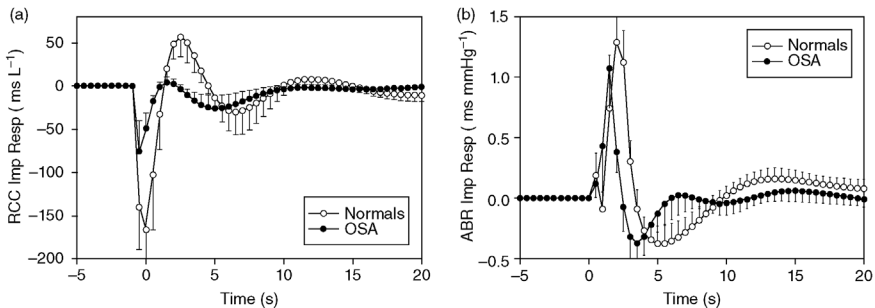
FIGURE 8.24   Respiratory–cardiac coupling (RCC, part (a)) and baroreflex (ABR, part (b)) impulse responses estimated from 11 controls (normal, shown as open circles) subjects and 11 subjects with obstructive sleep apnea (OSA, shown as closed circles) during wakefulness in the supine posture. (Reproduced with permission from Belozeroff et al. (2003).)

PROBLEMS

P8.1. One technique that has been used to assess lung mechanical function is known as the method of "forced oscillations." In one variant of this method, a loudspeaker system is used to generate random pressure perturbations (P_{ao}, considered the "input") that are directed into the subject's airways. The resulting fluctuations in airflow ($\dot{V}$, considered the "output") are measured. Using these input–output measurements, it is possible to deduce the quantities that represent airway resistance (R), airway inertance (L), and respiratory compliance (C) by assuming a linear model of respiratory mechanics, such as the linear model that we have considered previously (see Figure 4.1). Using the MATLAB script file `sensanl.m`, perform a sensitivity analysis to assess parameter identifiability. Plot sensitivity curves such as those displayed in Figure 8.7, assuming nominal parameter values of $R = 1.5$ cm H_2O s L^{-1}, $L = 0.01$ cm H_2O $s^2 L^{-1}$, and $C = 0.1$ L cm H_2O^{-1}. Use the MATLAB `randn` function to generate the white noise sequence that represents the applied forcing in P_{ao}. Assume a time-step of 0.01 s and a total duration of 25 s for each experimental trial. (*Note:* You will need to implement and solve the model differential equation in a function that will be called by `sensanl.m`.)

P8.2. The data set provided in the file `data_fo.mat` contains measurements of the input (labeled `Pao`) and output (labeled `Flow`) signals measured during an application of the method of forced oscillations, described in Problem P8.1. Assuming the respiratory mechanics model structure shown in Figure 4.1, estimate the model parameters (R, L, and C) from the input–output data. Use the optimization technique discussed in Section 8.2.5.2. It is expected that you will modify and apply the MATLAB script file `popt_llm.m`. Perform the minimization using different starting parameter estimates in order to obtain several sets of final parameter estimates. The differences in values of each parameter will give you some idea of the estimation error.

P8.3. Using the data set in `data_fo.mat` and assuming `Pao` to be the input and `Flow` to be the output, apply least-squares estimation (see Section 8.2.2) to deduce the impulse response of the corresponding system. It is expected that you will modify the MATLAB script file `sysid_ls.m`. The sampling interval is 0.01 s. Assume the number of points in the impulse response to be 50. Compute also the error band associated with the impulse response estimate.

P8.4. Use the SIMULINK model file `gmm_sim.slx` to generate 10 sets of insulin–glucose "data": In all cases, set the variance of the "measurement noise" at 36 mg^2 per 100 ml^2, but in each case set the random generator seed to a different integer. For each data set, use `gmm_est.m` to estimate the parameters p_1, p_2, p_3, and p_4 of the minimal model. Then, from the results of all 10

data sets, compute the mean and standard error associated with each of the model parameters.

P8.5. The data sets provided in the files – prbs1.mat, prbs2.mat, prbs3.mat, and prbs4.mat – represent measurements of P_{ACO2} and $\dot{V}_E$ obtained from four human subjects who were breathing from a gas mixture, the composition of which was alternated between air and 6% CO_2 in air on a pseudo-random binary basis. One set of these measurements is displayed in Figure 8.18. Using the MATLAB script file rcs_est.m, estimate in each case (a) the impulse response that characterizes the dynamics of gas exchange in the lungs; (b) the impulse response that characterizes the dynamics of the chemoreflexes; and (c) the lung-to-chemoreceptor delay. Assume the timescale to be expressed in numbers of breaths. By applying the fast Fourier transform to the impulse responses in (a) and (b), deduce the corresponding frequency responses. How much intersubject variability is there in the responses?

P8.6. Consider the linearized model of the pupillary light reflex displayed in Figure 6.13b. First, develop the SIMULINK implementation of this model, using the parameter values given in Section 6.6. Suppose we want to approximate the dynamics of this model using a simpler, open-loop model such as the generalized second-order system whose transfer function is displayed below (we will refer to this simpler model as "M2"):

$$\frac{\Delta A(s)}{\Delta I(s)} = \frac{G_{SS}\omega_n^2}{s^2 + 2\zeta\omega_n s + \omega_n^2} \qquad \text{(M2)}$$

where G_{SS} represents the steady-state gain, ζ the damping factor, and ω_n the natural frequency of M2. To do this, we determine the (unknown) model parameters (G_{ss}, ζ, and ω_n) that would give us the best fit between the output of M2 and the output of the SIMULINK model for the same input ΔI. Write a MATLAB program to accomplish this task (parameter estimation) using popt_rlc.m and fn_rlc.m as examples. For an input sequence $\Delta I(t)$ that consists of a "chirp" function (with starting frequency of 0.5 Hz and end frequency of 4 Hz over a duration of 100 s, and with an amplitude of 0.002 mlm mm^{-2}), use your SIMULINK program to simulate the response $\Delta A(t)$. Then, using $\Delta I(t)$ and $\Delta A(t)$, apply the optimization technique to estimate the parameters (G_{SS}, ζ, and ω_n) of M2. Since it is necessary to provide an initial estimate of the parameters with each optimization run, perform this parameter estimation process with at least 10 different combinations of initial parameter values.

(a) Display in a table, the initial parameter estimates, the final parameter estimates, and also the normalized mean square error* for each of the runs.

    (b) Plot an example of how the criterion function (J) changes with iteration number during the optimization process.

    (c) From the list of final parameter estimates, select what you think should be your "best" estimates of the parameters of M2 – explain your choice for these estimates.

    (d) Compare (plot) the best-fit (predicted) $\Delta A(t)$ with the simulated "data" ($\Delta A(t)$ from the SIMULINK model) as functions of time, so that visually we can see how well M2 is able to reproduce the dynamics of the pupillary light reflex model – In order to display the difference between predicted and simulated $\Delta A(t)$ clearly, show these comparisons over a span of 5 s for (i) a low-frequency region ($\sim$0.5 Hz); (ii) a moderate frequency region ($\sim$2 Hz); and (iii) a high-frequency region ($\sim$4 Hz).

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9

ESTIMATION AND CONTROL OF TIME-VARYING SYSTEMS

9.1 MODELING TIME-VARYING SYSTEMS: KEY CONCEPTS

Implicit in the physiological models that we have focused on thus far is the assumption that the model parameters are time invariant. On the one hand, this assumption may be approximately valid under "basal" steady-state conditions, when the key physiological variables are regulated so that internal conditions remain relatively stable and operate within narrow bounds, even when the system in question is perturbed by changes in the external environment. On the other hand, as we have demonstrated in the previous chapters, "homeostatic" conditions are established through the interplay of a multitude of dynamic factors that act to oppose or reinforce one another. As such, even in the "steady state," there can be significant fluctuations of various timescales around the equilibrium point of any given variable. For example, arterial blood pressure fluctuates between diastolic and systolic levels within each cardiac cycle, and although the simple Windkessel model assumes that peripheral resistance is a constant, this is clearly not true since peripheral resistance itself is altered by the baroreflex control of the peripheral vasculature. Breathing influences diastolic and systolic pressures, while over longer timescales, there are also fluctuations that result from dynamic changes in cardiac output and vascular resistance. And over even more extended timescales, blood pressure fluctuates as a by-product of the dynamic processes at play in renal autoregulation and thermoregulation. Thus, over a timescale of a few minutes, a

stationary model with constant total peripheral resistance may be sufficient to capture the dynamics of blood pressure. But over a longer timescale, such a time-invariant model is unlikely to be successful in predicting the fluctuations in blood pressure.

Figure 9.1 shows two physiological examples of time-varying or nonstationary behavior. In Figure 9.1a, mean arterial blood pressure in a chronically instrumented mouse displays diurnal fluctuations: Blood pressure is generally higher when the animal is active during the night and lower when the animal is likely to be sleeping during the day. Superimposed on the diurnal rhythm is a shorter periodicity (ultradian rhythm) that occurs when the mouse is in active state, most likely related to feeding. Finally, superimposed on these shorter fluctuations is what appears to be a downward trend over the 4 days of the recording. It is unclear if the downward trend is part of a low-frequency variation with periodicity of 16 or more days. In any case, if the blood pressure signal were to be the output of a Windkessel-type model, the parameters of the model would likely have to be time-varying in order to capture these nonstationary features. Figure 9.1b displays a different kind of time-varying behavior – a much faster type of nonstationarity. Beat-to-beat heart period or R–R interval (RRI, third panel), systolic blood pressure (SBP, fourth panel), and respiratory airflow (bottom panel) are recorded along with electroencephalogram (EEG – first panel) and muscle sympathetic nerve activity (MSNA – second panel) in a sleeping subject for about 30 s. At time $t = 0$, an acoustic stimulus is applied, arousing the subject from sleep transiently (as indicated by the increased fluctuations in EEG). This change in state elicits a surge in sympathetic activity (MSNA), as measured at the peroneal nerve just below the knee. The brief awakening decreases RRI (or equivalently, increases heart rate), increases SBP, and slightly alters respiratory airflow. After these brief changes, the system appears to settle back into sleep mode.

For nonstationarities that occur slowly, relative to the dynamics of the system in question, the simplest approach is to model the process as piecewise stationary. For example, frequency-domain (spectral) analysis of heart rate variability is generally carried out assuming conditions to be stationary. For this reason, one has to preclude data segments that contain overt transient fluctuations ("spikes" or artifactual behavior), and the consensus among most experts in the field is that one should not apply stationary analysis to RRI data segments that are longer than 5 min in duration (Task Force, 1996). If it is clear that one cannot adopt a piecewise stationary assumption, the next step up is to consider a time-varying linear model such as that illustrated in Figure 9.2. Here, the "external influence" is considered the source of nonstationary behavior. We assume that this external influence affects the linear model by altering the parameters of the model as well as through a separate additive contribution to the model output. The nonstationary influence on the model parameters can be characterized in the form of a time-varying impulse response, $h(t,\tau)$. Assuming that the time variable t associated with the nonstationarity is separable from the
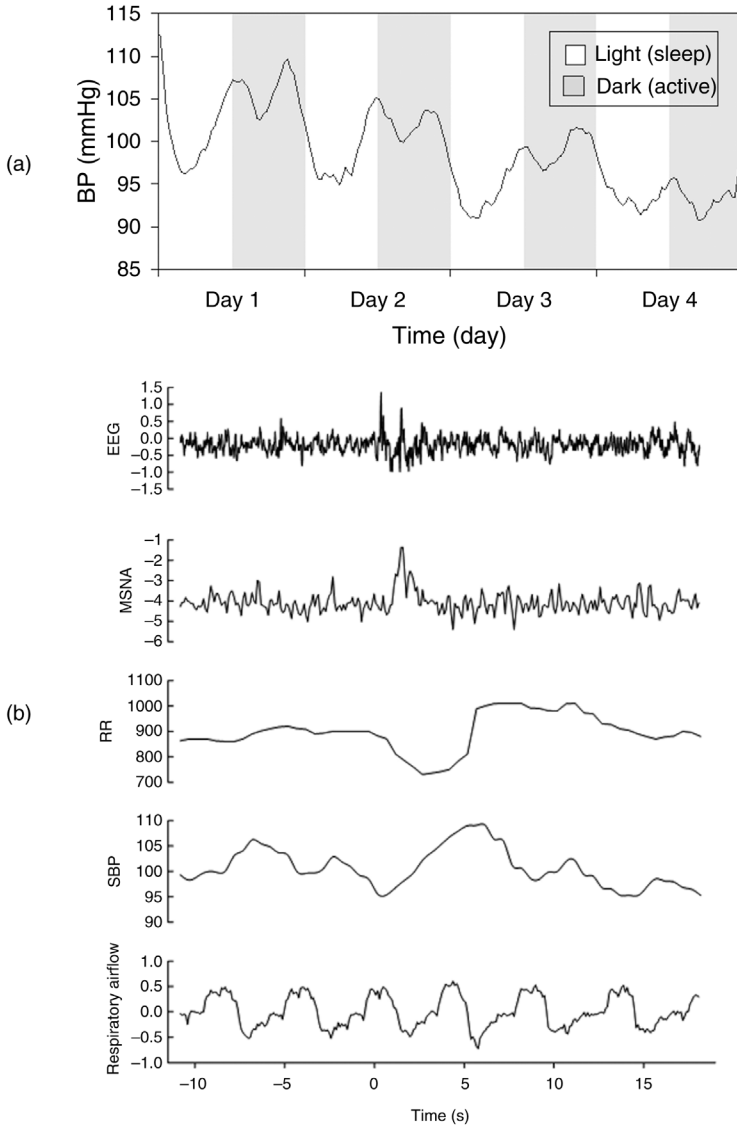
FIGURE 9.1   Physiological examples of time-varying or nonstationary behavior. (a) Mean arterial blood pressure in a chronically instrumented mouse displays diurnal fluctuations, being higher in the active period (dark) and lower during sleep (light). (Modified from Figure 1 of Chalacheva et al. (2013). (b) changes in heart period (RR), systolic blood pressure (SBP), respiration, and muscle sympathetic nerve activity in a human subject during transient arousal from sleep (indicated by the change in EEG pattern), following administration of an acoustic stimulus at time $t = 0$. (Modified from Figure 1 of Blasi et al. (2003).
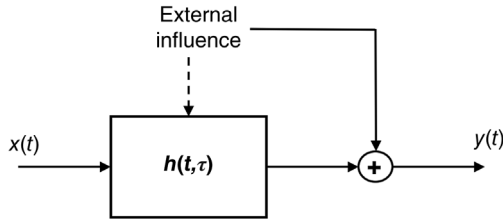
FIGURE 9.2 Schematic block diagram showing that nonstationary behavior may be the result of an external influence, and its effect on system dynamics modeled as a time-varying impulse response $h(t,\tau)$.

time variable τ associated with the dynamics of the linear system, one can visualize $h(t,\tau)$ to take the form of the three-dimensional structure as depicted in Figure 9.3 – This "structure" contains impulse responses (functions of τ that represent the system dynamics at the current time) that evolve with time t. The next question that arises is how can we estimate the time-varying impulse response from data, assuming that a time-varying linear model provides a good approximation to the underlying physiological process.
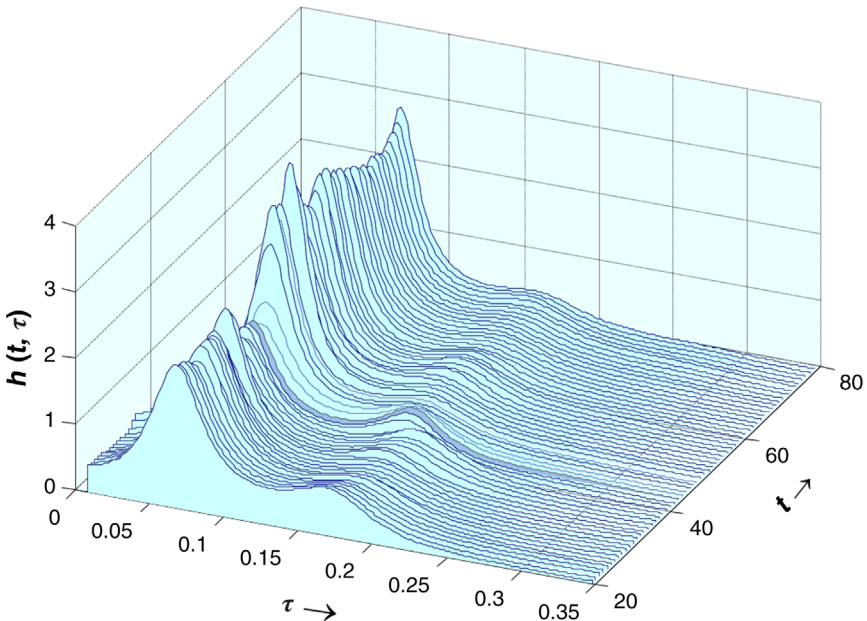
FIGURE 9.3 Three-dimensional visualization of a time-varying impulse response. τ represents the lag associated with the dynamics of the system, whereas time t is associated with the nonstationarity.

9.2 ESTIMATION OF MODELS WITH TIME-VARYING PARAMETERS

9.2.1 Optimal Estimation: The Wiener Filter

Consider the situation shown in Figure 9.4, in which we are trying to estimate the impulse response of the linear model that would best predict the output of a physiological system. The derivation to follow is easiest to demonstrate in discrete time. Thus, we have

$$\hat{y}(n) = \sum_{i=0}^{M-1} h(i)x(n-i) = \underline{\mathbf{h}}^T \underline{\mathbf{x}} \tag{9.1}$$

and

$$e(n) = y(n) - \hat{y}(n), \quad \text{where} \quad n = 0, 1, \ldots, N-1 \tag{9.2}$$

Note that the impulse response of the model is assumed to be nonzero over a finite duration of time (M samples), and it is represented by the $M \times 1$ column vector $\underline{\mathbf{h}}$. Also, $e(n)$ is the error between the measured physiological signal $y(n)$ and the model prediction $\hat{y}(n)$. Since $e(n)$ is stochastic, we can define the following cost function J based on the variance of the error (i.e., mean of the squared errors):

$$J(\underline{\mathbf{h}}) \equiv E\{e(n)^2\} \tag{9.3}$$

where $E\{\ \ldots\ \}$ represents the expectations operator, as introduced in Section 5.3.2.

Using Equations 9.1 and 9.2 in Equation 9.3, we obtain

$$J(\underline{\mathbf{h}}) = E\{(y_n - \underline{\mathbf{h}}^T)(y_n - \underline{\mathbf{h}}^T \underline{\mathbf{x}})\} \tag{9.4a}$$

$$J(\underline{\mathbf{h}}) = E\{(y_n^2 - 2\underline{\mathbf{h}}^T y_n \underline{\mathbf{x}} + \underline{\mathbf{h}}^T \underline{\mathbf{x}}\underline{\mathbf{x}}^T \underline{\mathbf{h}})\} \tag{9.4b}$$
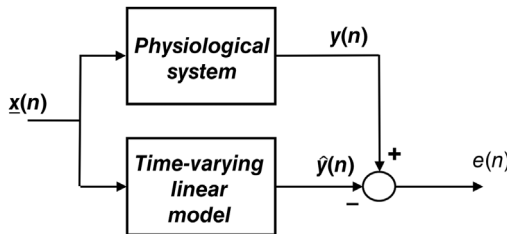
FIGURE 9.4 Using a time-varying linear system to model the behavior of a physiological system.

Applying the expectations operator to all the terms within the curly bracket, we get

$$J(\underline{h}) = \sigma_y^2 - 2\underline{h}^T\underline{r}_{yx} + \underline{h}^T\mathbf{R}_{xx}\underline{h} \tag{9.5}$$

where

$$\underline{r}_{yx} = E\{y_n\underline{r}\} = \begin{bmatrix} R_{yx}(0) & R_{yx}(1) & \cdots & R_{yx}(M-1) \end{bmatrix}^T \tag{9.6}$$

and

$$\mathbf{R}_{xx} = E\{\underline{x}\underline{x}^T\} = \begin{bmatrix} R_{xx}(0) & R_{xx}(1) & \cdots & R_{xx}(M-1) \\ \vdots & \vdots & \ddots & \vdots \\ R_{xx}(M-1) & R_{xx}(M-2) & \cdots & R_{xx}(0) \end{bmatrix} \tag{9.7}$$

Note that $\underline{r}_{yx}$ is a $(M \times 1)$ column vector and $\mathbf{R}_{xx}$ is an $(M \times M)$ matrix.

To obtain the "optimal" $\underline{h}$, we find the solution for $\underline{h}$ that yields the minimum J. Thus, we calculate the derivative of J with respect to $\underline{h}$, and set all elements in that vector to zero:

$$\frac{\partial J}{\partial \underline{h}} = -2\underline{r}_{yx} + 2\mathbf{R}_{xx}\underline{h} = 0 \tag{9.8}$$

Solving Equation 9.8 yields

$$\underline{h}_{\text{opt}} = \mathbf{R}_{xx}^{-1}\underline{r}_{yx} \tag{9.9}$$

Inserting Equation 9.9 into Equation 9.5 leads to the result

$$J_{\text{min}} = \sigma_y^2 - \underline{h}^T\underline{r}_{yx} \tag{9.10}$$

Equation 9.6 is known as the *Wiener–Hopf solution*, and the vector $\underline{h}_{\text{opt}}$ contains the impulse response of the *Wiener filter*.

9.2.2   Adaptive Estimation: The LMS Algorithm

The Wiener filter is a stationary system. To extend the filter for applicability to nonstationary systems, we allow $\underline{h}$ to change with time:

$$\underline{h}(n+1) = \underline{h}(n) + \Delta\underline{h}(n) \tag{9.11}$$

Assuming the current time $t = nT$, where T is the time-step between samples, Equation 9.11 shows how h is updated at time $t = (n+1)T$. There are several

approaches to determining the specific composition of the parameter update term $\Delta\underline{\mathbf{h}}(n)$.

We begin here with an algorithm based on steepest descent; this is the "least mean squares" (LMS) method, first developed by Widrow and Hoff. Starting with Equation 9.3, we can obtain an alternative expression for $\partial J/\partial\underline{\mathbf{h}}$:

$$\frac{\partial J}{\partial\underline{\mathbf{h}}} = 2e(n)\frac{\partial e(n)}{\partial\underline{\mathbf{h}}} \tag{9.12}$$

But since

$$e(n) = y(n) - \underline{\mathbf{h}}^T\underline{\mathbf{x}}(n) \tag{9.13}$$

we find that

$$\frac{\partial e(n)}{\partial\underline{\mathbf{h}}} = -\underline{\mathbf{x}}(n) \tag{9.14}$$

Substituting Equation 9.14 into Equation 9.12, we obtain

$$\frac{\partial J}{\partial\underline{\mathbf{h}}} = -2e(n)\underline{\mathbf{x}}(n) \tag{9.15}$$

In order to find the minimum J in an interactive manner, it would make most sense to choose $\Delta\underline{\mathbf{h}}(n)$ such that

$$J(\underline{\mathbf{h}}(n) + \Delta\underline{\mathbf{h}}(n)) < J(\underline{\mathbf{h}}(n)) \tag{9.16}$$

If we expand the left-hand side of Equation 9.16 as a Taylor series, and assuming that the norm of $\Delta\underline{\mathbf{h}}(n)$ is small, we obtain the following approximate result:

$$J(\underline{\mathbf{h}}(n) + \Delta\underline{\mathbf{h}}(n)) \approx J(\underline{\mathbf{h}}(n)) + \Delta\underline{\mathbf{h}}(n)^T\frac{\partial J(\underline{\mathbf{h}}(n))}{\partial\underline{\mathbf{h}}} \tag{9.17}$$

To guarantee the inequality in Equation 9.16, it is clear that the second term on the right-hand side of Equation 9.17 must be negative. One way to do this is to let

$$\Delta\underline{\mathbf{h}}(n) = -\mu\frac{\partial J(\underline{\mathbf{h}}(n))}{\partial\underline{\mathbf{h}}} \tag{9.18}$$

Note that μ is a positive constant that controls the size of the parameter update. μ needs to be small enough to suppress the influence of the higher order terms in the Taylor expansion (Equation 9.17) and thus guarantee convergence to the minimum
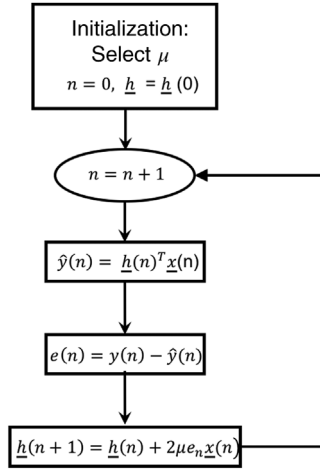
FIGURE 9.5   Flowchart for the LMS (least mean squares) algorithm for adaptive filtering.

J. In fact, it has been shown that the more noisy the observations $y(n)$, the smaller the μ needs to be.

Substituting Equation 9.18 into the second term in Equation 9.17 yields

$$\Delta\underline{\mathbf{h}}(n)^T \frac{\partial J(\underline{\mathbf{h}})}{\partial \underline{\mathbf{h}}} = -\mu \left[\frac{\partial J(\underline{\mathbf{h}}(n))}{\partial \underline{\mathbf{h}}}\right]^T \left[\frac{\partial J(\underline{\mathbf{h}}(n))}{\partial \underline{\mathbf{h}}}\right] \tag{9.19}$$

Since the norm of $\left[\frac{\partial J(\underline{\mathbf{h}}(n))}{\partial \underline{\mathbf{h}}}\right]^T \left[\frac{\partial J(\underline{\mathbf{h}}(n))}{\partial \underline{\mathbf{h}}}\right]$ must be positive, Equation 9.18 ensures that the inequality in Equation 9.16 is satisfied. In the LMS algorithm, we combine Equations 9.15 and 9.18 to give the result:

$$\Delta\underline{\mathbf{h}}(n) = 2\mu e(n)\underline{\mathbf{x}}(n) \tag{9.20}$$

The flowchart of the LMS algorithm is displayed in Figure 9.5.

9.2.3   Adaptive Estimation: The RLS Algorithm

Another widely used method for estimation of linear systems with time-varying parameters is the recursive least-squares (RLS) algorithm. In the RLS method, the criterion function to be minimized assumes the form:

$$J(\underline{\mathbf{h}}(n)) = \sum_{i=0}^{n} \lambda^{n-i} e(i)^2 \tag{9.21}$$

where $e(i)$ represents the error between observed and predicted values of the model output. λ is a constant factor (with values between 0 and 1). In effect, λ provides a means of assigning relative weights to more recent observations and less to observations further in the past. For instance, if λ is given a value close to zero, then $J(n)$ will be determined predominantly by $e(n)$ with virtually no influence from past errors. At the other extreme, when λ is equal to unity, $J(n)$ will be equal to the sum of squares of the residuals for the current and all past samples. In general, the weight given to the observations decays exponentially with time, the further back in the past they are from the current instant. For this reason, λ is often referred to as the *forgetting factor*. In practice, λ generally ranges from 0.95 to 0.9995. Lower values of λ tend to produce more variability in the estimates of the model parameters.

To find the minimum J, we differentiate J with respect to the vector of parameters, and set the derivative to equal the null vector:

$$\frac{\partial J}{\partial \underline{\mathbf{h}}} = -2 \sum_{i=0}^{n} \lambda^{n-i} e(i) \underline{\mathbf{x}}(i) = 0 \tag{9.22}$$

Replacing $e(i)$ in Equation 9.22 with Equation 9.13, we obtain

$$\sum_{i=0}^{n} \lambda^{n-i} \left\{ y(i) - \underline{\mathbf{h}}_{\text{opt}}^{T} \underline{\mathbf{x}}(i) \right\} \underline{\mathbf{x}}(i) = 0 \tag{9.23a}$$

Rearranging terms, Equation 9.23a becomes

$$\left[\sum_{i=0}^{n} \lambda^{n-i} \underline{\mathbf{x}}(i) \underline{\mathbf{x}}(i)^{T} \right] \underline{\mathbf{h}}_{\text{opt}}(n) = \sum_{i=0}^{n} \lambda^{n-i} y(i) \underline{\mathbf{x}}(i) \tag{9.23b}$$

From Equation 9.23b, we derive the following expression for the optimal value of the parameter vector:

$$\underline{\mathbf{h}}_{\text{opt}}(n) = \mathbf{R}^{-1}(n) \underline{\mathbf{r}}(n) \tag{9.24}$$

where

$$\mathbf{R}(n) = \sum_{i=0}^{n} \lambda^{n-i} \underline{\mathbf{x}}(i) \underline{\mathbf{x}}(i)^{T} \tag{9.25a}$$

and

$$\underline{\mathbf{r}}(n) = \sum_{i=0}^{n} \lambda^{n-i} y(i) \underline{\mathbf{x}}(i) \tag{9.26a}$$

We take the following steps to convert Equation 9.24 into a form that is recursive in n and thus implementable as an adaptive algorithm. Equation 9.25a can be rewritten as

$$\mathbf{R}(n) = \underline{\mathbf{x}}(n)\underline{\mathbf{x}}(n)^T = \lambda \sum_{i=0}^{n-1} \lambda^{n-i-1} \underline{\mathbf{x}}(i)\underline{\mathbf{x}}(i)^T \qquad (9.25b)$$

Then, applying Equation 9.25a again in the second term of the right-hand side of Equation 9.25b, we get

$$\mathbf{R}(n) = \underline{\mathbf{x}}(n)\underline{\mathbf{x}}(n)^T + \lambda \mathbf{R}(n-1) \qquad (9.25c)$$

In similar vein, Equation 9.26a can also be rewritten:

$$\underline{\mathbf{r}}(n) = y(n)\underline{\mathbf{x}}(n) + \lambda\underline{\mathbf{r}}(n-1) \qquad (9.26b)$$

If we define the matrix $\mathbf{P}(n)$ to be the inverse of $\mathbf{R}(n)$,

$$\mathbf{P}(n) \equiv \mathbf{R}^{-1} \qquad (9.27)$$

then, using the *matrix inversion lemma* (also known as the Sherman–Morrison–Woodbury formula), it can be shown (through a somewhat complicated proof) that $\mathbf{P}(n)$ can be expressed in the recursive form:

$$\mathbf{P}(n) = \frac{1}{\lambda}\left[\mathbf{P}(n-1) - \frac{\mathbf{P}(n-1)\underline{\mathbf{x}}(n)\underline{\mathbf{x}}(n)^T\mathbf{P}(n-1)}{\lambda + \underline{\mathbf{x}}(n)^T\mathbf{P}(n-1)\underline{\mathbf{x}}(n)}\right] \qquad (9.28)$$

Using Equation 9.26b and 9.28 in Equation 9.24, we obtain, after some algebraic manipulation, the following parameter update equation:

$$\underline{\mathbf{h}}(n) = \underline{\mathbf{h}}(n-1) + \underline{\mathbf{K}}(n)\varepsilon(n) \qquad (9.29)$$

where

$$\underline{\mathbf{K}}(n) = \frac{\mathbf{P}(n-1)}{\lambda + \underline{\mathbf{x}}(n)^T\mathbf{P}(n-1)\underline{\mathbf{x}}(n)}\underline{\mathbf{x}}(n) \qquad (9.30)$$

and

$$\varepsilon(n) = y(n) - \underline{\mathbf{h}}(n-1)^T\underline{\mathbf{x}}(n) \qquad (9.31)$$

It is important to distinguish the variable $\varepsilon(n)$, given by Equation 9.31 from the error $e(n)$ that is defined by Equation 9.13, even though the expressions look almost

identical. $\varepsilon(n)$ is the error between the current observation $y(n)$ and the model prediction based on the estimate of the model parameters computed at the previous time-step $\underline{h}(n-1)$). On the other hand, $e(n)$ is the error between $y(n)$ and the model prediction after the parameters of the model have been updated to the current time n. For this reason, $\varepsilon(n)$ is often referred to as the *a priori error*, whereas $e(n)$ is the *a posteriori error*. Equations 9.28–9.31 form the crux of the RLS algorithm, allowing estimates of the model parameters to be updated at each new time-step after the most current observation $y(n)$ becomes available: $\varepsilon(n)$ is calculated based on the previous estimate of the model parameters, and this error is multiplied by a "gain" vector $\underline{K}(n)$ and added to the previous parameter vector to obtain the latest estimate. $\underline{K}(n)$ is known as the *Kalman gain*. $\underline{K}(n)$ depends on $\mathbf{P}(n-1)$, which turns out to be the *parameter error covariance matrix* (see Section 8.3.2). Figure 9.6 displays the flowchart of the RLS estimation algorithm.

The RLS algorithm takes on a form that is very similar to the more ubiquitous *Kalman filter*, and one might think of RLS as a special case of the Kalman filter. The Kalman filter is generally derived assuming a state-space model structure, in which the model parameters evolve in time as the state equation, while the observation equation is used to relate the observed measurements to the model predictions. Thus, there are additional terms that represent the covariance matrix of the noise component of the state equation, as well as the covariance matrix of the observation noise. The RLS filter does not contain these parameters, but it does have the forgetting factor λ. The original version of the Kalman filter was
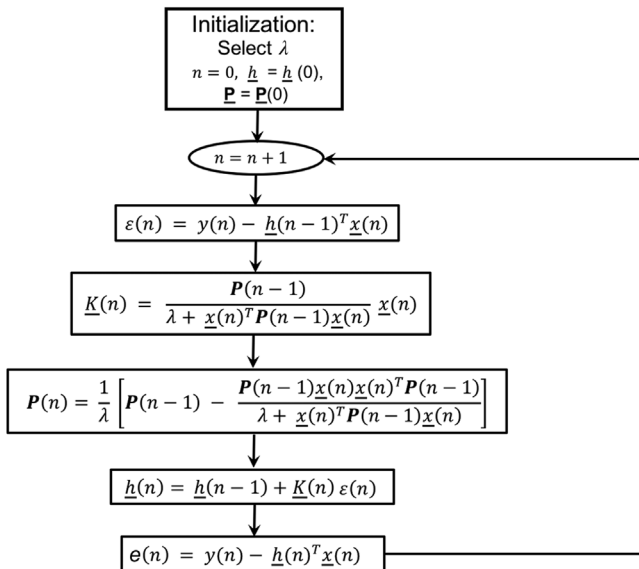
FIGURE 9.6 Flowchart for the RLS (recursive least squares) algorithm for estimating the parameters of a time-varying system.

not formulated as an adaptive filter, but it can be extended for application to time-varying systems.

9.3 ESTIMATION OF TIME-VARYING PHYSIOLOGICAL MODELS

9.3.1 Extending Adaptive Estimation Algorithms to Other Model Structures

In the previous sections, we assumed a finite impulse response (FIR) structure for the model to be estimated. As pointed out in Chapter 8, one problem with this kind of model structure is that there are usually many coefficients in an impulse response – especially when the system response is sluggish but the sampling rate is quite high. This generally leads to significant error in the estimated impulse response (see Section 8.2.2). To circumvent this problem, we can adopt model structures that can adequately characterize the underlying dynamics of the system being modeled with substantially fewer parameters. One type of "structure" that we introduced in the example in Section 8.5.2 was the ARX (autoregressive with exogenous input) model. Another approach that we can adopt is to employ orthonormal basis functions to reduce the level of parametrization, as outlined in Section 8.6. In the following examples, we demonstrate how the RLS algorithm can be applied using these model formulations.

9.3.2 Adaptive Estimation of Pulmonary Gas Exchange

We adopt the linearized model of CO_2 exchange in the lungs derived from Equation 6.42; this takes the form very similar to Equation 8.63, but we will assume here that the subject is breathing air and thus $P_{ICO2} = 0$.

$$\tau_{lung} \frac{d\left(\Delta P_{ACO_2}\right)}{dt} + \Delta P_{ACO_2} = -G_2 \Delta \dot{V}_E \qquad (9.32a)$$

where

$$\tau_{lung} = \frac{V_{lung}}{\dot{V}_E - \dot{V}_D + 863 Q K_{CO_2}} \qquad (9.32b)$$

and

$$G_2 = \frac{P_{ACO_2}}{\dot{V}_E - \dot{V}_D + 863 Q K_{CO_2}} \qquad (9.32c)$$

Since the measurements are made on a breath-by-breath basis, we can use any of the methods introduced in Chapter 7 to convert the continuous-time system in Equation 9.32a into its discrete-time equivalent. As in Section 8.5.2, we obtain the following ARX representation:

$$\Delta P_{ACO_2}(n) + \alpha \Delta P_{ACO_2}(n-1) = \beta \Delta \dot{V}_E(n) + e(n) \tag{9.33a}$$

where

$$\alpha = -e^{-T/\tau_{\text{lung}}} \tag{9.33b}$$

and

$$\beta = -\frac{G_2}{\tau_{\text{lung}}} \tag{9.33c}$$

and n represents the current breath. Note that T is the average breath duration. And α and β are related to the parameters in the continuous-time model via Equations 9.33b and 9.33c, if we employ the impulse invariance method. To allow the model to accommodate time-varying effects in the parameters, we let α and β be functions of time (in breaths). Thus, we have the following time-varying ARX model:

$$\Delta P_{ACO_2}(n) + \alpha(n)\Delta P_{ACO_2}(n-1) = \beta(n)\Delta \dot{V}_E(n) + e(n) \tag{9.34}$$

We can rewrite Equation 9.34 in a form similar to Equation 9.31:

$$e(n) = y(n) - \underline{\theta}(n)^T \underline{x}(n) \tag{9.35a}$$

where

$$y(n) = \Delta P_{ACO_2}(n) \tag{9.35b}$$

$$\underline{\theta}(n) = [-\alpha(n)\beta(n)]' \tag{9.35c}$$

and

$$\underline{x}(n) = \left[\Delta P_{ACO_2}(n-1)\Delta \dot{V}_E(n)\right]' \tag{9.35d}$$

The criterion function for minimization is defined as

$$J\big(\underline{\theta}(n)\big) = \sum_{i=0}^{n} \lambda^{n-i} e(i)^2 \tag{9.36}$$

The estimation procedure using RLS follows the same sequence as was presented in Section 9.2.3 and illustrated in the flowchart in Figure 9.6. The steps are as follows:

(a) Assume a value for the forgetting factor λ, generally between 0.97 and 0.995. Start with initial educated guesses for the parameters (α and β) to be estimated, as well as the parameter error covariance matrix $\mathbf{P}$.

(b) Using previous parameter estimates, generate the predicted y (y_{pred}), and thus compute the *a priori* error ε:

$$\varepsilon(n) = y(n) - \underline{\boldsymbol{\theta}}(n-1)^T \mathbf{x}(n) \tag{9.37a}$$

(c) Compute the Kalman gain $K(n)$ from $\varepsilon(n)$ and $\mathbf{P}(n-1)$ using Equation 9.30.

(d) Obtain updated $\mathbf{P}(n)$ using Equation 9.28.

(e) Calculate the updated estimate of the parameter vector, using

$$\underline{\boldsymbol{\theta}}(n) = \underline{\boldsymbol{\theta}}(n-1) + \underline{\mathbf{K}}(n)\varepsilon(n) \tag{9.37b}$$

(f) Update the prediction error (*a posteriori* error) using the updated parameters in

$$e(n) = y(n) - \underline{\boldsymbol{\theta}}(n)^T \mathbf{x}(n) \tag{9.37c}$$

(g) Increment n by 1 and repeat the sequence starting at step (b).
In this example, the parameter error covariance matrix $\mathbf{P}$ is 2×2 and the Kalman gain vector is 2×1 in dimensions. The MATLAB script file that incorporates this algorithm is named TVmodel_RLS.m.

