An Invitation to Mathematical Physics

and its History

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Contents

1 Introduction ........................................... 15

1.1 Early Science and Mathematics ......................... 17
1.1.1 Lecture 1: (Week 1) Three Streams from the Pythagorean theorem 18
1.1.2 Pythagorean Triplets ................................ 18
1.1.3 What is mathematics? ................................ 19
1.1.4 Early physics as mathematics .......................... 21
1.1.5 The birth of modern mathematics ....................... 22
1.1.6 The Three Pythagorean Streams ....................... 23

1.2 Stream 1: Number Systems (10 Lectures) ................. 25
1.2.1 Lecture 2: The Taxonomy of Numbers: \( \mathbb{P}, \mathbb{N}, \mathbb{Z}, \mathbb{Q}, \mathbb{F}, \mathbb{I}, \mathbb{R}, \mathbb{C} \) ......... 25
1.2.2 Lecture 3: The role of physics in mathematics ............ 33
1.2.3 Lecture 4: (Week 2) Prime numbers ..................... 37
1.2.4 Lecture 5: The Euclidean algorithm ..................... 40
1.2.5 Lecture 6: (Week 3) Continued fractions ................. 42
1.2.6 Lecture 7: Pythagorean triplets (Euclid’s formula) ....... 45
1.2.7 Lecture 8: Pell’s Equation ............................. 46
1.2.8 Lecture 9: (Week 4) Fibonacci sequence ................. 49
1.2.9 Lecture 10: Exam I .................................... 51

1.3 Stream 2: Algebraic Equations (12 Lectures) .......... 52
1.3.1 Lecture 11: Algebra and geometry as physics .......... 52
1.3.2 Taylor Series ............................................ 60
1.3.3 Lecture 12: (Week 5) Polynomial root classification ....... 66
1.3.4 Lecture 13: Residue expansions of rational functions (e.g., impedance) 70
1.3.5 Lecture 14: Introduction to Analytic Geometry ......... 71
1.3.6 Lecture 15: (Week 6) Gaussian Elimination .......... 77
1.3.7 Lecture 16: Transmission (ABCD) matrix composition method 79
1.3.8 Impedance matrix ........................................ 81
1.3.9 Lecture 17: Riemann Sphere ......................... 82
1.3.10 Lecture 18: (Week 7) Complex analytic mappings (Domain-coloring) 85
1.3.11 Lecture 19: Signals: Fourier and Laplace transforms .... 87
1.3.12 Lecture 20: Systems: Laplace transforms .......... 91
1.3.13 Lecture 21: (Week 8) Network (System) Postulates .... 95
1.3.14 Lecture 22: Exam II .................................... 97

1.4 Stream 3a: Scalar Calculus (11 Lectures) ............... 98
1.4.1 Lecture 23: Fundamental theorems of calculus .......... 100
1.4.2 Lecture 24: (Week 9) Cauchy-Riemann conditions ....... 101
1.4.3 Lecture 25: Complex Analytic functions and Brune Impedance .......... 103
## CONTENTS