To demonstrate the application of TVmodel_RLS.m, we use the ARX model given in Equation 9.33a to simulate the ventilation–CO_2 exchange process, and subsequently employ the RLS algorithm to estimate the original parameters of the ARX model. The advantage of testing the estimation algorithm with "data" simulated by a model is that it allows us to check the estimated parameters against "ground truth." In the example shown in Figure 9.7, the "data"-generating model assumes ARX parameter values of $\alpha = -0.7$ and $\beta = -0.14$ for the first half of the simulation (breaths 1 through 250) and an abrupt change in these parameters to $\alpha = -0.9$ and $\beta = -0.28$ in the second half of the simulation (breaths 251 through 500). The ARX model is assumed to incorporate Gaussian white noise ($e(n)$ in Equation 9.33a) with standard deviation of 0.2 mmHg, while the ventilation input also takes the form of a Gaussian white noise sequence (not correlated with the former white noise sequence) with standard deviation of $1\,\mathrm{L\,min^{-1}}$, representing breath-by-breath variability in
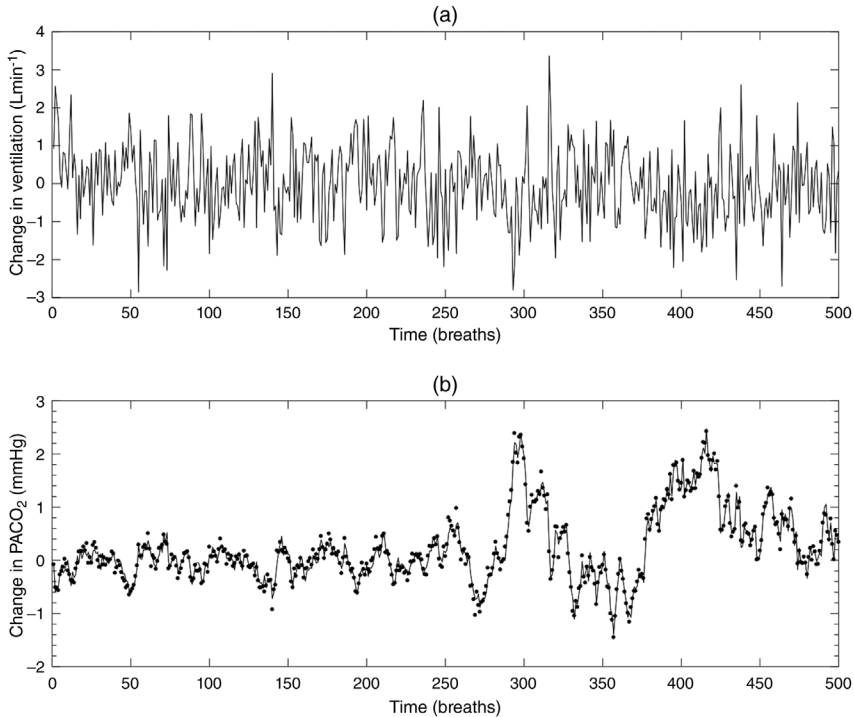
FIGURE 9.7 Application of RLS for estimation of the parameters of a time-varying model of lung gas exchange. Ventilation (part (a)) is considered the input to the model, and changes in alveolar P_{CO2} are the output (part (b), closed circles). The predicted change in P_{ACO2} is shown as the black continuous tracing that tracks the closed circles rather closely.

spontaneous breathing. In reality, changes in gas exchange are unlikely to be as large and as abrupt as simulated in Figure 9.7. However, applying the RLS estimation algorithm to the simulated "data set" offers a way in which we can determine how quickly the parameter estimates will adapt toward the "true values," given an assumed value of the forgetting factor λ. In this example, we employed a value of 0.98 for λ. The breath-to-breath values of the simulated "noisy data," that is, ΔP_{ACO2}, are displayed as closed circles in Figure 9.7b. The model predicted ΔP_{ACO2} is shown as the continuous dark tracing. The estimated time-varying ARX model parameters are displayed as the continuous tracings in Figure 9.8, with α in part (a) and β in part (b). The "true" values of α and β are displayed in Figure 9.8 as broken lines. Note that the parameter estimates begin far away from their "true" values and converge toward these values after 10–20 breaths. Following the abrupt change in values of the "true" parameter values, the estimate of α adapts to its new value within 50 breaths, but the estimate of β
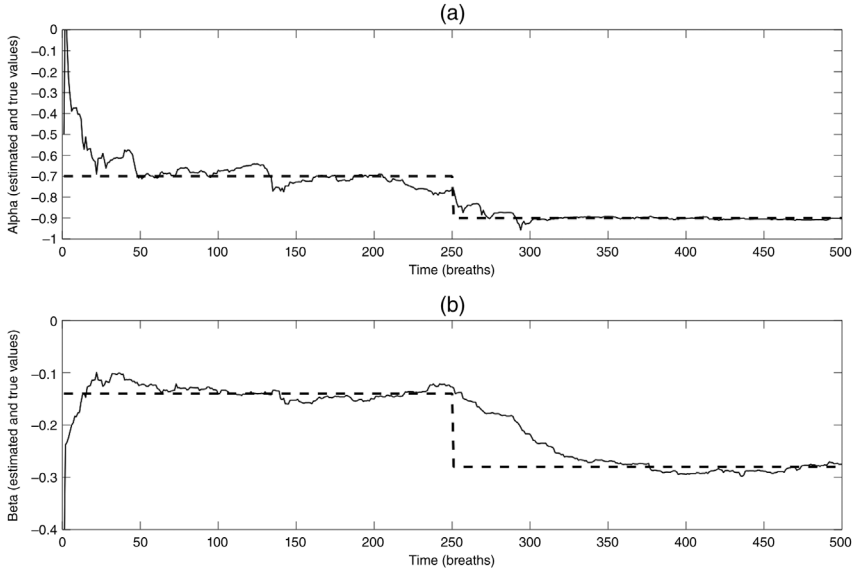
FIGURE 9.8   Plots of the estimated parameters α (part (a)) and β (part (b)) of the time-varying ARX model against their corresponding "true" values.

converges more slowly. Convergence behavior will differ depending on λ and the level of noise in the data. Smaller values of λ lead to faster convergence but also considerably more variability from one point to the next.

9.3.3   Quantifying Transient Changes in Autonomic Cardiovascular Control

In Section 8.6.3, we introduced the minimal model that characterized the respiratory and baroreflex modulation of heart period and employed the Laguerre function expansion technique to estimate the impulse responses of these two main mechanisms of heart rate variability. This approach can be easily adapted for applicability to nonstationary (time-varying) conditions. As illustrated in Figure 9.1b, cardiovascular and respiratory fluctuations that accompany arousal from sleep are a common example of time-varying behavior. This raises the question of whether these cardiorespiratory fluctuations are also associated with transient changes in the baroreflex and respiratory–cardiac coupling mechanisms that mediate heart rate variability. The model discussed in Section 8.6.3 can allow us to examine this question if we relax the assumption of stationarity and let the RCC and ABR impulse responses be time-varying:

$$\Delta\mathrm{RRI}(n) = \sum_{i=0}^{p-1} h_{\mathrm{RCC}}(i,n)V(n-N_{\mathrm{RCC}}-i) + \sum_{i=0}^{p-1} h_{\mathrm{ABR}}(i,n)\Delta\mathrm{SBP}(n-N_{\mathrm{ABR}}-i) + e(n)$$

$$(9.38)$$

The time-varying impulse responses are each expanded as the weighted sum of basis functions, but here the coefficients are allowed to be time-varying:

$$h_{\text{RCC}}(i,n) = \sum_{k=0}^{q} c_{\text{RCC}}(k,n)B_k(i), \quad 0 \le i \le p-1 \qquad (9.39)$$

$$h_{\text{ABR}}(i,n) = \sum_{k=0}^{q} c_{\text{ABR}}(k,n)B_k(i), \quad 0 \le i \le p-1 \qquad (9.40)$$

Substituting Equations 9.39 and 9.40 into Equation 9.38 yields

$$\Delta\text{RRI}(n) = \sum_{k=0}^{q} c_{\text{RCC}}(k,n)v_{\text{RCC}}(k,n) + \sum_{k=0}^{q} c_{\text{ABR}}(k,n)v_{\text{ABR}}(k,n) + e(n) \quad (9.41)$$

where

$$v_{\text{RCC}}(k,n) = \sum_{i=0}^{p-1} B_k(i)V(n-i-N_{\text{RCC}}) \qquad (9.42)$$

and

$$v_{\text{ABR}}(k,n) = \sum_{i=0}^{p-1} B_k(i)\Delta\text{SBP}(n-i-N_{\text{ABR}}) \qquad (9.43)$$

$v_{\text{RCC}}(k,n)$ and $v_{\text{ABR}}(k,n)$ are the new input time series that have been derived from having the respiration and ΔSBP inputs convolved with each member of the family of $q+1$ basis functions. Thus, Equation 9.41 takes on a form equivalent to Equation 9.37c in which the time-varying coefficients $c_{\text{RCC}}(k,n)$ and $c_{\text{ABR}}(k,n)$ form the elements of the time-varying parameter vector $\underline{\theta}(n)$, while input vector is composed of $v_{\text{RCC}}(k,n)$ and $v_{\text{ABR}}(k,n)$. The RLS algorithm is applied to estimate $\underline{\theta}(n)$ in a manner similar to what was described previously in Section 9.3.2. Having estimated the time-varying coefficients $c_{\text{RCC}}(k,n)$ and $c_{\text{ABR}}(k,n)$, the corresponding time-varying impulse responses $h_{\text{RCC}}(i,n)$ and $h_{\text{ABR}}(i,n)$ can be derived using Equations 9.39 and 9.40. The final outcome of these calculations would take the form of 3D plots such as that displayed in Figure 9.3, with "slices" that represent the impulse response at any given time and the stacks of these slices representing how the impulse response in question varies with time. Since it is generally easier to display changes in 2D format, one can extract salient features from the impulse response and track how these features change with time. One feature that is simple to extract is the peak amplitude or impulse response magnitude (IRM), which we have defined as the magnitude of the difference between the first successive maximum and
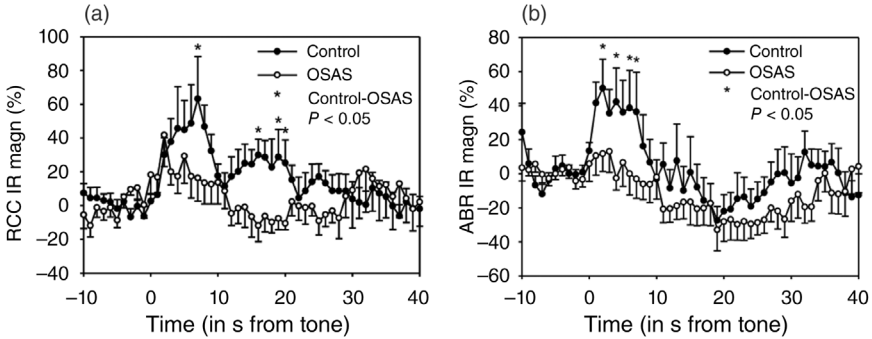
FIGURE 9.9   Changes in the estimated RCC (part (a)) and ABR (part (a)) impulse response magnitudes from a group of normal controls (closed circles) and subjects with obstructive sleep apnea syndrome (OSAS) following transient arousal from sleep with an acoustic stimulus.

minimum values of the impulse response (see Section 4.4.2). Figure 9.9 displays the results obtained by applying the above adaptive estimation technique to respiration, ΔSBP and ΔRRI data collected from a group of sleeping subjects with obstructive sleep apnea (OSA) and normal controls during which they were transiently aroused from sleep by a brief acoustic stimulus (Khoo and Blasi, 2013). The subjects with OSA were placed on continuous positive airway pressure during the study to stabilize their upper airways, allowing them to attain a steady level of quiet sleep before the acoustic stimulus was applied. As shown in Figure 9.9, the estimated RCC and ABR impulse responses displayed a substantial increase ($\sim$40–50% above baseline) over a 10 s duration following the start of the tone-induced arousal in the control subjects, but not in the OSA subjects.

The MATLAB program `TVmodel_RLS.m` can be modified to incorporate the basis function expansion technique discussed in this section and the reader is challenged to develop the equivalent code for the time-varying two-input model of heart rate variability in Problem P9.2. An important detail that should be mentioned is that the model order q (number of basis functions to use) and the delays in the model, N_{RCC} and N_{ABR}, should first be determined based on the stationary version of the model (see Section 8.6.3), using "baseline" data before the point when time-varying changes are expected to occur. This can be done using MDL as the criterion function to minimize. This provides a "fixed" model structure whose parameters are allowed to be time-varying when the algorithm is applied to the section of data to be analyzed. The alternative path of allowing the model structure to evolve with time would inject unpredictable complications into the parameter estimation process. The reader who is interested in employing the above method of time-varying model estimation is referred to the published studies by Blasi et al. (2006) and Chaicharn et al. (2009).

9.4 ADAPTIVE CONTROL OF PHYSIOLOGICAL SYSTEMS

9.4.1 General Considerations

The basic features that distinguish adaptive control from simple feedback control are the addition of a parameter estimator to determine the changes in dynamics of the unknown plant and a control law that uses an optimization algorithm to select the control signal that is optimally adjusted for the altered plant dynamics. Most adaptive control schemes require a model of the plant. Therefore, the accuracy and reliability with which this model characterizes plant dynamics are key factors that govern how well the adaptive control system will work in practice.

The two major types of adaptive controllers employed in online physiological control and closed-loop drug delivery schemes are illustrated in Figure 9.10. In the clinical setting, there is always considerable variability in plant dynamics across subjects as well as within an individual subject at different times. The *multiple*
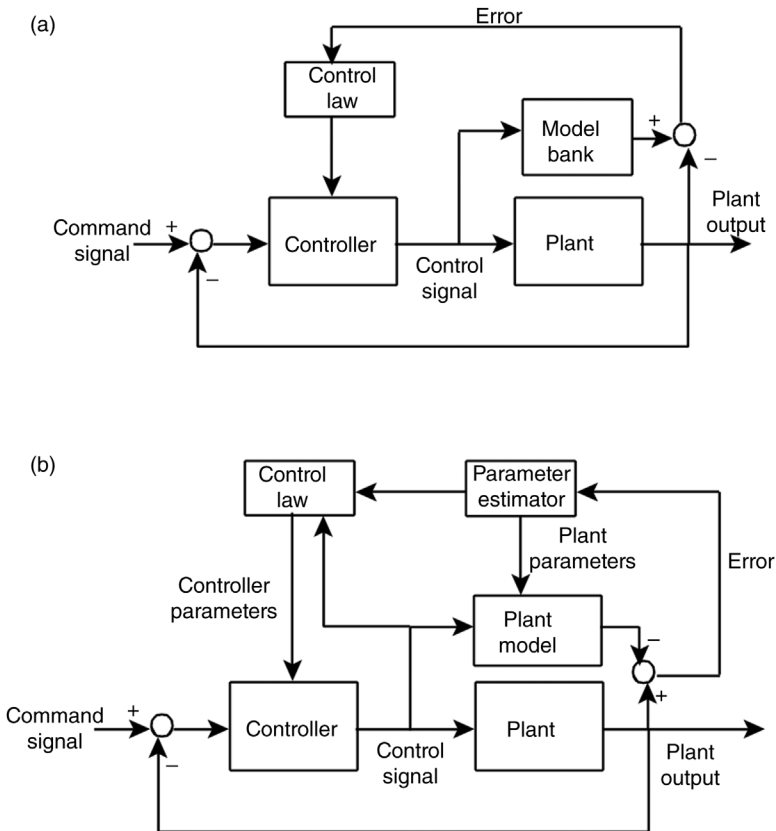
FIGURE 9.10 Schematic block diagrams of (a) a multiple-model adaptive control (MMAC) system, and (b) a model reference adaptive control (MRAC) system.

model adaptive control (MMAC) system, shown in Figure 9.10a, allows a finite range of representations of plant states by containing a model bank. Constraints are placed on the parameters of each model employed, so the controller responses remain reasonable and bounded. One disadvantage of this approach is that it requires significant knowledge of the plant dynamics. Another is that the controller may not be able to handle plant behavior that lies beyond the range specified in the model bank. The *model reference adaptive control* (MRAC) system, on the other hand, uses a single general model of the plant (Figure 9.10b). Thus, it can be more versatile. However, there is no guarantee of stability for the parameter estimates and for the physiological variable being controlled. Both these types of adaptive control schemes have been used in a variety of closed-loop drug delivery applications, including blood pressure control, neuromuscular blockade, and control of blood glucose level (Katona, 1982; Martin et al., 1987; Olkkola and Schwilden, 1991; Fischer et al., 1987). A detailed consideration of adaptive control theory and its applications lies beyond the scope of this text. For this, the reader is referred to a number of excellent volumes, such as Harris and Billings (1981), Astrom and Wittenmark (1989), and Haykin (2013).

9.4.2   Adaptive Buffering of Fluctuations in Arterial P_{CO_2}

In this section, we consider to some degree of detail a biomedical example that illustrates how adaptive control theory can be implemented in practice. The problem at hand concerns the considerable amount of breath-to-breath variability that has been observed in spontaneous ventilation. Accompanying this ventilatory varia-bility are the corresponding fluctuations in alveolar and, therefore, arterial P_{CO_2}. Modarreszadeh et al. (1993) addressed the issue of buffering these fluctuations in arterial P_{CO_2} in an optimal manner by changing the CO_2 composition of the inhaled gas (F_{ICO_2}) breath-by-breath basis. Figure 9.11 shows a block diagram of the scheme employed for achieving this goal. The bottom portion of the block diagram represents the respiratory control system. Fluctuations in arterial P_{CO_2}, which we assume are measured in the form of fluctuations in the end-tidal CO_2 fraction (F_{ETCO_2}), result from "ventilatory noise" entering the closed-loop system as well as changes in gas exchange dynamics in the lungs. A simple linearized model of the gas exchange process is assumed, and based on measurements of ventilation ($\dot{V}_E$) and F_{ETCO_2}, the plant model parameters are identified. However, since gas exchange dynamics can change with sleep–wake state or other conditions of the subject, the estimation of plant model parameters has to be performed adaptively. At any given breath, the estimated plant model parameters are used along with existing measurements of $\dot{V}_E$ and F_{ETCO_2} to predict what F_{ICO_2} should be applied next to minimize the fluctuation of F_{ETCO_2} in the next breath.

9.4.2.1   *Plant Model*   Since the purpose of the scheme is to minimize the fluctuations in F_{ETCO_2} about its mean level, we define the new variables ΔF_{ETCO_2}, $\Delta \dot{V}_E$, and ΔF_{ICO_2} as the deviations in F_{ETCO_2}, $\dot{V}_E$, and F_{ICO_2}, respectively, about
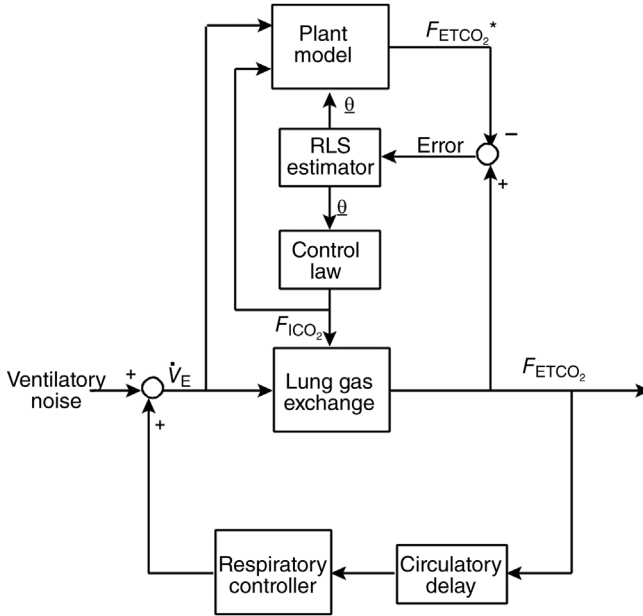
FIGURE 9.11   Control scheme for adaptive buffering of spontaneous fluctuations in end-tidal CO_2 in humans.

their corresponding means. Since negative F_{ICO2} cannot be realized in practice, it is assumed that the "resting" or mean level of F_{ICO2}, prior to the application of adaptive control, is 2.5% and not zero (i.e., the subject is breathing a gas mixture that resembles air but contains a small amount of CO_2). The linearized plant model takes a form very similar to that given in Equation 8.66:

$$\Delta F_{ETCO_2}(n) = a(n-1)\Delta F_{ETCO_2}(n-1) + b(n-1)\Delta \dot{V}_E(n-1)$$
$$+ c(n-1)\Delta F_{ICO_2}(n-1) + e(n) \tag{9.44}$$

Apart from the fact that ΔF_{ETCO2} and ΔF_{ICO2} are now used in place of ΔP_{ACO2} and ΔP_{ICO2}, respectively, a large difference between Equations 9.44 and 8.66 is that the unknown parameters here (a, b, and c) are assumed to be time-varying. n refers to the current breath. Thus, $\Delta F_{ETCO2}(n-1)$ and $\Delta \dot{V}_E(n-1)$ represent the changes in F_{ETCO2} and $\dot{V}_E$ at the *previous* breath. However, the exception is with $\Delta F_{ICO2}(n-1)$, which represents the change in F_{ICO2} of the *current* breath. This peculiar assignment of breath number is made because, in the real-time implementation of this scheme, the measurement of F_{ICO2} in the preceding inspiration is already available for use in the algorithm during the expiratory phase of the nth breath. By the same token, $\Delta \dot{V}_E(n-1)$ has to be used in place of $\Delta \dot{V}_E(n)$ in Equation 9.44 (note the use of $\Delta \dot{V}_E(n)$ in Equation 8.66), because the computation of $\Delta \dot{V}_E(n)$ requires knowledge of the tidal volume and total period of the nth breath, and the latter measurement is not

available until the end of the expiratory phase in the current breath (n). Finally, $e(n)$ in Equation 9.44 represents the error between the measured ΔF_{ETCO2} at breath n and the corresponding value predicted by using the plant model. Thus, the predicted ΔF_{ETCO2} ($\Delta F^*_{FETCO_2}$) is given by

$$\Delta F^*_{ETCO_2}(n) = a(n-1)\Delta F_{ETCO_2}(n-1) + b(n-1)\Delta \dot{V}_E(n-1)$$
$$+ c(n-1)\Delta F_{ICO_2}(n-1) \tag{9.45}$$

9.4.2.2 Plant Model Parameter Estimation

The plant model parameters are estimated adaptively, so that with each new breath, the new set of measurements obtained can be used to update our knowledge about the gas exchange process. To achieve this task, Modarreszadeh et al. (1993) selected the *recursive least-squares* (RLS) method, discussed in Section 9.2.3.

Since the estimation involves three parameters, we can recast Equation 9.45 in the form of a vector equation:

$$\Delta F^*_{ETCO_2}(n) = \underline{\theta}(n)'\underline{y}(n-1) \tag{9.46}$$

where

$$\underline{\theta}(n) = \begin{bmatrix} a(n) & b(n) & c(n) \end{bmatrix}' \tag{9.47}$$

and

$$\underline{y}(n-1) = \begin{bmatrix} \Delta F_{ETCO_2}(n-1) & \Delta \dot{V}_E(n-1) & \Delta F_{ICO_2}(n-1) \end{bmatrix}' \tag{9.48}$$

Thus, Equation 9.21 can be rewritten as

$$J(n) = \sum_{i=0}^{n} \lambda^{n-i} \left(\Delta F_{ETCO_2}(n) - \underline{\theta}(n)'\underline{y}(n-1) \right)^2 \tag{9.49}$$

The next steps are the same as described in Section 9.2.3, yielding the following results that have been rewritten in a form directly applicable to this particular problem. The parameter error covariance matrix is updated in the following way:

$$\mathbf{P}(n) = \frac{1}{\lambda} \left[\mathbf{P}(n-1) - \frac{\mathbf{P}(n-1)\underline{y}(n)\underline{y}(n)^T\mathbf{P}(n-1)}{\lambda + \underline{y}(n)^T\mathbf{P}(n-1)\underline{y}(n)} \right] \tag{9.50}$$

The Kalman gain vector, which determines the relative contribution of the prediction error $e(n)$ to the estimate of the parameter vector $\underline{\theta}(n)$ takes the following form:

$$\underline{K}(n) = \frac{\mathbf{P}(n-1)}{\lambda + \underline{y}(n)^T\mathbf{P}(n-1)\underline{y}(n)}\underline{y}(n) \tag{9.51}$$

The parameter vector update equation is given by

$$\underline{\theta}(n) = \underline{\theta}(n-1) + \underline{K}(n)e(n) \tag{9.52}$$

9.4.2.3 Adaptive Control Law

Having estimated the most current values of the plant model parameters, the next task in each time-step (breath) would be to determine the optimal value of ΔF_{ICO2} in the current breath (i.e., $\Delta F_{ICO2}(n-1)$) that minimizes the predicted ΔF_{ETCO2} at the end of the expiration phase of the current breath (i.e., $\Delta F_{ETCO2}(n)$). The criterion function to be minimized, in this case, was selected by Modarreszadeh et al. (1993) to be the following:

$$I(n) = \left(\alpha\Delta F^{*}_{ETCO_2}(n)\right)^2 + \left(\beta\Delta F_{ICO_2}(n-1)\right)^2 \tag{9.53}$$

The reason for including the term with $\Delta F_{ICO2}(n-1)$ in Equation 9.53 was to minimize the fluctuations in inhaled CO_2 concentration along with those in alveolar CO_2. This allows for a solution that does not lead to minimal ΔF_{ETCO2} at the "expense" of employing large ΔF_{ICO2}. This is important from a practical point of view, since high values of F_{ICO2} can be a source of unpleasant sensation to the subject. The relative contributions of ΔF_{ETCO2} and ΔF_{ICO2} to the criterion function are determined by the weights α and β. Based on preliminary tests, the authors chose values of 1 for α and 0.5 for β.

To determine the "optimal" $\Delta F_{ICO2}(n-1)$, Equation 9.45 is substituted into Equation 9.53, and this is differentiated with respect to $\Delta F_{ICO2}(n-1)$. The result of the differentiation is set equal to zero. After rearranging terms, we obtain

$$\Delta F_{ICO_2}(n-1) = \frac{-\alpha^2 c(n-1)}{\alpha^2 c(n-1)^2 + \beta^2}\left(\alpha(n-1)\Delta F_{ETCO_2}(n-1) + b(n-1)\Delta\dot{V}_E(n-1)\right) \tag{9.54}$$

In Equation 9.54, note that the plant parameter estimates from the previous breath (i.e., $a(n-1)$, $b(n-1)$, and $c(n-1)$) are used since, from Equations 9.51 and 9.52, the parameter estimates from the current breath (i.e., $a(n)$, $b(n)$, and $c(n)$) are determined in part by $\Delta F_{ICO2}(n-1)$.

The overall algorithm for the adaptive control scheme is displayed in the form of a flowchart in Figure 9.12. When the algorithm is initiated from starting conditions ($n=0$), it is useful to employ initial values for the plant parameters that are not too far from their "true" values. For this reason, Modarreszadeh et al. estimated these values beforehand by measuring the subjects' responses to a dynamic CO_2 stimulus that was altered in pseudorandom binary fashion (see Section 8.5.2.1). The average values of a, b, and c in eight subjects were found to be 0.66, -0.02, and 0.69, respectively. The value for b assumes that ventilation is measured in $L\,min^{-1}$ and F_{ETCO2} is expressed as a percentage. A starting value for the parameter error
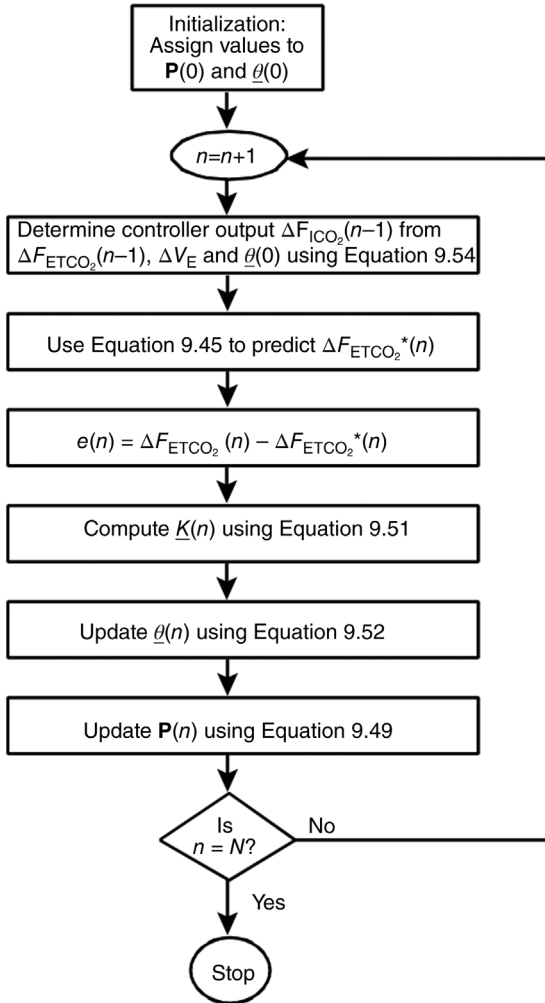
FIGURE 9.12   Flowchart of adaptive control algorithm.

covariance matrix **P** is also required. A common procedure is to set **P**(0) equal to the identity matrix scaled by a factor of 100.

9.4.2.4 *Performance of the Adaptive Controller*   Breath-by-breath measurements of F_{ETCO2} in a normal subject during spontaneous breathing are shown in Figure 9.13a. These can be compared with the corresponding measurements of F_{ETCO2} in the same subject during adaptive buffering of the end-tidal CO_2 fluctuations, shown in Figure 9.13b. It is clear that the adaptive controller produced a significant reduction of the fluctuations in F_{ETCO2}, particularly in the lower
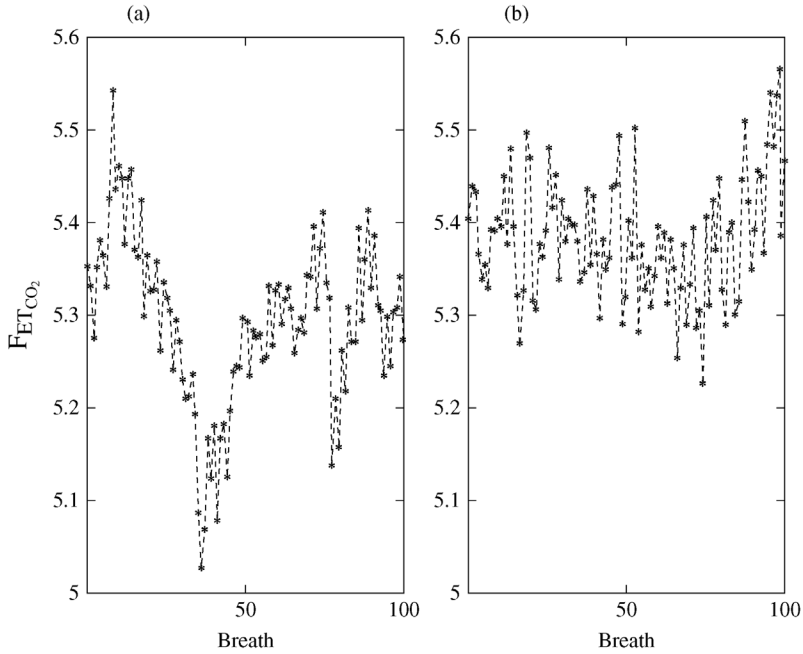
FIGURE 9.13 (a) Breath-by-breath measurements of F_{ETCO2} in an adult human, showing considerable spontaneous variability. (b) F_{ETCO2} measurements in the same subject during application of the adaptive buffering scheme. (Reproduced from Modarreszadeh et al. (1993).)

frequency region. It would have been possible to reduce ΔF_{ETCO2} further by setting β in the criterion function I to zero. But this would be achieved at the expense of incurring larger fluctuations in F_{ICO2}.

The properties of this adaptive control scheme can be studied further by executing the MATLAB script file `acs_CO2.m` included with this book. However, instead of obtaining measurements for a real human subject, a simple dynamic simulation of the chemoreflex control system is used to generate "data" and to interact with the adaptive controller. The fluctuations in $\dot{V}_E$, F_{ICO2}, and F_{ETCO2} are assumed to occur around their corresponding mean levels.

PROBLEMS

P.9.1. By applying the matrix inversion lemma, determine how Equation 9.28 is derived from Equation 9.25c.

P.9.2. Develop a MATLAB program that implements the LMS algorithm discussed in Section 9.2.2 and illustrated by the flowchart in Figure 9.5.

P.9.3. Starting with the MATLAB script files `laguerest.m` from Chapter 8 and `TVmodel_RLS.m` from this chapter, use the content of Section 9.3.3 to develop the time-varying two-input model of heart rate variability discussed in that chapter.

P.9.4. Using the MATLAB script file `acs_CO2.m`, determine how the ratio of the variance (=standard deviation^2) of the fluctuations in F_{ETCO2} during adaptive buffering to the corresponding variance of F_{ETCO2} during spontaneous breathing (no buffering) would change as the weighting factor β for F_{ICO2} in the control law is changed from 0 to 1; plot this relative variance as a function of β. Determine also how these changes in β would affect the fluctuations in F_{ICO2}. Use increments of 0.1 in β.

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10

NONLINEAR ANALYSIS OF PHYSIOLOGICAL CONTROL SYSTEMS

10.1 NONLINEAR VERSUS LINEAR CLOSED-LOOP SYSTEMS

Thus far, the methods we have employed to analyze the dynamics of physiological control systems have been primarily linear methods, although it is clear that nonlinearity is the rule and not the exception in biology. As a first approximation, linear models work surprisingly well in many instances, but one can find many more instances in which the nonlinear features are critical for the functioning of the system in question. Classic examples of this include the mechanism through which the nerve action potential is generated, as modeled by the Hodgkin–Huxley equations, and various phenomena associated with nonlinear oscillators, such as frequency entrainment and phase resetting. These will be discussed in detail later.

A key disadvantage in the analysis of nonlinear systems is that the principle of superposition can no longer be applied. This has profound consequences, for it means that in contrast to linear systems where the dynamics can be fully character-ized in terms of the impulse response, the same concise means of description cannot be applied to the nonlinear system. Another consequence of the inapplicability of the superposition principle is that local solutions cannot be extrapolated to the global scale. As an example, consider the comparison of the linear lung mechanics model described in Chapters 4 and 5 with a version that contains nonlinear feedback. Both models are illustrated in Figure 10.1. The responses of the linear and nonlinear models to input steps in P_{ao} are displayed in Figure 10.2a and b,
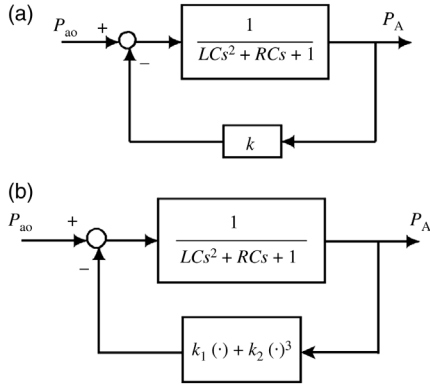
FIGURE 10.1 (a) Linear lung mechanics model. (b) Lung mechanics model with nonlinear feedback.
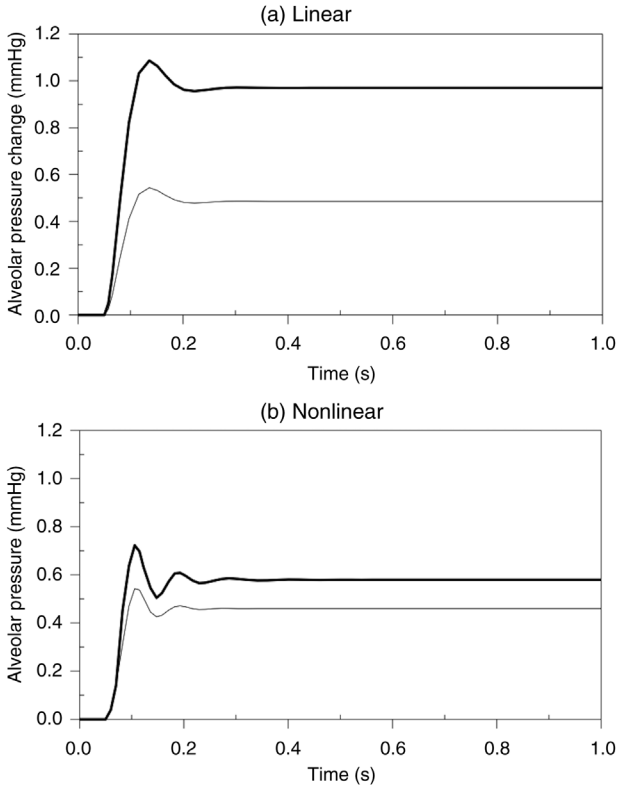
FIGURE 10.2 (a) Responses of the linear lung mechanics model to a unit step (light tracing) and a step of magnitude 2 in P_{ao}. (b) Responses of the model with nonlinear feedback to the same inputs, showing that the principle of superposition is no longer valid in nonlinear systems.

respectively. In each case, the responses in P_A to step inputs of amplitude 1 and 2 mmHg in P_{ao} (starting at time 0.5 s) are shown as the light and bold tracings, respectively. In the linear case, the response to a step input that is twice as large leads to a proportionately scaled version of the response to the unit step. However, in the nonlinear case, the steady-state response to the larger step is clearly less than twice the steady-state response to the unit step in P_{ao}. Furthermore, the response to the larger step is more oscillatory than the unit step response. Thus, knowing the unit impulse response or unit step response of the nonlinear system does not enable us to predict the responses to input steps of other amplitudes.

There is another major difference between the dynamics of linear and nonlinear systems. As one might recall from Chapter 5, the dynamics of linear systems can also be characterized in terms of their frequency responses, since sinusoidal perturbation of a linear system results in a sinusoidal output of the same frequency. In nonlinear systems, however, sinusoidal perturbation can give rise to a response that contains not only the fundamental frequency of the perturbation but also higher harmonics of that frequency. Figure 10.3 shows a comparison between responses elicited from the linear and nonlinear versions of the lung mechanics model displayed in Figure 10.1. The linear response to an input sine wave of unit
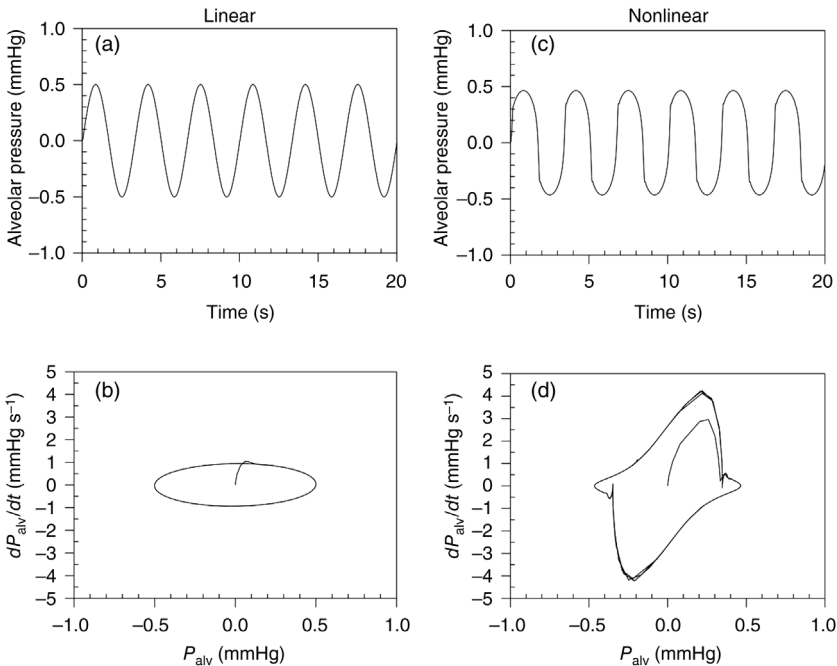
FIGURE 10.3 Responses of the linear (a) and nonlinear (c) lung mechanics models to sinusoidal forcing in P_{ao}. The corresponding phase-plane plots are displayed in parts (b) and (d), respectively.

amplitude and frequency 0.3 Hz shows a sinusoidal output of amplitude 0.5 and the same frequency (Figure 10.3a). The same input forcing produces a nonlinear response of the same fundamental frequency and with an amplitude of approximately 0.5. The shape of the response is clearly nonsinusoidal but appears more "squarish," since it contains higher frequency components (Figure 10.3c). The difference becomes much more apparent when these responses are viewed in terms of their corresponding *phase-plane* plots. As described further in the next section, signals generated by systems that are governed by second-order differential equations can be completely characterized by plotting the first time-derivative versus the variable in question. In the examples given, dP_{alv}/dt is plotted against P_{alv}. In the linear case, the phase-plane plot of the system response is an ellipse (Figure 10.3b). In the nonlinear case, the phase-plane plot also shows a closed-loop figure, but the structure of the plot is much more irregular than the ellipse (Figure 10.3d).

Another feature that illustrates the dynamic complexity of nonlinear systems is that, under certain conditions, the response to periodic stimulation at a given frequency can change dramatically if the *amplitude* of the stimulus is varied. Again, we illustrate this point with the example of the closed-loop nonlinear lung mechanics model of Figure 10.1b. When the nonlinear model is stimulated by a sinusoidal perturbation in P_{ao} of frequency 0.16 Hz and amplitude 10 mmHg, the response, as depicted by the time series and phase-plane plots in the top panel of Figure 10.4, is essentially a very high-frequency oscillation that rides on top of the (slower) fundamental frequency. When the forcing amplitude is decreased to 1 mmHg (middle panel of Figure 10.4), the frequency of the "fast" oscillatory component is decreased and the response contains a mixture of periodic and aperiodic components. Finally, when the forcing amplitude is decreased to less than 0.01 mmHg (bottom panel of Figure 10.4), the fluctuations in P_{alv} become aperiodic and appear unpredictable. This type of dynamic behavior is known as *deterministic chaos*, since the seemingly random motion is generated by a perfectly deterministic model with no explicit noise input.

10.2 PHASE-PLANE ANALYSIS

Consider the motion of a simple linear spring–mass system, which is characterized by the following second-order differential equation:

$$m\frac{d^2x}{dt^2} + kx = 0 \tag{10.1}$$

The steady-state solution to the above equation is given by

$$x(t) = A \sin\left(\sqrt{\frac{k}{m}}t + \phi\right) \tag{10.2}$$
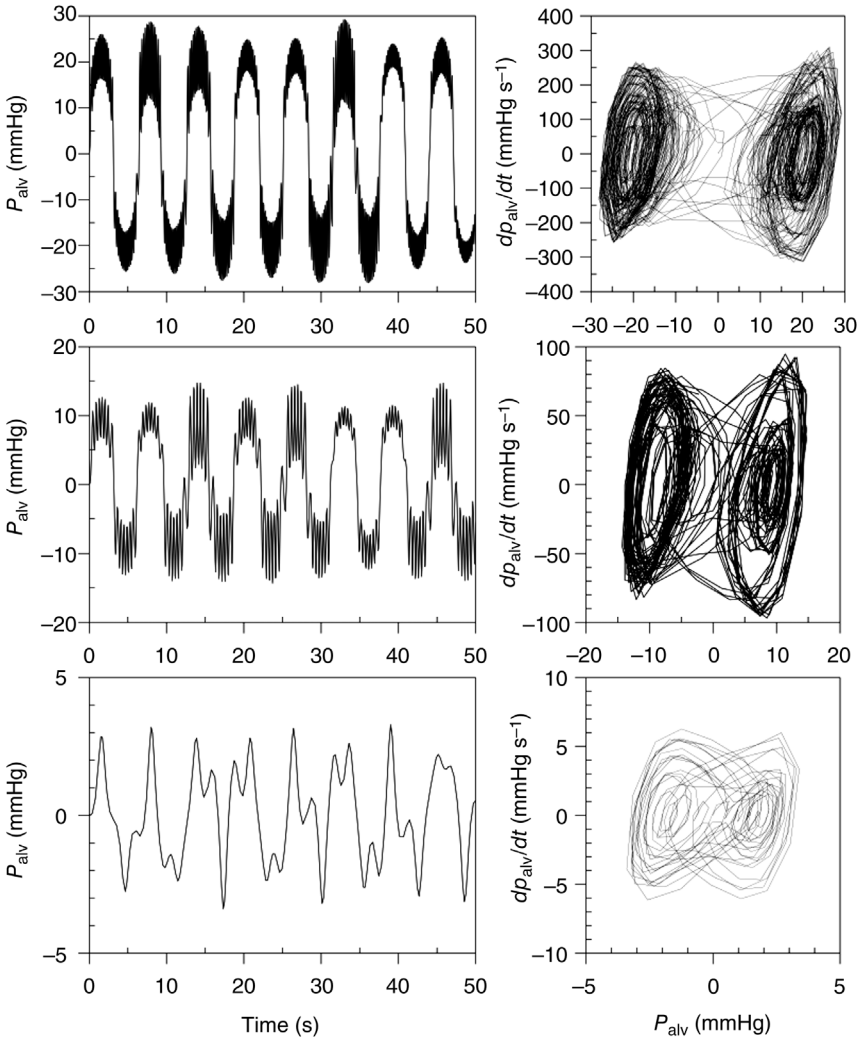
FIGURE 10.4 Changes in the dynamics of the sinusoidally forced nonlinear lung mechanics model from almost periodic to chaotic as the forcing amplitude is reduced (top to bottom panels). Forcing frequency is 0.16 Hz.

where the constants A and ϕ are determined by the initial conditions, that is, the initial position and velocity of the mass. If we differentiate Equation 10.2 with respect to time, we obtain the velocity $y(t)$ of the mass:

$$y(t) = \dot{x}(t) = \sqrt{\frac{k}{m}} A \cos\left(\sqrt{\frac{k}{m}} t + \phi\right) \qquad (10.3)$$

Then, normalizing Equation 10.2 by A and Equation 10.3 by $(k/m)^2 A$, and using the trigonometric equality,

$$\sin^2\left(\sqrt{\frac{k}{m}}t + \phi\right) + \cos^2\left(\sqrt{\frac{k}{m}}t + \phi\right) = 1 \tag{10.4}$$

we obtain the following relation between $x(t)$ and $y(t)$:

$$x^2 + \frac{y^2}{k/m} = A^2 \tag{10.5}$$

Equation 10.5 allows the motion of the mass to be completely characterized by a knowledge of the instantaneous position and velocity of the mass, given that the initial position of the mass is also known. As such, x and y represent the *state* of the system. Note that although x and y are functions of time, Equation 10.5 contains no explicit terms in time. Thus, when y is plotted against x, a "stationary" ellipse appears, similar to that displayed in Figure 10.3b. For different values of A, ellipses of different sizes are generated. Each of these ellipses is known as a *trajectory* of the system. And the plane formed by the position and velocity axes is the *phase plane*.

10.2.1   Local Stability: Singular Points

The above example of a system with second-order dynamics can also be expressed in terms of a set of coupled first-order differential equations involving the position variable $x(t)$ and the velocity variable $y(t)$:

$$\frac{dx}{dt} = y \tag{10.6}$$

$$\frac{dy}{dt} = -\frac{k}{m}x \tag{10.7}$$

On the phase plane, the locus of points in which dx/dt (velocity) or dy/dt (acceleration) becomes zero is known as a *nullcline*. In general, the locus of points in the phase plane through which phase trajectories pass with constant slope is termed an *isocline*. The x-nullcline ($dx/dt = 0$) is a special case of an isocline that has infinite slope, while the y-nullcline ($dy/dt = 0$) is an isocline that has zero slope. In the linear oscillator, the x-nullcline coincides with the x-axis of the phase plane, while the y-nullcline lies on the y-axis. At the point where both nullclines intersect (i.e., at the origin), dx/dt and dy/dt both are simultaneously zero. This corresponds to an *equilibrium point*, a point at which there is no motion. Equations 10.6 and 9.7 can also be represented in the form of the following differential equation, in which there is no longer any explicit dependence on time:

$$\frac{dy}{dx} = \frac{-(k/m)x}{y} \tag{10.8}$$

At the equilibrium point, the numerator and denominator on the right-hand side of Equation 10.8 each becomes zero. As such, equilibrium points are also referred to as *singular points* in phase-plane terminology. Although there is "no motion" at the singular points, they do not necessarily represent stable points of equilibrium. The type of stability in the vicinity of each singular point can yield useful information about overall system dynamics.

Consider a second-order system that can be characterized by the following phase-plane equations (in which $y = dx/dt$):

$$\frac{dx}{dt} = F(x, y) \tag{10.9}$$

$$\frac{dy}{dt} = G(x, y) \tag{10.10}$$

where F and G can be nonlinear functions of x and y. Suppose one of the singular points is at (x_0, y_0). Consider the dynamics of motion at a point (x, y) located in the proximity of the singular point, where

$$x = x_0 + u \tag{10.11}$$

$$y = y_0 + v \tag{10.12}$$

If we use Equations 10.11 and 10.12 to substitute for x and y, respectively, in Equations 10.9 and 10.10, and perform a Taylor expansion about (x_0, y_0), we obtain, after ignoring terms higher than first order, the following expressions are obtained for the local dynamics around (x_0, y_0):

$$\frac{du}{dt} = F_x u + F_y v = \frac{\partial F}{\partial x} u + \frac{\partial F}{\partial y} v \tag{10.13}$$

$$\frac{dv}{dt} = G_x u + G_y v = \frac{\partial G}{\partial x} u + \frac{\partial G}{\partial y} v \tag{10.14}$$

where the partial derivative terms (F_x, F_y, G_x, G_y) are all evaluated at the singular point (x_0, y_0).

Equations 10.13 and 10.14 can be combined in order to eliminate v, resulting in the following linear second-order differential equation:

$$\frac{d^2 u}{dt^2} - (F_x + G_y) \frac{du}{dt} + (F_x G_y - G_x F_y) u = 0 \tag{10.15}$$

The solution to Equation 10.15 is given by

$$u = A_1 e^{\alpha_1 t} + A_2 e^{\alpha_2 t} \tag{10.16}$$

where the constants A_1 and A_2 depend on the initial conditions, and α_1 and α_2 are given by the roots of the following quadratic equation:

$$\alpha^2 - \left(F_x + G_y\right)\alpha + \left(F_x G_y - G_x F_y\right) = 0 \tag{10.17}$$

The solution for v takes a form similar to that of Equation 10.16, except that the coefficient of each exponential term will be different from the corresponding term in Equation 10.16.

From Equation 10.16, it can be seen that the singular point in question will be *stable* only if the real parts of α_1 and α_2 are both *negative*. For this to be the case, two conditions must hold:

(A) The sum of the roots must be negative, that is,

$$F_x + G_y < 0 \tag{10.18}$$

(B) The product of the roots must be positive, that is,

$$F_x G_y - G_x F_y > 0 \tag{10.19}$$

Even if the singular point is stable, there is the additional question of whether the associated dynamics is oscillatory. For nonoscillatory dynamics, both roots must be real (i.e., have no imaginary parts):

(C)
$$\left(F_x + G_y\right)^2 - 4\left(F_x G_y - G_x F_y\right) > 0 \tag{10.20}$$

Thus, depending on the values of the roots of Equation 10.17, one can have singular points with a variety of dynamics:

(1) *Both roots real and negative*: Here, conditions A, B and C are all satisfied. This singular point represents a *stable node*: The decay toward this equilibrium point is nonoscillatory.

(2) *Both roots complex with negative real parts*: Conditions A and B are satisfied but not condition C. The equilibrium point is stable, but the decay toward it is oscillatory. This kind of singularity is known as a *stable focus*.

(3) *Both roots real and positive*: Here, conditions B and C are satisfied but not condition A. This produces an *unstable node*: Any infinitesimal perturbation will cause the state point to move away from the singularity but the motion will not be oscillatory.

(4) *Both roots complex with positive real parts*: Only condition B is satisfied. This produces an *unstable focus*: Any infinitesimal perturbation will cause the state point to move away from the singularity with oscillatory dynamics.

(5) *Both roots imaginary (zero real parts)*: Since the roots must be conjugate, their sum in this case is zero but their product is a positive real value. Thus, only condition B is satisfied. This leads to a *center*, which is considered neutrally stable. The singular point associated with the linear spring–mass system in Equation 10.1 is an example of a center.

(6) *Both roots real, one positive and one negative*: Here, condition C is satisfied, but not condition B. However, condition A may or may not hold, depending on the magnitude of the negative root relative to that of the positive root. This gives rise to a peculiar type of unstable equilibrium point known as a *saddle point*.

10.2.2   Method of Isoclines

While the complete phase portrait of any given second-order system can be arrived at by simply solving the set of coupled first-order differential equations (Equations 10.9 and 10.10), a good understanding of the dynamics of the system can often be obtained by applying an approximate, semigraphical analysis known as the *method of isoclines*. We illustrate the application of this method here by considering the dynamics of a simple nonlinear system: the pendulum. We assume that this pendulum consists of a heavy steel disk linked by a weightless rigid rod to a vertical fixture (Figure 10.5). If we apply Newton's second law to the motion of the bob in the direction tangential to the rod, we obtain the following second-order differential equation:

$$mL\frac{d^2\theta}{dt^2} = -mg\sin\theta \qquad (10.21a)$$

where θ, the angular displacement of the pendulum, is as shown in Figure 10.5. The above equation simplifies to

$$\frac{d^2\theta}{dt^2} + K\sin\theta = 0 \qquad (10.21b)$$

where $K\ (=g/L)$ is a constant.

To apply the method of isoclines, we express Equation 10.21b in the form of the equivalent phase-plane equations. Thus, we have

$$\frac{d\theta}{dt} = \phi \qquad (10.22)$$

$$\frac{d\phi}{dt} = -K\sin\theta \qquad (10.23)$$

These can also be combined to give

$$\frac{d\phi}{d\theta} = -\frac{K\sin\theta}{\phi} \qquad (10.24)$$

$\theta=-\pi$ $\theta=\pi$

L

θ

$\theta = 0$

$mgL\sin\theta$

θ
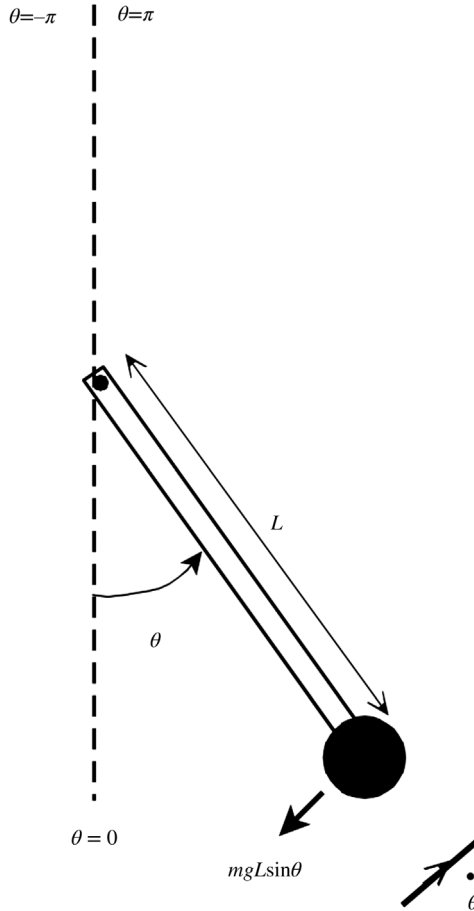
FIGURE 10.5 Idealized rigid pendulum, showing tangential acceleration and tangential component of gravitational force.

The θ-nullcline (i.e., points along which $d\theta/dt=0$) is defined by the line $\phi=0$, which corresponds to the θ-axis. The ϕ-nullcline (i.e., along which $d\phi/dt=0$) is given by $\sin\theta=0$. Since θ can take on values between $-\pi$ and π radians, there are three possible solutions for $\sin\theta=0$ in this range and therefore three ϕ-nullclines: $\theta=0$, $\theta=-\pi$, and $\theta=\pi$. However, it should be noted that $\theta=-\pi$ and $\theta=\pi$ correspond to the same physical configuration for the pendulum, that is, when the bob is vertically above the hinge (see Figure 10.5). Intersection of the θ-nullcline with the three ϕ-nullclines yields three singular points. Treating θ as the horizontal axis and ϕ as the vertical axis, these singular points are located at coordinates (0,0), $(-\pi,0)$, and $(\pi,0)$, as indicated by the filled circles in Figure 10.6. The θ- and
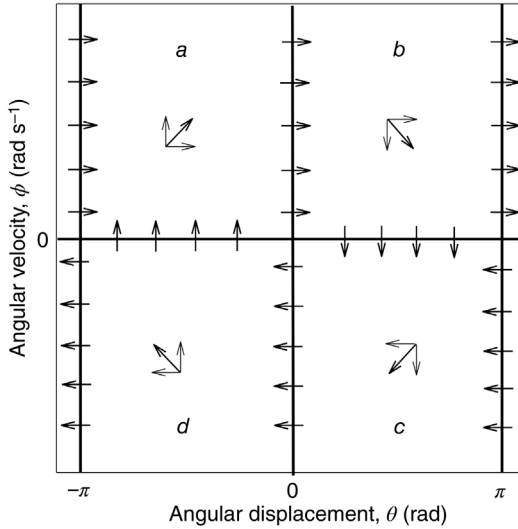
FIGURE 10.6   Illustration of the method of isoclines. Bold lines represent the nullclines of the system. Arrows indicate the direction of the phase-plane trajectories. *a*, *b*, *c*, and *d* represent four regions of the phase space in which the general flow directions are different.

ϕ-nullclines also divide up the phase plane into four regions, labeled *a*, *b*, *c*, and *d* in Figure 10.6. In region *a*, θ is negative and ϕ is positive, so by Equations 10.22 and 10.23, $d\theta/dt > 0$ and $d\phi/dt > 0$; thus, the trajectories in this region will generally be directed upward and to the right. In region *b*, both θ and ϕ are positive, so from Equations 10.22 and 10.23, $d\theta/dt > 0$ and $d\phi/dt < 0$; therefore, the flow is now directed downward and to the right. In region *c*, θ is positive and ϕ is negative, so $d\theta/dt < 0$ and $d\phi/dt < 0$. Finally, using similar considerations, it may be shown that in region *d*, the flow is directed upward and to the left. In Figure 10.6, we have also included arrows to indicate the directions of the flows on the nullclines. Thus, the overall "picture" we obtain from this approximate analysis is that for $\phi > 0$, the trajectories generally flow from left to right, while for $\phi < 0$, they flow from right to left. Furthermore, there is also a tendency for the flow to rotate in a clockwise manner around the origin.

Figure 10.7 shows a set of phase trajectories for the pendulum system, computed by solving Equation 10.21b numerically for $K = 39.5$. Each phase trajectory is obtained by assigning different values to the initial conditions for θ and ϕ before computing the numerical solution. It is clear that the trajectories shown here are consistent with the inferences made using the method of isoclines. For small starting values of θ and ϕ, the motion of the pendulum is a sinusoidal function of time, oscillating about $\theta = 0$ rad. Applying Equations 10.18–10.20 to this example, we
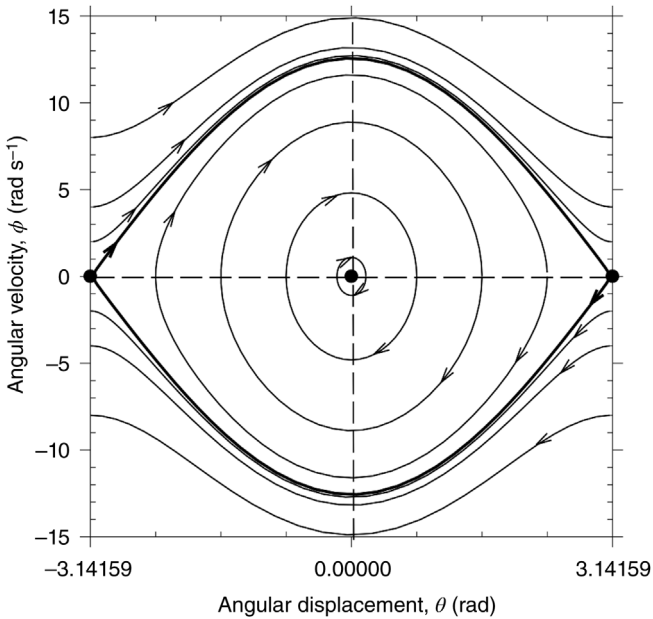
FIGURE 10.7   Phase-plane portrait of the dynamics of the rigid pendulum. Filled circles represent the singular points, while the bold trajectories represent the separatrices that divide the oscillatory mode (inside) of the pendulum from the rotating mode (outside).

find that the singular point (0,0) corresponds to a center (imaginary roots). As the starting value for θ approaches $-\pi$ or π, the oscillations become less sinusoidal in character. Finally, if the initial condition for θ is set equal to $-\pi$ or π, the pendulum in principle would achieve an equilibrium position with its bob directly above its hinge. Theoretically, one can imagine that a virtually imperceptible nudge in either direction would send the pendulum swinging in that direction, making a 2π rotation until it comes to rest again with its bob balanced directly above its hinge. Thus, the singular points $(-\pi,0)$ and $(\pi,0)$ correspond to saddle points that are attracting when the phase trajectories approach them from one direction but repelling for trajectories in the orthogonal direction. Application of Equations 10.18–10.20 will allow a verification that $(-\pi,0)$ and $(\pi,0)$ are saddle points.

For phase trajectories that begin at $\theta=-\pi$ or $\theta=\pi$ with nonzero velocity (i.e., $\phi \neq 0$), ϕ remains uniformly positive or negative over the whole range of θ values. This implies that the motion of the pendulum now is no longer oscillatory; instead it simply rotates either in clockwise or anticlockwise manner around its hinge. In Figure 10.7, it is clear that the phase trajectories that lead into or away from the singular points $(-\pi,0)$ and $(\pi,0)$ define the boundaries that separate the oscillatory type of motion from the rotational type of motion. These trajectories, which divide the phase plane into regions of differing dynamic modes, are known as *separatrices*.

10.3 NONLINEAR OSCILLATORS

10.3.1 Limit Cycles

The only type of singularity associated with periodic oscillations that was discussed in the previous section is the *center*. The motion associated with a center takes the form of phase trajectories that close in themselves and enclose the singularity. Which particular phase trajectory is taken depends on the initial conditions that preceded the dynamics. For example, in the case of the rigid pendulum, applying an impulsive disturbance to the bob at the end of its swing can increase or decrease the swing amplitude, depending on the relative direction of the disturbance. On the phase portrait, this corresponds to a sudden change in phase trajectory to another of the concentric elliptical (or circular) orbits that enclose the center singularity. In Figure 10.8a, the pendulum bob is given a knock directed toward the equilibrium point ($\theta = 0$) at the end of its swing. This allows it to pass the $\theta = 0$ position with increased velocity, and consequently allows it to achieve an oscillation of larger amplitude. In the phase-plane diagram, this is represented by a change in trajectory from *a* to *b* (Figure 10.8a). Since no damping is present, the state point will not return to the original phase trajectory unless another externally imposed disturbance forces it to do so.

Many physiological oscillators exhibit a behavior that is quite different from that displayed in Figure 10.8a. On the phase plane, these oscillations assume the form of a stable, closed trajectory called a *limit cycle*. What distinguishes the limit cycle from the type of oscillation discussed previously is that, although external perturbations can move the state point away from the limit cycle trajectory, it eventually always rejoins the original trajectory. This is illustrated in Figure 10.8b. Whether the state point is moved to a location outside the limit cycle (*a* in Figure 10.8b) or a location inside the limit cycle (*b* in Figure 10.8b), the original oscillatory behavior is always reestablished after some time.

10.3.2 The van der Pol Oscillator

In 1928, van der Pol and van der Mark proposed the first dynamic model of oscillatory activity in the heart. Their model consisted of the following second-order nonlinear differential equation:

$$\frac{d^2x}{dt^2} - c\left(1 - x^2\right)\frac{dx}{dt} + x = 0 \tag{10.25}$$

where the constant $c > 0$. The phase-plane properties of the van der Pol equation are most conveniently explored by applying *Lienard's transformation*, that is:

$$y = \frac{1}{c}\frac{dx}{dt} + \frac{x^3}{3} - x \tag{10.26a}$$
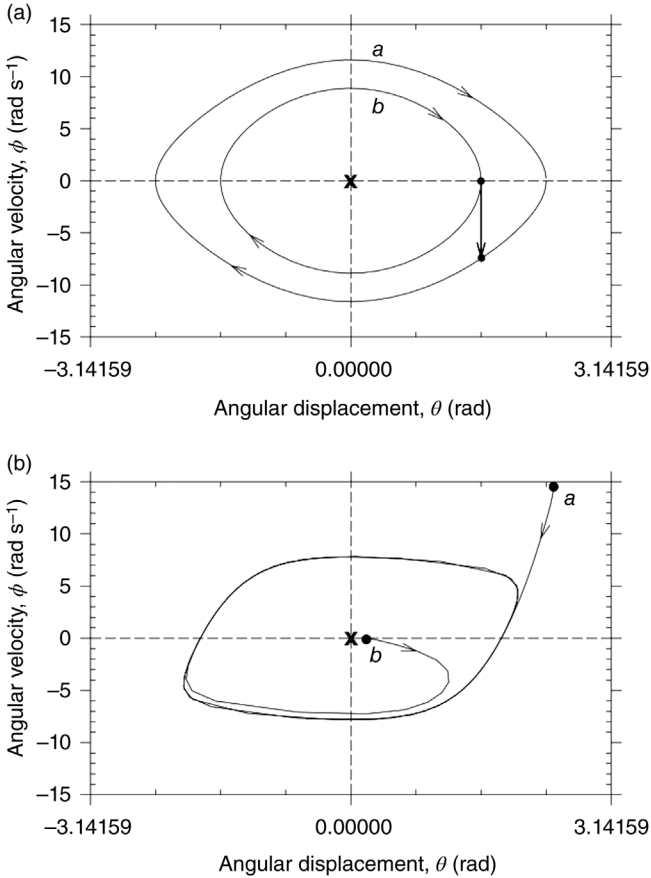
FIGURE 10.8   Differences between a non-limit-cycle oscillator (a) and a limit-cycle oscillator (b). In the former, external disturbance (bold arrow) moves the phase trajectory to a different orbit (*b* versus *a*) around the center. In the stable limit cycle, the state point always returns to its original trajectory even after an external disturbance moves it to a different location.

Differentiating Equation 10.26a with respect to time and substituting the result into Equation 10.25, we obtain

$$\frac{dy}{dt} = -\frac{x}{c} \tag{10.27}$$

Rearranging Equation 10.26a, we have

$$\frac{dx}{dt} = c\left(y - \frac{x^3}{3} + x\right) \tag{10.26b}$$

Equations 10.26b and 10.27 form a set of coupled first-order differential equations that do not have a closed-form analytic solution. However, the techniques of Section 10.2 can be employed to provide a rough picture of the phase portrait of this dynamic system.

First, we deduce the phase-plane locations of the nullclines. The x-nullcline ($dx/dt = 0$) corresponds to the locus defined by the cubic function:

$$y = \frac{x^3}{3} - x \tag{10.28}$$

The y-nullcline ($dy/dt = 0$) is given by the vertical axis, or

$$x = 0 \tag{10.29}$$

The x- and y-nullclines intersect at only one point, that is, at the origin $(0, 0)$. We determine the nature of the singular point found at $(0, 0)$ by evaluating the coefficients of Equation 10.17 and subsequently the roots of the characteristic equation. Assuming that

$$F(x, y) = c\left(y - \frac{x^3}{3} + x\right) \tag{10.30}$$

and that

$$G(x, y) = -\frac{x}{c} \tag{10.31}$$

the characteristic equation describing the dynamics in the vicinity of $(0, 0)$ takes the form:

$$\alpha^2 - c\alpha + 1 = 0 \tag{10.32}$$

Condition A is clearly not satisfied, while condition B is valid for all values of c. Whether condition C is satisfied depends on the value of c. When $c \geq 2$, the roots of Equation 10.32 will be real and positive, in which case the singular point would be an unstable node. However, when $c < 2$, the roots become complex with positive real parts; in this case, the singular point is an unstable focus. Therefore, for all feasible values of c, the equilibrium point at the origin will be an unstable one.

The nullclines divide the phase plane into four major regions, as shown in Figure 10.9. In region a, $x > 0$ and $y > x^3/3 - x$. Therefore, from Equation 10.26b, $dx/dt > 0$ and from Equation 10.27, $dy/dt < 0$. This means that the phase trajectories here in general would be directed downward and to the right. In region b, $x > 0$ and $y < x^3/3 - x$, so $dx/dt < 0$ and $dy/dt < 0$, and the phase trajectories would tend to point downward and to the left. Applying similar considerations, the pattern of flow
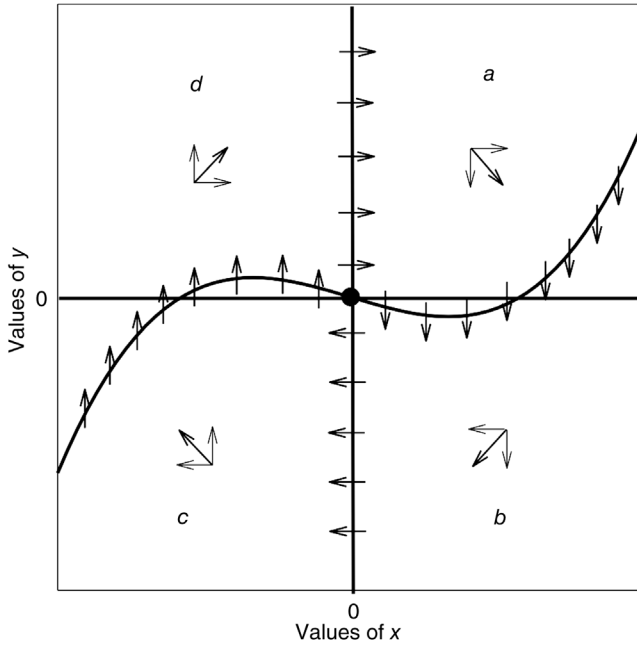
FIGURE 10.9   Method of isoclines applied to the van der Pol model. Arrows indicate direction of phase-plane trajectories. a, b, c, and d represent four regions in which the general flow directions are different. The nullclines ($y = x^3/3 - x$ and $x = 0$) are shown as bold lines. The filled circle at the origin represents the only singular point for this system.

is upward and to the left in region c, and upward and to the right in region d. For consistency, the directions of flow on the nullclines must be as displayed in Figure 10.9. Thus, in general, there is a clockwise flow of phase trajectories around the origin; however, at the same time, because of the unstable node or focus, the trajectories are also directed away from the origin.

To complete the picture, we numerically integrate Equations 10.26b and 10.27 for given values of c, but with several different initial conditions. This can be achieved quite easily by implementing the system defined by Equations 10.26b and 10.27 as a SIMULINK model. This model, the source code for which may be found in the SIMULINK file `vdpmod.slx`, is displayed in Figure 10.10. Figure 10.11a shows an example of the oscillatory dynamics generated by the van der Pol model for $c = 3$. This saw-toothed type of waveform is commonly referred to as a *relaxation oscillation*. A more complete representation of van der Pol dynamics is displayed in Figure 10.11b, which shows the portraits for six phase trajectories that originate from different starting points (or initial conditions, shown as filled circles) in the phase space. The x-nullcline (shown as the dotted curve) is displayed for reference. Note that phase trajectories that originate from inside the limit cycle move away from the singular point toward the limit cycle, whereas the phase
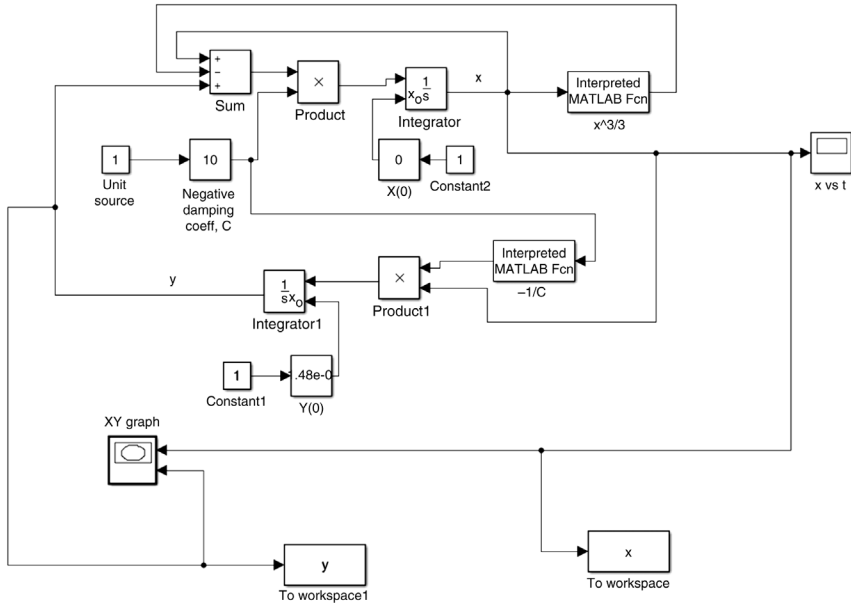
FIGURE 10.10 SIMULINK implementation of the van der Pol oscillator. This model is contained in SIMULINK model file `vdpmod.slx`.

trajectories from outside the limit cycle move inward toward the limit cycle. All trajectories tend to circulate around the origin in a clockwise pattern, as predicted in Figure 10.9.

One feature that distinguishes a nonlinear oscillator, such as the van der Pol system, from a linear oscillatory system is that the former can exhibit the phenomenon of *entrainment* or *phase-locking*. The coupling of two linear systems with different natural oscillatory frequencies leads to *beating*, in which the combined output shows the original two frequencies of oscillation plus a new oscillation that corresponds to the difference between the two frequencies. However, when a nonlinear oscillator is driven by an external periodic stimulus whose frequency is quite different from the former, the output of the oscillatory system will contain a mixture of components that result from the interaction of the driving periodicity and the natural oscillation. As the driving frequency approaches the natural frequency of the nonlinear oscillator, there will be a range of frequencies over which the nonlinear system will adopt the frequency of the driving stimulus. This is illustrated for the case of the van der Pol oscillator in Figure 10.12. Here, the system of equations becomes

$$\frac{dx}{dt} = c\left(y - \frac{x^3}{3} + x\right) \tag{10.33}$$
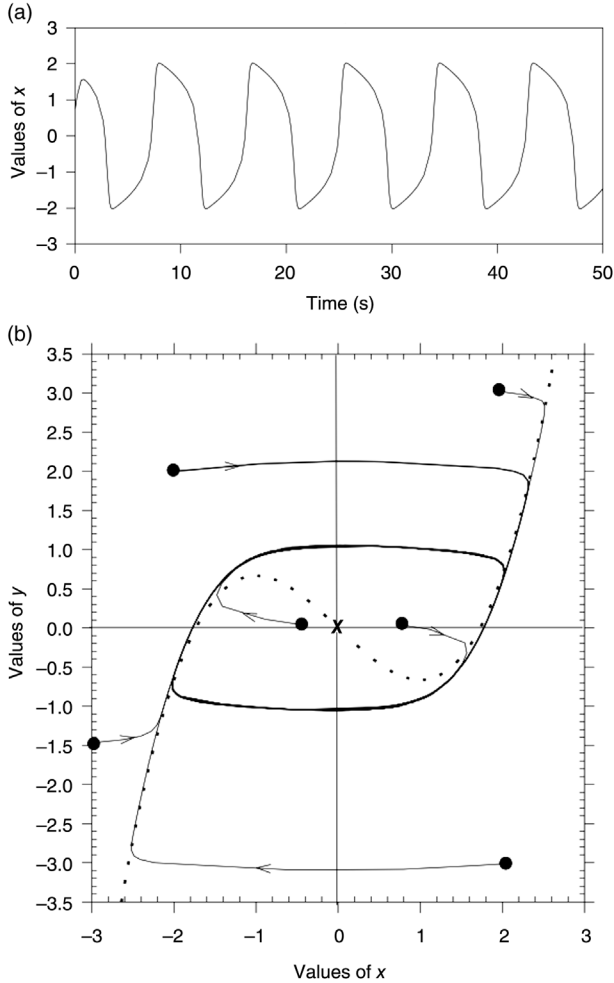
FIGURE 10.11 (a) Time-course of oscillatory activity generated by the van der Pol model. (b) Phase portrait showing limit cycle formed by several trajectories initiated at different starting points (shown as filled circles). The x-nullcline is shown as the dotted curve, while the singular point at the origin is marked as a cross. Note consistency of flows with Figure 10.10.

and

$$\frac{dy}{dt} = -\frac{x}{c} + \frac{B}{c}\sin(2\pi f t) \tag{10.34}$$

where B and f represent the amplitude and frequency, respectively, of the periodic stimulus. As shown in Figure 10.12a, the natural frequency of the van der Pol
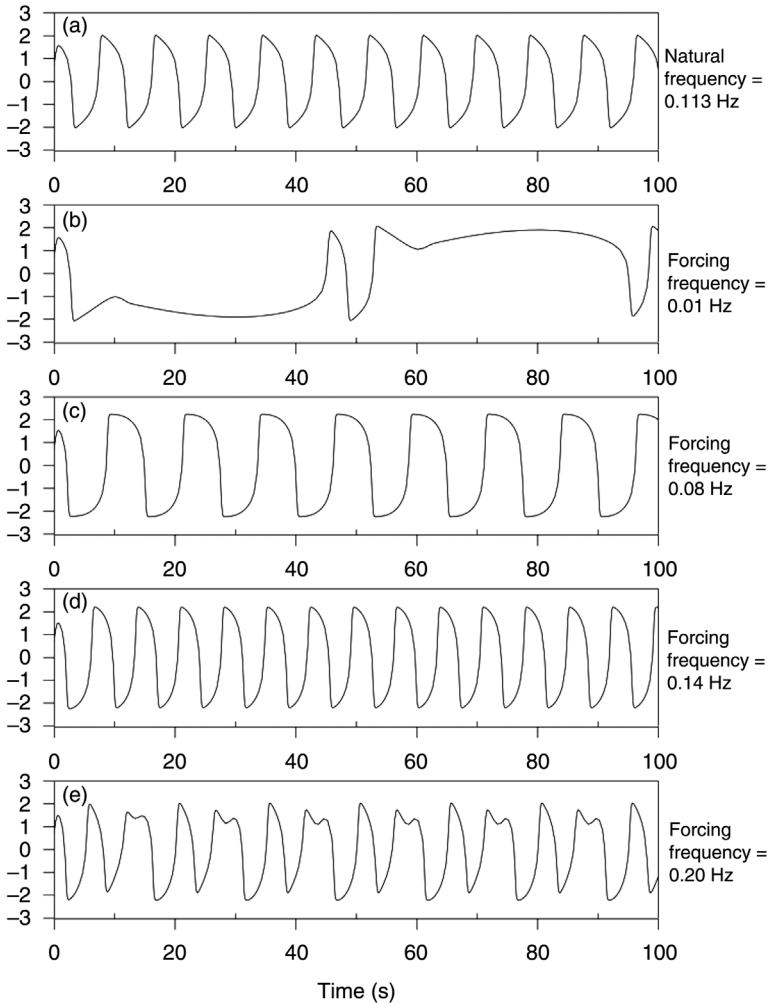
FIGURE 10.12   Responses of the van der Pol oscillator to external sinusoidal forcing. The top panel (a) shows spontaneous oscillations at 0.113 Hz. The system becomes entrained to the frequency of the external forcing when the former is close to the natural frequency of the oscillator.

oscillator is 0.113 Hz. When this system is stimulated periodically at frequencies that are substantially lower (Figure 10.12b) or higher (Figure 10.12e) than 0.113 Hz, the result is a mixture of the forcing and natural frequencies. However, at frequencies close to 0.113 Hz, the van der Pol system adopts the frequencies of the driving stimulus (Figure 10.12b and c). Frequency entrainment is an important phenomenon from a practical standpoint, since it forms the basis on which heart pacemakers work. Frequency entrainment also explains the synchronization of

many biological rhythms to the light–dark cycle, the coupling between respiration and blood pressure, as well as the synchronization of central pattern generators during walking and running.

10.3.3   Modeling Cardiac Dysrhythmias

Under normal circumstances, the cardiac cycle originates as electrical activity generated by the sinoatrial node. This impulse spreads through the atrial muscula-ture, the atrioventricular node, and finally through the Purkinje network of conducting fibers to elicit ventricular contraction. A common class of disorders, known as *atrioventricular heart block*, can occur, in which the relative timing between atrial and ventricular contractions becomes impaired. One line of thought postulates that these dysrhythmias are the result of dynamic interaction among two or more coupled nonlinear oscillators in heart tissue. One of the simplest oscillator models that can demonstrate this type of phenomena is the *Poincaré oscillator*. This dynamic system is characterized by the following set of differential equations:

$$\frac{dr}{dt} = ar(1 - r) \tag{10.35}$$

and

$$\frac{d\Phi}{dt} = 2\pi \tag{10.36}$$

where r represents the radial coordinate and Φ ($-\infty < \Phi < \infty$) the angular coordi-nate (in radians) of the state point in the phase plane. These dynamics give rise to a limit cycle that rotates anticlockwise on the unit circle (Figure 10.13). As such, it is more convenient to define the new angular coordinate ϕ as follows:

$$\phi = \frac{\Phi}{2\pi} (\text{mod}\,1) \tag{10.37}$$

so that $0 \leq \phi < 1$. ϕ is also known as the (normalized) *phase* of the oscillation.

Guevara and Glass (1982) considered what would occur if this oscillator were perturbed by an isolated, brief stimulus in the limit where $a \to \infty$. This is illustrated in Figure 10.13. The stimulus is represented by the heavy arrow that shifts the state point from its prestimulus location, corresponding to phase ϕ (shown in black), to its poststimulus location, corresponding to phase θ (shown in gray). The length of the arrow represents b, the magnitude of the stimulus. Because a is infinite, the new state point moves instantaneously along the radial direction back to the limit cycle. It can be seen that when $0 < \phi < 0.5$, the state point is pushed back to a location that it had previously traversed; thus, the perturbation causes a *phase delay*. On the other hand, when $0.5 < \phi < 1$, the same stimulus would push the state point to a location
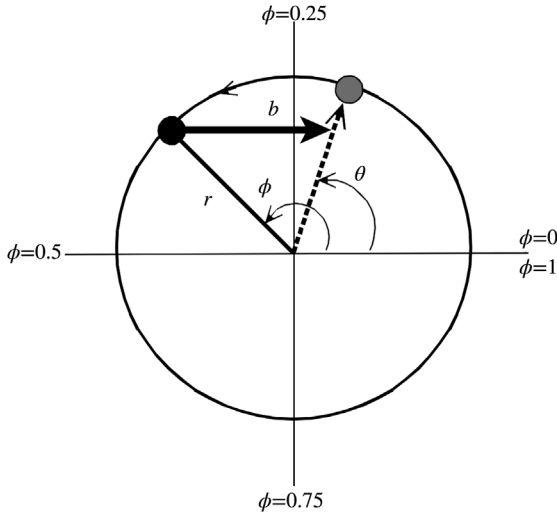
FIGURE 10.13   The Poincaré oscillator. Application of a brief stimulus of magnitude *b* leads to a resetting of the phase from ϕ to θ (old state point shown as black circle, new state point shown as gray circle). When $0 < \phi < 0.5$, the perturbation produces a delay in phase, but when $0.5 < \phi < 1$, the same stimulus causes an advance in phase.

further along the limit cycle; in this case, the perturbation causes a *phase advance*. This type of phenomenon is known as *phase resetting*. By careful consideration of the geometrical details of Figure 10.13, it can be shown that the new phase θ is related to the old phase ϕ through the following relationship:

$$\cos(2\pi\theta) = \frac{b + \cos(2\pi\phi)}{\sqrt{1 + 2b\cos(2\pi\phi) + b^2}} \qquad (10.38)$$

However, depending on the magnitude of the stimulus *b*, Equation 10.38 can yield very different looking functions that relate θ to ϕ; these functions are termed *phase transition curves*. As illustrated in Figure 10.14, when $b < 1$ (*weak resetting*), there is a phase delay (i.e., $\theta < \phi$) for the range $0 < \phi < 0.5$ and a phase advance (i.e., $\theta > \phi$) for $0.5 < \phi < 1$, as noted earlier. Since the average slope of the phase transition curve is unity, this type of phase resetting is also commonly referred to as *Type 1* resetting. However, when $b > 1$ (*strong resetting*), the effect of the perturbation on the trajectory of the state point becomes interesting and somewhat surprising at first glance. As ϕ increases from zero toward 0.5, θ initially increases but subsequently decreases so that when ϕ attains the value of 0.5, θ becomes zero. The reason for this form of relationship may be better understood if one considers what happens for the case when ϕ equals 0.5: The perturbation forces the state point to a location on the horizontal axis that is past the center of the circle. The closest point on the limit cycle to this new state location is at $\phi = 0$. When ϕ increases
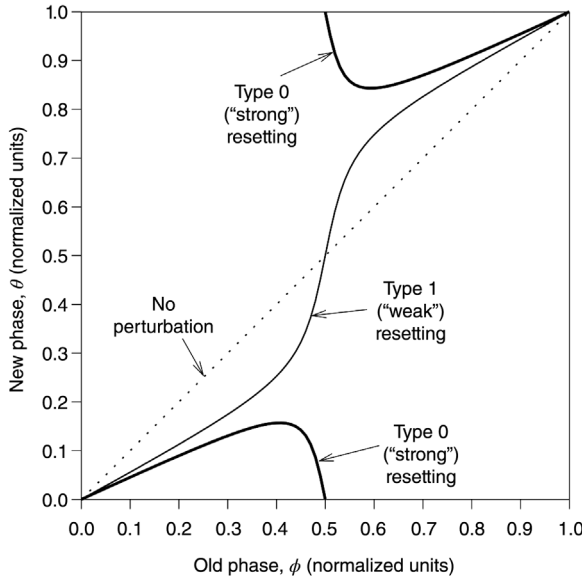
FIGURE 10.14   Phase transition curves for the Poincaré oscillator. The dotted (identity) line represents the case in which there is no perturbation. When the system is perturbed by a brief stimulus of magnitude b $(b < 1)$, Type 1 or weak resetting occurs (light solid curve). However, when $b > 1$, Type 0 or strong resetting occurs (bold solid curve) in which there is an apparent discontinuity at $\phi = 0.5$.

beyond 0.5, perturbation of the state point leads to a resetting of phase to points that begin at $\phi = 1$, decrease below 1, but eventually increase back toward 1. As a result, the phase transition curve shows an apparent discontinuity and the average slope becomes zero (Figure 10.14, bold curve). This kind of resetting is known as *Type 0* resetting.