<table>
<thead>
<tr>
<th>Section</th>
<th>Lecture</th>
<th>Title</th>
<th>Page</th>
</tr>
</thead>
<tbody>
<tr>
<td>3.3.2</td>
<td>Lec 16</td>
<td>Matrix composition: Bilinear and ABCD</td>
<td>191</td>
</tr>
<tr>
<td>3.3.3</td>
<td>Lec 16a</td>
<td>Transmission line example</td>
<td>193</td>
</tr>
<tr>
<td>3.3.4</td>
<td>Lec 17</td>
<td>Introduction to the Branch cut and Riemann sheets</td>
<td>196</td>
</tr>
<tr>
<td>3.4</td>
<td>Week 7</td>
<td></td>
<td>196</td>
</tr>
<tr>
<td>3.4.1</td>
<td>Lec 18</td>
<td>Complex analytic mappings (domain coloring)</td>
<td>196</td>
</tr>
<tr>
<td>3.4.2</td>
<td>Lec 20</td>
<td>Laplace transform and the Riemann $\zeta(s)$ (Zeta) function (I)</td>
<td>198</td>
</tr>
<tr>
<td>3.4.3</td>
<td>Lec 20a</td>
<td>Filter classification</td>
<td>198</td>
</tr>
<tr>
<td>3.4.4</td>
<td>Lec 20b</td>
<td>Laplace Transform of the Riemann $\zeta(s)$ Zeta function (II)</td>
<td>204</td>
</tr>
<tr>
<td>3.5</td>
<td>Week 8</td>
<td></td>
<td>205</td>
</tr>
<tr>
<td>3.5.1</td>
<td>Lec 21</td>
<td>The ten postulates of System of algebraic Networks</td>
<td>205</td>
</tr>
<tr>
<td>3.5.2</td>
<td>Lec 22</td>
<td>Exam II (Evening)</td>
<td>211</td>
</tr>
<tr>
<td>4</td>
<td>Week 8-continued</td>
<td></td>
<td>213</td>
</tr>
<tr>
<td>4.1.1</td>
<td>Lec 23</td>
<td>Integration in the complex plane</td>
<td>213</td>
</tr>
<tr>
<td>4.2</td>
<td>Week 9</td>
<td></td>
<td>214</td>
</tr>
<tr>
<td>4.2.1</td>
<td>Lec 24</td>
<td></td>
<td>214</td>
</tr>
<tr>
<td>4.2.2</td>
<td>Lec 25a</td>
<td>Complex analytic functions and Brune impedance (I)</td>
<td>214</td>
</tr>
<tr>
<td>4.2.3</td>
<td>Lec 25b</td>
<td>Complex analytic functions and Brune impedance (II)</td>
<td>214</td>
</tr>
<tr>
<td>4.2.4</td>
<td>Lec 26</td>
<td>Summary review</td>
<td>214</td>
</tr>
<tr>
<td>4.3</td>
<td>Week 10</td>
<td></td>
<td>214</td>
</tr>
<tr>
<td>4.3.1</td>
<td>Lec 27</td>
<td>Three Cauchy Theorems: I</td>
<td>214</td>
</tr>
<tr>
<td>4.3.2</td>
<td>Lec 28</td>
<td>Three Cauchy Theorems: II, III</td>
<td>214</td>
</tr>
<tr>
<td>4.3.3</td>
<td>Lec 29</td>
<td>Inverse Laplace Transform I</td>
<td>214</td>
</tr>
<tr>
<td>4.4</td>
<td>Week 11</td>
<td></td>
<td>214</td>
</tr>
<tr>
<td>4.4.1</td>
<td>Lec 30</td>
<td>Inverse Laplace Transforms II</td>
<td>214</td>
</tr>
<tr>
<td>4.4.2</td>
<td>Lec 31</td>
<td>Laplace Transforms properties</td>
<td>214</td>
</tr>
<tr>
<td>4.4.3</td>
<td>Lec 32</td>
<td>Review for Exam III</td>
<td>214</td>
</tr>
<tr>
<td>5</td>
<td>Week 13</td>
<td></td>
<td>215</td>
</tr>
<tr>
<td>5.1.1</td>
<td>Lec 33</td>
<td>Grad, Div, Curl and Laplacian</td>
<td>215</td>
</tr>
<tr>
<td>5.1.2</td>
<td>Lec 34</td>
<td>Scalar wave equation (I)</td>
<td>215</td>
</tr>
<tr>
<td>5.1.3</td>
<td>Lec 35</td>
<td>Scalar wave equation (II)</td>
<td>215</td>
</tr>
<tr>
<td>5.1.4</td>
<td></td>
<td>Generalized impedance</td>
<td>218</td>
</tr>
<tr>
<td>5.1.5</td>
<td>Lec 37</td>
<td>Webster Horn Equation</td>
<td>218</td>
</tr>
<tr>
<td>4.1.2</td>
<td></td>
<td>Laplacian operator in $N$ dimensions</td>
<td>220</td>
</tr>
<tr>
<td>5.2</td>
<td>Week 14</td>
<td></td>
<td>226</td>
</tr>
<tr>
<td>5.2.1</td>
<td></td>
<td>Week 14</td>
<td>226</td>
</tr>
<tr>
<td>5.2.2</td>
<td>Lec 39</td>
<td>Geometry of gradient, divergence and curl vector operators</td>
<td>226</td>
</tr>
<tr>
<td>5.2.3</td>
<td>Lec 40</td>
<td></td>
<td>226</td>
</tr>
<tr>
<td>5.2.4</td>
<td>Lec 41</td>
<td>The Quasi-static approximation and more applications</td>
<td>226</td>
</tr>
<tr>
<td>5.3</td>
<td>Week 15</td>
<td></td>
<td>227</td>
</tr>
<tr>
<td>5.3.1</td>
<td>Lec 42</td>
<td>The Quasi-static approximation and more applications</td>
<td>227</td>
</tr>
<tr>
<td>5.3.2</td>
<td>Lec 43</td>
<td>Summary review of Fund Thms of Mathematics</td>
<td>228</td>
</tr>
</tbody>
</table>
A Notation 229
  A.1 Number systems .................................................. 229
  A.1.1 Units .............................................................. 229
  A.1.2 Symbols and functions ........................................ 229
  A.1.3 Special symbols common to mathematical: ................. 229
  A.1.4 Greek letters .................................................... 230
  A.1.5 Table of double-bold number notation ..................... 231
  A.2 Vectors ............................................................ 232
  A.2.1 Vectors in \( \mathbb{R}^3 \) ......................................... 233
  A.3 Matrices ............................................................ 234
  A.4 Gaussian elimination ............................................ 235
  A.5 Periodic functions ................................................ 237
  A.6 Differential equations vs. Polynomials ...................... 237