Equation 10.38 characterizes the effect on the Poincaré oscillator of a single, isolated stimulus, delivered when the phase of the oscillation is ϕ. This can be extended to produce a corresponding formula that characterizes how a periodic train of impulses would affect the behavior of the oscillator. If ϕ_i is the phase of the oscillator immediately prior to the ith stimulus, then the phase just before the next stimulus occurs is given by

$$\phi_{i+1} = \frac{1}{2\pi} \cos^{-1} \left(\frac{b + \cos(2\pi\phi_i)}{1 + 2b \cos(2\pi\phi_i) + b^2} \right) + \frac{T_s}{T_0} \qquad (10.39)$$

where T_0 is the period of the limit cycle and T_s is the interval between successive stimuli. Guevara and Glass showed that the nonlinear finite difference equation represented by Equation 10.39 can give rise to dynamics that are qualitatively similar to the dysrhythmias that have been observed in the electrocardiogram. To
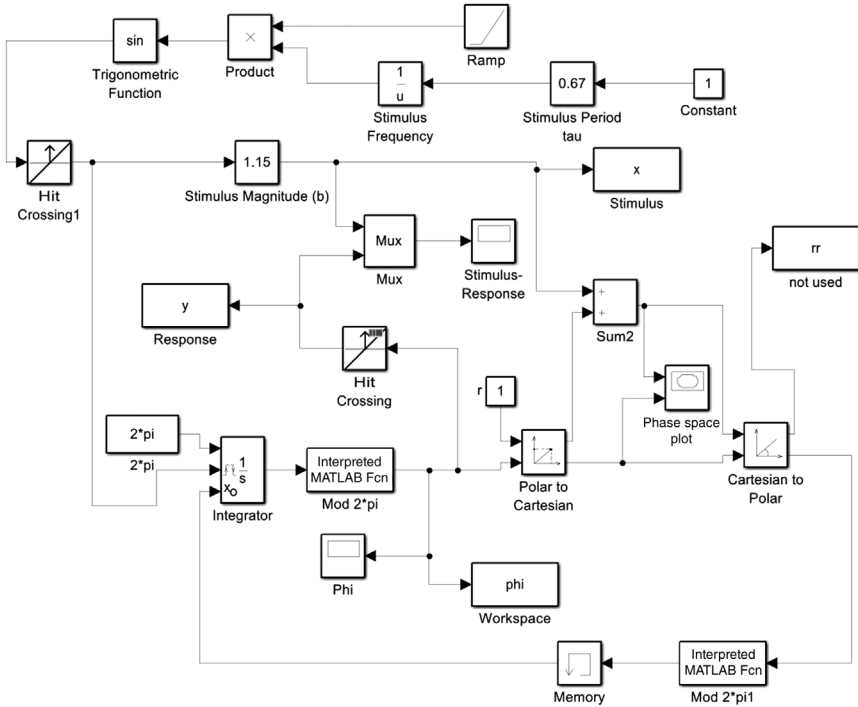
FIGURE 10.15 SIMULINK implementation of the Poincaré oscillator model `poincare.slx`.

reproduce their simulations, we have developed the SIMULINK implementation (labeled `poincare.slx`) of the Poincaré model, as shown in Figure 10.15. Since the constant *a* in Equation 10.35 is taken to be infinitely large, the state point is assumed to instantaneously return to the limit cycle after each perturbation by the external stimulus. As such, *r* is assumed to be always equal to unity, and the radial dynamics in Equation 10.35 are neglected. In this model, we have also assumed that whenever the rotating arm of the oscillator passes through $\phi = 0$, the system will generate a unit impulse (simulating a neural spike). The "stimulus period" (`tau` in Figure 10.15) that has to be specified prior to running the simulation is normalized with respect to the natural period of the limit cycle, that is, `tau` $= T_s/T_0$.

Figure 10.16 displays some examples of the response of the Poincaré oscillator to a periodic stimulation of magnitude $b = 1.13$. In Figure 10.16a, the period of the stimulation is 75% as long as the natural oscillatory cycle of the model. Entrainment occurs so that the Poincaré oscillator "fires" at approximately the same frequency as the external periodic stimulus. This kind of entrainment is also called *1:1 phase locking*. When the normalized stimulation period is reduced to 0.69 (Figure 10.16b), the Poincaré oscillator now alternates between a long interspike interval and a short interspike interval. For every two stimulus spikes, the system responds
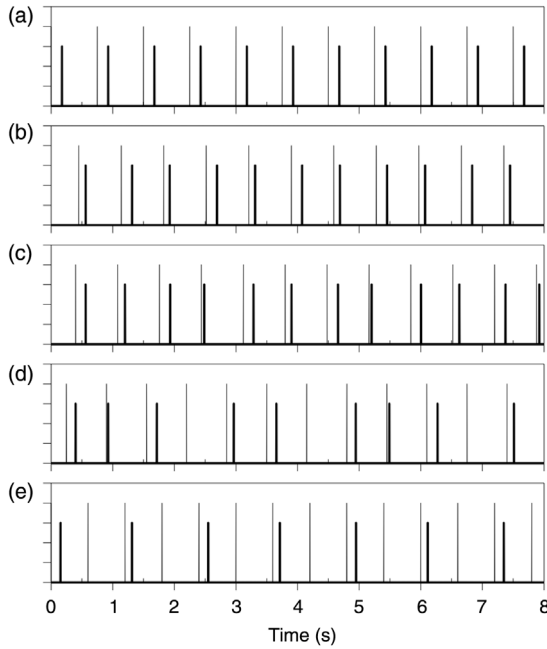
FIGURE 10.16   Responses (heavy bars) of the Poincaré oscillator to periodic stimulation (light bars) of magnitude $b = 1.13$. The different panels represent responses to different stimulation periods: $T_s/T_0 = 0.75$, 0.69, 0.68, 0.65, and 0.60 in panels (a)–(e), respectively. See text for further details.

with one long interval and one short interval; then, the pattern repeats itself. This type of phenomenon is known as *2:2 phase locking*. With further decrease of T_s/T_0 to 0.68 (Figure 10.16c), four stimulus spikes give rise to four response impulses, but the time relationship between each stimulus spike and its corresponding response is different for the four pairs, producing *4:4 phase-locking*. In Figure 10.16d, where T_s/T_0 is decreased to 0.65, the periodicity in the response disappears. Now the response spikes appear in an unpredictable fashion, giving rise also to skipped beats (e.g., at $t \sim 2.5$ s and $t \sim 4.5$ s in Figure 10.16d). Guevara and Glass have argued that this pattern reflects chaotic dynamics arising in this highly nonlinear system. Finally, in Figure 10.16e, decreasing T_s/T_0 to 0.6 leads to a re-emergence of periodicity in the response. However, under these conditions, four stimulus spikes correspond to only two response spikes, producing what is known as *4:2 phase locking*.

   The changes in dynamics exhibited by the Poincaré oscillator or any nonlinear system that occurs abruptly as a system parameter is decreased or increased are commonly referred to as *bifurcations*. In the examples considered above, the bifurcations occurred at the points where changes in the value of the stimulus period led to sudden changes from one type of phase-locking to another mode. Bifurcations
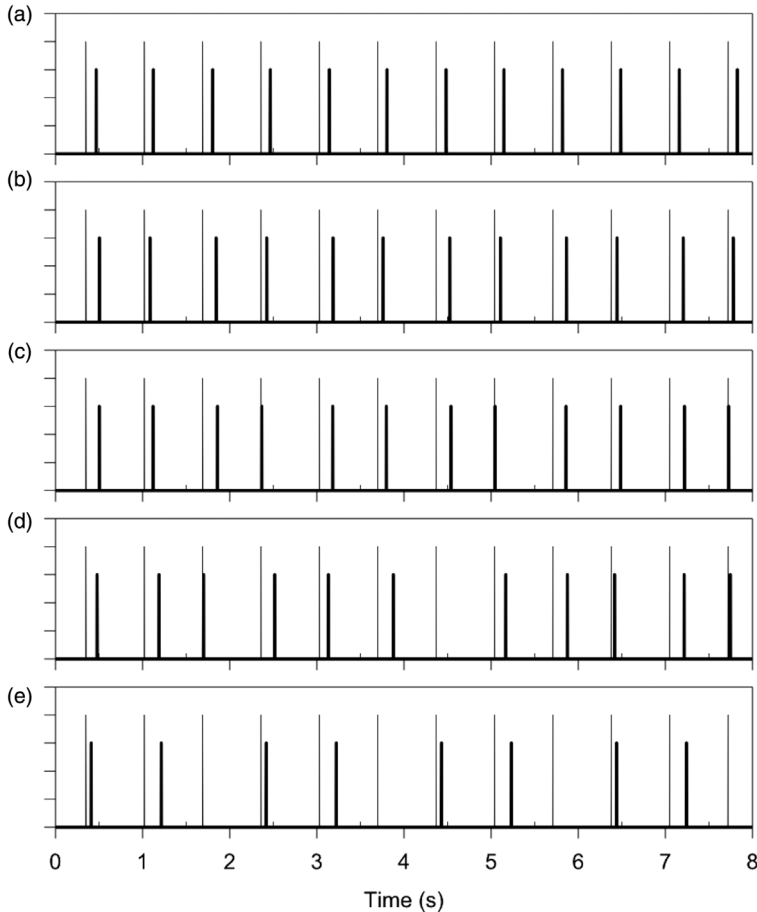
FIGURE 10.17 Responses (heavy bars) of the Poincaré oscillator to periodic stimulation (light bars) of period $T_s/T_0 = 0.67$. The different panels represent responses to different stimulation magnitudes: $b = 1.40$, 1.22, 1.15, 1.12, and 1.02 in panels (a)–(e), respectively. See text for further details.

also occur when the magnitude of the stimulus (b) is continually varied. Figure 10.17 shows examples of the model response to periodic stimulation when T_s/T_0 is kept constant at 0.67, but the stimulus magnitude is varied. In Figure 10.17a, when $b = 1.4$, there is 1:1 phase-locking. Decreasing b to 1.22 leads to 2:2 phase-locking (Figure 10.17b), and subsequently, 4:4 phase-locking when $b = 1.15$ (Figure 10.17c). Decreasing b a little further to 1.12 produces chaotic dynamics (Figure 10.17d). Finally, with b decreased to 1.02, the system once again exhibits periodic dynamics (Figure 10.17e). There is now 3:2 phase-locking of the type where the stimulus–response spike interval becomes progressively longer until the Poincaré oscillator misses a beat (e.g., at $t \sim 2\,\mathrm{s}$ and $t \sim 4\,\mathrm{s}$ in Figure 10.17e).

This kind of pattern is similar to the clinically observed electrocardiographic phenomenon known as "second-degree AV (atrioventricular node) block with Wenckebach periodicity."

10.4 THE DESCRIBING FUNCTION METHOD

10.4.1 Methodology

The describing function method, sometimes also known as the method of *harmonic balance*, is useful in determining the conditions that produce limit cycles in relatively simple nonlinear systems. It may be viewed as an extension of the Nyquist stability criterion discussed in Chapter 6. The method assumes a closed-loop nonlinear model of the type displayed in Figure 10.18. This system can be decomposed into two parts: a linear portion that contains dynamic features, $G(s)$, and a static nonlinear component, characterized by the function $F(\varepsilon)$. It is also assumed that this system is oscillating at some fundamental frequency ω (given in radians per unit time) without any input perturbation (i.e., in Figure 10.18, $u = 0$). In general, the output x of the nonlinear subsystem will be a periodic oscillation with fundamental frequency plus its harmonic components. This can be expressed as a Fourier series:

$$x(t) = X_0 + \sum_{n=1}^{\infty} (a_n \sin (n\omega t) + b_n \cos (n\omega t)) \tag{10.40a}$$

On the other hand, if we assume the linear "plant" subsystem to be low-pass in nature, the harmonics in x will be filtered out and the output y of the linear subsystem is likely to be approximately sinusoidal in form. Since ε is equal to the negative of y in the absence of any external input, we can assume that

$$\varepsilon(t) = A \sin (\omega t) \tag{10.41}$$
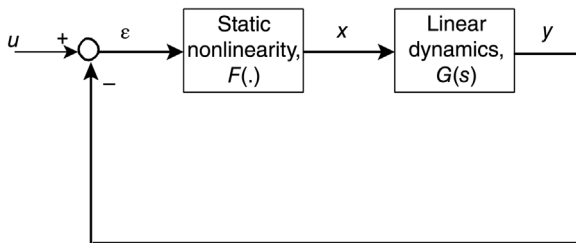
FIGURE 10.18   Closed-loop nonlinear control system with static nonlinearity and linear dynamic components.

And for purposes of assessing stability of the closed-loop system, we focus only on the fundamental component of $x(t)$, so that from Equation 10.40a we obtain

$$x(t) \approx X_0 + a_1 \sin (\omega t) + b_1 \cos (\omega t) \qquad (10.40\text{b})$$

The *describing function* DF of the nonlinearity $F(\cdot)$ is defined as the complex coefficient for the fundamental frequency output divided by the input signal amplitude. The mathematical definition is as follows:

$$\text{DF}(A) = \frac{a_1 + jb_1}{A} \qquad (10.42)$$

In Equation 10.42, DF is shown explicitly to be a function of the input amplitude A. Although, in principle, DF can also be a function of frequency, this dependence is rare under most practical circumstances. In general, the nonlinearity $F(\cdot)$ is assumed to be static and, therefore, independent of frequency.

If we let $\theta = \omega t$, then by making use of the property of orthogonality for $\sin \theta$, we can deduce a_1 by multiplying both sides of Equation 10.41 by $\sin \theta$ and then integrating over the range of 0–2π. After simplification and rearrangement of terms, we obtain

$$a_1 = \frac{1}{\pi} \int_0^{2\pi} x(\theta) \sin \theta d\theta \qquad (10.43)$$

Similarly, it can be shown that

$$b_1 = \frac{1}{\pi} \int_0^{2\pi} x(\theta) \cos \theta d\theta \qquad (10.44)$$

Since $x(\theta)$ is periodic, changing the range of integration in Equations 10.43 and 10.44 will not alter the values of a_1 and b_1. For reasons that will become self-evident as our discussion proceeds, we choose to change the integral limits in Equations 10.43 and 10.44 to the range $-\pi$ to $3\pi/2$. Also, we introduce the following change of variable:

$$z = \sin \theta \qquad (10.45)$$

so that the differentials dx and $d\theta$ are related by

$$dz = \cos \theta d\theta = \pm \sqrt{1 - z^2} \, d\theta \qquad (10.46)$$

Note that the square-root term in Equation 10.46 will take on positive values when $-\pi/2 < \theta < \pi/2$ and negative values when $\pi/2 < \theta < 3\pi/2$. Thus, Equation 10.44 becomes

$$b_1 = \frac{1}{\pi}\left(\int_{-\pi/2}^{\pi/2} x(\theta)\cos\theta d\theta + \int_{\pi/2}^{3\pi/2} x(\theta)\cos\theta d\theta\right)$$

$$= \frac{1}{\pi}\left(\int_{-1}^{1} x(z)dz + \int_{1}^{-1} x(z)dz\right) = 0 \qquad (10.47)$$

The result derived from Equation 10.47 is important, as it implies that the imaginary part of DF will be zero as long the nonlinear function $F(\cdot)$ is *single-valued*. If there is hysteresis in the nonlinearity, the two integrals in Equation 10.47 (second line) would not be equal in magnitude and opposite in sign, and as a consequence, b_1 would not be zero.

The same analysis applied to Equation 10.43 yields the following result for a_1:

$$a_1 = \frac{1}{\pi}\left(\int_{-\pi/2}^{\pi/2} x(\theta)\sin\theta d\theta + \int_{\pi/2}^{3\pi/2} x(\theta)\sin\theta d\theta\right)$$

$$= \frac{1}{\pi}\left(\int_{-1}^{1} x(z)\frac{z}{\sqrt{1-z^2}}dz + \int_{1}^{-1} x(z)\frac{-z}{\sqrt{1-z^2}}dz\right) \qquad (10.48)$$

$$= \frac{2}{\pi}\int_{-1}^{1} x(z)\frac{z}{\sqrt{1-z^2}}dz$$

The following expression, analogous to the Nyquist stability criterion, provides the conditions under which a limit cycle of amplitude A and angular frequency ω might exist:

$$1 + DF(A)G(j\omega) = 0 \qquad (10.49a)$$

Note the similarity in form between the above equation and Equation 6.8, which characterizes the condition in which any *linear* closed-loop system becomes unstable. One might consider Equation 10.49a to be an extension of the Nyquist criterion (see Section 6.4) to a particular class of nonlinear closed-loop systems.

Equation 10.49a can be rearranged into the following form:

$$G(j\omega) = -\frac{1}{DF(A)} \qquad (10.49b)$$

Equation 10.49b can be solved graphically by plotting $G(j\omega)$ on the Nyquist plane and determining the values of A and ω from the point of intersection of the $G(j\omega)$ curve and the line running along part of the real axis, represented by $-1/DF(A)$.

10.4.2 Application: Periodic Breathing with Apnea

To illustrate a specific application of the describing function method, we turn to the model of Cheyne–Stokes breathing discussed in Section 6.7. As one might recall, this was a linearized model. However, for our present purposes, we will introduce a *thresholding nonlinearity* into the model by assuming that the controller output will become zero once the operating level of P_{aCO_2} falls below a certain value B. In other words, the simulated episodes of Cheyne–Stokes breathing would include periods of apnea. A schematic block diagram of this model is shown in Figure 10.19a, and examples of the waveforms in P_{aCO2} and $\dot{V}_E$ that one would expect to find are displayed in Figure 10.19b.

For simplicity, in the current example we will assume that there is only one chemoreflex loop in the system and that, unlike the example considered in Section 6.7, the controller responds instantaneously to changes in P_{aCO2}. Suppose the controller response is given by

$$
\begin{aligned}
\dot{V}_E &= S_{CO_2}\left(P_{aCO_2} - B\right), \quad P_{aCO_2} > B \\
&= 0, \quad P_{aCO_2} \le B
\end{aligned}
\tag{10.50}
$$

where S_{CO2} is the slope of the steady-state ventilatory response to CO_2. We assume also that during periodic breathing, the P_{aCO2} waveform can be characterized by the following expression:

$$
P_{aCO_2} = A \sin\theta + P_M
\tag{10.51}
$$

where $\theta = \omega t$ and P_M represents the mean level of the arterial P_{CO2} signal. Substituting Equation 10.51 into Equation 10.50, we obtain

$$
\begin{aligned}
\dot{V}_E &= S_{CO_2}(A \sin\theta + P_M - B), \quad 0 < \theta < \theta_1 \quad \text{or} \quad \theta_2 < \theta < 2\pi \\
&= 0, \quad \theta_1 \le \theta \le \theta_2
\end{aligned}
\tag{10.52}
$$

where θ_1 and θ_2 represent the two points in the periodic breathing cycle at which P_{aCO2} crosses the apneic threshold B (see Figure 10.19b). Thus, θ_1 and θ_2 can be computed from Equation 10.51:

$$
\theta_{1,2} = \sin^{-1}\left(\frac{B - P_M}{A}\right)
\tag{10.53}
$$

(a)

(b)
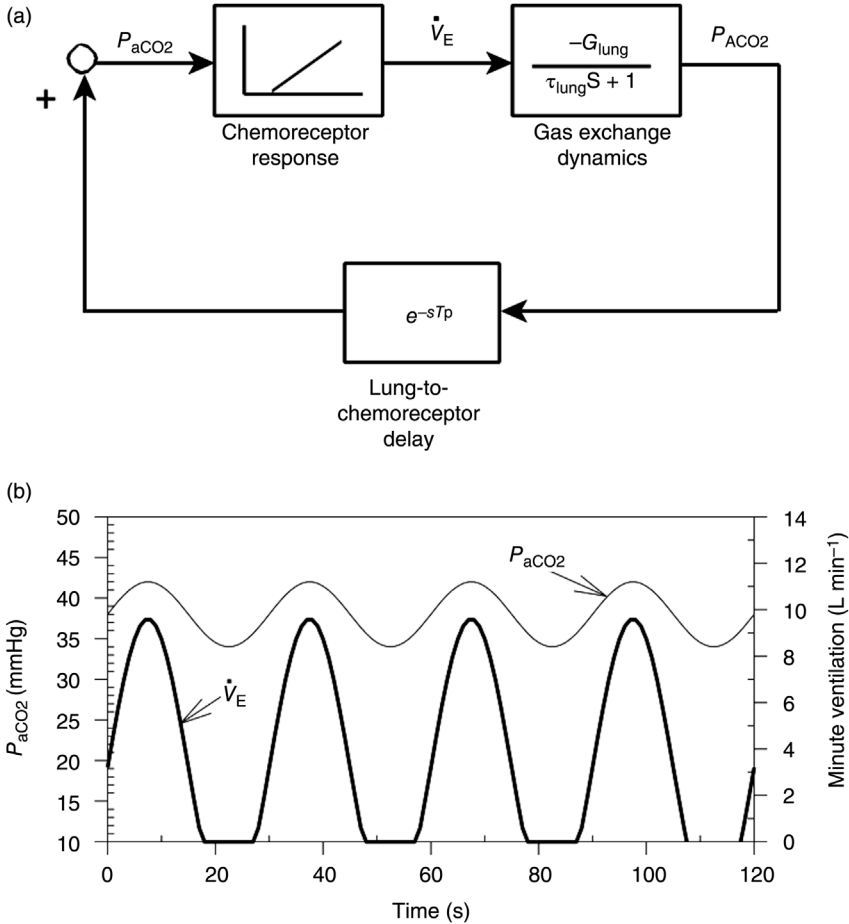
FIGURE 10.19 (a) Model of periodic breathing with apnea. (b) Input (P_{aCO2}) to and output ($\dot{V}_E$) from the thresholding nonlinearity.

If we assume, as illustrated in Figure 10.19b, that $B < P_M$, then $\sin\theta_1 < 0$, implying that θ_1 will be in the third quadrant ($\pi < \theta_1 < 3\pi/2$). If we define

$$\theta_0 = \sin^{-1}\left(\frac{|B - P_M|}{A}\right) \tag{10.54}$$

so that if $0 < \theta_0 < \pi/2$, then

$$\theta_1 = \pi + \theta_0 \tag{10.55a}$$

and

$$\theta_2 = 2\pi - \theta_0 \tag{10.55b}$$

We know from the previous section that since Equation 10.50 is a single-valued function, the imaginary part (b_1) of DF must equal zero. In order to evaluate the real part a_1, we substitute the expressions in Equation 10.52, 10.55a, and 10.55b into Equation 10.43 to obtain

$$
\begin{aligned}
a_1 &= \frac{1}{\pi}\left(\int_0^{\theta_1} S_{CO_2}(A\sin\theta + P_M - B)\sin\theta\, d\theta + \int_{\theta_2}^{2\pi} S_{CO_2}(A\sin\theta + P_M - B)\sin\theta\, d\theta\right) \\
&= \frac{S_{CO_2}A}{2\pi}\left(\pi + 2\theta_0 - \sin(2\theta_0) + \frac{4(P_M - B)}{A}\cos\theta_0\right)
\end{aligned}
\tag{10.56}
$$

Therefore, the describing function of the nonlinear chemoreceptor characteristic is

$$DF(A) \equiv \frac{a_1}{A} = \frac{S_{CO_2}}{2\pi}\left(\pi + 2\theta_0 - \sin(2\theta_0) + \frac{4(P_M - B)}{A}\cos\theta_0\right) \tag{10.57}$$

where θ_0 is determined from Equation 10.54. The linear dynamic portion of this model (see Figure 10.19b and Section 6.7) is characterized by the following frequency response:

$$G(j\omega) = \frac{G_{lung}}{\left(1 + j\omega\tau_{lung}\right)}e^{-j\omega T_d} \tag{10.58}$$

where T_d represents the lung-to-chemoreceptor delay and G_{lung} and τ_{lung} were defined in Equations 6.45 and 6.46.

The existence of a limit cycle is predicted if the locus of $G(j\omega)$ on the Nyquist plane intersects with the locus defined by $-1/DF(A)$. Figure 10.20 shows the solution obtained using Matlab m-file df_resp.m for the case of the patient with congestive heart failure. As in Section 6.7, the parameter values used to represent this type of subject were $V_{lung} = 2.5\,L$, $K_{CO2} = 0.0065\,mmHg^{-1}$, $\dot{V}_E = 0.12\,L\,s^{-1}$, $\dot{V}_D = 0.03\,L\,s^{-1}$, $P_{ICO2} = 0$, and $P_{aCO2} = P_{ACO2} = 40\,mmHg$, $Q = 0.05\,L\,s^{-1}$, $B = 37$ mmHg, and $S_{CO2} = 0.02\,L\,s^{-1}\,mmHg^{-1}$. Using Equations 6.45 and 6.46, it can be determined that these parameters produced values for G_{lung} and τ_{lung} of 108 mmHg s$\,L^{-1}$ and 6.75 s, respectively. The circulatory delay (T_D) employed was 14.2 s. Since the function $-1/DF(A)$ consists only of real values, its locus merely retraces a portion of the real axis. The point at which the two functions intersect (indicated by the open circle in Figure 10.20) corresponds to a frequency of 0.026 Hz, which
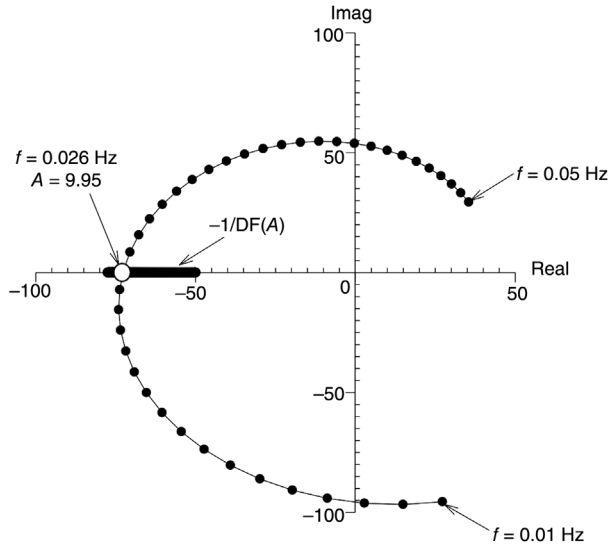
FIGURE 10.20   Illustration of the describing function method for determining the periodicity and amplitude of Cheyne–Stokes breathing in congestive heart failure. Thick line on real axis represents the locus of the function $-1/DF(A)$. Intersection of this locus with that of the linear transfer function $G(j\omega)$ yields solution for limit cycle.

translates into a periodicity of 38.5 s. This point also yields a value of 9.95 mmHg for the amplitude of the oscillation in P_{aCO2}. This calculation was undertaken merely to illustrate how the describing function method can be applied. As a predictor, it grossly underestimates the periodicity associated with Cheyne–Stokes breathing since it does not take into account the contributions of both chemoreflex loops and also ignores the response time associated with the chemoreflex. Incorporating these factors would produce more realistic predictions, but would make the expression for $G(j\omega)$ much more complicated.

10.5   MODELS OF NEURONAL DYNAMICS

We saw in Section 10.3 the utility of employing the van der Pol and Poincaré models as theoretical constructs for characterizing the dynamic behavior observed in physiological oscillators, such as cardiac and circadian pacemakers. However, they represent only the class of systems that are spontaneously oscillating. There is an even larger class of systems that do not spontaneously oscillate, but which can oscillate given sufficient stimulation. These systems provide a better description of the properties of nerve and muscle tissue. As the following discussion will show, these models may be viewed as closed-loop systems with both negative and positive feedback.

10.5.1 The Hodgkin–Huxley Model

The first relatively complete mathematical model of neuronal membrane dynamics was published by Hodgkin and Huxley in 1952. This work laid the foundation for further development of a quantitative approach to understanding the biophysical mechanism of action potential generation and was the seminal achievement that won them the Nobel Prize in 1963. Their model was based largely on empirical findings obtained through application of the voltage–clamp technique, which we discussed briefly in Section 8.4.4. This melding of physical intuition, modeling principles, and excellent experimental design is a classic example of first-class bioengineering research.

Under resting conditions, the intracellular space of the nerve cell is on the order of 60 mV more negative relative to the extracellular fluid. This net equilibrium potential is determined by the ionic concentration gradients across the slightly permeable membrane as well as by the effect of active transport by the sodium–potassium pump. There is a higher concentration of potassium ions inside the cell versus a higher concentration of sodium and chloride ions on the outside. However, the membrane permeabilities to sodium and potassium are strongly dependent on the membrane potential. Depolarization of the membrane potential leads to rapid changes in sodium permeability and a somewhat slower time-course in potassium permeability. The Hodgkin–Huxley model postulates that it is the initial rapid influx of sodium ions and the subsequent outflow of potassium ions that account for the generation of the action potential that follows the depolarizing stimulus. The chloride ions do not play much of a role but account primarily for a small leakage current into the cell. The electrical circuit analog of this model is displayed in Figure 10.21a. E_K, E_{Na}, and E_{Cl} represent the Nernst potentials for potassium, sodium, and chloride, respectively. Based on their measurements on the squid giant axon, Hodgkin and Huxley employed values of -77, 50, and -54 mV for E_K, E_{Na}, and E_{Cl}, respectively. (It should be noted that they assumed the membrane potential difference to be measured outside relative to inside; thus, their sign convention was opposite to what has generally been adopted since their early work.) C represents the membrane capacitance, which is on the order of $1\,\mu F\,cm^{-2}$. g_K, g_{Na}, and g_{Cl} represent the respective conductance for potassium, sodium, and chloride ions that correspond to the resistive elements displayed in Figure 10.21a; because of their voltage dependence, g_K and g_{Na} have been shown as variable resistors.

Application of Kirchoff's law to the circuit in Figure 10.21a yields the following equation relating the total membrane current I to the potential difference V across the membrane:

$$I = C\frac{dV}{dt} + g_k(V - E_k) + g_{Na}(V - E_{Na}) + g_{Cl}(V - E_{Cl}) \tag{10.59}$$

The dependence of g_{Na} on membrane voltage is characterized by the following expressions:

$$g_{Na} = G_{Na}m^3h \tag{10.60}$$
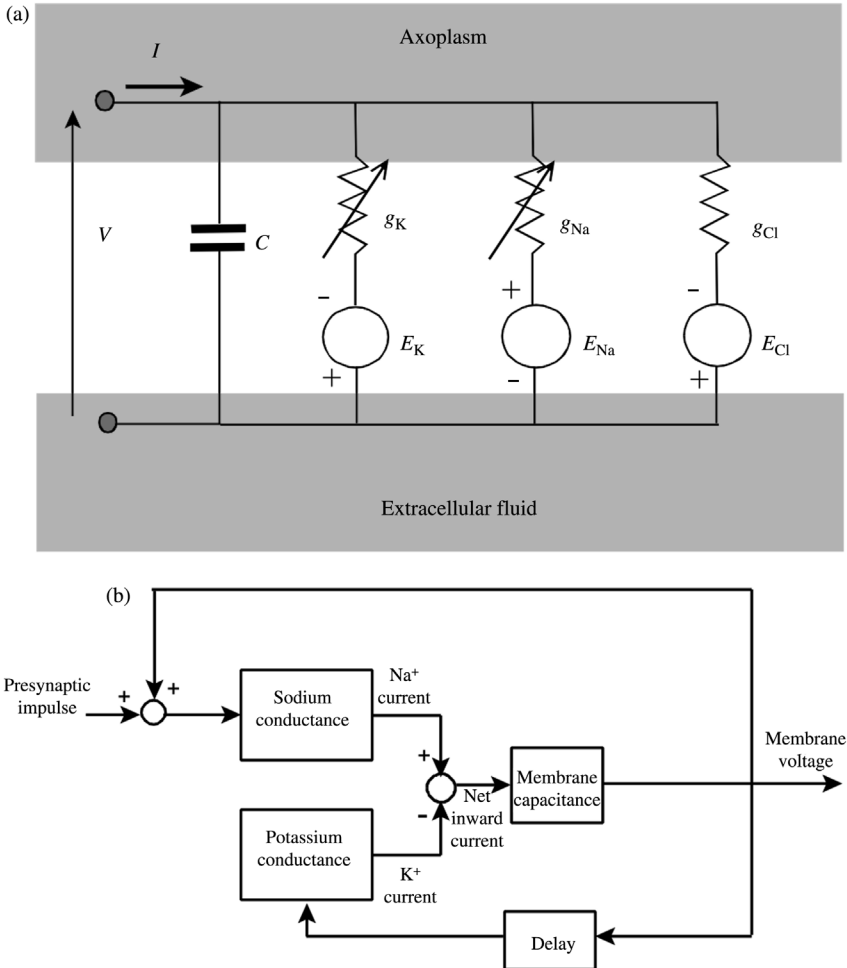
FIGURE 10.21 (a) Circuit analog of the Hodgkin–Huxley nerve membrane model. (b) The Hodgkin–Huxley model as a closed-loop system with negative and positive feedback.

where G_{Na} is a constant and assigned the value of 120 millimho cm^{-2} by Hodgkin and Huxley. The time-course of g_{Na} is assumed to be the result of interaction between two processes, one represented by the "activation" state variable m and the other by the "inactivation" state variable h, where m and h each may vary from 0 to 1. These state variables each obey first-order dynamics:

$$\frac{dm}{dt} = \alpha_m(1 - m) - \beta_m m \tag{10.61}$$

and

$$\frac{dh}{dt} = \alpha_h(1 - h) - \beta_h h \tag{10.62}$$

where the rate "constants" are voltage-dependent quantities defined by the following:

$$\alpha_m = 0.1(25 - V)\left(e^{(25-V)/10} - 1\right)^{-1} \tag{10.63}$$

$$\beta_m = 0.125e^{-V/80} \tag{10.64}$$

$$\alpha_h = 0.07e^{-V/20} \tag{10.65}$$

$$\beta_h = 1/\left(e^{(30-V)/10} - 1\right) \tag{10.66}$$

The potassium conductance follows similar but somewhat simpler dynamics:

$$g_K = G_K n^4 \tag{10.67}$$

where G_K is the constant and given the value of 36 millimho cm^{-2} in Hodgkin and Huxley's simulations. The single state variable n is assumed to obey the following first-order differential equation:

$$\frac{dn}{dt} = \alpha_n(1 - n) - \beta_n n \tag{10.68}$$

where

$$\alpha_n = 0.01(10 - V)\left(e^{(10-V)/10} - 1\right)^{-1} \tag{10.69}$$

and

$$\beta_n = 0.125e^{-V/80} \tag{10.70}$$

Finally, the membrane conductance for chloride ions is assumed to be constant and equal to 0.3 millimhos cm^{-2}.

Equations 10.59 through 10.70 constitute the Hodgkin–Huxley model. Functionally, the dynamic behavior represented by this set of equations can also be modeled in terms of the closed-loop system shown in Figure 10.21b. A depolarizing stimulus that exceeds the threshold produces an increase in sodium conductance, which allows sodium ions to enter the intracellular space. This leads to further depolarization and greater increase in sodium conductance. This positive feedback effect is responsible for the rising phase of the action potential. However, fortunately, there is a built-in inactivation mechanism (represented by h) that now begins to reverse the depolarization process. This reversal is aided by the negative feedback effect of the increase in potassium conductance, which follows a
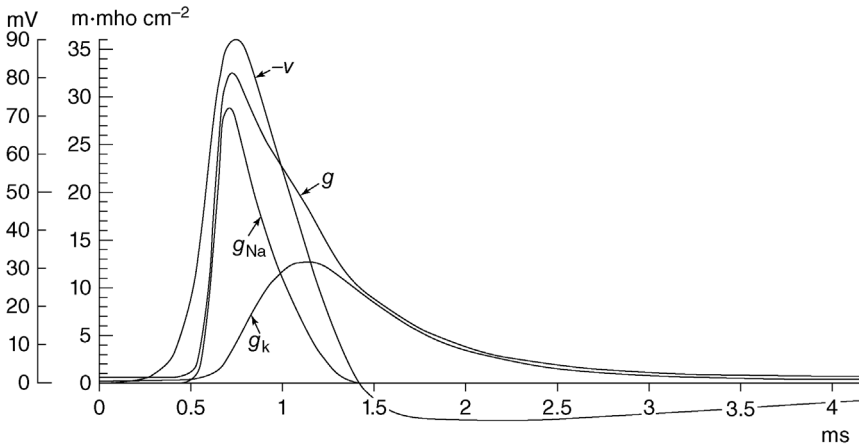
FIGURE 10.22   Time-courses of ionic conductances and membrane potential during an action potential, as predicted by numerical solution of the Hodgkin–Huxley equations. Membrane voltage is displayed as predominantly negative because Hodgkin and Huxley referenced all voltages to the intracellular fluid. (Reproduced from Hodgkin and Huxley (1952).)

time-course slower than that of the sodium conductance. The outflow of potassium ions leads to further repolarization of the membrane potential. Thus, the action potential is now in its declining phase. Because the potassium conductance remains above its resting level even after sodium conductance has returned to equilibrium, the nerve cell continues to be slightly hyperpolarized for a few more milliseconds following the end of the action potential. Figure 10.22, reproduced from the original Hodgkin–Huxley paper, shows the time-courses for V, g_{Na}, and g_K as predicted by the model to occur during an action potential. The curve labeled g represents the time-course of the overall membrane conductance. Multiplying this function with V allows us to predict the time-course for the net membrane current during the action potential.

10.5.2   The Bonhoeffer–van der Pol Model

Although the Hodgkin–Huxley equation set is able to reproduce many features of neuronal dynamics, it constitutes a somewhat unwieldy model, containing several state variables and a large number of empirical constants. Fitzhugh (1961) considered the Bonhoeffer–van der Pol (BvP) model as a simplified alternative, demonstrating the similarity of the phase-space characteristics of the former to the reduced phase-space behavior of the Hodgkin–Huxley model. The differential equations representing the BvP model are very similar to those of the van der Pol, which was discussed in Section 10.3.2. These are

$$\frac{dx}{dt} = c\left(y - \frac{x^3}{3} + x + z\right) \tag{10.71}$$

and

$$\frac{dy}{dt} = -\frac{1}{c}(x - a + by) \tag{10.72}$$

where a, b, and c are constants that satisfy the following constraints:

$$1 - \frac{2b}{3} < a < 1 \tag{10.73}$$

$$0 < b < 1 \tag{10.74}$$

$$b < c^2 \tag{10.75}$$

The variable z in Equation 10.71 represents the magnitude of the stimulus applied to the model. This can consist of two components. The first is the steady-state level of the stimulus (i.e., $z =$ constant); as we will demonstrate below, since this enters into Equation 10.71 explicitly, it can change the dynamics of the model quite dramatically. The second component of z is the transient contribution, which generally takes the form of a brief pulse of given magnitude.

By setting dx/dt and dy/dt in Equations 10.71 and 10.72 to zero, we can obtain expressions for the x- and y-nullclines, respectively. The x-nullcline is given by the cubic equation:

$$y = \frac{x^3}{3} - x - z \tag{10.76}$$

which is the same as the x-nullcline for the van der Pol model except that, here, the vertical position (in phase space) of the cubic curve is controlled by the stimulus level (z). The y-nullcline is given by

$$y = \frac{(a - x)}{b} \tag{10.77}$$

which is a straight line with negative slope. The conditions specified in Equations 10.73 through 10.75 guarantee that there will be only one intersection between the two nullclines. An example of the phase-plane diagram for $a = 0.7$, $b = 0.8$, $c = 3$, and $z = 0$ is displayed in Figure 10.23a. In this case, the single equilibrium point is located at coordinates $x = 1.2$ and $y = -0.625$ (point e in Figure 10.23a). This singular point may be shown, using the analysis technique of Section 9.2.1, to be a stable focus. To the left of the y-nullcline ($y < (a - x)/b$), evaluation of Equation 10.72 shows that dy/dt must be positive. Thus, the trajectories that cross the x-nullcline in this region must be directed upward. Conversely, in the region to the right of the y-nullcline, the trajectories that cross the x-nullcline must point downward. In the region above the x-nullcline, evaluation of Equation 10.71 shows that dx/dt is positive and, therefore, the horizontal arrows on the y-nullcline in this
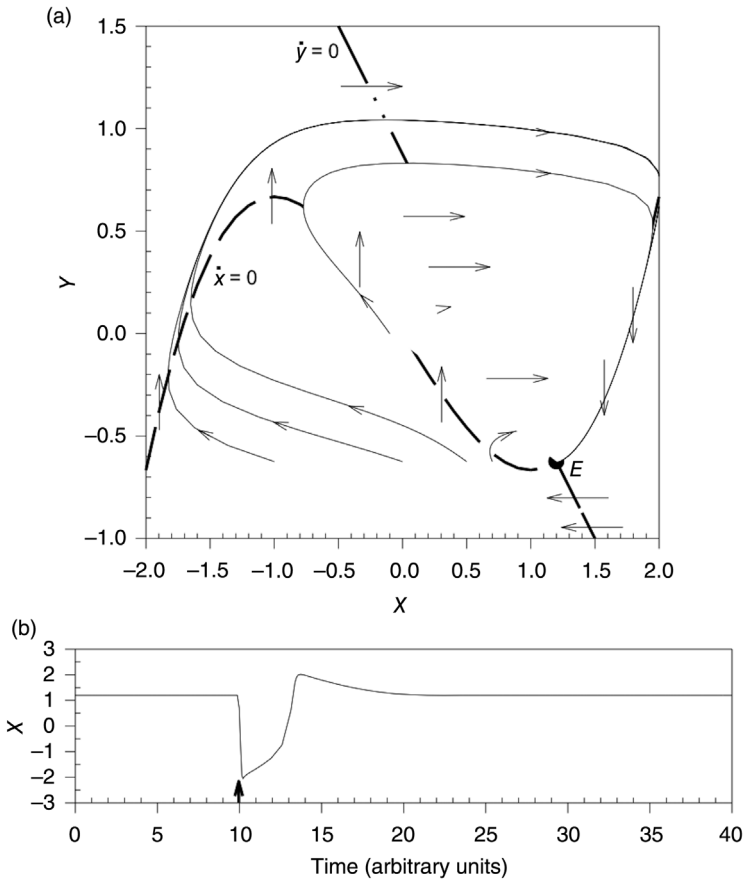
FIGURE 10.23 (a) Phase-plane diagram of the Bonhoeffer–van der Pol model with steady-state stimulus level (z) set equal to zero. Bold dashed curve represents x-nullcline ($dx/dt = 0$, vertical arrows); bold chained line represents y-nullcline ($dy/dt = 0$, horizontal arrows). Other curves are sample phase trajectories. E is the stable singular point; $a = 0.7$, $b = 0.8$, and $c = 3$. (b) Time response of BVP model to impulsive disturbance delivered at $t = 10$ (bold arrow).

region must be directed rightward. By similar reasoning, the trajectories that cross the y-nullcline below the x-nullcline must be directed leftward. Thus, the general flow is in a clockwise direction, as shown in Figure 10.23a.

The time-course of x, which simulates membrane voltage, for one of the sample phase trajectories is displayed in Figure 10.23b. Application of a brief pulse of magnitude ($\Delta z =$) -5 units at time $t = 10$ produces an abrupt decrease in x, followed later by a more gradual recovery, a small overshoot, and finally a slow return to baseline. Thus, this pattern of x simulates an inverted action potential (compare this with Figure 10.22). For this reason, we will refer to negative changes in x or z as
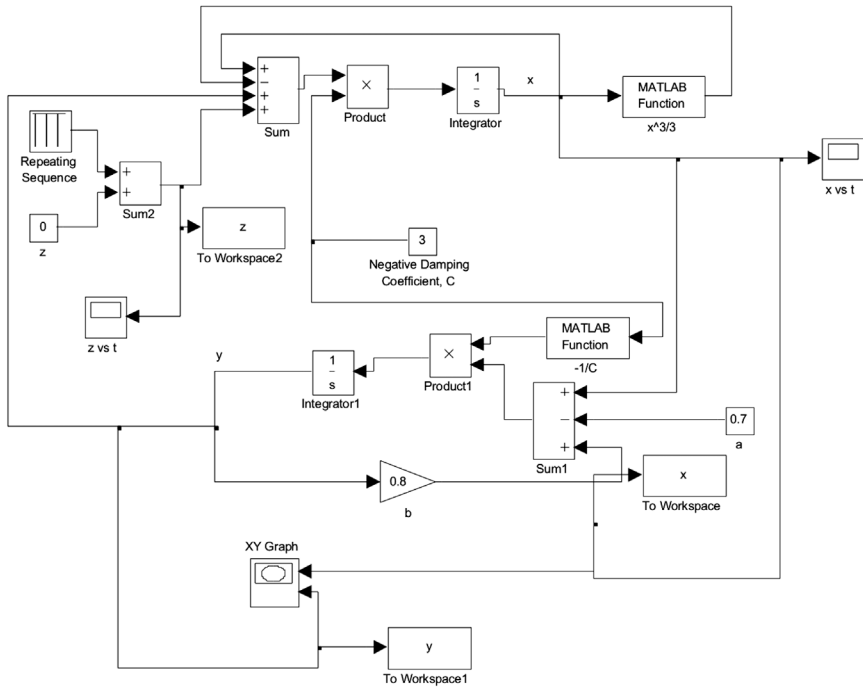
FIGURE 10.24 SIMULINK implementation (`bvpmod.slx`) of the Bonhoeffer–van der Pol model for parameter values: $c = 3$, $a = 0.7$, and $b = 0.8$. x represents membrane voltage (with inverted sign), while z represents the applied stimulus, which includes a constant level plus instantaneous impulses applied at times specified in the repeating sequence block.

"depolarizing" and positive changes as "repolarizing." The simulation result shown in Figure 10.23b and others that are displayed in subsequent figures were produced by a SIMULINK implementation of the BvP model, named `bvpmod.slx`. The diagram of the SIMULINK configuration for this model is shown in Figure 10.24. A `Repeating Sequence` block is used to generate the brief "shocks" (pulses in z) that are applied to the model. It should be noted from Figure 10.23a that, in general, only negative pulses in z (if sufficiently strong) would generate action potentials, since these displace the state point to the left of the equilibrium point.

One basic property of nerve and muscle tissue is the "all or none" phenomenon of thresholding. A pulse of insufficient magnitude, when applied to the nerve or muscle cell, produces only a small "depolarization" but does not elicit a full-fledged action potential. However, if the stimulus magnitude is increased above threshold, the action potential becomes unstoppable. The BvP model shows this type of behavior, as illustrated in Figure 10.25. In Figure 10.25a, a brief pulse of magnitude -1.0 produces only subthreshold behavior – a small and brief depolarization before x returns to baseline. In the phase-plane diagram, this is represented by a small displacement of the state point to the left of the equilibrium point E. As long as the
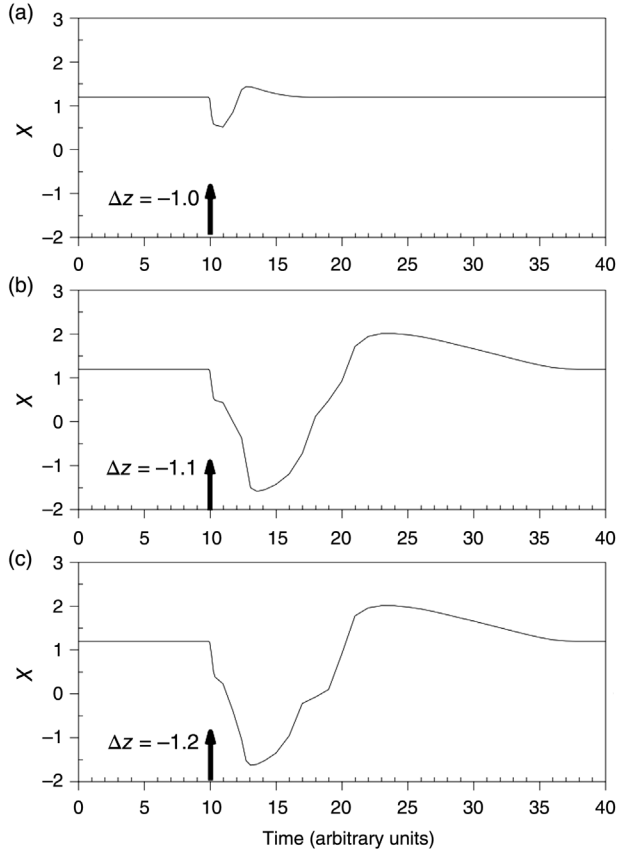
FIGURE 10.25 Demonstration of the threshold property of the Bonhoeffer–van der Pol model. Brief pulses of magnitudes -1.0 (a), -1.1 (b), and -1.2 (c) are applied through the variable z (baseline level of $z = 0$) at $t = 10$. Only the latter two cases are suprathreshold and produce "action potentials."

displaced state point falls to the right of the negatively sloped portion of the x-nullcline or not too far to the left of it, the resulting phase trajectory will follow a small loop that leads back into E. However, if the pulse is large enough to push the state point sufficiently leftward of the x-nullcline, the subsequent phase trajectory will be one that moves leftward and upward, turns to the right, and then moves back toward E. The corresponding time-course of x would be the action potential displayed in Figure 10.25b. Increasing the stimulus pulse magnitude does not alter the size of the action potential, as shown in Figure 10.25c and Figure 10.23a.

Another fundamental neuronal property is the presence of a refractory period. If a depolarizing stimulus is applied to a nerve cell too soon after the firing of an action potential, this stimulus would not elicit another action potential. The BvP model
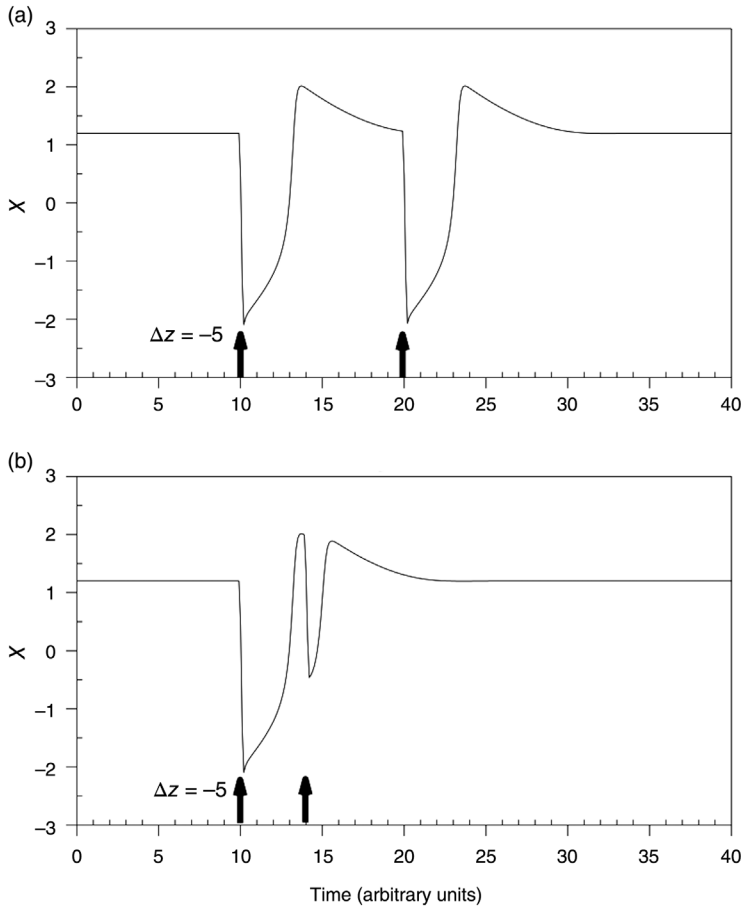
FIGURE 10.26 Demonstration of the refractory property of the Bonhoeffer–van der Pol model. (a) Brief pulses of magnitude -5 applied at $t = 10$ and $t = 20$ produce separate "action potentials." (b) The brief pulse applied at $t = 10$ produces an action potential, but the succeeding pulse at $t \sim 14$, during the "refractory interval," does not.

also exhibits this kind of behavior. Figure 10.26a shows the effect of stimulating the BvP model with two brief pulses of magnitude -5 units, spaced 10 time units apart. The second pulse occurs after much of the response to the first pulse has already taken place. Consequently, this second pulse leads to another action potential. In Figure 10.26b, the second pulse is applied only four time units after the application of the first pulse. This occurs during the early stages of "repolarization." The net result is a small and brief depolarization, but a second action potential does not take place. This behavior can be better understood if one turns again to the phase-plane diagram in Figure 10.23. It can be seen that any stimulus that displaces the state point horizontally to the left (negative Δz) when the latter is on the final two portions

of the phase trajectory will not change its subsequent movement much, because the state point will basically follow its original course back toward E.

Thus far, we have examined how the BvP model responds to brief pulses in z. What is the effect of changing the baseline level of z, which until now has been assumed to equal zero? From Equation 10.76, it is clear that giving z a nonzero value would shift the vertical position of the x-nullcline: Positive z-values would move the cubic curve downward, while negative z-values, corresponding to step depolarization, would move it upward. Figure 10.27 shows how the model would
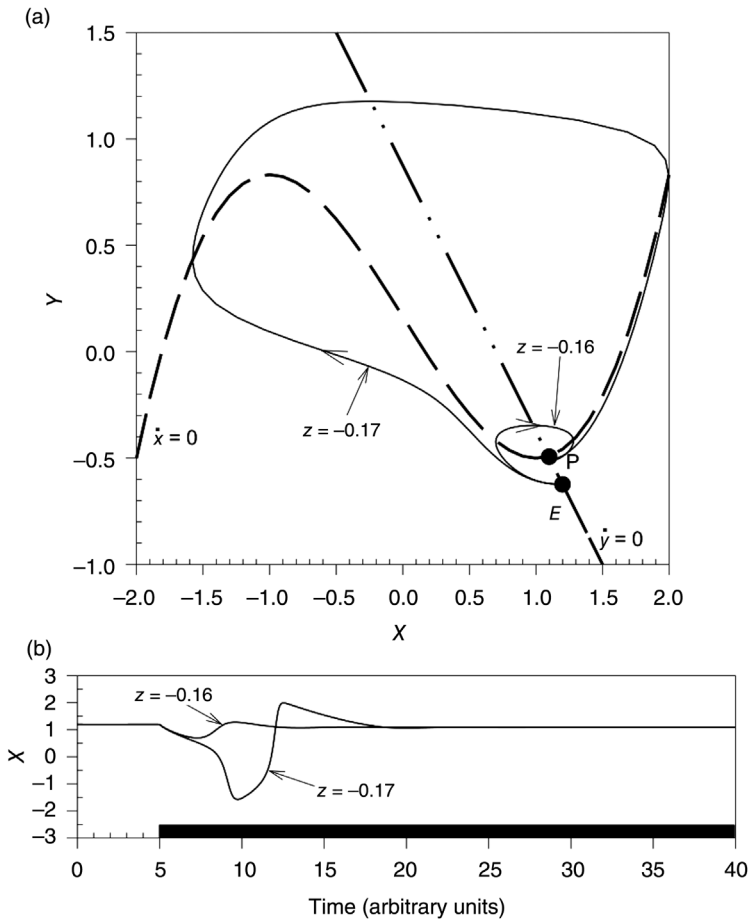
FIGURE 10.27 Responses of the Bonhoeffer–van der Pol model to "step depolarizations." (a) Phase-plane diagram showing the change in singular point from E to P following application of step in z of -0.16 units. The resulting response in x is subthreshold. When the step is made slightly more negative ($z = -0.17$), an action potential is generated. The singular point corresponding to $z = -0.17$ is located very close to P and therefore is not shown separately. (b) Time-courses of x following step depolarizations applied at $t = 5$ (indicated by black horizontal bar).

respond when subjected to step changes in z from zero to -0.16 and -0.17. In the phase-plane diagram (Figure 10.27a), the state-point starts off at E, prior to application of the step stimulus. When the step is applied, the x-nullcline moves upward so that the new singular point (labeled P) is now located at $x = 1.1$ and $y = -0.5$. The resulting response in x, however, is subthreshold (Figure 10.27b), and the corresponding phase trajectory is a small loop that begins at E and ends at P (Figure 10.27a). If the step in z is made only slightly more negative ($z = -0.17$), the response becomes quite different. Now, the phase trajectory that describes the dynamics of x between $z = 0$ and $z = -0.17$ takes the form of the large loop that corresponds to the generation of an action potential (Figure 10.27b). In both cases, the new singular points remain stable. Consequently, following the occurrence of the action potential, the state variable x settles down to a new steady level. However, when the applied steps in z are made sufficiently negative, in addition to being displaced further upward, the new singular point (U) also becomes unstable (Figure 10.28). Instead of converging to the new "equilibrium" level, x simply oscillates around it (Figure 10.28b). In the phase-plane diagram (Figure 10.28a), this corresponds to the phase trajectory that begins at E but eventually gets trapped in the limit cycle that encloses the unstable focus U. Thus, the BvP model predicts that when sufficiently large step depolarizations are applied, an infinite periodic train of action potentials will be generated. However, both BvP and Hodgkin–Huxley models are not able to simulate finite trains of action potentials, a phenomenon commonly observed in experimental nerve preparations.

10.6 NONPARAMETRIC IDENTIFICATION OF NONLINEAR SYSTEMS

The methods for analyzing nonlinear systems that we have discussed so far assume the existence of a structured model – in the form of a set of ordinary differential equations or difference equations – that is employed to characterize the dynamics of the phenomenon being studied. In order to estimate the parameters of the structured model, a common approach is to apply an optimization technique, in a manner similar to what we had discussed in Section 8.2.5. However, it can be extremely challenging to come up with a nonlinear model primarily based on data and with only very basic assumptions about the underlying process. Under such circumstances, it may be useful to extract as much information about the characteristics of the dynamic nonlinearities as possible first, using a "black-box" approach (see Section 8.1). Due to the diversity of types of nonlinear systems, our focus in this section will be limited to one approach that has been widely applied to analyze physiological systems that have stable nonlinear dynamic input–output characteristics – models that employ functional expansions or Volterra–Wiener kernels. There have been many advances in this modeling approach over the past few decades, and confining our discussion to only one section of one chapter clearly does not do justice to this area of work. Our goal here is only to provide the reader with a brief overview of the key concepts and how these are applied to a
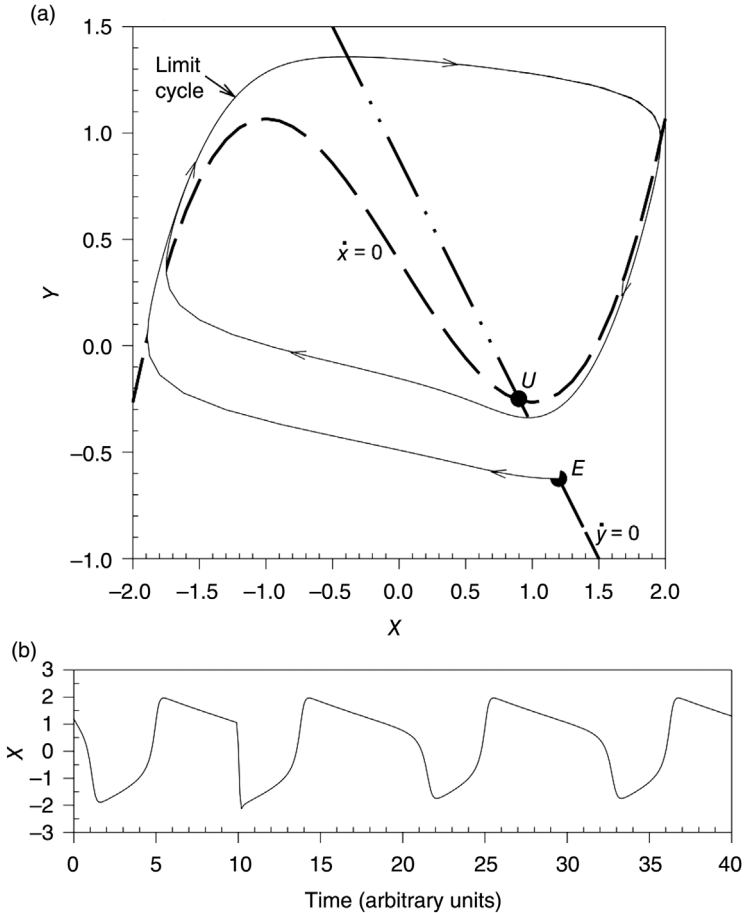
FIGURE 10.28 Conditions that lead to a train of "action potentials" from the Bonhoeffer–van der Pol model. (a) Simulating a "step depolarization" by setting $z = -0.4$ shifts the x-nullcline upward and produces a new unstable singular point (U). The net result is the creation of a stable limit cycle that encloses the unstable focus. (b) Time-course of x during application of depolarizing step.

physiological system. The interested reader is referred to the many excellent volumes and review articles have been published that provide more comprehensive and expert coverage of the material. These include, among others, Marmarelis and Marmarelis (1978), Schetzen (1989), Westwick and Kearney (2003), and Marmarelis (2004).

10.6.1   Volterra–Wiener Kernel Approach

The Volterra series may be considered an extension of the Taylor series expansion to functionals that can collectively represent a nonlinear dynamic system. The

Volterra representation of the response $y(t)$ of a continuous-time nonlinear system to an input time series $u(t)$ is given by

$$y(t) = \int_{-\infty}^{\infty} h_1(\tau_1)u(t - \tau_1)d\tau_1 + \int_{-\infty}^{\infty}\int_{-\infty}^{\infty} h_2(\tau_1, \tau_2)u(t - \tau_1)u(t - \tau_2)d\tau_1 d\tau_2$$

$$+ \int_{-\infty}^{\infty} \cdots \int_{-\infty}^{\infty} h_M(\tau_1, \ldots, \tau_M)u(t - \tau_1)\cdots u(t - \tau_M)d\tau_1 \cdots d\tau_M + \cdots + e(t)$$

$$(10.78)$$

The above equation assumes that the means of both the input and output signals have been removed and that $x(t)$ and $y(t)$ represent changes from those means. In Equation 10.78, the function $h_M(\tau_1, \ldots \tau_M)$ is known as the Mth order Volterra kernel. When the highest order of the Volterra series equals 1, Equation 10.78 defaults to the canonical characterization of a linear system. Under such circumstances, the first-order Volterra kernel is simply the impulse response of the system. Thus, one might consider $h_M(\tau_1, \ldots \tau_M)$ a "higher order impulse response" of the nonlinear system. For practical applications, the discrete-time representation of the Volterra or Wiener series with a finite number of terms is more frequently employed:

$$y(n) = \sum_{k_1=0}^{p-1} h_1(k_1)u(n - k_1) + \sum_{k_1=0}^{p-1}\sum_{k_2=0}^{p-1} h_2(k_1, k_2)u(n - k_1)u(n - k_2)$$

$$+ \sum_{k_1=0}^{p-1} \cdots \sum_{k_M=0}^{p-1} h_M(k_1, \ldots, k_M)u(n - k_1)\cdots u(n - k_M) + \cdots + e(n)$$

$$(10.79)$$

To provide further insight into what the higher order kernels of a nonlinear dynamic system represent, consider the case of a system that is fully characterized by a Volterra series where the highest order is 2. This would be a special case of Equation 10.79 where $M = 2$. Suppose further that the first-order kernel $h_1(k_1)$ takes the form of an exponential function and the second-order kernel $h_2(k_1, k_2)$ takes the form of the "2D" exponential function displayed in Figure 10.29. Then, the response of this nonlinear system to two unit impulses δ_1 and δ_2 occurring at the times indicated by the heavy arrows is predicted through Equation 10.79 to be the thin black tracing displayed in Figure 10.30. The observation noise $e(n)$ is assumed to be zero in this example. The contribution of only the first-order kernel to the overall response is represented by the thick gray tracing. Thus, the difference between the thin and thick curves represents the contribution of $h_2(k_1, k_2)$ alone. This difference therefore is the
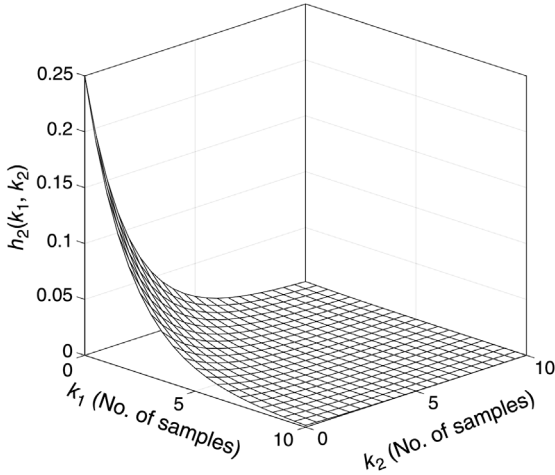
FIGURE 10.29 Example of a second-order discrete-time Volterra kernel $h_2 (k_1, k_2)$. Each point on the kernel represents the nonlinear (second-order) response to a unit impulse occurring k_1 samples earlier and another unit impulse occurring k_2 samples earlier.
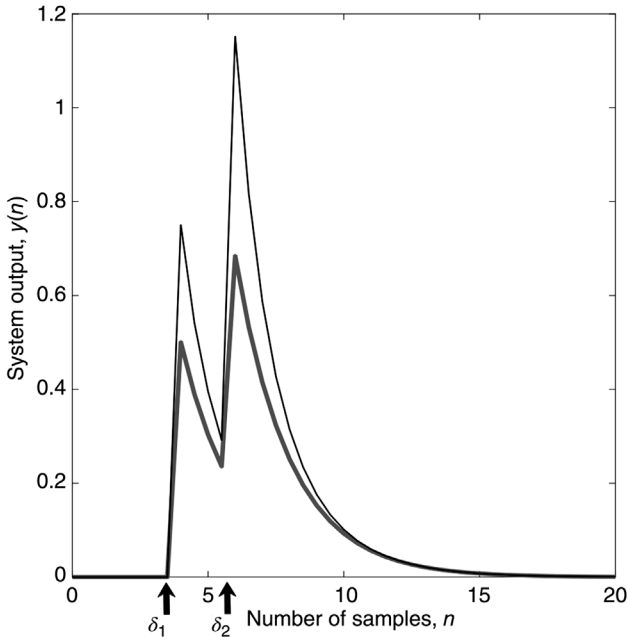
FIGURE 10.30 Response of a nonlinear system with an exponentially decaying first-order kernel and second-order kernel $h_2 (k_1, k_2)$ (as shown in Figure 10.29) to unit impulses δ_1 and δ_2 (displayed as black arrows). The thin black tracing represents the total system output, while the thick gray tracing represents the contribution from the linear component alone.

result of the nonlinear properties of this particular dynamic system. Notice, however, that in the duration between δ_1 and δ_2 (i.e., following the first impulse but before the second impulse), the difference between the two curves is attributed to the nonlinear portion of the response to δ_1 only. But following the occurrence of δ_2, the difference between the curves reflects the continuing nonlinear contribution in the response to δ_1, the (new) nonlinear contribution in the response to δ_2, and also the contribution from the interaction between the responses to δ_1 and δ_2. Any point on the $h_2(k_1,k_2)$ surface represents the response of the system to a unit impulse that occurred k_1 samples ago *and* a unit impulse that occurred k_2 samples ago, after one has accounted for the contributions from the linear responses of these two unit impulses.

In general, the highest order of the Volterra series should be infinite in order for the representation to be complete. This kind of representation immediately points to two key difficulties associated with the practical utility of Volterra series. The first is the issue of convergence of the series, similar to what one faces with the Taylor series – the contribution of the Mth order functional, in general, should diminish as M increases toward infinity. The second problem is that estimation of the Volterra kernels is complicated by the fact that the basis functionals of the series are correlated with one another. Wiener (1958) introduced his approach to circumvent these limitations by orthogonalizing the Volterra series in the case where the input is Gaussian white noise of given power level. This orthogonalization procedure allows each of the new (Wiener) kernels to be estimated one at a time. Lee and Schetzen (1965) introduced a practicable method for estimating the Volterra kernels from the above Volterra series by using cross-correlation to first estimate the Wiener kernels and subsequently converting these results into Volterra kernels. However, the cross-correlation method requires long data records, strict whiteness of the input, and a heavy computational burden associated with the estimation of higher order kernels. These strict input requirements are relaxed with the "exact orthogonalization technique" based on least-squares fitting (Korenberg, 1988).

Another method that alleviates most of the deficiencies inherent in the cross-correlation technique is the Laguerre expansion of kernels technique (LEK), introduced by Marmarelis (1993). Here, the unknown Laguerre expansion coefficients are estimated using least-squares, and Gram–Schmidt orthogonalization with respect to Gaussian white noise is not utilized, thus relaxing the requirement for strict whiteness of the input. This leads to increased estimation accuracy in the presence of noise, reduces the requirement for long data records, and allows relatively accurate kernel estimation when the input deviates from the theoretical requirement for white-noise stimuli. The basis expansion method discussed in Section 8.6.2 for linear systems is essentially a special case of the LEK. In LEK, the Volterra kernels are expanded into weighted sums of Laguerre basis functions. The expansions for the first- and second-order kernels are as

shown below:

$$h_1(k_1) = \sum_{i_1=0}^{q} c_1(i_1)L_{i_1}(k_1), \quad 0 \le k_1 \le p - 1 \qquad (10.80)$$

$$h_2(k_1, k_2) = \sum_{i_1=0}^{q} \sum_{i_2=0}^{q} c_2(i_1, i_2)L_{i_1}(k_1)L_{i_2}(k_2), \quad 0 \le k_1, \quad k_2 \le p - 1 \qquad (10.81)$$

where $L_i(k)$ are the Laguerre basis functions, as defined in Equation 8.84. The above expressions are substituted in Equation 10.79 to yield

$$y(n) = \sum_{i_1=0}^{q} c_1(i_1)v_{i_1}(n) + \sum_{i_1=0}^{q} \sum_{i_2=0}^{q} c_2(i_1, i_2)v_{i_1}(n)v_{i_2}(n) + \cdots + e(n) \qquad (10.82)$$

where

$$v_{i_1}(n) = \sum_{k=0}^{p-1} L_{i_1}(k)u(n - k) \qquad (10.83)$$

and

$$v_{i_2}(n) = \sum_{k=0}^{p-1} L_{i_2}(k)u(n - k) \qquad (10.84)$$

Since $v_{i1}(n)$ and $v_{i2}(n)$ are known time series (essentially Laguerre-filtered versions of the input), Equation 10.82 takes the form of a multiple linear regression problem whose unknown coefficients $c_1(i_1)$ and $c_2(i_1,i_2)$ can be solved using least-squares minimization, analogous to what was discussed in Section 8.6.2. Of course, as the number of basis functions (q) and the order of the model (M) increase, statistical reliability of this estimation procedure decreases. The MATLAB function `stationaryVolterra2_1in1out.m` provided along with the other programs in this book implements the LEK for a single-input single-output second-order Volterra model.

10.6.2 Nonlinear Model of Baroreflex and Respiratory Modulated Heart Rate

The dual-input model of heart rate variability discussed in Section 8.6.3 has been extended to incorporate nonlinearity up to the second order in the baroreflex and respiratory–cardiac coupling mechanisms (Jo et al., 2007). In addition, the interaction between the baroreflex and respiratory–cardiac coupling mechanisms is also included in the extended model. The mathematical representation of the

model is as follows:

$$
\begin{aligned}
\Delta \mathrm{RRI}(n) = {} & \sum_{i=0}^{p-1} h_{\mathrm{RCC}}(i) V(n - N_{\mathrm{RCC}} - i) + \sum_{i=0}^{p-1} h_{\mathrm{ABR}}(i) \Delta \mathrm{SBP}(n - N_{\mathrm{ABR}} - i) \\
& + \sum_{i=0}^{p-1} \sum_{j=0}^{p-1} h_{\mathrm{RCC2}}(i,j) V(n - N_{\mathrm{RCC}} - i) V(n - N_{\mathrm{RCC}} - j) \\
& + \sum_{i=0}^{p-1} \sum_{j=0}^{p-1} h_{\mathrm{ABR2}}(i,j) \Delta \mathrm{SBP}(n - N_{\mathrm{ABR}} - i) \Delta \mathrm{SBP}(n - N_{\mathrm{ABR}} - j) \\
& + \sum_{i=0}^{p-1} \sum_{j=0}^{p-1} h_{\mathrm{ABR-RCC}}(i,j) V(n - N_{\mathrm{RCC}} - i) \Delta \mathrm{SBP}(n - N_{\mathrm{ABR}} - j) + e(n)
\end{aligned}
$$

$$(10.85)$$

The above model is displayed schematically in Figure 10.31. Thus, apart from the contributions from the linear impulse responses h_{RCC} and h_{ABR}, there are contributions from the nonlinear portions of the RCC and ABR mechanisms (i.e., h_{RCC2} and h_{ABR2}), as well as a contribution from the interaction between RCC and ABR ("cross-kernel" $h_{\mathrm{ABR-RCC}}$). Note that the second-order self-kernels (h_{RCC2} and h_{ABR2}) are symmetric functions of their arguments, whereas the second-order cross-kernel ($h_{\mathrm{ABR-RCC}}$) is asymmetric with respect to its arguments. Other parameters of this model have been defined previously in Section 8.6.3. $e(n)$ represents the variability of ΔRRI not explained by the model.

To estimate the kernels of this model from the measured responses in ΔRRI, the LEK technique is applied to expand each kernel into an orthonormal set of
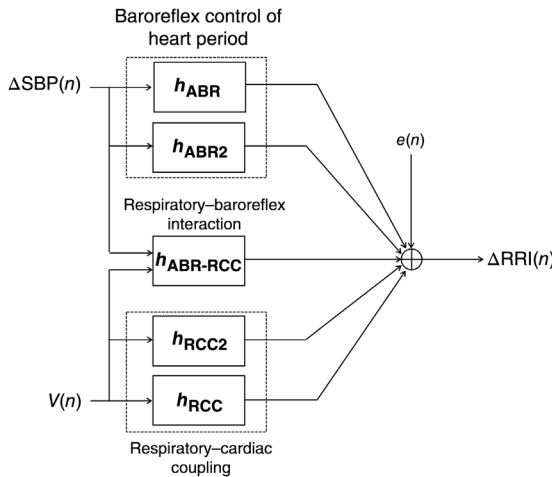
FIGURE 10.31 Schematic representation of the nonlinear dual-input model of heart rate variability (Jo et al., 2007).

discrete-time Laguerre functions, as displayed below:

$$h_{\text{RCC}}(n) = \sum_{i_1=0}^{q_{\text{RCC}}} c_{\text{RCC}}(i_1) L_{i_1}(n), \quad 0 \leq n \leq p - 1 \tag{10.86}$$

$$h_{\text{ABR}}(n) = \sum_{i_1=0}^{q_{\text{ABR}}} c_{\text{ABR}}(i_1) L_{i_1}(n), \quad 0 \leq n \leq p - 1 \tag{10.87}$$

$$h_{\text{RCC2}}(n_1, n_2) = \sum_{i_1=0}^{q_{\text{RCC}}} \sum_{i_2=0}^{q_{\text{RCC}}} c_{\text{RCC2}}(i_1, i_2) L_{i_1}(n_1) L_{i_2}(n_2), \quad 0 \leq n_1, n_2 \leq p - 1 \tag{10.88}$$

$$h_{\text{ABR2}}(n_1, n_2) = \sum_{i_1=0}^{q_{\text{ABR}}} \sum_{i_2=0}^{q_{\text{ABR}}} c_{\text{ABR2}}(i_1, i_2) L_{i_1}(n_1) L_{i_2}(n_2), \quad 0 \leq n_1, \quad n_2 \leq p - 1 \tag{10.89}$$

$$h_{\text{ABR–RCC}}(n_1, n_2) = \sum_{i_1=0}^{q_{\text{RCC}}} \sum_{i_2=0}^{q_{\text{ABR}}} c_{\text{ABR–RCC}}(i_1, i_2) L_{i_1}(n_1) L_{i_2}(n_2), \quad 0 \leq n_1, \quad n_2 \leq p - 1 \tag{10.90}$$

In Equations 10.86–10.90, $\{c_{\text{RCC}}(\cdot), c_{\text{ABR}}(\cdot), c_{\text{RCC2}}(\cdot), c_{\text{ABR2}}(\cdot), c_{\text{ABR-RCC}}(\cdot)\}$ are the sets of the unknown expansion coefficients, which are to be estimated from the input–output data. $L_i(n)$ represents the ith order Laguerre basis function, the form of which was given in Equation 8.84. q_{ABR} and q_{RCC} are the number of Laguerre functions used in the expansion of the kernels for the ABR and RCC mechanisms, and these are allowed to be different across individual data sets.

Substitution of Equations 10.86–10.90 into Equation 10.85 enables the conversion of Equation 10.85 into the form:

$$\Delta\text{RRI}(n) = \sum_{i_1=0}^{q_{\text{RCC}}} c_{\text{RCC}}(i_1) v_{\text{RCC}}(i_1, n) + \sum_{i_1=0}^{q_{\text{ABR}}} c_{\text{ABR}}(i_1) v_{\text{ABR}}(i_1, n)$$

$$+ \sum_{i_1=0}^{q_{\text{RCC}}} \sum_{i_2=0}^{q_{\text{RCC}}} c_{\text{RCC2}}(i_1, i_2) v_{\text{RCC}}(i_1, n) v_{\text{RCC}}(i_2, n)$$

$$+ \sum_{i_1=0}^{q_{\text{ABR}}} \sum_{i_2=0}^{q_{\text{ABR}}} c_{\text{ABR2}}(i_1, i_2) v_{\text{ABR}}(i_1, n) v_{\text{ABR}}(i_2, n) \tag{10.91}$$

$$+ \sum_{i_1=0}^{q_{\text{RCC}}} \sum_{i_2=0}^{q_{\text{ABR}}} c_{\text{ABR–RCC}}(i_1, i_2) v_{\text{RCC}}(i_1, n) v_{\text{ABR}}(i_2, n) + e(n)$$

where

$$v_{\text{RCC}}(i, n) = \sum_{k=0}^{p-1} L_i(k) V(n - k - N_{\text{RCC}}), \quad 0 \le i \le q_{\text{RCC}} \tag{10.92}$$

and

$$v_{\text{ABR}}(i, n) = \sum_{k=0}^{p-1} L_i(k) \Delta\text{SBP}(n - k - N_{\text{ABR}}), \quad 0 \le i \le q_{\text{ABR}} \tag{10.93}$$

Equation 10.91 turns out to be simply an extension of Equation 10.82 with more terms. As in Section 10.6.1, we can treat this as a multiple linear regression problem and solve for the unknown Laguerre coefficients using least-squares minimization. Since there are many combinations of q_{RCC}, q_{ABR}, N_{RCC}, N_{ABR}, and α, the least-squares minimization would have to be carried out for each of these "model candidates." The "optimal" solution is arrived at by selecting the model candidate with the lowest MDL, as discussed in Section 8.5.3.

Jo et al. (2007) applied the nonlinear model to data obtained from normal controls and subjects with obstructive sleep apnea (OSA) during ventilator-assisted, randomized breathing in wakefulness and sleep. Adding the second-order and interaction kernels to the linear model produced a significant improvement in the goodness of fit between the model-predicted ΔRRI and the corresponding data. When only linear terms were included, the model accounted for less than 50% of the total variance in the data. After including the nonlinear terms, the proportion of total variance explained by the model increased to approximately 70%. These findings suggest that nonlinearities account for a significant portion of heart rate variability. The estimated second-order kernels were found to be different in the OSA subjects relative to normal controls. Figure 10.32 shows sample plots of the second-order ABR kernel, h_{ABR2}, in one of the controls (panel (a)) and one of the OSA subjects (panel (b)). It is clear that the overall magnitude of h_{ABR2} was lower in the OSA subject compared to the normal control. Indeed, they found statistically significant differences between the two subject groups. The interaction kernel $h_{\text{ABR-RCC}}$ was also found to be significantly reduced in the OSA group.

10.6.3 Interpretations of Kernels

One of the drawbacks of using Volterra kernels to characterize nonlinear dynamics in physiological systems is the difficulty in interpreting the resulting kernel model in terms of the structure and function of the underlying physiological system. However, it should be possible in many situations to extract "meaning" from the estimated kernels by using them to simulate the conditions under which the system has been studied previously. Then, one can compare the simulated responses
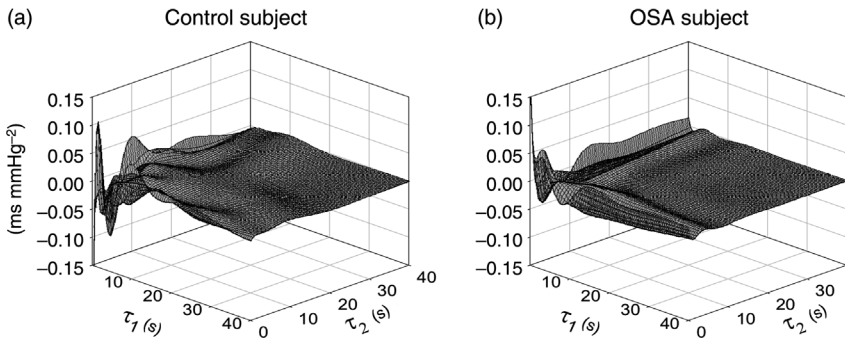
FIGURE 10.32 Sample plots of the second-order ABR kernel, h_{ABR2}, estimated in a normal subject (a) and a subject with obstructive sleep apnea (OSA) (b) in the study of Jo et al. (2007). The study found that the OSA group had significantly smaller magnitudes of h_{ABR2} than controls. (Modified with permission from Figure 6 of Jo et al. (2007).)

of the model with the observed responses to the same kinds of stimuli that were employed in the previous studies.

Consider, for instance, the Volterra kernels estimated for the nonlinear heart rate variability model of Jo et al. (2007). It has been previously observed by others that the frequency response relating respiration to RRI is dependent on tidal volume. The second-order RCC kernel h_{RCC2} was tested to determine whether its presence introduces tidal volume dependence into the frequency response of RRI with respect to respiration. The model was convolved with a respiration input taking on the form of a chirp signal with frequency increasing from 1 to 45 cycles per minute. The simulations were conducted with tidal volumes ranging from 0.5 to 3 L, and the frequency response of the system was determined for each tidal volume condition. The simulations confirmed that increasing tidal volume does indeed increase the RCC gain at any given frequency when h_{RCC2} was included in the model, but volume dependence of the frequency response did not occur when the model did not contain h_{RCC2}.

It is also well established that the steady-state ABR response is not linear, showing saturation at higher blood pressure levels. In other order to test the hypothesis that h_{ABR2} contributes to this phenomenon, Jo et al. convolved the model with step functions in ΔSBP of different magnitudes ranging from 1 to 5 mmHg. They found that, with h_{ABR2} present in the model, the saturation effect was indeed observed.

Finally, it has been shown that neck suction when applied during inspiration produces smaller increases in RRI than when applied during expiration, indicating that ABR gain is modulated by respiratory phase. The model was stimulated with brief pulses of ΔSBP that were triggered at various phases of the inspiratory/expiratory cycle. The simulations demonstrated that ΔSBP pulses delivered during expiration exert a greater effect on ΔRRI than when these pulses were delivered during inspiration, consistent with reported

experimental data on humans. This effect was only present when the cross-kernel $h_{ABR\text{-}RCC}$ was included in the model.

10.6.4 Higher Order Nonlinearities and Block-Structured Models

One can appreciate from the examples in Sections 8.6.3 and 10.6.2 that the number of unknown parameters (i.e., all the Laguerre coefficients and time delays) increases rapidly as the order of the model increases, even with the incorporation of the LEK technique. Studies that have applied the Volterra–Wiener approach to physiological systems have generally employed kernels only as high as the third order, due to problems that arise from computational burden and interpretability of the results. Block cascade models offer one avenue for accommodating stronger nonlinearities in compact form (Billings and Fakhouri, 1982; Hunter and Korenberg, 1986). The "Wiener cascade" or L–N system is one of the most commonly used block cascade models; it consists of a linear dynamic system whose output feeds into a static nonlinearity. Its counterpart is the "Hammerstein cascade" or N–L system, in which the static nonlinearity precedes the linear dynamic system. A more general class is the "sandwich" or "L–N–L" model in which linear dynamic systems precede and succeed a static nonlinearity. Artificial neural networks may be considered to be a greater generalization of such models. However, one of the main drawbacks with these block-structured models is that the parameters generally have to be estimated using iterative optimization techniques.