B Linear algebra of 2x2 matrices 239
  B.1 Notation ............................................................ 239
  B.1.1 Gaussian elimination exercises ............................. 241
  B.2 Inverse of the 2x2 matrix ...................................... 242
  B.2.1 Derivation of the inverse of a 2x2 matrix ................ 243

C Symbolic analysis of \( AE = EA \) 245
  C.1 General case ....................................................... 245
  C.2 Symmetry .......................................................... 245
  C.2.1 Reciprocal systems .......................................... 245
  C.2.2 Hermitian ...................................................... 245
  C.2.3 Reversible systems .......................................... 246
  C.2.4 Transmission matrices and symmetry ...................... 246

D Analysis of Pell equation (N=2) 247
  D.1 Pell equation eigenvalue-eigenvector analysis .............. 247
  D.1.1 Pell equation for \( N = 3 \) ................................. 250

E Stillwell’s Intersection vs. composition 253
Abstract

An understanding of physics requires knowledge of mathematics. The contrary is not true. By definition, pure mathematics contains no physics. Yet historically, mathematics has a rich history filled with physical applications. Mathematics was developed by people with intent of making things work. In my view, as an engineer, I see these creators of early mathematics, as budding engineers. This book is an attempt to tell their story, of the development of mathematical physics, as viewed by an engineer.

This is not strictly a book on mathematics, nor is it a book on physics, rather it is about the marriage of math and physics, presented roughly in the chronological order that they were developed. This work was largely done by those who would today be viewed as Engineers. Hopefully this book contains the information that a well informed engineer needs to know, to do a great job.

The book is broken down into three topics, called streams, presented as five chapters: 1) Introduction, 2) Number systems, 3) Algebra Equations, 4) Scalar Calculus, and 5) Vector Calculus. The material is delivered as 43 “Lectures” spread out over a semester of 15 weeks, three lectures per week, with a 3 lecture time-out for administrative duties. Eleven problem sets are provided for each week’s assignment. These problems are written out in \LaTeX{}, with built in solutions, that may be expressed by un-commenting one line. Once the home-works are turned in, each student is given the solution. The students rated these Assignments as the most important part of the course. There is a built-in interplay between these assignments and the lectures. On many occasions I solved the homework in class, as motivation for coming to class.