The "hybrid" approach advocated by Marmarelis (2004) follows along the lines of the Wiener system by decomposing the nonlinear dynamics into two operational layers. Such a system is illustrated schematically in Figure 10.33 for a single-input single-output discrete-time system. The input $u(n)$ feeds into a linear filter bank whose impulse response functions form a complete and orthogonal basis (B_1 through B_M) that spans the system dynamics. The second layer consists of a multi-input static nonlinearity that is the recipient of the outputs from the first layer and transforms them into the output of the whole system. In the case of the LEK
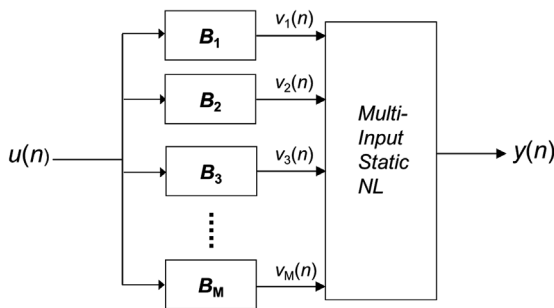
FIGURE 10.33 The nonlinear dynamic system as the cascade of a filter bank of orthogonal linear dynamic filters with a multi-input static nonlinearity, as proposed by Marmarelis (2004).

technique, the Laguerre functions of successively increasing order constitute the filters B_1 through B_M in Figure 10.33. The output of each filter is given by an equation similar to Equation 10.83 or 10.84. The output from one filter is subsequently multiplied by itself and the outputs from the other filters once or more times, depending on the order of the system. And the resulting products are weighted by the coefficients associated with the Laguerre filters involved and finally summed to produce the system output. Thus, the Volterra models described in Sections 10.6.1 and 10.6.2 are covered in this general signal processing scheme. The beauty of this conceptual scheme is that it allows for a means of capturing the "essential" dynamic characteristics of a given nonlinear system by transforming the system into one with a smaller number of "necessary" filters. This simplifying transformation can be achieved to derive the "principal dynamic modes" (PDM) of the system through the eigen-decomposition of the matrix constructed from the Laguerre expansion coefficients. This approach does not totally solve the problem of "interpretability" associated with Volterra kernels, but it does offer a useful means of capturing the "essence" of the observed dynamics, thus casting the findings in a more understandable form by nonmathematical researchers. A clear and detailed exposition of the PDM approach can be found in Marmarelis (2004).

PROBLEMS

P10.1. Consider the dynamical system defined by the following pair of coupled first-order differential equations:

$$\frac{dx}{dt} = x - \frac{x^3}{3} + y - 15$$

$$\frac{dy}{dt} = \frac{1 - 0.6x - 0.48y}{5.4}$$

Sketch the phase-plane diagram for this system, showing the x- and y-nullclines, as well as the general directions of flow for the phase trajectories. Find out where the singular point in this system is located and determine its stability characteristics.

P10.2. A genetically engineered form of white blood cell has been developed for therapeutic use in fighting the bird flu virus. When introduced into the body, these white blood cells feed on other cells that have been invaded by the virus. A simple model that characterizes the growth of these two cell populations is given below:

$$\frac{dx}{dt} = x - cxy$$

$$\frac{dy}{dt} = cxy - y$$

Here, x represents the fraction of lung cells that have been infected by the virus, while y represents the fraction of the genetically engineered white blood cells. c is a (positive) constant parameter. Note that, without therapeutic intervention ($y = 0$), x would grow exponentially. In contrast, the therapeutic cells are self-limiting but the growth rate (dy/dt) of their population is enhanced by the number of virally infected lung cells.

(a) For the case in which $c = 2$, determine the values of the cell populations at which there will be no net growth in the population of both cell types.

(b) Determine whether the equilibrium points determined in (a) are stable or unstable. In addition, deduce the nature of stability at these points (i.e., are they stable/unstable foci or nodes, or saddle points, etc.).

(c) Sketch the phase-plane diagram for this system (when $c = 2$), showing clearly the x- and y-nullclines, as well as the general directions (using arrows) of flow for the phase trajectories.

(d) When the populations of both the flu-infected cells and the white blood cells are very small (i.e., both x and y are close to zero), the dynamics become approximately linear. Under such circumstances, how would you expect x and y to change with time? Provide sketches of the time-courses of x and y, as well as a brief explanation for your answer. Assume $c = 2$ here as well.

P10.3. Develop the SIMULINK model of the system given in Problem P10.1. Determine the phase portrait of this dynamical system by computing the phase trajectories from several different starting locations on the phase plane.

P10.4. Use the describing function method to determine whether a limit cycle exists for the respiratory control model discussed in Section 10.4.2 if the controller response is characterized by the following equations:

$$\dot{V}_E = 0.02\left(P_{aCO_2} - 37\right), \quad P_{aCO_2} > 39$$
$$= 0, \quad P_{aCO_2} \le 39$$

where $\dot{V}_E$ is given in units of $L\,s^{-1}$. This controller response function differs from that given in Equation 10.50 in that there is an abrupt silencing of chemoreceptor output when P_{aCO2} is decreased from values slightly greater than $39\,mmHg$ to values below $39\,mmHg$. Assume in your computations the values for the other parameters as given in Section 10.4.2. If the limit cycle exists, determine the periodicity and amplitude of the oscillation.

P10.5. In the example of describing function analysis given in Section 10.4.2, the predicted oscillation period for Cheyne–Stokes breathing was on the order

of 40 s. This is substantially shorter than the ~60 s cycle time that is more frequently observed. Determine whether the inclusion of additional factors, such as circulatory mixing and chemoreceptor response time, can account for the much of the difference. To do this, extend the overall transfer function of the linear component so that it takes on the following frequency response:

$$G(j\omega) = \frac{G_{lung}}{\left(1 + j\omega\tau_{lung}\right)\left(1 + j\omega\tau_{circ}\right)\left(1 + j\omega\tau_{chemo}\right)} e^{-j\omega T_d}$$

Assume that $\tau_{circ} = 2$ s and $\tau_{chemo} = 15$ s. For other parameters, use the values employed in Section 10.4.2.

P10.6. In the Poincaré oscillator model, described in Section 10.3.3, a neural spike is generated each time the traveling "particle" passes through the $\phi = 0$ point on the unit circle (see Figure 10.13). It will normally not generate another neural spike until 1 revolution later when the "particle" passes through $\phi = 0$ again.

(a) Suppose at $\phi = 0.625$ (i.e., midway between $\phi = 0.5$ and $\phi = 0.75$), the oscillator is hit by an external stimulus (b in Figure 10.13) of magnitude 2. Determine graphically what would be the "new phase" θ of the oscillator immediately following the perturbation? Note that both ϕ and θ are each normalized to 1 revolution (i.e., $\phi = 1$ represents 360°) – so provide your answer for θ in normalized units.

(b) Suppose the natural period of this oscillator is 1 s. What would be the duration (in seconds) between the last natural spike generated by this oscillator and the occurrence of the spike that follows the perturbation by the stimulus? Does the spike that occurs following perturbation appear earlier or later than what would have been expected if there was no perturbation? Assume that the parameter a in Equation 10.35, describing the radial dynamics of the particle, is positive and extremely large (i.e., effectively infinite).

(c) Note that in Equation 10.35, dr/dt becomes zero when $r = 0$ (i.e., when the particle is at the origin) or when $r = 1$ (when the particle is located on the limit cycle). Is the origin ($r = 0$) a stable or unstable equilibrium point? Explain how you have arrived at your answer.

P10.7. By decreasing the normalized period (T_s/T_0) of the stimulation from 0.95 to 0.5 in the SIMULINK program `poincare.slx`, explore the changes in the phase relationship between the stimulus and response changes in the Poincaré oscillator during Type 1 resetting with the magnitude of the stimulus, b, set equal to 0.9.

P10.8. Modify the SIMULINK implementation (`vdpmod.slx`) of the van der Pol model so that it can be driven by an external periodic stimulus of
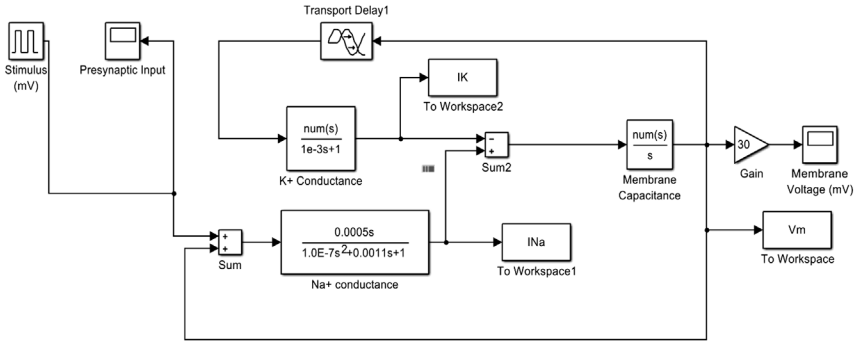
FIGURE P10.1   Linearized SIMULINK representation (`linhhmod.slx`) of the Hodgkin–Huxley model.

magnitude B and frequency f. Thus, modify this model so that its dynamics are described by Equations 10.33 and 10.34. First, set B equal to 2 and vary f in small increments from 0.01 to 0.2 Hz. Then, set B equal to 0.5 and vary f over the same range of frequencies. In each case, determine the band of frequencies over which entrainment occurs. Does the strength of the external forcing affect the entrainment band?

P10.9. Figure P10.1 shows the SIMULINK implementation (`linhhmod.slx`) of the linearized, closed-loop representation of the Hodgkin–Huxley model displayed in Figure 10.21b. Use this model to determine the membrane voltage response to a brief depolarization pulse of 60 mV. Does this model display the properties of thresholding and refractoriness? Modify the model to simulate voltage clamp experiments, in which the membrane voltage is constrained to follow a step change of +60 mV. Determine the time-courses of the sodium and potassium currents following the step depolarization.

P10.10. Figure P10.2 displays the control block diagram of a closed-loop system that contains a dynamic linear forward block and a static nonlinear feedback block. Note that since the input to the static nonlinear feedback block is x, the output of this block would be $h(x)$. Thus, if the input z to this closed-loop system is zero, the dynamics of $x(t)$ can be characterized by the following ordinary differential equation:

$$\frac{d^2x}{dt^2} + \frac{dx}{dt} + h(x) = 0$$

Consider the case where the function $h(x)$ is defined as

$$h(x) = x^2 - x - 2$$
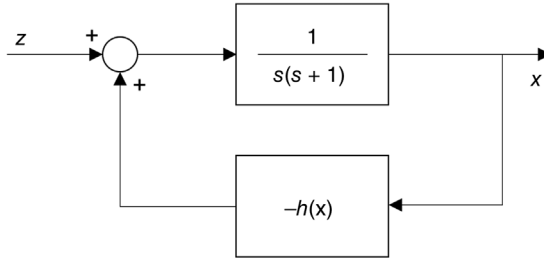
FIGURE P10.2 Closed-loop control system with nonlinear feedback element.

(a) By introducing the variable y ($=dx/dt$), convert the differential equation above into the form where you can perform a phase-plane (y versus x) analysis of the dynamics of this system.

(b) Determine the x- and y-nullclines and singular point(s) of this system.

(c) Determine the stability properties (i.e., stable or unstable; node or focus or center or saddle point) of the system's singular point(s).

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11

COMPLEX DYNAMICS IN PHYSIOLOGICAL CONTROL SYSTEMS

11.1 SPONTANEOUS VARIABILITY

The examples presented in Chapter 10 showed that the presence of nonlinearity can dramatically increase the range of dynamic behavior exhibited by relatively simple open- or closed-loop systems. Linear systems are, by definition, constrained to obey the principle of superposition, which makes prediction of their future behavior relatively easy, unless the output measurements are heavily contaminated by random noise. By contrast, nonlinear systems can produce responses that are difficult to predict, even in the absence of noise. However, unless one subjects these signals to careful analysis, it is often difficult to distinguish one kind of "complexity" from the other. This is very much the case with the naturally occurring fluctuations, or *spontaneous variability*, exhibited by physiological control systems of all kinds. An important question that continues to stimulate bioengineering research is whether these spontaneous physiological fluctuations represent the effect of random perturbations on the underlying system or whether they are the result of the complex dynamics generated by a deterministic, nonlinear system. It is likely that both random and nonlinear influences contribute in various degrees to these natural variations. The purpose of this chapter is to examine in greater detail some of the possible mechanisms through which complex dynamics can arise.

The spontaneous variability exhibited by physiological systems ranges from highly periodic and regular waveforms to highly complex temporal structures. To
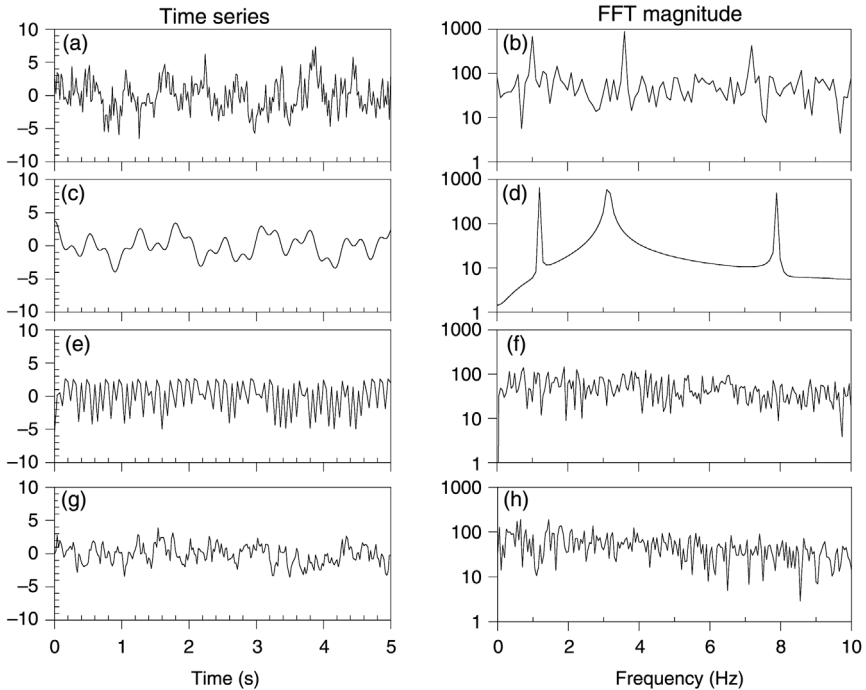
FIGURE 11.1   Time series plots and corresponding Fourier spectra of four types of system responses.

illustrate the breadth of complex physiological patterns, we examine four very different time series displayed in Figure 11.1 (panels a, c, e, and g). From visual inspection, all four contain different degrees of regularity and irregularity. To aid in distinguishing one signal from the other, it is useful to also examine the frequency content of these signals. Panels b, d, f, and h display the Fourier transform magnitudes of the four signals; these were computed using the `fft` command in MATLAB and subsequently applying the `abs` command to each result:

(1) Although it is difficult to pick this out from the display in the time domain, the signal displayed in Figure 11.1a actually contains three sinusoidal components with center frequencies of 1.2, 3.6, and 7.2 Hz, as the Fourier spectrum reveals in Figure 11.1b. However, the regularity of this signal is obscured by the substantial amount of random noise that accompanies it. On the other hand, linear filtering techniques can easily strip away the noise and reveal the underlying periodic nature of this signal.

(2) The second signal displayed in Figure 11.1c does not exhibit the "noisiness" of the first signal and looks "almost" periodic, but is not. This is a *quasiperiodic* signal with no noise added. As its Fourier spectrum

(Figure 11.1d) shows, this time series, like the first in Figure 11.1a, is actually composed of three sinusoidal components. However, the three frequencies are *incommensurate* with one another, that is, the ratio between any two of the frequencies is not a rational number. Thus, the signal exhibits *recurrence* (i.e., segments of the signal regularly "resemble" each other but are not completely reproducible) but not strict periodicity.

(3) The third time series displayed in Figure 11.1e appears quite "noisy," but the "noisiness" is different from that in Figure 11.1a. In fact, this signal was generated by a completely deterministic system and contains no random noise. At the same time, however, it is also quite different from the quasiperiodic signal in Figure 11.1c. The corresponding Fourier spectrum in Figure 11.1f shows no dominant periodicities, unlike the previous two signals. Instead, the power of the signal is spread over a broadband of frequencies. This is an example of a *chaotic* signal.

(4) The final time series displayed in Figure 11.1g looks like a hybrid between the first and second signals. However, its Fourier spectrum is broadband and more similar to that of the chaotic signal (Figure 11.1f) than to the spectra of the other two time series. In fact, this signal was generated by low-pass filtering white noise. This kind of signal is commonly referred to a *colored* or *correlated* noise.

The above discussion was aimed at highlighting the fact that it is not always easy to determine the nature of the system that produces spontaneous variability. Oscillatory systems that generate periodic signals are the easiest to recognize. However, measurement or system noise can "bury" the weaker oscillatory components (Figure 11.1a). Oscillations with incommensurate frequencies can lead to quasiperiodic outputs that look "almost" but not exactly periodic (Figure 11.1b). However, such signals are easily distinguishable from others by the presence of significant peaks in their Fourier spectra. Although the chaotic signal is very different from the periodic or quasiperiodic signal, a common feature is that the values taken by any of these signals are always bounded. In phase-space representation, one would observe that all the corresponding points fall only in certain defined regions, leaving other areas void. The multidimensional object formed by these filled regions is called the *attractor*. By contrast, the random signal would have a representation that fills the entire phase space.

In the rest of this chapter, we will highlight a number of control mechanisms that can give rise to the spontaneous variability that is so much a trademark of physiological signals. Nonlinear component properties and delays in the transmission of feedback information provide a potent source of periodic, quasiperiodic, and chaotic behavior. There are several oscillatory components in the body, and the interactions among these oscillators as well as the effect of external rhythms, such as the environmental light, can produce complex and unexpected consequences. Moreover, most physiological parameters are time-varying. Thus, one would expect these "nonstationarities" to exert an important influence on the physiological

system in question. Finally, most people would agree on the existence of truly random processes at various hierarchical levels in physiology. The propagation of noise generated by these processes through the feedback loops of a given system can also lead to highly complex behavior.

11.2   NONLINEAR CONTROL SYSTEMS WITH DELAYED FEEDBACK

11.2.1   The Logistic Equation

The logistic equation, proposed by May (1976) as a model of population growth, is one of the best examples of how a simple nonlinear process can lead to highly complicated dynamics. In May's population model, x_n represents the (normalized) population of the current generation of a given species. x_n depends on two opposing influences. The first is a factor tending to increase the species population – This, of course, depends on the birth date and the population of the previous generation of this species. The second factor represents the limiting influence to growth resulting from a finite supply of food or energy. This second factor is what imparts nonlinearity to the model. Thus, the model can be written as

$$x_n = \alpha x_{n-1} - \alpha x_{n-1}^2 = \alpha(1 - x_{n-1})x_{n-1} \qquad (11.1)$$

This process can also be cast in the form of a nonlinear feedback control system, as illustrated in Figure 11.2. Note that, in this case, the parameter α can be thought of as a "gain factor" in the feedback control system.

The dynamic behavior of the logistic equation depends on the magnitude of α. When α is less than 3, x_n simply converges to a steady-state equilibrium value after a number of iterations. However, when α increases above 3 but is below 3.4495, this system exhibits a simple oscillation with a period of two time steps; this is known as a *period-2* oscillation (Figure 11.3a). When α increases further and falls into the range of 3.4495–3.5441, the behavior changes character and becomes a *period-4*
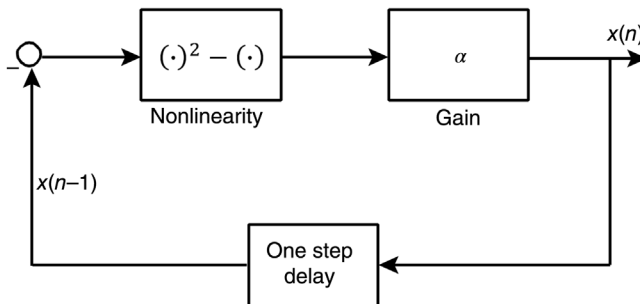
FIGURE 11.2   The logistic equation as a nonlinear feedback control system.
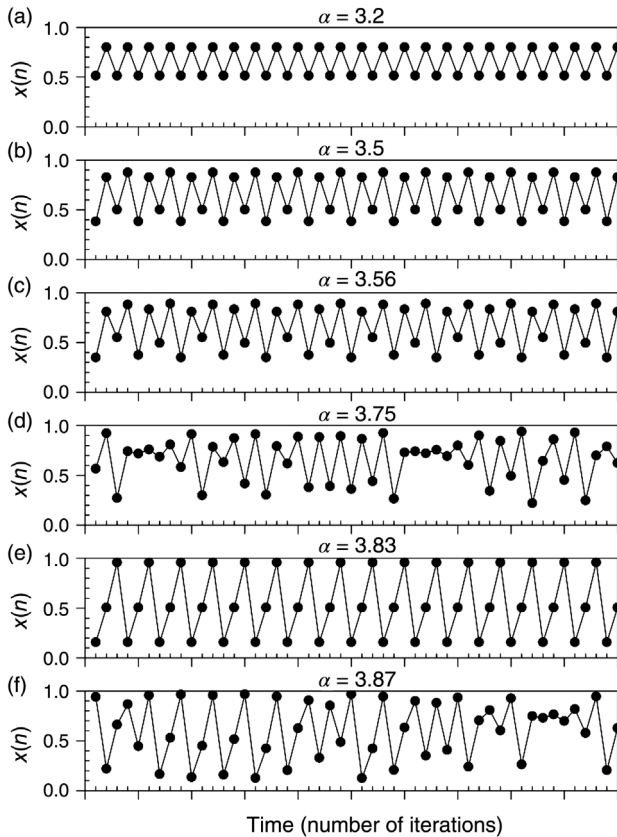
FIGURE 11.3 Dynamic behavior of the logistic equation as α is increased, showing the period-doubling route to chaos.

oscillation (Figure 11.3b). Subsequently, when α is increased to any value between 3.5441 and 3.5644, the cycling continues but now the oscillation is a *period-8* cycle (Figure 11.3c). With further increases of α, there are successive doublings of the periodicity. However, when α increases beyond 3.57, the periodic behavior disappears and chaotic dynamics emerges instead (Figure 11.3d). This kind of change in dynamics of the system is a classic illustration of what has been termed *the period-doubling route to chaos*. The reader should be cautioned that the logistic system does not always behave chaotically when α is between 3.57 and 4. In certain subranges of α, the system reverts to an orderly periodic behavior, such as the period-3 cycling shown in Figure 11.3e when $\alpha = 3.83$. Then, just as abruptly, in other regions, the behavior can become chaotic again (e.g., Figure 11.3f, when $\alpha = 3.87$).

As one may recall from Section 10.3.3, the points at which the dynamics of the system changes abruptly, as the "gain" α is increased, are known as *bifurcations*. A
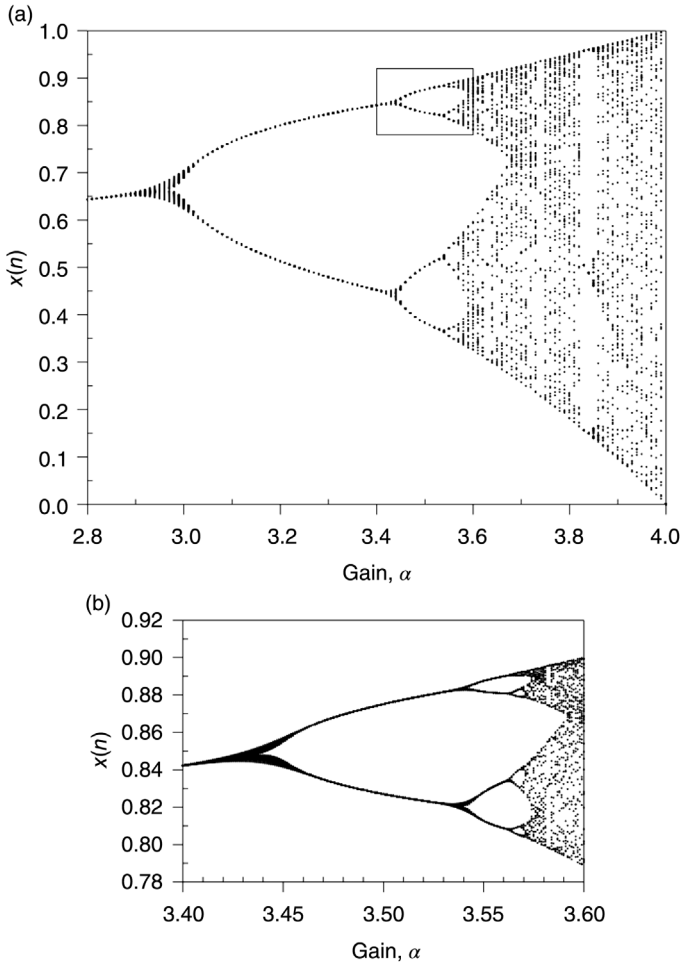
FIGURE 11.4 (a) Bifurcation diagram of the logistic equation. (b) Expanded view of the logistic map within the rectangular region shown in part (a).

useful way of displaying the dependence of these bifurcations on α is to plot the solutions for x_n (after allowing for the transients to fade away) as a function of α. The result is a *bifurcation diagram*. The bifurcation diagram for the logistic equation, frequently called the *logistic map*, is shown in Figure 11.4a for values of α that range from 2.8 to 4. Here, the bifurcations are particularly evident as the points at which a given locus of points "pitchforks" into two daughter branches. One feature of the logistic map that is quite apparent to any observer is the shortening of distances between successive bifurcations. For instance, the range of α-values over which period-2 oscillations occur is $3.4495 - 3 = 0.4495$, whereas the range in α over which period-4 oscillations occur is $3.5441 - 3.4495 = 0.0946$; subsequently, the range in α

over which period-8 oscillations occur becomes 0.0203. Note that the ratio of the α-range for period-2 oscillations to the α-range for period-4 oscillations is approximately 4.75, whereas the ratio of the α-range for period-4 oscillations to the corresponding range for period-8 oscillations is about 4.66. Feigenbaum (1980) showed theoretically that the ratio of the range in α for period-n oscillations to the corresponding range for period-$2n$ oscillations approaches a value of approximately 4.6692 when n approaches infinity. For obvious reasons, this limiting value is known as *Feigenbaum's number*. Another interesting feature of the logistic map is that it exhibits self-similarity. By this, we mean that if we were to "zoom" into a selected section of the bifurcation diagram (as illustrated by the small rectangle in Figure 11.4a), examination of the resulting view would yield a structure that is similar to the original diagram. This kind of self-similarity is characteristic of *fractal* structures. Further discussion of fractals falls beyond the scope of this volume, and the interested reader is encouraged to look up a number of excellent sources, such as Feder (1988), West (1990), and Bassingthwaighte et al. (1994).

The key property that distinguishes the chaotic waveform from other periodic and quasiperiodic signals is its *sensitivity to initial conditions*. This property is illustrated in Figure 11.5, which shows two possible outputs (bold versus thin tracing) generated by the logistic equation. Both were produced using exactly the same value of α. The only difference is that in the bold tracing, the initial value (i.e., x_0) was set equal to exactly 0.1, whereas for the thin tracing, the initial state was
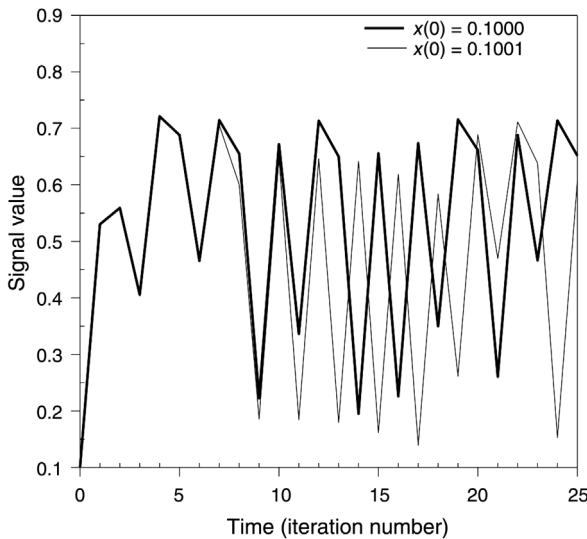
FIGURE 11.5 Illustration of "sensitivity to initial conditions" in a chaotic signal. Both signals (bold and thin tracings) were generated by the logistic model. In the bold tracing, initial value was set to 0.1; in the thin tracing, initial value was set equal to 0.1001. The two time-courses diverge considerably from one another after ~10 iterations.

assigned the value of 0.1001. For the first few iterations, both trajectories are extremely close. However, after the tenth iteration, the thin tracing begins to diverge further and further away from the bold tracing, so that eventually two totally distinct trajectories emerge. In the other types of deterministic signal, two state points that are initially close to each other will always remain so. The logistic system is an example of a process that is both fractal and, when $\alpha \geq 3.57$, sensitive to initial conditions. However, it is important to remember that not all fractal processes are chaotic and not all chaotic processes are fractal.

Sensitivity to initial conditions implies that it is impossible to predict the future values of a chaotic signal in the long run. Because we can only measure this signal with finite precision, the initially very small difference between our initial measurement and the true signal value at that point would, over time, grow exponentially. Referring again to Figure 11.5, if we take the thin tracing to be our prediction of the "true" signal (bold tracing), it is clear that the predicted values after the tenth iteration are totally off the mark. The rate at which two initially nearby trajectories become increasingly separated from each other as time progresses is quantified by the dominant *Lyapunov exponent*. The presence of a positive Lyapunov exponent indicates that the underlying system is chaotic. However, the computation of the Lyapunov exponent itself is difficult and the statistical reliability of the solution depends on the quality and quantity of the available data. The reader is again referred to other excellent references that exclusively cover the topic of chaos, such as Thompson and Stewart (1986), Moon (1987), and West (1990).

11.2.2   Regulation of Neutrophil Density

The white blood cell counts of patients with *chronic myeloid leukemia* (CML) are known to fluctuate wildly about elevated levels. These fluctuations are roughly periodic with cycle durations that range from 30 to 70 days. Mackey and Glass (1977) speculated that these oscillations may be related to changes in the dynamic properties of the physiological control system that regulates the balance between production and destruction of the neutrophils that circulate throughout the body. They proposed the following differential-delay equation to account for the dynamics of this regulatory process:

$$\frac{dx}{dt} = \frac{\beta \theta^n x(t - T_d)}{\theta^n + x(t - T_d)^n} - \gamma x(t) \tag{11.2}$$

where $x(t)$ represents the neutrophil density in blood at time t and T_d is the "maturation time," that is, the delay between the time the new neutrophils are produced by the stem cells in the marrow and the time the mature neutrophils are released into the circulation. The parameter γ, assumed constant, represents the rate at which the cells are destroyed due to a variety of factors. The parameters θ and n determine the relationship between the neutrophil production rate and the past neutrophil density, while the parameter β represents a scaling factor. Figure 11.6
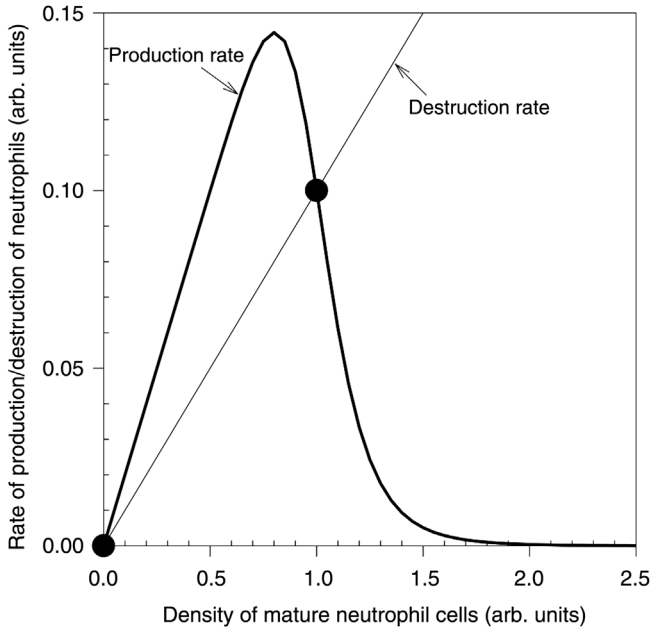
FIGURE 11.6 The steady-state relationships characterizing the production (thick curve) and destruction (thin line) of neutrophil cells as functions of the circulating neutrophil density in blood. There are two steady-state equilibrium points (filled circles).

shows this nonlinear function (thick curve) when θ, n, and β are assigned the values of 1, 10, and 0.2, respectively. Also shown is the linear function relating destruction rate to neutrophil density; here, a value of 0.1 is assumed for γ. Over a large range of neutrophil densities, poietin feedback control exerts its effects by reducing the production of new neutrophils when the circulating neutrophil density increases. However, as the neutrophil density decreases toward zero, it is assumed that the production rate also falls to zero. Under these conditions, the type of feedback therefore changes from negative to positive.

As shown in Figure 11.6, the intersection between the straight line representing destruction rate as a function of neutrophil density and the nonlinear curve representing production rate yields the steady-state (equilibrium) solutions for x. There is a trivial solution at the origin. The other solution (located at $x = 1$, $y = 0.1$) can be shown, using the method of phase-plane analysis (see Chapter 9), to be stable. To examine the dynamic behavior of this system in the presence of feedback time delay, we solve Equation 11.2 through computer simulation: The SIMULINK implementation (named `hematop.slx`) of the model is shown in Figure 11.7. Some simulation results are displayed in Figure 11.8. When T_d is small (=2 days), as would be expected under normal circumstances, the system is stable and, following any perturbation in neutrophil density (e.g., due to blood loss), there is a rapid return to the stable steady-state level (Figure 11.8a).
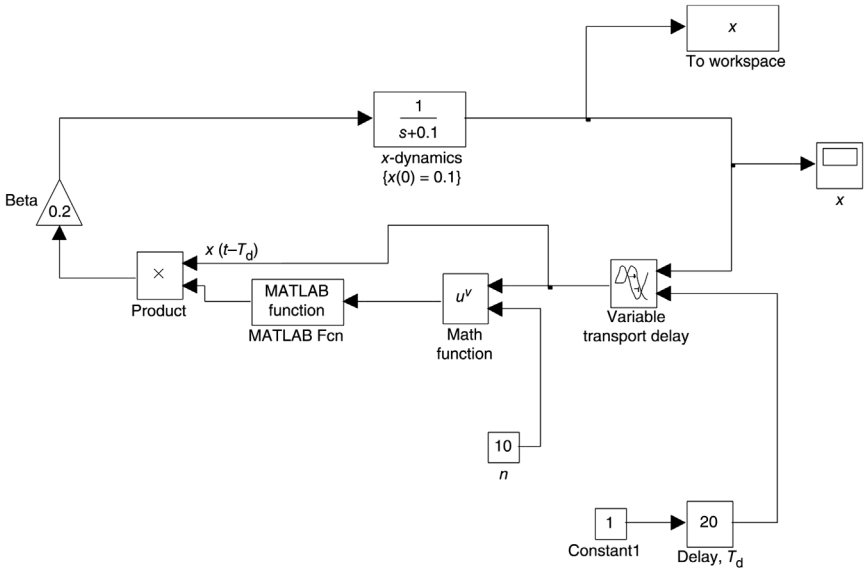
FIGURE 11.7   SIMULINK implementation of the neutrophil density regulation model of MacKey and Glass (1977).

However, if the maturation time is increased threefold to 6 days, following an initial perturbation, x does not return to its previous stable steady state. Instead, it oscillates with a small amplitude around the original equilibrium level (Figure 11.8b). The period of this oscillation is 18 days, which falls in the range of periodicities (17–28 days) observed in humans who have disorder known as *cyclical neutropenia*

If T_d is increased to 20 days to simulate the conditions of CML, large fluctuations in neutrophil density occur, as illustrated in Figure 11.8c. These fluctuations look somewhat periodic but they are actually quite irregular, displaying a significant degree of "noisiness." The irregular, yet almost periodic, waveform of Figure 11.8c resembles the pattern of white blood cell count that has been observed in some patients with CML. Note, however, that the differential-delay equation (Equation 11.2) that produced this time-course is absolutely deterministic – No random noise has been added to the simulation output. In other words, the neutrophil time-course in Figure 11.8c is chaotic. To confirm that this indeed reflects chaotic behavior, we need to determine that the dynamic evolution of the predicted neutrophil density is sensitive to small changes in initial conditions. Figure 11.9 shows the result of just such a determination. Solution of Equation 10.2 using the SIMULINK model in Figure 11.7 is performed using two sets of initial conditions. The first assumes the initial neutrophil density, $x(0)$, to be 1.22; in the second simulation, $x(0)$ is set equal to 1.21. As Figure 11.9 shows, after an initial period of remaining very close together, the two simulated time-courses begin to diverge after $t = 100$ days.
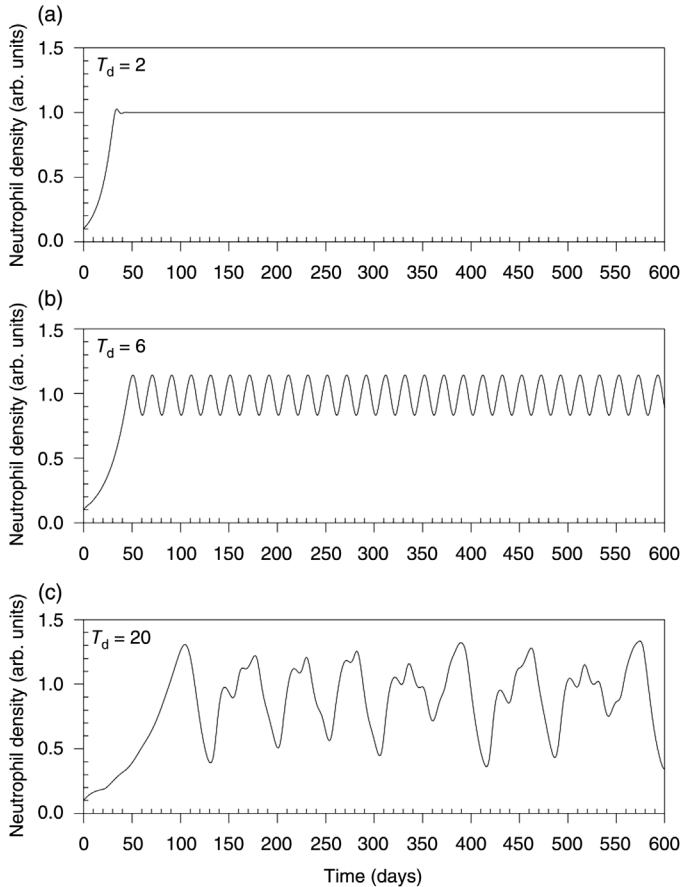
FIGURE 11.8 Spontaneous dynamics of neutrophil density predicted by the MacKey-Glass model as the cell maturation time, T_d, increases from the "normal" value of 2 days (a) to 6 days (b), and in chronic myeloid leukemia to 20 days (c).

Eventually, each trajectory appears to follow its own course, although both remain within a bounded range of values. This is the hallmark of chaotic behavior.

11.2.3 Model of Cardiovascular Variability

Beat-to-beat fluctuations in the duration of the cardiac cycle, arterial blood pressure, and cardiac output are well-known phenomena. Are these simply manifestations of intrinsically noisy processes? Recent applications of the tools of nonlinear dynamics to cardiovascular measurements suggest that deterministic chaos may be the underlying mechanism. In this section, we examine a model published recently by Cavalcanti and Belardinelli (1996) that postulates that the spontaneous variability in
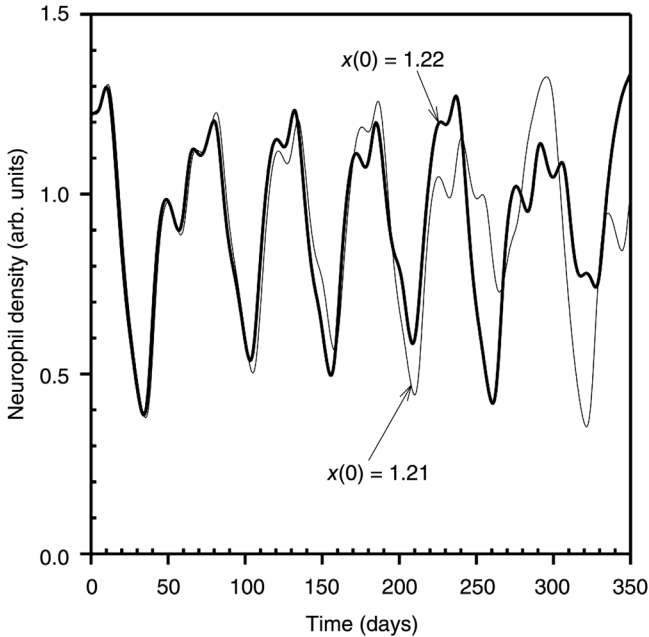
FIGURE 11.9   Sensitivity of the neutrophil regulation model to initial conditions indicates that the model becomes chaotic when the cell maturation delay, T_d, is increased to 20 days.

heart rate and blood pressure results from chaotic behavior occurring in the baroreflex control system. Functionally, the model contains two feedback loops: one representing the effect of the baroreflex on heart rate and the other representing the effect of the baroreflex on cardiac contractility, which in turn affects stroke volume.

To highlight the importance of interaction between the two feedback loops, we begin by discussing a simplified variant of this model. The schematic block diagram that represents this model is shown in Figure 11.10a, and the corresponding SIMULINK implementation of the model (`cvvar1.slx`) appears in Figure 11.10b. For a given cardiac output, circulatory mechanics determines the corresponding level of arterial blood pressure. In the model, circulatory mechanics is characterized by a three-element *Windkessel* model. The latter consists of a resistance, representing the aortic characteristic impedance (r), placed in series with a parallel combination of the peripheral resistance (R) and the total arterial compliance (C). The differential equation describing the three-element Windkessel model is given by

$$RC\frac{dP}{dt} + P = rRC\frac{dQ}{dt} + (R + r)Q \tag{11.3}$$

where P and Q represent arterial blood pressure and cardiac output, respectively. The resistances and compliance are assumed constant and given the following
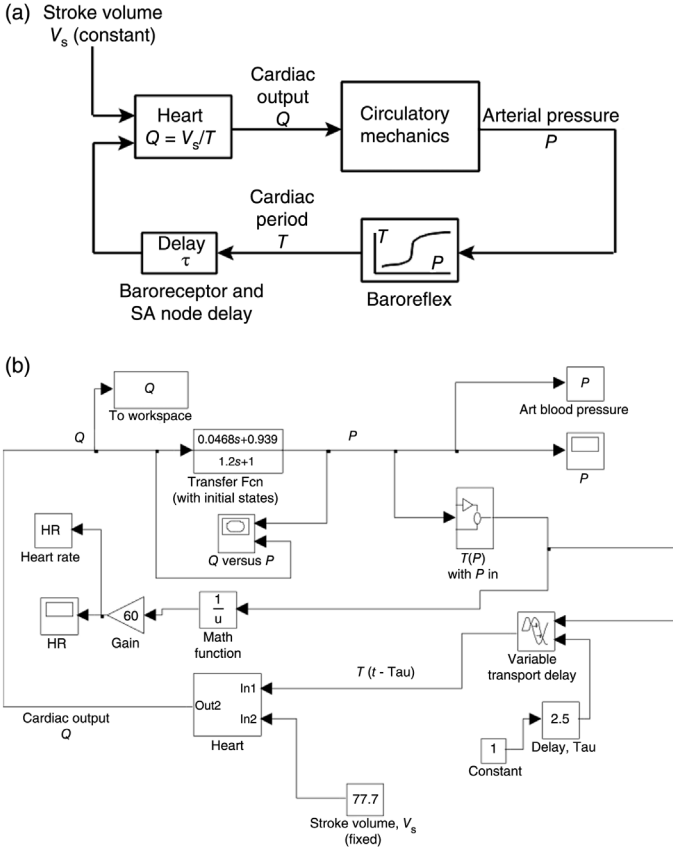
FIGURE 11.10 (a) Cardiovascular variability model with single feedback loop. (b) SIMULINK implementation (`cvvar1.slx`) of the above model.

values: $C = 1.333$ mL mmHg^{-1}, $R = 0.900$ mmHg s mL^{-1}, and $r = 0.039$ mmHg s mL^{-1}. Thus, the circulatory mechanics subsystem is linear.

Changes in P are sensed by the baroreceptors that relay this information back to the vasomotor center in the brainstem. The vasomotor responds with changes in vagal and sympathetic nerve activity that, in turn, modulate the cardiac period T (and thus, heart rate $= 1/T$). In this simplified version of the Calvacanti model, we assume stroke volume (V_s) to be independent of P. The steady-state characteristic of this baroreflex response is shown in Figure 11.11a. As P decreases, T also decreases, meaning that cardiac output increases, since $Q = V_s/T$ and V_s is constant. Conversely, as P increases, T also increases, decreasing Q. This accounts for the negative feedback effect of the baroreflex. However, when P falls below 80 mmHg or rises above 100 mmHg, the sensitivity of T to further changes in P drops substantially. The baroreflex response therefore saturates at low and high levels of
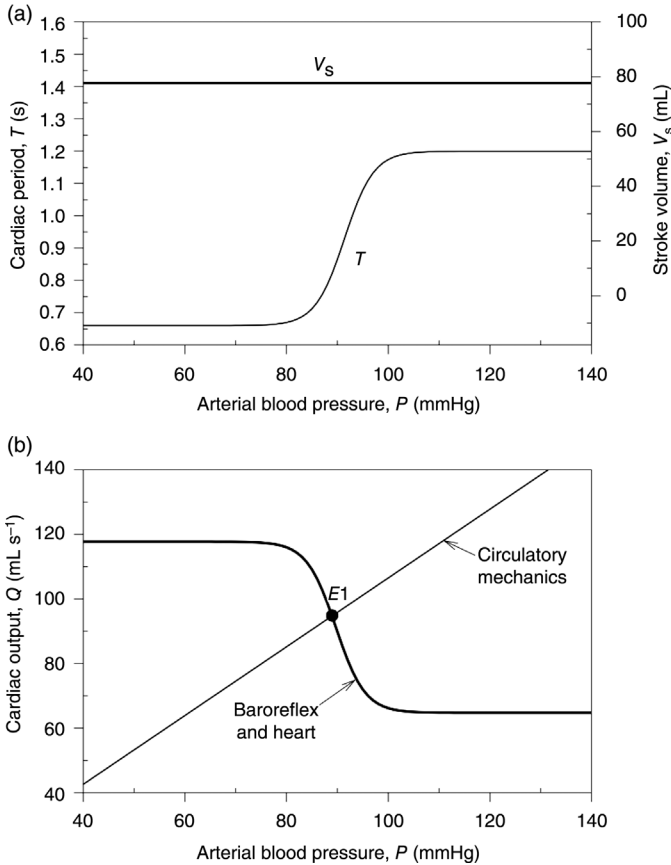
FIGURE 11.11   (a) Steady-state characteristics of the baroreflex; stroke volume is assumed to be fixed and independent of arterial blood pressure. (b) Steady-state properties of the baroreflex and heart combined (bold curve), shown together with the arterial pressure–cardiac output dependence determined by circulatory mechanics. Intersection of these two functions yields a single equilibrium point $E1$.

P. The dependence of T on P is given by the following equation:

$$T(P) = T_{min} + \frac{T_{max} - T_{min}}{1 + \gamma e^{-\alpha P / P_e}} \tag{11.4}$$

where T_{min} and T_{max} are the lowest and highest possible values for T, assumed to be 0.66 and 1.2 s, respectively. P_e is the equilibrium level of arterial pressure and is equal to 89 mmHg. The constants α and γ control the range and slope of the linear portion of the T–P curve; they are assigned values of 31 and 6.7×10^{13}, respectively.

The model does not take into account the dynamics of the sinoatrial node, as in the model of Saul et al. (see Section 5.4). However, an overall delay τ is incorporated. This delay represents the combined lag associated with the baroreceptor response time and the response times of the sinoatrial node to vagal and sympathetic stimulation. In the discussions that follow, we will examine how the dynamics of this baroreflex model are affected by τ as it is given different values that span the range of feasible delays.

Before proceeding with our exploration of the dynamics of this model, it is useful to apply the approach presented in Chapter 3 and to determine the steady-state operating point that results from the matching of the feedforward (circulatory mechanics) and feedback (baroreflex + delay + heart) portions of this closed-loop system. This is shown in terms of the variables Q and P in Figure 11.11b. Intersection of the curve representing the baroreflex and heart (bold curve) with the straight line representing linear circulatory mechanics yields the equilibrium point $E1$ at which $P = P_e = 89$ mmHg and $Q = 78.8$ mL s^{-1}. Applying the phase-plane method of analysis (Section 10.2), it can be determined that this equilibrium point is stable. Since these considerations involve only the steady state, the delay plays no role in the determination of the equilibrium point.

Figure 11.12 shows some simulation results obtained with the SIMULINK model `ccvar1.slx`. The evolution of the time-course of instantaneous heart rate ($= 60/T$, expressed in beats per minute) is displayed in each of the left panels (a, c, e, and g) of Figure 11.12; the transient effects of starting the simulations with arbitrary initial conditions have been removed from these plots. The corresponding phase-space plots, with Q plotted against P, are shown in the panels on the right. When τ is small, for example, 0.5 s (as in Figure 11.12a), the response rapidly converges to the equilibrium level; this is represented as a single dot in the phase space (Figure 11.12b). As τ is increased, the system becomes oscillatory, as demonstrated by the periodic waveform (Figure 11.12c) and limit cycle behavior in the phase-space plot (Figure 11.12d). This periodic behavior persists with further increases in τ (Figure 11.12e-h). The cycle duration of the oscillation, however, increases as τ is increased. Again, this example demonstrates that prolongation of the time delays inherent in a closed-loop system constitutes a highly destabilizing effect.

A schematic block diagram representing the complete version of the model is displayed in Figure 11.13a. The SIMULINK implementation of this model, `cvvar2.slx`, is shown in Figure 11.13b. Here, stroke volume V_s is assumed to be a function of arterial pressure. Figure 11.14a illustrates the dependence of V_s on P, which is also basically sigmoidal in form as in the relation between T and P. Above 90 mmHg, V_s remains relatively constant. Below 80 mmHg, V_s decreases steeply with decreases in P. This is due to the concomitant increase in heart rate, which reduces the time for ventricular filling. As P decreases even further, V_s decreases toward zero, as the heart begins to fail. Incorporation of the dependence of V_s on P produces a dual feedback-loop model. Furthermore, since cardiac output is the ratio of V_s to T, this introduces a nonlinear interaction between the two feedback loops.
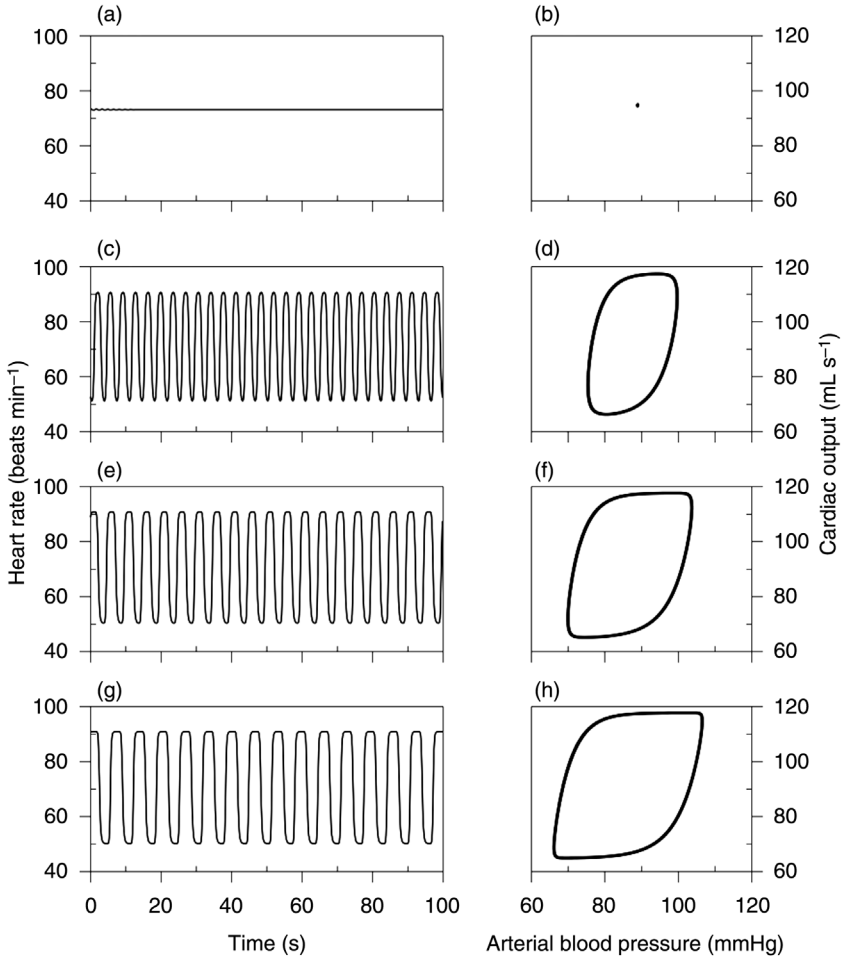
FIGURE 11.12   Simulation results produced by `ccvar1.slx`. Left panels show predicted time-courses of heart rate; right panels show *x*–*y* plots of cardiac output (= heart rate × stroke volume) versus arterial blood pressure. *A* and *B*: $\tau = 0.5$ s; *C* and *D*: $\tau = 1.2$ s; *E* and *F*: $\tau = 1.8$ s; and *G* and *H*: $\tau = 2.5$ s. As the delay increases, the system becomes oscillatory with increasing cycle duration.

Figure 11.14b displays the steady-state relations between cardiac output and *P* for both the circulatory mechanics subsystem (thin line) and the subsystem representing the baroreflex and heart (thick curve). Notice that as *P* decreases from physiological levels, *Q* attains a peak and subsequently decreases monotonically toward zero. Consequently, intersection of the two curves yields two equilibrium points, *E*1 and *E*2, unlike the single feedback-loop case. Stability analysis shows that *E*1 is stable and corresponds to the same equilibrium point that
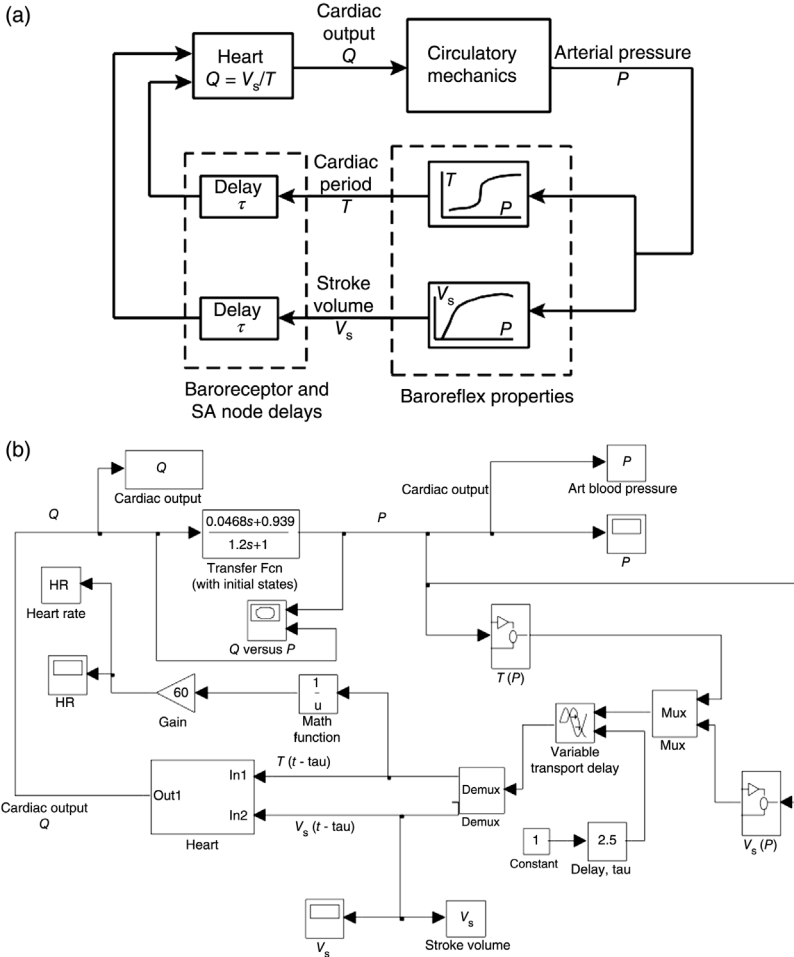
FIGURE 11.13 (a) Cardiovascular variability model with two interacting feedback loops. (b) SIMULINK implementation (`cvvar2.slx`) of the above model.

we found for the simpler version of the model. However, it can be shown that $E2$ is an unstable fixed point. This is basically due to the fact that $E2$ falls on the portion of the baroreflex–heart Q–P curve where the slope now has become positive, instead of negative as for the case of $E1$. In other words, the system contains negative feedback over the range of high P values, but becomes one with positive feedback when P falls below 80 mmHg.

What is the dynamic behavior of this model when the time delay is taken into account? Figure 11.15 shows some examples of simulations performed with the SIMULINK program `ccvar2.slx`. When τ is kept small at, say, 0.5 s, the system trajectory converges rapidly to the steady state represented by $E1$, after the initial
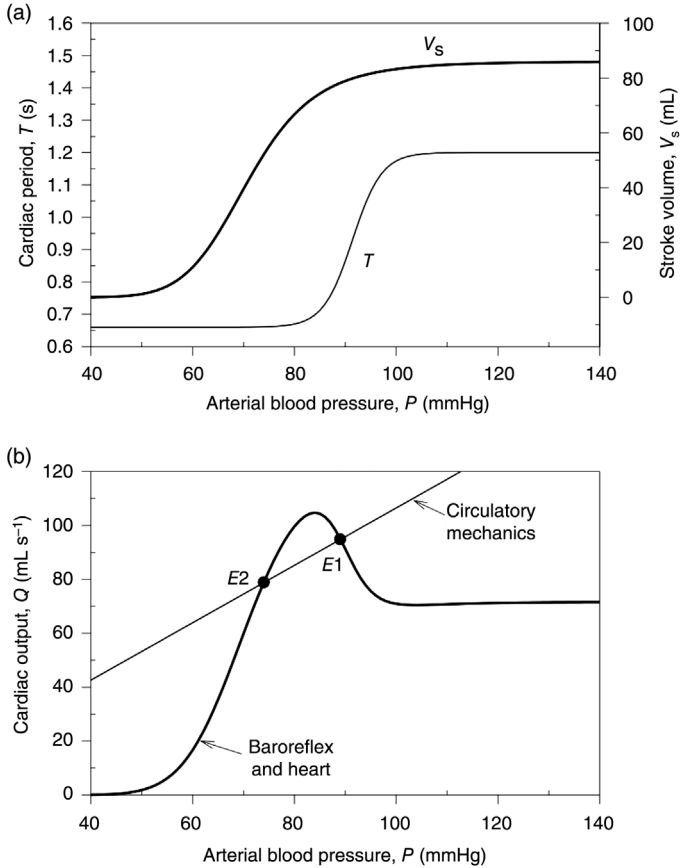
FIGURE 11.14 (a) Steady-state characteristics of the baroreflex; *both* cardiac period and stroke volume are assumed to be dependent on arterial blood pressure. (b) Steady-state properties of the baroreflex and heart combined (bold curve), shown together with the arterial pressure–cardiac output dependence determined by circulatory mechanics. Intersection of these two functions yields two nontrivial equilibrium points, $E1$ and $E2$.

transient that depends on starting conditions (Figure 11.15a and b). As τ is increased, a periodic oscillation develops (Figure 11.15c and d for $\tau = 1.2$ s) as in the single feedback-loop case; in this case, the frequency of the oscillation is 0.28 Hz. However, with further increases in τ, *period-doubling* occurs. At $\tau = 1.8$ s, the system remains periodic; however, the oscillations now contain multiple frequencies at a subharmonic and superharmonics of 0.19 Hz (Figure 11.15e). The corresponding phase-space plot shows a complicated double loop (Figure 11.15f). When τ is increased beyond 2 s, the oscillations turn into chaos, as illustrated in the example in Figure 11.15g and h; in this case, $\tau = 2.5$ s. Figure 11.16a provides a closer look at the relative time-courses of the
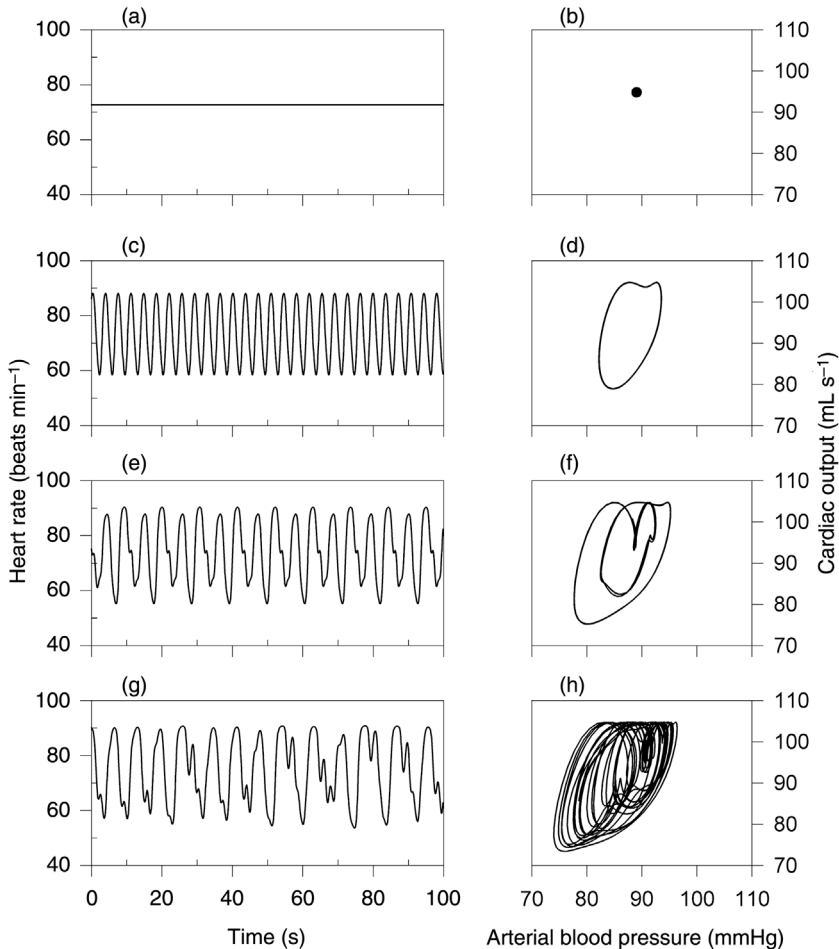
FIGURE 11.15 Simulation results produced by `ccvar2.slx`. Left panels show predicted time-courses of heart rate; right panels show x–y plots of cardiac output versus arterial blood pressure. (a and b) $\tau = 0.5$ s. (c and d) $\tau = 1.2$ s. (e and f) $\tau = 1.8$ s. (g and h) $\tau = 2.5$ s. As the delay is increased, the system becomes periodic; then period-doubling occurs, and finally, chaos sets in.

system variables P, heart rate (or equivalently, T), and V_s when the system is in chaotic mode, with τ set equal to 2.5 s. When these three state variables are plotted in three-dimensional format, the picture of the chaotic attractor emerges (Figure 11.16b).

As in the neutrophil regulation model of Section 11.2.2, the preceding discussion demonstrates that chaotic behavior can be produced in systems with relatively low-order dynamics when the important ingredients of time delay and mixed feedback are present. However, the predictions made by this model are not so consistent with
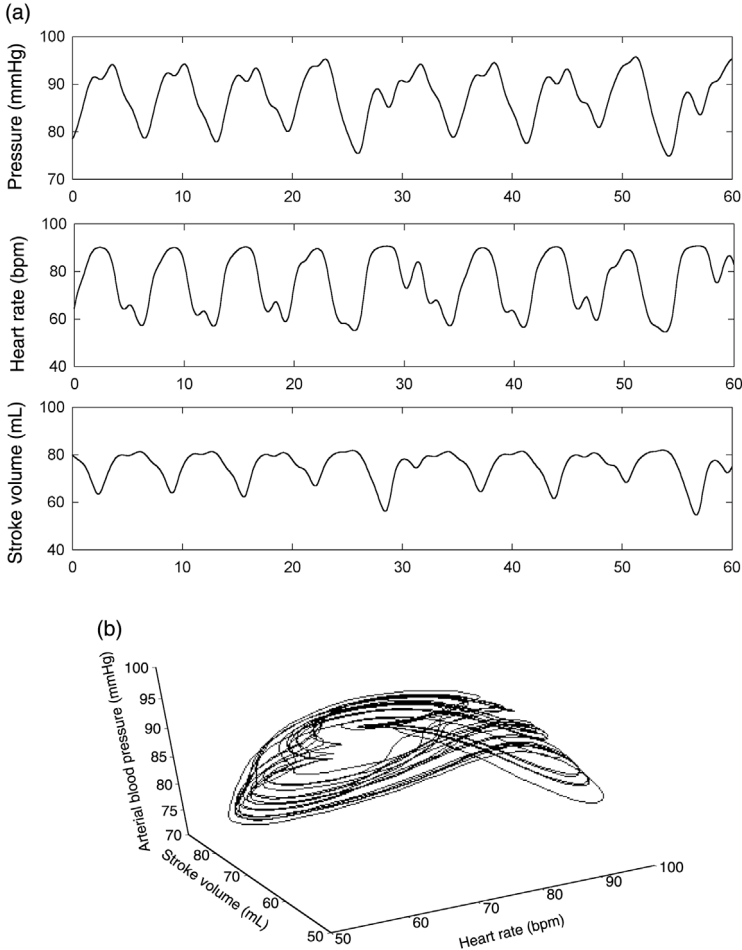
FIGURE 11.16　(a) Time courses of arterial blood pressure, heart rate, and stroke volume when the system is in chaotic mode ($\tau = 2.5$ s). (b) Three-dimensional view of the chaotic attractor.

empirical observations, which tend to show more chaotic behavior in heart rate variability in normals (where τ would be expected to be small) and less chaotic, more periodic behavior in patients who suffer from myocardial infarction (where we would expect τ to be increased due to the dominant influence of the sympathetic nervous system). One reason could be that this model is highly simplistic in not explicitly incorporating the dynamics of the sinoatrial node, the baroreflex, and the heart. Another important factor is that the influence of breathing on arterial pressure and heart rate fluctuations is ignored. Finally, the assumed dependencies relating P to T and V_s may be too simplistic.

11.3   COUPLED NONLINEAR OSCILLATORS: MODEL OF CIRCADIAN RHYTHMS

In Section 10.3.2, we pointed to the phenomenon of "entrainment" as a distinguishing feature of nonlinear oscillators that are coupled together. Since the body contains a number of pacemakers and we have evolved to adapt to the 24 h rhythm of the light–dark cycle, it follows that several of these circadian oscillators must be entrained to the external *zeitgeber* (or, if translated literally from German, "time giver"). There is a large body of evidence to support this notion. For example, in subjects who have been isolated from all external time cues for over 2 months, the sleep–wake and body temperature rhythms become *internally desynchronized*, with the temperature oscillator assuming a periodicity that is slightly longer than 24 h and the sleep–wake cycle being prolonged to approximately 30 h. Other physiological rhythms in these subjects then tend to be entrained to either the temperature or the sleep–wake cycle. Kronauer et al. (1982) have proposed a model consisting of two coupled van der Pol oscillators to represent the temperature and sleep–wake pacemakers. Under normal circumstances, both oscillators are entrained to the external 24 h zeitgeber. However, under conditions that simulate temporal isolation, the model exhibits the complex variations in periodicities and relative phasing between the temperature and sleep–wake rhythms that closely resemble empirical measurements.

The schematic diagram in Figure 11.17a shows the temperature and sleep–wake oscillators and the mutual coupling between them. Kronauer and coworkers found that it was necessary to assume that the synchronizing zeitgeber is applied directly to the sleep–wake oscillator instead of the temperature system in order to obtain realistic phase relations between them during zeitgeber entrainment. We represent the outputs of the temperature and sleep–wake oscillators by x and y, respectively. The zeitgeber output is represented by z. The model is characterized by the following pair of van der Pol equations:

$$k^2\ddot{x} + k\mu_x(x^2 - 1)\dot{x} + \omega_x^2 x + F_{yx}k\dot{y} = 0 \qquad (11.5)$$

and

$$k^2\ddot{y} + k\mu_y(y^2 - 1)\dot{y} + \omega_y^2 y + F_{xy}k\dot{x} = F_{zy}\sin\left(\frac{\omega_z t}{k}\right) \qquad (11.6)$$

In the above equations, the *scaling factor* k ($= 24/2\pi$) is introduced so that the intrinsic periods of the (uncoupled) temperature and sleep–wake oscillators would equal 24 h if the angular frequencies ω_x and ω_y, respectively, were each set equal to unity. Similarly, ω_z is set equal to unity so that the zeitgeber period is 24 h. The parameters μ_x and μ_y represent the "stiffness" of the temperature and sleep–wake oscillators, respectively. They determine the time constants of the transient duration of adjustment in phase of each oscillator to that of the zeitgeber following release

(a)
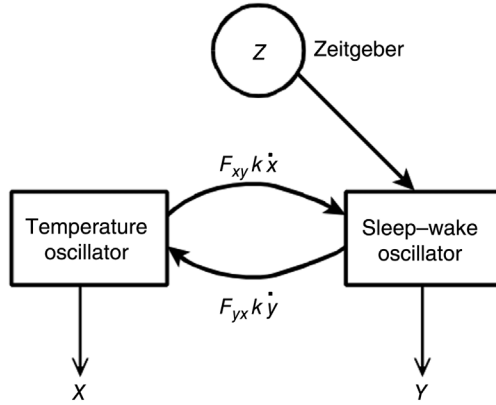
(b)
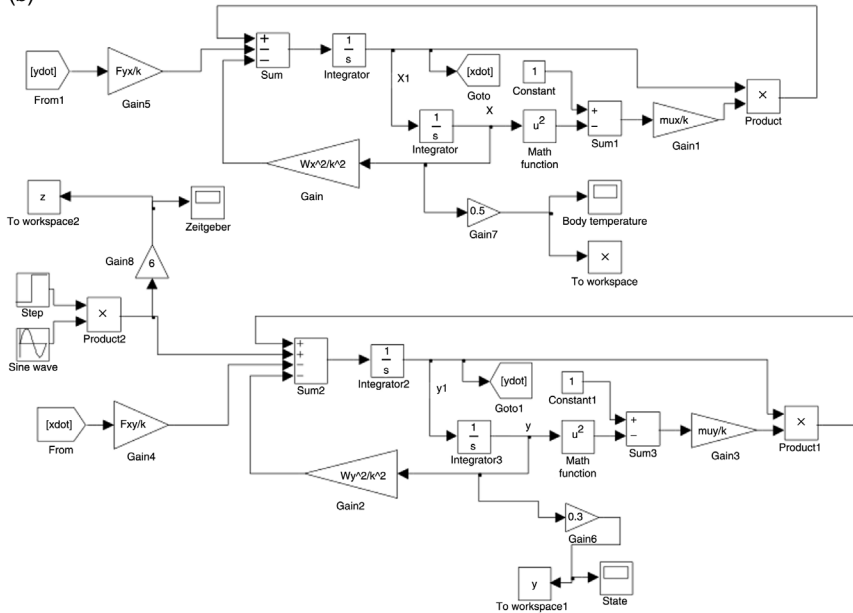
FIGURE 11.17   (a) Schematic representation of the circadian model of Kronauer et al. (1982). (b) SIMULINK implementation (`circad.slxl`) of the circadian model.

from entrainment or after re-entrainment. In our simulations, μ_x and μ_y are each assigned the value of 0.1. F_{yx} and F_{xy} represent the strengths of the coupling between the temperature and sleep–wake oscillators. F_{yx} and F_{xy} are assigned the values of −0.04 and −0.16, respectively. The relative magnitudes of these values imply that the temperature oscillator has a stronger influence on the sleep–wake oscillator than vice versa. F_{zy} is assigned the value of unity. The SIMULINK implementation of this model (`circad.slx`) is shown in Figure 11.17b. Note the
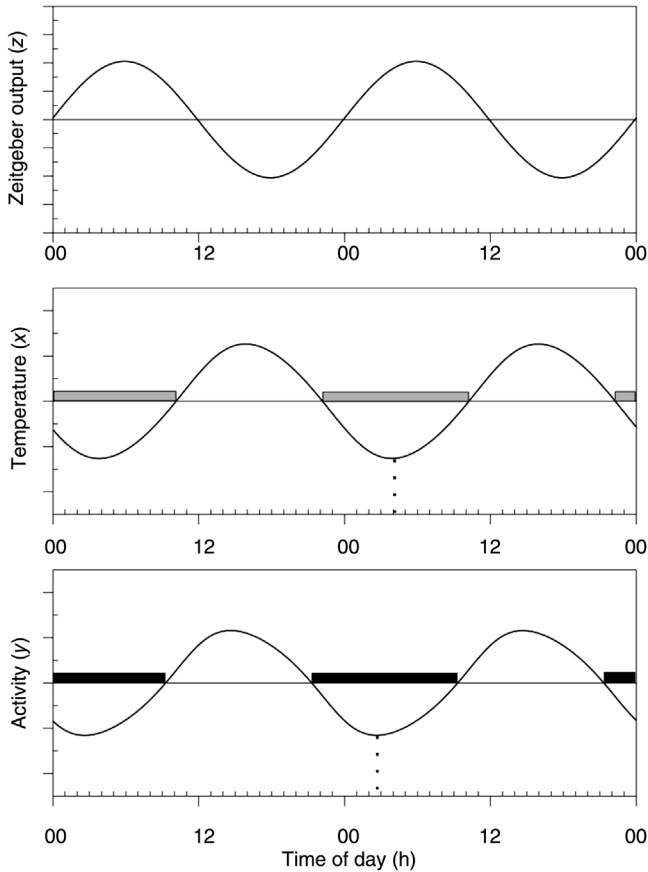
FIGURE 11.18   Phase relations between the oscillations in core body temperature (x) and activity (y) during entrainment by the external 24 h zeitgeber (z). Time is shown in hours starting at midnight ("00"); "12" represents noon. The gray and dark horizontal bars indicate the durations over which temperature and activity, respectively, are below their mean levels. Dotted lines indicate times at which temperature and activity are at their lowest levels.

use in this model file of `Go to` and `From` blocks in order to couple the two oscillators to each other without creating a mess in signal lines.

Figure 11.18 shows the simulated behavior of temperature (x, middle panel) and activity (y, bottom panel) during entrainment by the zeitgeber (z, top panel). The selected model parameters allow the establishment of the following phase relations among the three oscillations. The positive half of the z-oscillation corresponds to the 12 h duration between midnight and noon; these times are represented as 00 and 12, respectively, in Figure 11.18 and all subsequent graphs. The negative half of the x-oscillation corresponds to the duration over which core body temperature is below its mean level. Similarly, the negative half of the y-oscillation represents the

duration over which activity is below average. Kronauer used the middle two-thirds of this duration to represent the period of sleep. However, for simplicity, we will take the entire below-average activity duration to correspond to sleep. The model simulation shows that the start of the "sleep period" (represented by the dark bars in Figure 11.18, bottom panel) occurs at approximately 9:30 p.m. Core body temperature starts to fall below its mean level (gray bars in Figure 11.18, middle panel) some 40 min later, at about 10:20 p.m. Thus, temperature attains its lowest value at approximately 4 a.m., some 6.5 h after sleep onset. This is reasonably consistent with empirical observations.

In the simulation shown in Figure 11.18, the following intrinsic frequencies of the temperature and sleep–wake oscillators were assigned: $\omega_x = 0.99$ and $\omega_y = 0.92$. These frequencies correspond to periodicities of 24.37 h for temperature and 26.09 h for activity. To simulate the "free run" condition (i.e., release from zeitgeber entrainment), F_{zy} was set equal to zero. In addition, ω_y was assumed to decrease linearly so that by the end of the 100th day after the start of free run, ω_y would become 0.78, which corresponds to a period of 30.8 h. Kronauer found this latter assumption to be necessary to produce a better match between the simulation results and empirical data.

Figure 11.19 shows the relative phasing between x (light waveforms) and y (bold waveforms) during selected segments of the free run duration. In the first 5 days following release from zeitgeber drive, the time of sleep onset, which previously preceded the drop in temperature below its mean level, can be seen to be delayed progressively (top panel, Figure 11.19). On day 5 after release into free run, the point of lowest activity (*mid-sleep*) occurs approximately 3 h *after* the point of lowest core body temperature. This gradual delaying of the period of sleep relative to that of lower-than-average body temperature continues until approximately day 35. The day-to-day periods (τ_x and τ_y) of the two oscillators are shown in Figure 11.20, along with their intrinsic periods (T_x and T_y). Note that immediately following release into free run, the temperature oscillator abruptly increases its period by about 2 h, but almost as abruptly, moves back toward τ_x. From day 36 through day 70, the "drift" in relative phase between x and y ceases to occur; instead, there is a tendency for both oscillators to arrive at a compromise cycle duration and relatively constant phase relationship. This results in the tendency for τ_x and τ_ψ to fluctuate around each other (Figure 11.20). This stage of the free run is referred to as *phase-trapping*, and is illustrated in the middle panel of Figure 11.19. After the 70th day, a new stage begins, known as *internal desynchronization*. Here, each oscillator tends to track its own intrinsic period, as Figure 11.20 quite dramatically illustrates. The oscillations in τ_x and τ_ψ reflect the fact that both oscillators still exert some influence on the other and are not totally independent. During internal desynchronization, the sleep and low-temperature periods can become out-of-phase with one another (Figure 11.19, bottom panel). The sleep–wake cycle duration alternates between shorter periods of $\sim$26 h and very long periods of $\sim$35 h. Again, these simulated results resemble what has been observed in temporally isolated humans.
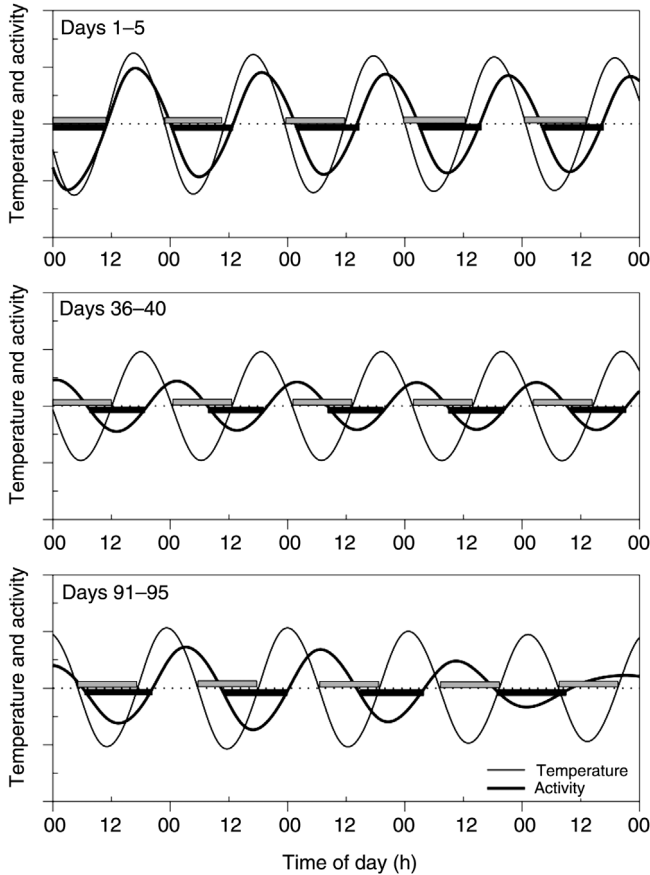
FIGURE 11.19 Phase relations between the temperature and activity oscillators following release from 24 h entrainment by the external zeitgeber. Each panel shows 5 days of simulated behavior. Light waveforms represent temperature, and bold waveforms represent activity. Gray and dark bars represent durations over which temperature and activity, respectively, are below their corresponding mean levels.

11.4 TIME-VARYING PHYSIOLOGICAL CLOSED-LOOP SYSTEMS: SLEEP APNEA MODEL

In the previous section, we saw how direct input "forcing" from the temperature oscillator affected the dynamics of the sleep–wake (or activity) oscillator, and vice versa. The coupling between related systems is frequently not as direct. For instance, changes in sleep–wake state are known to affect chemoreflex gain, cardiac output, and circulatory delay. These factors are not state variables, but instead comprise the "parameters" in any model of respiratory control. In the previous models of respiratory control that we have discussed (see Sections 3.7, 6.7, and 8.5.2), these
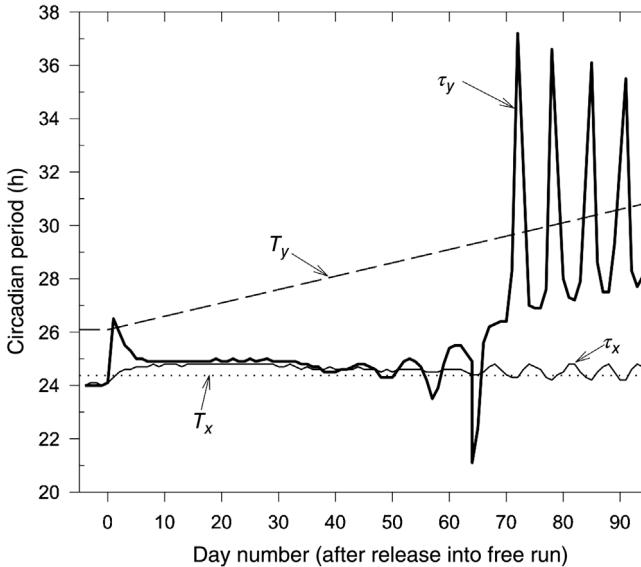
FIGURE 11.20   Dynamic changes in the periods of the temperature (τ_x, light tracing) and activity (τ_y, bold tracing) oscillators following release from entrainment by the external 24 h zeitgeber. The dotted and dashed lines represent the intrinsic periods of the temperature (T_x) and activity (T_y) oscillators. Both τ_x and τ_y assume values between 24 and 25 h throughout much of the duration of free run. However, after 70 days, there is desynchronization between the two oscillators and τ_y takes on much larger values.

parameters were always assumed to take on constant values. A model in which one or more of the key parameters change with time is said to be *time-varying* or *nonstationary*.

In this section, we examine the dynamics that can result from a simple model of obstructive sleep apnea (OSA). A primary mechanism that leads to the obstructive apnea in patients with OSA is the pronounced decrease in tone of the upper airway muscles when sleep sets in. Compounding the effect of this mechanism is the added predisposing factor that the upper airway passage in these patients is already anatomically narrower than in normal individuals. As such, when negative intra-luminal pressure is applied to the upper airway during inspiration, the net result is a tendency for the "floppy" airway to collapse, thereby obstructing airflow. There are, of course, many other factors involved, but what we have just described is the basic chain of events that generally occurs during the periodic episodes of sleep apnea. A schematic diagram of the model is shown in Figure 11.21. Figure 11.22a shows the SIMULINK implementation of this model (`osa.slx`). The OSA model is similar to the respiratory control models that we have considered previously, except for three major added features:

    (1) The first is the addition of the "upper airway conductance" component. In the model, this takes the form of a "gain" that transforms the total respiratory
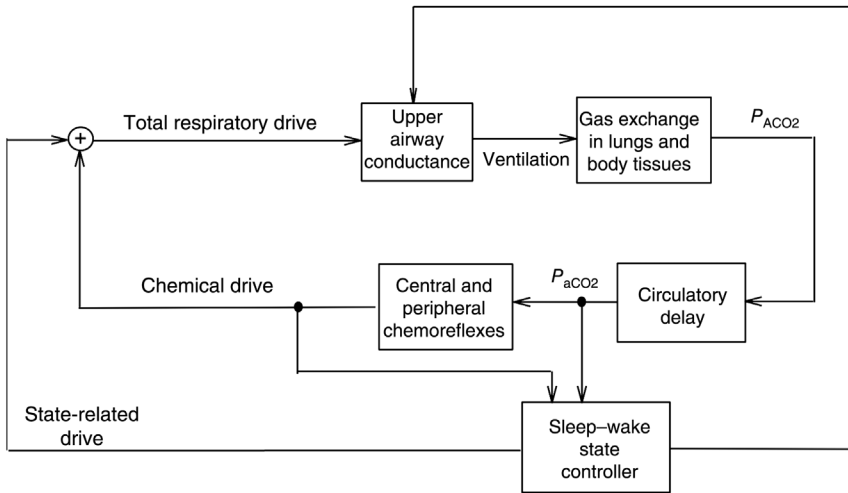
FIGURE 11.21 Model of state–chemoreflex interaction in ventilatory control.

drive into ventilation (or airflow). However, this gain is a time-varying parameter. During wakefulness, upper airway conductance is assumed to be equal to unity so that all of the total respiratory drive is converted directly into ventilation. During the transition from wake to sleep, we assume that the upper airway conductance decreases in proportion to the "state-related drive," so that when sleep is attained, conductance becomes zero (i.e., the upper airway is fully obstructed). Thus, the upper airway conductance, as we have defined it, is a normalized quantity that can only assume values between zero and unity. The SIMULINK subsystem block that represents this model component is shown in Figure 11.22b.