There were four exams, one at the end of each of the three sections, and a final. The first exams is in class, two others and the final were evening exams. Each of the exams and assignments are \LaTeX{} files, with solutions encoded with a one line software switch. When the exam was returned by the student, the full solution was provided, while the exam was fresh in the students mind, providing a teaching moment. The Exams are largely based on the Assignments. It is my philosophy that, in principle, the students could see the exam in advance of taking it. In a sense they do, since each exam was based on the homework.

Author’s Personal Statement

An expert is someone who has made all possible mistakes in a small field. I don’t know if I would be called and expert, but I certainly have made my share of mistakes. I openly state that “I love making mistakes, because I learn so much from them.” One might call that the “expert’s corollary.”

This book has been written out of both my love for the topic of mathematical physics, and a frustration for wanting to share many key concepts, and many new ideas on these basic concepts. Over the years I have developed a certain physical sense of math, along with a related mathematical sense of physics. While doing my research,\(^1\) I have come across what I feel are certain conceptual

\(^1\)https://auditorymodels.org/index.php/Main/Publications
holes that need filling, and sense many deep relationships between math and physics, that remain unidentified. While what we presently teach is not wrong, it is missing these relationships. What is lacking is an intuition for how math “works.” A good scientist “listens” to their data. In the same way we need to start listening to the language of mathematics. We need to let mathematics guide us toward our engineering goals.

It is my suspicion that over the centuries many others have had similar insights, and like me, have been unable to convey this slight misdirection. I hope these views will be useful to young open minds.

As summarized in Fig. 1, this marriage of math, physics and engineering (STEM) will help us make progress in understanding the physical world. We must turn to mathematics and physics when trying to understand the universe. I have arrived in my views following a lifelong attempt to understand human communication, i.e., the perception and decoding of human speech sounds. This research arose from my 32 years at Bell Labs in the Acoustics Research Department. There such lifelong pursuits were not only possible, they were openly encouraged. The idea was that if you were successful at something, take it as far as you can. But on the other side, don’t do something well, that’s not worth doing. People got fired for the latter. I should have left for University after a mere 20 years, but the job was just too cushy.

In this text it is my goal to clarify conceptual errors while telling the story of physics and mathematics. My views have been inspired by classic works, as documented in the bibliography. This book was inspired by my reading of Stillwell (2002), through Chapter 21 (Fig. 2) (p. 11). Somewhere in Chapter 22 I switched to the third edition (Stillwell, 2010), at which point I realized

---

2 I started around December 1970, fresh out of Graduate school, and retired in December 2002.
I had much more to master. It became clear that by teaching this material to first year engineers I could absorb the advanced material at a reasonable pace. This book soon followed.

**Back Cover Summary**

This is foremost a math book, but not the typical math book. *First*, this book is for the engineering minded, for those who need to understand math to do engineering, to learn how things work. In that sense the book is more about physics and engineering than mathematics. Math skills are essential for making progress in building things, be it pyramids or computers, as clearly shown by the many great civilizations of the Chinese, Egyptians, Arabs (people of Mesopotamia), and Greeks and Romans.

*Second*, this is a book about the math that developed to explain physics, to allow people to engineer complex things. To sail around the world one needs to know how to navigate. This requires a model of the planets and stars. You can only know where you are on earth once you understand where earth is relative to the sun, planets, Milky Way way and the distant stars. The answer to such a deep questions will depend on who you ask. The utility and accuracy of that answer depends critically on the depth of understanding of how the worlds and heavens work. Who is qualified to answer such question? It is best answered by those who study mathematics applied to the physical world.

The English astronomer Edmond Halley (1656–1742) asked Newton (1643–1727) for the equation that describes the orbit of the planets. Halley was obviously interested in comets. Newton immediately answered “an ellipse.” It is said that Halley was stunned by the response (Stillwell, 2010, p. 176), as this was what had been experimentally observed by Kepler (c1619), and thus he knew Newton must have some deeper insight.

When Halley asked Newton to explain how he knew, Newton responded “I calculated it.” But when challenged to show the calculation, Newton was unable to reproduce it. This open challenge eventually led to Newton’s grand treatise, *Philosophiae Naturalis Principia Mathematica* (July 5, 1687). It had a humble beginning, as a letter to Halley, explaining how to calculate the orbits of the planets. To do this Newton needed mathematics, a tool he had mastered. It is widely accepted that Isaac Newton and Gottfried Leibniz invented calculus. But the early record shows that perhaps Bhāskara II (1114–1185 AD) had mastered this art well before Newton.³

*Third*, the main goal of this book is to teach motivated engineers mathematics, in a way that it can be understood, mastered and remembered. How can this impossible goal be achieved? The answered is to fill in the gaps with “who did what, and when.” An historical story may be easily mastered.

To be an expert in a field, one must know its history. This includes who the people were, what they did, and the credibility of their story. Do you believe the Pope or Galileo on the roles of the sun and the earth? The observables provided by science are clearly on Galileo’s side. Who were those first engineers? They are names we all know: Archimedes, Pythagoras, Leonardo da Vinci, Galileo, Newton, .... All of these individuals had mastered mathematics. This book teaches the tools taught to every engineer. Do not memorize complex formulas, rather make the equations “obvious” by learning the simplicity of the underlying concepts.

*Forth*, when most educators look at the table of contents their immediate reaction is “You have one lecture on a topic that we spend a week on (in our math/physics/engineering) class.” And “You have too much material crammed into one semester.” The first sentence is correct, the second is not.

³[http://www-history.mcs.st-and.ac.uk/Projects/Pearce/Chapters/Ch8_5.html](http://www-history.mcs.st-and.ac.uk/Projects/Pearce/Chapters/Ch8_5.html)
Tracking the students who have taken the course, and looking at their grades, along with personal interviews, demonstrates this is simply not true [http://www.istem.illinois.edu/news/jont.allen.html](http://www.istem.illinois.edu/news/jont.allen.html).

**Credits**

Besides thanking my parents, I would like to credit John Stillwell for his constructive, historical summary of mathematics. My close friend and colleague Steve Levinson somehow drew me into this project, without my even knowing it. My brilliant graduate student Sarah Robinson was constantly at my side, grading home-works and exams, and tutoring the students. Without her, I would not have survived the first semester the material was taught. Her proof-reading skills are amazing. Thank you Sarah for your infinite help. Finally I would like to thank John D’Angelo for putting up with my many silly questions. When it comes to the heavy hitting, John was always there to provide a brilliant explanation that I could easily translate into Engineer’eese (Matheering?) (i.e., Engineer language).

To write this book I had to master the language of mathematics (John’s language). I had already mastered the language of engineering, and a good part of physics. One of my secondary goals is to bring down this scientific Tower of Babble by unifying the terminology and removing the jargon.

Finally I would like to thank my wife (Sheau Feng Jeng aka Patricia Allen) for her unbelievable support and love. She delivered constant peace of mind, without which this project could never have been started, much less finish.

There are many others who played important roles, but they must remain anonymous, out of my fear of offending someone I forgot to mention. Some I must mention include Paul Fahey and Chris Shera.

–Jont Allen, Mahomet IL, Dec. 24, 2015 (Jan 1, 2018)
# Mathematics and its History *(Stillwell, 2002)*

## Contents

1. The Theorem of Pythagoras ........................................... 1
2. Greek Geometry .......................................................... 17
3. Greek Number Theory .................................................. 37
4. Infinity in Greek Mathematics ......................................... 53
5. Number Theory in Asia ................................................... 68
6. Polynomial Equations ................................................... 87
7. Analytic Geometry .......................................................... 109
8. Projective Geometry ....................................................... 126
9. Calculus ......................................................................... 157
10. Infinite Series ............................................................... 181
11. The Number Theory Revival ............................................ 203
12. Elliptic Functions ............................................................ 224
13. Mechanics ....................................................................... 243
14. Complex Numbers in Algebra .......................................... 275
15. Complex Numbers and Curves .......................................... 295
16. Complex Numbers and Functions ...................................... 313
17. Differential Geometry ...................................................... 335
18. Non-Euclidean Geometry ................................................ 359
19. Group Theory .................................................................. 382
20. Hypercomplex Numbers .................................................. 415
21. Algebraic Number Theory ................................................. 438
22. Topology ......................................................................... 467
23. Simple Groups ............................................................... 495
24. Sets, Logic, and Computation .......................................... 525
25. Combinatorics ............................................................... 553