(2) Aside from the time-varying effect of state changes on upper airway conductance, the model also includes the direct effect of sleep–wake state on respiratory drive. There is much empirical evidence that suggests that a "wakefulness drive" (or "wakefulness stimulus") to breathing that is present during the awake state is withdrawn or inhibited as sleep sets in. This state-related drive is separate from the chemical drive that depends on feedback from the chemoreceptors. Thus, in the model, we have assumed that total respiratory drive consists of the sum of the combined chemoreflex or chemical drive and this state-related drive. During the transition from wake to sleep, we assume a simple linear decrease in this state-related drive, so that it becomes zero when stable sleep has been achieved. As we will demonstrate later, the duration over which this linear decrease (or equivalently, the wake-to-sleep transition) occurs, τ, plays an important role in determining the ventilatory and state dynamics that accompany sleep.
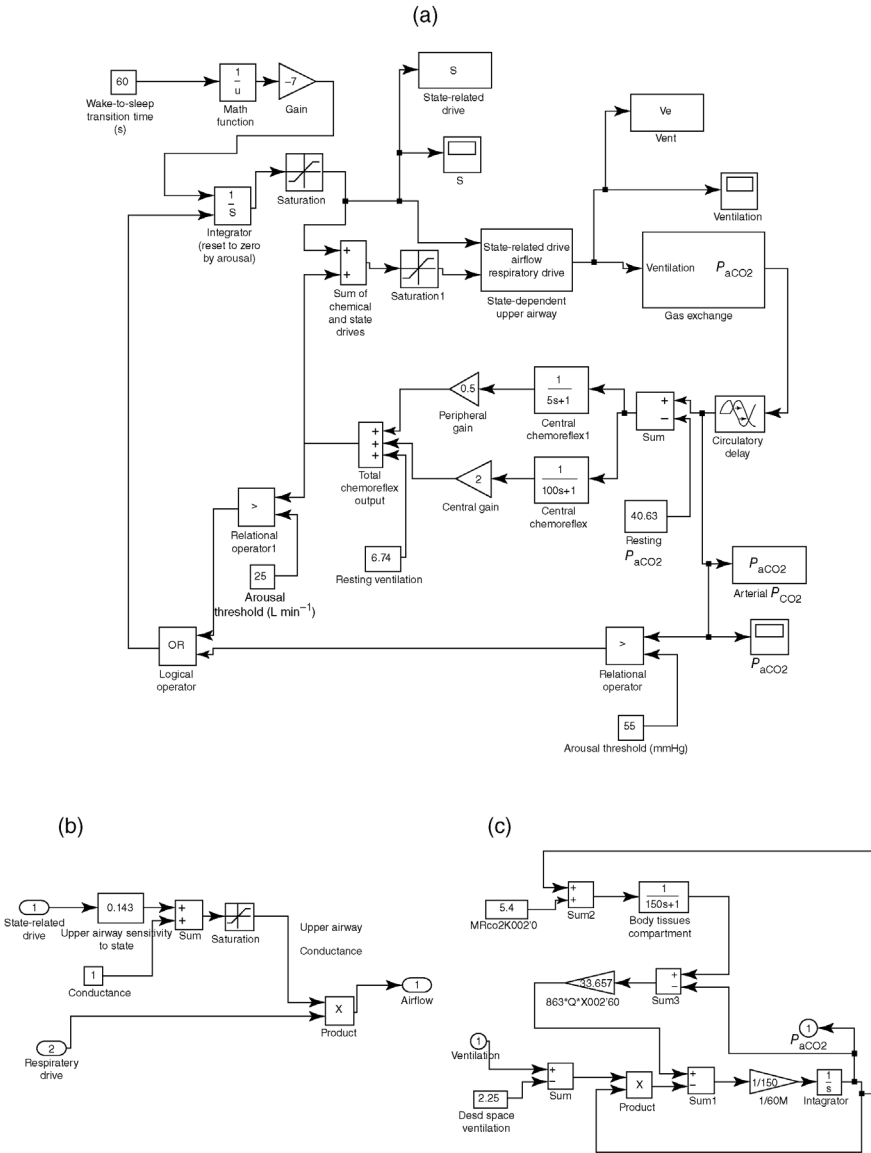
FIGURE 11.22 (a) SIMULINK implementation (`osa.slx`) of the model of state–chemoreflex interactions in obstructive sleep apnea. (b) Subsystem showing how upper airway conductance is dependent on state. (c) Subsystem that characterizes CO_2 exchange in the lungs and body tissues.

(3) The final key component in this model is the "sleep–wake state controller." This controller determines the time-course of the decrease in state-related drive during the transition from wake to sleep. It also receives two forms of feedback from the other parts of the model. First, it monitors the arterial P_{CO2} (P_{aCO2}). Second, it monitors the chemical drive, that is, the combined output of the central and peripheral chemoreflexes. During sleep or during the transition from wake to sleep, if P_{aCO2} exceeds 55 mmHg *or* the chemical drive exceeds 25 L min^{-1}, the controller will revert the current state back to wakefulness and restart the transition from wake to sleep. This automatic mechanism simulates, to a first approximation, the arousals that are known to occur when the stimulation of respiratory drive exceeds certain *arousal thresholds*. The arousal mechanism is a potent protective defense against asphyxiation during sleep. However, at the same time, it is the reason why sleep architecture is so severely disrupted in subjects who have OSA.

The equations characterizing the rest of the model are the same as those described in Section 6.7.1, except for the addition of the body tissues compartment in the gas exchange subsystem. CO_2 exchange in this compartment is characterized by the differential equation (Equation 11.7):

$$\frac{V_T}{Q}\frac{dP_{vCO_2}}{dt} = \left(P_{aCO_2} - P_{vCO_2}\right) + \frac{\dot{V}_{CO_2}}{QK_{CO_2}} \tag{11.7}$$

where P_{vCO2} is the mixed venous P_{CO2}, $\dot{V}_{CO2}$ is the metabolic production rate of CO_2, Q is the cardiac output, and V_T is the effective volume of the body tissues compartment. K_{CO2} is the slope of the CO_2 dissociation curve (approximated as a straight line) for blood. In the simulations, we assume that $V_T = 15$ L, $Q = 0.1$ L s^{-1}, $K_{CO2} = 0.0065$ mmHg^{-1}, and $\dot{V}_{CO2} = 210$ mL min^{-1}. The SIMULINK implementation of the gas exchange subsystem is shown in Figure 11.22c. The values of the other parameters are the same as those employed in Section 6.7.1.

Two simulations with the model showing how sleep onset would affect subsequent ventilatory and state variability are displayed in Figure 11.23. In both cases, the wake-to-sleep transition time τ is assumed to be 60 s. The first example (top panel of Figure 11.23) represents a "normal" subject, in which we have assumed upper airway conductance to be unchanged from wake to sleep. (In reality, upper airway resistance increases – or equivalently, upper airway conductance decreases – quite significantly during sleep, even in normals, although not to the point of collapsing the airway.) During the transition from wake to sleep (first 60 s), ventilation ($\dot{V}_E$) decreases and P_{aCO2} increases as a consequence of the reduction in state-related drive. However, after the transition period, there is some recovery of P_{aCO2} and $\dot{V}_E$ toward their original equilibrium levels. Finally, the new equilibrium level during sleep is established with P_{aCO2} about 2 mmHg higher than during wakefulness and $\dot{V}_E$ about 1 L min^{-1} lower. The main point here is that the wake-to-sleep transition occurs
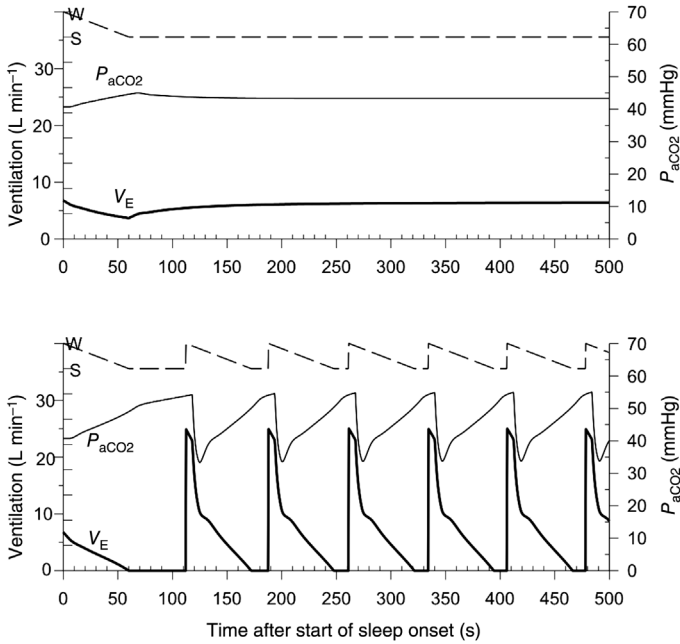
FIGURE 11.23   Sample simulations generated by `osa.slx`, showing effect of sleep onset on ventilation and P_{aCO2}, as well as subsequent sleep–wake state. In the "normal subject" (top panel), stable levels of ventilation, P_{aCO2}, and sleep are attained. However, in the subject with obstructive sleep apnea, there are alternating episodes of upper airway obstruction and arousal-induced hyperpnea.

smoothly and without incident. However, in the second simulation representing OSA subject (lower panel of Figure 11.23), the result is quite different. Here, as the transition from wake to sleep occurs, the rate of decrease in $\dot{V}_E$ takes place more rapidly than in the normal subject. This is due to the fact that, in addition to the decrease in state-related drive, there is also a concomitant decrease in upper airway conductance. P_{aCO2} also rises at a substantially faster rate. When sleep is established, upper airway conductance becomes zero, and consequently there is approximately 50 s of obstructive apnea. During this interval, P_{aCO2} continues to rise until it exceeds the arousal threshold (55 mmHg). When arousal is triggered, there is an abrupt restoration of the state-related drive as well as a sudden increase in upper airway conductance from zero to unity. This abrupt return to the wake state briefly produces a large increase in $\dot{V}_E$. However, subsequently, a new transition from wake to sleep occurs, leading to obstructive apnea and then another arousal. Hence, a periodic alternation between apnea and hyperpnea, which coincides with the alternation between sleep and wake, occurs. The periodicity of this cyclic behavior is approximately 76 s in duration, which falls in the range of cycle times observed in OSA.
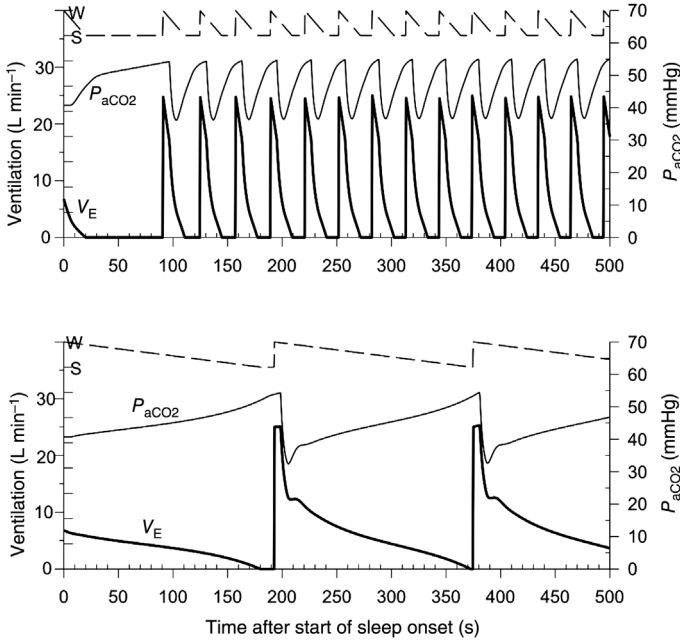
FIGURE 11.24 Simulations showing the effect of wake-to-sleep transition time on ventilatory and state stability following sleep onset. Top panel shows results for a rapid wake-to-sleep transition (20 s). Lower panel shows the effects of a slow wake-to-sleep transition (180 s). Although there is periodic ventilation in both cases, the rate of occurrence of arousals is much lower in the case with slow wake-to-sleep transition time.

Figure 11.24 shows the effect of altering the wake-to-sleep transition time on the subsequent dynamics of respiration and sleep–wake state. The first example (top panel of Figure 11.24) simulates the case in which $\tau = 20$ s. Although this represents the fast end of the spectrum, it is not entirely unrealistic: EEG measurements in sleep-deprived individuals do show a very rapid change in pattern that reflects fast sleep onset; these changes can occur over time spans as short as a few breaths. The rapid sleep onset leads to a long period of obstructive apnea, during which P_{aCO2} and chemical drive build up to a point where arousal is triggered. This is followed by a brief hyperpnea and then a subsequent rapid wake-to-sleep transition, and a repetition of the same cycle of events. The period between these apneas (or equivalently, arousals) is 34 s. An alternative means of expressing this result is to convert it into the corresponding *apnea index*, that is, the number of apneas per hour. In this case, the apnea index is about 106. This is at the high (although not impossible) end of the severity scale for OSA. On the other hand, if τ is large (i.e., wake-to-sleep transition is very slow), the arousals would be much less frequent and the apneas substantially shorted in duration, as

represented in the example in the lower panel of Figure 11.24. Here, $\tau = 180\,s$ and the corresponding periodicity is about $182\,s$.

The previous examples assumed central and peripheral chemoreflex sensitivities of 2 and $0.5\,L\,min^{-1}\,mmHg^{-1}$, respectively. Consider what happens when the peripheral chemoreflex gain is increased to $2\,L\,min^{-1}\,mmHg^{-1}$, simulating what would occur if the subject became hypoxic. The top panel of Figure 11.25 shows the result for the simulated "normal" (i.e., upper airway conductance unchanged by sleep onset). Due to the enhanced loop gain, the disturbance produced by the wake-to-sleep transition becomes progressively amplified until periodic breathing results. However, the cycling in $\dot{V}_E$ and P_{aCO2} is mediated completely by the chemoreflex loops and does not involve the arousal mechanism. As such, a stable stage of sleep is attained while the periodic respiration persists. The periodicity in this case is approximately $18\,s$, similar to the cycle times reported for periodic breathing at altitude. In the corresponding simulation for the OSA subject (lower panel, Figure 11.25), a more complex pattern occurs. Following the initial transition
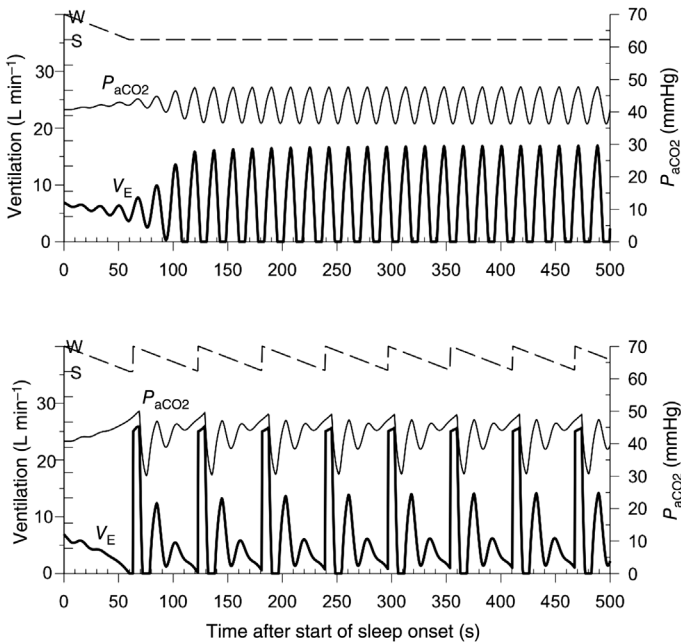
FIGURE 11.25 Simulations showing the dynamics of ventilatory and state variability following sleep onset in "subjects" with high chemoreflex gain. The top panel shows the results for a "normal" with no upper airway obstruction: Periodic breathing occurs, but the episodes do not elicit periodic arousal. The lower panel shows the corresponding results for the obstructive sleep apnea patient. Periodic arousals that are accompanied by brief durations of strong hyperpnea occur together with chemoreflex-mediated oscillations that do not involve arousal.

from wake to sleep, there is a brief interval of upper airway obstruction that is punctuated by an arousal. This produces a burst of hyperpnea that drives the P_{aCO2} down to the point where central apnea (i.e., period of zero respiratory drive) occurs. The central apnea is followed, in turn, by an overshoot in $\dot{V}_E$, a subsequent undershoot, and a second (but not so large) overshoot. The key point is that these latter oscillations are mediated by the chemoreflexes and take place while a new wake-to-sleep transition occurs. Thus, in this case, there is a periodic alternation between the large arousal-induced hyperpneas and the chemoreflex-mediated oscillations in ventilation. The arousals are spaced approximately 1 min apart, while the chemoreflex oscillations have a period of roughly 20 s. This example underscores the kind of interaction that can occur between the arousal and chemoreflex feedback loops as a consequence of the time-varying model parameter representing upper airway conductance.

11.5   PROPAGATION OF SYSTEM NOISE IN FEEDBACK LOOPS

We have seen thus far that a variety of factors, such as mixed feedback, time delays, nonlinear coupling, and time-varying properties, can contribute to the complex dynamical behavior of physiological control systems. These factors, however, share a common feature: They are all deterministic characteristics. Random influences are clearly present in physiological systems of all hierarchical levels. We will show through a simple example in this section that one potential source of the spontaneous variability observed in physiological systems may be the responses elicited in these systems by random input perturbations.

   We turn once again to the chemoreflex control of respiration model presented in Section 6.7.1. The natural variability in ventilation in wakefulness or sleep is known to be quite substantial, even when the respiratory control system is operating under clearly stable conditions. We propose that some, if not a large part, of this spontaneous variability may be due to the continual perturbation of the chemoreflex dynamics by random noise inputs. Figure 11.26a shows the proposed scheme through which "system noise" might enter the closed-loop structure of the respiratory control system. Since the primary structures that generate the drive to breathe are neural systems, it is reasonable to assume that respiratory drive, which is ultimately converted into ventilation, consists of the chemoreflex-mediated chemical drive plus some random influences that represent neural noise. Another obvious source of "noise" is the gas exchange process itself. Regional inhomogeneities in ventilation and perfusion of the lungs, as well as temporal fluctuations in cardiac output and the circulatory delays, can give rise to noise that contaminates the time-evolution of alveolar P_{CO2} (P_{ACO2}) and consequently P_{aCO2}. These reasons have prompted us to select the sites shown in Figure 11.26a as the points in the closed-loop structure at which noise enters the system. The SIMULINK implementation of this model (`noisycls. slx`) is displayed in Figure 11.26b. One detail that is of special importance in

(a)

(b)
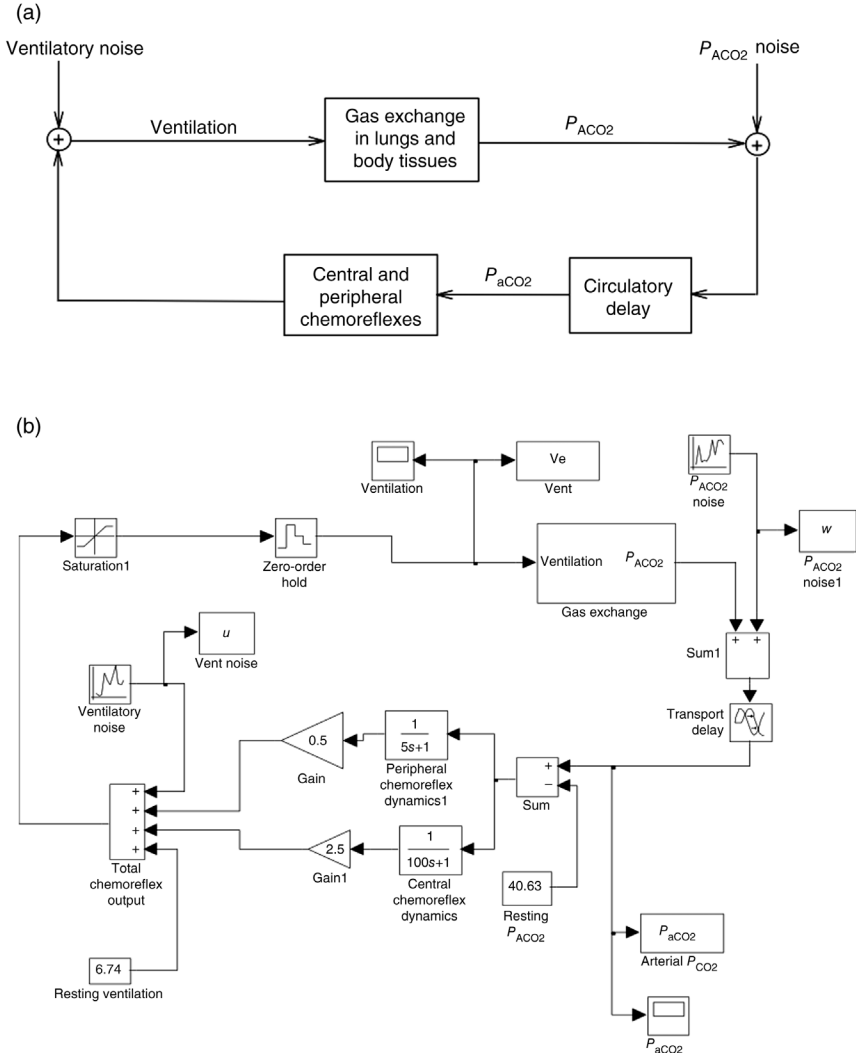
FIGURE 11.26 (a) Schematic diagram showing the "sites" at which noise might enter the closed-loop respiratory control system. (b) SIMULINK implementation (`noisycls.slx`) of the closed-loop respiratory control model with noise inputs.

this example relates to the method by which "breath-by-breath" values of ventilation and P_{aCO_2} are simulated, since the underlying equations assume a continuous-time process. To generate "breaths," the total respiratory output (chemoreflex drive plus "neural noise") is sent through a `Zero-Order Hold` block, which samples the continuous respiratory drive waveform every 4 s and
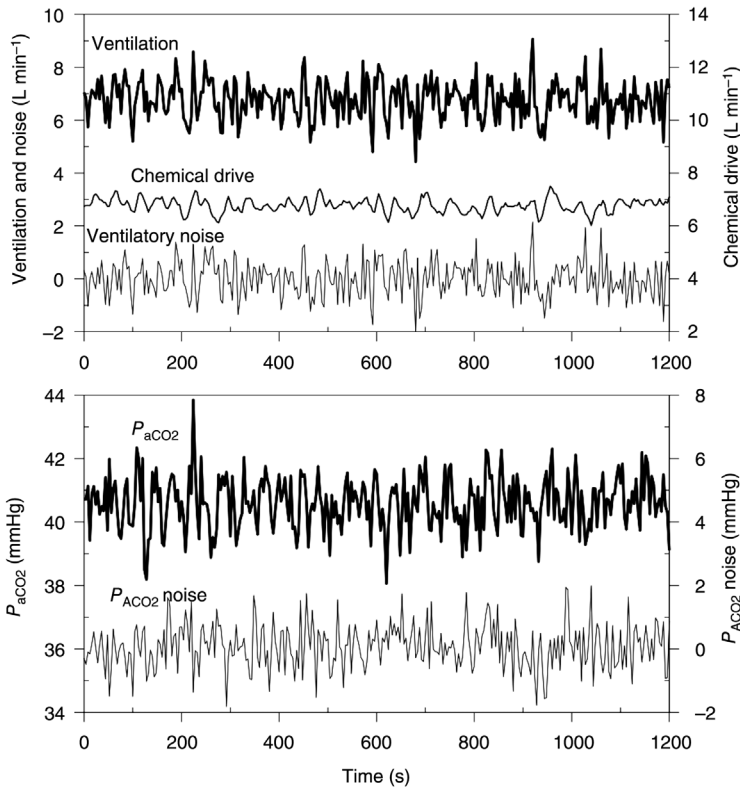
FIGURE 11.27 Spontaneous variability in ventilation (top panel, bold tracing) and P_{aCO2} (lower panel, bold tracing) resulting from the propagation of noise (both panels, light tracings) in the chemoreflex loops of the respiratory control model. "Chemical drive" in the top panel refers to the combined output of the central and peripheral chemoreflexes in response to the "noisy" P_{aCO2}.

holds each sampled value constant for the following 4 s. Then, ventilation, P_{aCO2}, and the noise processes are sampled at 4 s intervals and saved to the MATLAB workspace for further analysis.

Figure 11.27 shows samples of the spontaneously varying waveforms in ventilation (top panel) and P_{aCO2} (lower panel) generated by the model. The coefficient of variation (standard deviation divided by mean value) of the simulated ventilation time-course is approximately 12%, which falls in the range commonly observed in resting humans. The peripheral and central chemoreflex gains are assigned values of 0.5 and $2\,L\,min^{-1}\,mmHg^{-1}$, respectively. One may note the strong similarity between the ventilation time-course and the ventilatory noise waveform, since the former is simply the sum of the latter and the combined central and peripheral chemoreflex drives. Ventilation is causally related to current and past

values of the ventilatory noise input because of the propagation of the noise around the chemoreflex loops; however, the ventilatory noise is not causally related to past values of ventilation. The waveform labeled "chemical drive" represents the combined response of the chemoreflexes to the variations in P_{aCO2} elicited in part by the past spontaneous fluctuations in ventilation as well as by the noise entering the system through the gas exchange process (P_{ACO2} noise). The coefficient of variation of the P_{aCO2} waveform in the lower panel of Figure 11.27 is approximately 2%.

The corresponding spectra of these fluctuations in ventilation and P_{aCO2} show enhanced power in the 0.01–0.05 Hz range (bold tracings in Figure 11.28). These contrast with the much flatter (broadband) spectra of the noise inputs (light tracings
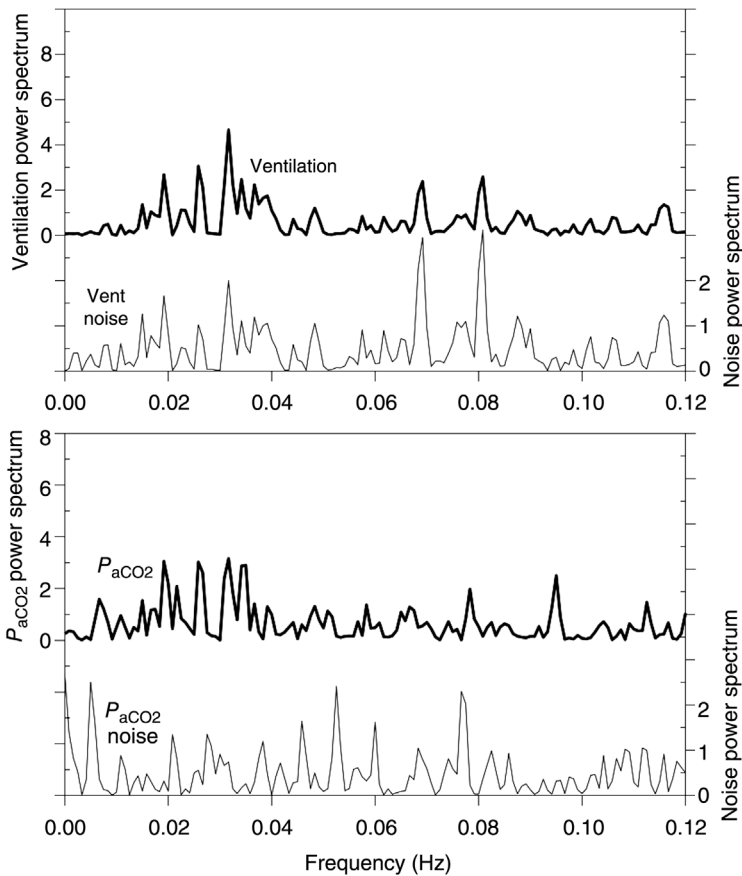
FIGURE 11.28 Power spectra of the spontaneous fluctuations in ventilation (top panel, bold tracing) and P_{aCO2} (lower panel, bold tracing) shown in Figure 11.27. Note the increased power in the 0.01–0.05 Hz range. This contrasts with the much flatter (broadband) spectra of the random noise inputs (light tracings).

in Figure 11.28). Thus, the propagation of these noise inputs through the feedback loops of the closed-loop system gives rise to certain oscillations that appear within a bandwidth that is consistent with the dynamic characteristics (i.e., gains, component lags, and delays) of the system. In this case, the bandwidth of frequencies 0.01–0.05 Hz corresponds to periodicities in the range of 20–100 s. These are compatible with the oscillation cycle durations that have been observed and reflect the frequency range for "resonance" in the human respiratory control system. When the peripheral chemoreflex gain is increased by 140% to $1.2\,\text{L}\,\text{min}^{-1}\,\text{mmHg}^{-1}$, the resulting simulation clearly shows enhanced oscillatory activity in both ventilation and P_{aCO2} (Figure 11.29). Furthermore, bursts of oscillations in ventilation and P_{aCO2} occur somewhat randomly, giving the appearance that the underlying system contains time variations in loop gain, although the model parameters, in fact, have been assigned constant values. The increased oscillatory activity is clearly evident
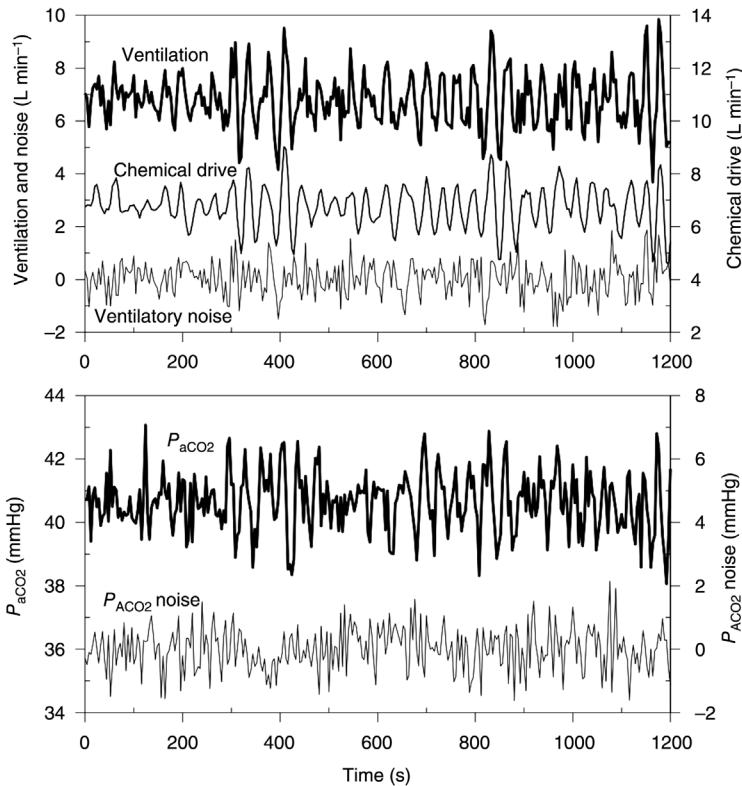
FIGURE 11.29   Spontaneous variability in ventilation (top panel) and P_{aCO2} (lower panel) produced by the propagation of random noise through the chemoreflex loops. In this case, peripheral chemoreflex gain has been increased to 1.5 times its nominal value. Notice the much more oscillatory pattern and the appearance of "bursts" of oscillations.
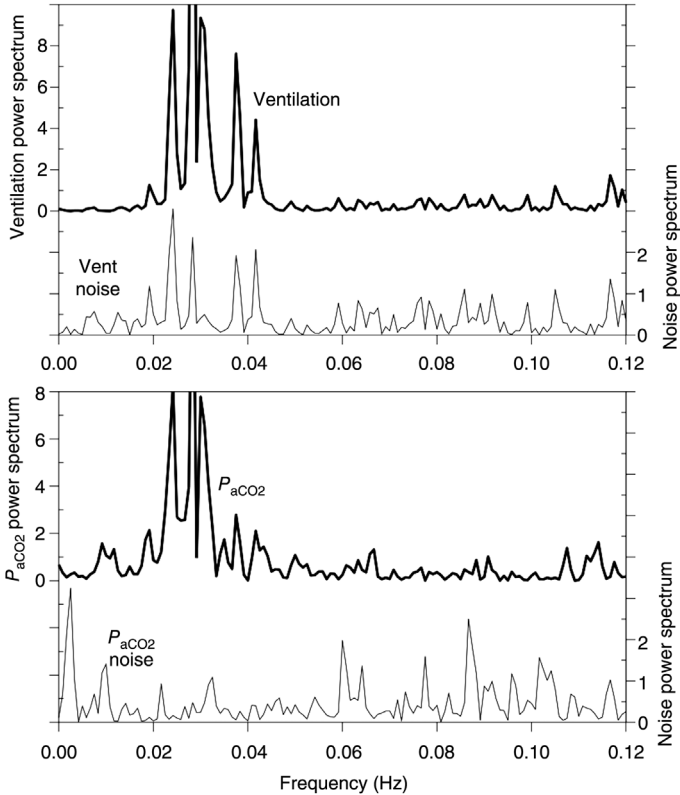
FIGURE 11.30   Power spectra of the fluctuations in ventilation (top panel) and P_{aCO2} (lower panel) when peripheral chemoreflex gain has been increased by 140% of its nominal value. Note the substantially increased power in the 0.02–0.04 Hz range (compared to the original condition in Figure 11.28), reflecting the amplification of oscillatory activity visible in Figure 11.29.

in the corresponding power spectra shown in Figure 11.30 (for comparison, see Figure 11.28). However, the spectral composition of the fluctuations in both ventilation and P_{aCO2} here is concentrated primarily in a narrower band of frequencies that range from ~0.02 to 0.04 Hz.

The example presented above demonstrates that the propagation of random influences through the feedback loops of a closed-loop control system can give rise to temporally correlated fluctuations in the system variables. These fluctuations take the form of oscillations that have frequencies consistent with the stability properties of the closed-loop system in question. "Bursts" of oscillations, which represent somewhat lower frequency phenomena, can also appear. In this way, an intrinsically stable closed-loop system can appear to be intermittently unstable as these oscillatory bursts take place.

PROBLEMS

P11.1. Develop a MATLAB or SIMULINK program to simulate the "logistic equation" given by

$$x_n = (1 - x_{n-1}) x_{n-1}$$

Use this program to generate time series for different values of the parameter α that range from 2.9 to 4 in increments of 0.01. Demonstrate that, prior to exhibiting chaotic behavior, this nonlinear dynamical system undergoes several stages of period-doubling. Display the magnitude spectrum of the fast Fourier transforms of each time series to determine how the system dynamical behavior changes in the frequency domain.

P11.2. Explore the changes in behavior of the neutrophil regulation model (`hematop.slx`) as the shape of the neutrophil production rate function changes with the parameter n increasing from 5 to 20 (see Equation. 11.2). Present phase-plane plots (i.e., dx/dt versus x) of the dynamics for each value of n employed. For your simulations, use the following values for the other parameters: $\beta = 2$, $\theta = 1$, $\gamma = 1$, and $T_d = 2$.

P11.3. Determine how the dynamics of the model of cardiovascular variability (`cvvar2.slx`) would be affected by conditions that simulate (a) vagal blockade, and (b) β-adrenergic blockade. Simulate vagal blockade by adding a low-pass filter of unit gain and time constant of 10 s to the feedback loop for cardiac period (T). To simulate β-adrenergic blockade, employ a low-pass filter of unit gain and time constant of 0.8 s. In each case, determine how the phase-space plots would change as the delay τ is increased in increments of 0.1 s from 0.5 to 2.5 s. In each case, does the model still exhibit chaotic behavior for certain ranges of τ?

P11.4. Use the Kronauer model of circadian oscillators (`circad.slx`) to simulate the effect of "jet travel" on the temperature and sleep–wake cycles. First entrain the model to the 24 h light–dark cycle until a steady state has been attained. Then, "expose" the model to a 24 h period of light (to simulate a 12 h flight in continuous daylight from Los Angeles to Tokyo) before it returns to the regular 24 h light–dark cycle. Determine how this "disturbance" affects the time of sleep onset, the duration of sleep, and the phase relation between the activity and temperature oscillations in the few days that follow.

P11.5. Using the model of obstructive sleep apnea (`osa.slx`), determine the dynamics of ventilation and sleep state following sleep onset in a patient whose symptoms have been improved by treatment. Modify the model so that upper airway conductance does not decrease to zero but only falls to

one-quarter of its waking value during sleep. Assume a range of wake-to-sleep transition times from 20 to 120 s. What are the cycle durations of the periodic breathing episodes in each case? How would oxygen administration (simulated by eliminating the peripheral chemoreflex gain) affect the ventilation-state dynamics in such a patient?

P11.6 Modify the model of chemoreflex ventilatory control (`noisycls.slx`) to determine how random fluctuations in cardiac output (Q) might lead to spontaneous variations in ventilation and P_{aCO2}. Assume a nominal value of $0.1\,\mathrm{L\,s^{-1}}$ for Q and a coefficient of variation of 3% for its random fluctuations. Determine the coefficients of variation for the resulting fluctuations in ventilation and P_{aCO2}, as well as the corresponding spectra.

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Appendix A

COMMONLY USED LAPLACE TRANSFORM PAIRS

Waveform, $x(t)$	Laplace Transform, $X(s)$
Unit impulse, $\delta(t)$	1
Unit step, $u(t)$	$1/s$
$e^{-at}u(t)$	$\dfrac{1}{s+a}$
$\dfrac{t^k}{k!}e^{-at}u(t)$	$\dfrac{1}{(s+a)^{k+1}}$
$tu(t)$	$\dfrac{1}{s^2}$
$\dfrac{t^k}{k!}u(t)$	$\dfrac{1}{s^{k+1}}$
$\sin(\omega t)u(t)$	$\dfrac{\omega}{s^2+\omega^2}$
$\cos(\omega t)u(t)$	$\dfrac{s}{s^2+\omega^2}$
$e^{-at}\sin(\omega t)u(t)$	$\dfrac{\omega}{(s+a)^2+\omega^2}$

<div align="right">(continued)</div>

(*Continued*)

Waveform, $x(t)$	Laplace Transform, $X(s)$
$e^{-at}\cos{(\omega t)}u(t)$	$\dfrac{s+a}{(s+a)^2+\omega^2}$
$t\sin{(\omega t)}u(t)$	$\dfrac{2\omega s}{(s^2+\omega^2)^2}$
$t\cos{(\omega t)}u(t)$	$\dfrac{s^2-\omega^2}{(s^2+\omega^2)^2}$

Appendix B

LIST OF MATLAB AND SIMULINK PROGRAMS

TABLE B.1   MATLAB Script Files

Chapter	Program Name	Description
4	nmr_var.m	Assigns values to parameters of msreflex.slx prior to simulation
4	tra_llm.m	Transient response analysis of linearized lung mechanics model
5	fda_llm.m	Frequency-domain analysis of linearized lung mechanics model
5	rsa_tf.m	Computes transfer functions from simulated outputs of rsa.slx
5	rsa_var.m	Assigns values to parameters of RSA model prior to simulation
6	gpmargin.m	Computes gain and phase margins of given linear system

(continued)

TABLE B.1 (*Continued*)

Chapter	Program Name	Description
6	nyq_resp.m	Nyquist stability analysis of respiratory control model
6	pupil.m	Nyquist stability analysis of pupillary reflex model
7	CT2DTsys_impresp.m	Converts continuous-time to discrete-time linear system
8	fn_gmm.m	Computes goodness of fit between glucose–insulin model and data
8	fn_llm.m	Computes criterion function for linearized lung mechanics model
8	fn_rlc.m	Predicts output of R–L–C model for given input
8	gmm_est.m	Estimates parameters of glucose–insulin minimal model
8	laguerest.m	Estimates impulse response from input-output data using Laguerre expansion technique
8	popt_llm.m	Estimation of lung mechanics model using optimization technique
8	prbs.m	Generates pseudorandom binary sequence
8	rcs_est.m	Estimation of respiratory control model parameters
8	sensanl.m	Performs sensitivity analysis for given model
8	sss_llm.m	Simulation using state-space formulation of lung mechanics model
8	sysid_ls.m	Nonparametric system identification using least squares
9	acs_CO2.m	Simulates adaptive buffering of spontaneous fluctuations in ventilation
9	TVmodel_RLS.m	Adaptive parameter estimation using recursive least squares
10	df_resp.m	Stability analysis of respiratory control model using describing function
10	stationaryVolterra2_1in1out.m	Estimate optimal kernels using basis function expansion technique

TABLE B.2   SIMULINK Model Files

Chapter	Program Name	Description
2	`respm1.slx`	Simulation of patient–ventilator system
2	`respm2.slx`	Simulation of patient–ventilator system (alternative model)
3	`msrflx.slx`	Neuromuscular stretch reflex model (steady state)
3	`respss.slx`	Simulation of steady-state respiratory control
4	`glucose.slx`	Simulation of glucose–insulin regulation (Stolwijk and Hardy model)
4	`nmreflex.slx`	Simulation of neuromuscular reflex model
5	`fdallm.slx`	Computes frequency response of linearized lung mechanics model
5	`rsa.slx`	Simulation of respiratory sinus arrhythmia (Saul model)
7	`IPFM.slx`	Simulation of integral pulse frequency modulation model
8	`gmm_sim.slx`	Simulation of minimal model of glucose–insulin dynamics (Bergman)
10	`bvpmod.slx`	Simulation of Bonhoeffer–van der Pol model
10	`linhhmod.slx`	Simplified and linearized version of Hodgkin–Huxley model
10	`poincare.slx`	Simulation of cardiac dysrhythmias using Poincaré oscillator
10	`vdpmod.slx`	Simulation of the van der Pol oscillator
11	`circad.slx`	Simulation of Kronauer circadian rhythms model
11	`cvvar1.slx`	Simulation of cardiovascular variability (stroke volume constant)
11	`cvvar2.slx`	Simulation of cardiovascular variability (stroke volume variable)
11	`hematop.slx`	Simulation of neutrophil density regulation
11	`noisycls.slx`	Simulation of spontaneous variability in control of ventilation
11	`osa.slx`	Simulation of ventilatory instability in obstructive sleep apnea

B.1   HOW TO DOWNLOAD THE MATLAB AND SIMULINK FILES

The programs accompanying this edition are compatible with MATLAB/ SIMULINK release version R2016b. The files listed in Tables B.1 and B.2 can be downloaded, along with some accompanying data files (*.mat), from the following website Wiley-IEEE Press:

www.wiley.com/go/khoo/controlsystems2e

Index

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