Bibliography ........................................................................ 589

Index ................................................................................. 629

*Figure 2: Table of contents of *Stillwell (2002)*
Preface

It is widely acknowledged that interdisciplinary science is the backbone of modern scientific investigation. This is embodied in the STEM (Science, Technology, Engineering, and Mathematics) programs. Contemporary research is about connecting different areas of knowledge, thus it requires an understanding of cross-disciplines. However, while STEM is being taught, interdisciplinary science is not, due to its inherent complexity and breadth. Furthermore there are few people to teach it. Mathematics, Engineering and Physics (MEP) are at the core of such studies.

STEM vs. MEP

Mathematics is about rigor. Mathematicians specifically attend to the definitions of increasingly general concepts. Thus mathematics advances slowly, as these complex definitions must be collectively agreed upon. Since it embraces rigor, mathematics shuns controversy, the opposite of uncertainty. Physics explores the fringes of uncertainty. Physicists love controversy. Engineering addresses the advancement the technology. Engineers, much like mathematicians, are uncomfortable with uncertainty, but are trained to deal with it.

To create such an interdisciplinary STEM program, a unified MEP curriculum is needed. In my view this unification could (should) take place based on a core mathematical training, from a historical perspective, starting with Euclid or before (i.e., Chinese mathematics), up to modern information theory and logic. As a bare minimum, the fundamental theorems of mathematics (arithmetic, algebra, calculus, vector calculus, etc.) need to be appreciated by every MEP student. The core of this curriculum is outlined in Table 1.1 (p. 24).

If first year students are taught a common MEP methodology and vocabulary, presented in terms of the history of mathematics, they will be equipped to

1. Exercise interdisciplinary science (STEM)
2. Communicate with other MEP trained (STEM) students and professors.

The goal is a comprehensive understanding of the fundamental concepts of mathematics, defined as those required for engineering. We assume that students with this deep understanding will end up being in the top 0.1% of Engineering. Time will tell if and when this assumption is proved to be correct.

The key tool is methodology. The traditional approach is a five to eight course sequence: Calc I, II, III, Linear Algebra IV, DiffEq V, Real analysis VI, Complex analysis VII and Number theory VIII, over a time frame of three or more years (six semesters). This was the way I learned math. Following this regime I felt I had not fully mastered the material, so I started over. I now consider myself to be self-taught. We need a more effective teaching method. I am not suggesting we replace the standard 6 semester math curriculum, rather I am suggesting replacing Calc I, II with a mathematical physics course, based on the historical thread, designed for those students who have
demonstrated advanced ability. One needs more than a high school education to succeed in college engineering courses. One thing missing in our education system right now is teaching critical thinking.

By teaching mathematics in the context of history, the student can fully appreciate the underlying principles. Including the mathematical history provides a uniform terminology for understanding the fundamentals of mathematics. The present teaching method, using abstract proofs, with no (or few) figures or physical principles, by design, removes intuition and the motivation that was available to the creators of these early theories. This present six semester approach does not function for many students, leaving them with poor, or even no intuition.

Postscript Dec 5, 2017  How to cram five semester of math into one semester, and leave the students with something they can remember? Here are some examples:

Maxwell’s equations are a fundamental challenging topic, presented in one lecture (Section 1.5.12). Here is how it works:

1. The development starts with Sections 1.4.1-1.4.9, which develops complex integration, and ends with the Laplace transform

2. Section 1.5.1 develops analytic field theory, Sect. 1.5.2 introduces $\nabla(), \nabla \cdot (), \nabla \times (), \nabla^2$ based on the scalar $A \cdot B$ and vector products $A \times B$ of two vectors.

3. In Sect. 1.5.11 the second order operators are introduced, and given a physical meaning (based on ideas from fluid motion), but most important they are given memorable names (DoC, CoG, Dog, God, and CoC). Using this technique the student and both understand, and remember the relationships between these various second order vector calculus operations.

4. This has carefully set the stage for ME (Sect. 1.5.12), which starts with the names and units of the electrical and magnetic field intensity (strength, or force seen by charge) $E, H$, and electric and magnetic flux (flow) $D, B$.

5. The ratio of the force over the flow is an capacitance (e.g., compliance) per unit area ($\epsilon_o [F/m^2]$) and the inductance (e.g., mass) per unit area ($\mu_o [H/m^2]$) as summarized on page 154.

6. In the Lectures, exercises and assignments, the meanings of these equations are further explored, in integral form. After Sect. 1.5.2-1.5.12, the students are fully conversent with MEs. The proof are the final exam grade distributions.

Postscript Dec 15, 2017  This book is coming to completion, and I’m appreciating the Feynman lectures as I bring this to a halt. We all know (I hope you know) that Feynman had a special lecture style, that was both entertaining, and informative. He died in 1988, at the age of 70. I am 75, and so far in good health. Not only was he a fantastic communicator, he had depth of understanding, and was not afraid to question the present understanding of physics. He was always on a quest. Let us all be on a constant quest. The thought that we have already figured out the universe, is absurd. We have a lot to learn, and we need to understand where we came from before we can know where we are going.

– Jont Allen
Chapter 1

Introduction

Much of early Chinese mathematics centered around the love of art and music, due to the sensations of light and sound. Exploring our physiological senses requires a scientific understanding of vision and hearing, as first explored by Newton (1687) and Helmholtz (1863a)\(^1\) (Stillwell, 2010, p. 261). Our senses of color and musical pitch are determined by the frequencies of light and sound. The Chinese and Pythagoreans are well known for their early contributions to music theory.

Pythagoras strongly believed that “all is integer,” meaning that every number, and every mathematical and physical concept, could be explained by integral relationships. It is likely that this belief was based on Chinese mathematics from thousands of years earlier. It is also known that his ideas about the importance of integers were based on what was known about music theory. For example it was known that the relationships between the musical notes (pitches) obey natural integral relationships.

Two applications of number theory, relevant today, are:

- Public-private key encryption, which requires computationally intensive prime factoring of large integers
- IEEE Floating point\(^2\)

As acknowledged by Stillwell (2010, p. 16), the Pythagorean view is relevant today

> With the digital computer, digital audio, and digital video coding everything, at least approximately into sequences of whole numbers, we are closer than ever to a world in which “all is number.”

Mersenne (1588-1647) contributed to our understanding of the relationship between the wavelength and the length of musical instruments. These results were extended by Galileo’s father, and then by Galileo himself (1564-1642). Many of these musical contributions resulted in new mathematics, such as the discovery of the wave equation by Newton (c1687), followed by its one-dimensional general solution by d’Alembert (c1747).

By the time of Newton there was a basic understanding that sound and light traveled at very different speeds (thus why not the velocities of different falling weights?).

Ole Rømer first demonstrated in 1676 that light travels at a finite speed (as opposed to instantaneously) by studying the apparent motion of Jupiter’s moon Io. In 1865,

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\(^1\)https://en.wikipedia.org/wiki/Acoustics
\(^2\)https://en.wikipedia.org/wiki/IEEE_floating_point\#Formats
James Clerk Maxwell proposed that light was an electromagnetic wave, and therefore traveled at the speed \( c \) appearing in his theory of electromagnetism.\(^3\)

In 1589 Galileo famously conceptualized an experiment where he suggested dropping two different weights from the Leaning Tower of Pisa, and showed that they must take the same time to hit the ground. Conceptually this is an important experiment, driven by a mathematical argument in which he considered the two weights to be connected by an elastic cord (i.e., gravity). This resulted in the concept of conservation of energy, one of the cornerstones of physical theory since that time.

While Newton may be best known for his studies on light, he was the first to predict the speed of sound. However his theory was in error\(^4\) by \( \sqrt{c_p/c_v} = \sqrt{1.4} = 1.183 \). This famous error would not be resolved for over two hundred years, awaiting the formulation of thermodynamics and the equi-partition theorem, by Maxwell and Boltzmann, and others. What was needed was the concept of constant-heat, or an adiabatic process. For audio frequencies (0.02-20 [kHz]), the small temperature gradients cannot diffuse the distance of a wavelength in one cycle (Pierce, 1981; Boyer and Merzbach, 2011), “trapping” the heat energy in the wave. There were several other physical enigmas, such as the observation that sound disappears in a vacuum and that a vacuum cannot draw water up a column by more than 34 feet.

There are other outstanding examples where physiology impacted mathematics. Leonardo da Vinci is well known for his studies of the human body. Helmholtz’s theories of music and the perception of sound are excellent examples of under-appreciated fundamental mathematical contributions (Helmholtz, 1863a). Lord Kelvin (aka William Thompson),\(^5\) was one of the first true engineer-scientists, equally acknowledged as a mathematical physicist, well known for his interdisciplinary research, and knighted by Queen Victoria in 1866. Lord Kelvin coined the term thermodynamics, a science more fully developed by Maxwell (the same Maxwell of electrodynamics).  

\(^3\)https://en.wikipedia.org/wiki/Speed_of_light

\(^4\)The square root of the ratio of the specific heat capacity at constant pressure \( c_p \) to that at constant volume \( c_v \).  

\(^5\)Lord Kelvin was one of half a dozen interdisciplinary mathematical physicists, all working about the same time, that made a fundamental change in our scientific understanding. Others include Helmholtz, Stokes, Green, Heaviside, Rayleigh and Maxwell.
Thus the interdisciplinary nature of science has played many key roles in the development of thermodynamics. Lord Rayleigh’s (William Strutt) book on the theory of sound (Rayleigh, 1896) is a classic text, read even today by those who study acoustics.

It seems that we have retracted from this venerable interdisciplinary view of science, by splitting the disciplines into into smaller parts, whenever we perceived a natural educational boundary. Reforging these natural connections into the curriculum, is essential for the proper training of STEM students.

WEEK 1

1.1 Early Science and Mathematics

The first 5,000 years is not well documented, but the basic record is clear, as outlined in Fig. 1.2. Thanks to Euclid and later Diophantus (c250 CE), we have some vague understanding of Chinese thought. For example, Euclid’s formula (Eq. 1.9, p. 46; Sec. 2.2.2, Fig. 2.3, p. 171) provides a method for computing Pythagorean triplets, a formula known long before Euclid (Stillwell, 2010, pp. 4-9).

Chinese Bells and stringed musical instruments were exquisitely developed with tonal quality, as documented by ancient physical artifacts (Fletcher and Rossing, 2008). In fact this development was so rich that one must question why the Chinese failed to initiate the industrial revolution. Specifically, why did Europe eventually dominate with its innovation when it was the Chinese who did the extensive early invention?

According to Lin (1995) this is known as the Needham question:

“Why did modern science, the mathematization of hypotheses about Nature, with all its implications for advanced technology, take its meteoric rise only in the West at the time of Galileo[,] but] had not developed in Chinese civilization or Indian civilization?"

Needham cites the many developments in China:

“Gunpowder, the magnetic compass, and paper and printing, which Francis Bacon considered as the three most important inventions facilitating the West’s transformation from the Dark Ages to the modern world, were invented in China.” (Lin, 1995)

“Needham’s works attribute significant weight to the impact of Confucianism and Taoism on the pace of Chinese scientific discovery, and emphasizes what it describes as the ‘diffusionist’ approach of Chinese science as opposed to a perceived independent inventiveness in the western world. Needham held that the notion that the Chinese script had inhibited scientific thought was ‘grossly overrated’ ” (Grosswiler, 2004).

Lin was focused on military applications, missing the importance of non-military contributions. A large fraction of mathematics was developed to better understand the solar system, acoustics,
musical instruments and the theory of sound and light. Eventually the universe became a popular

Regarding the “Needham question,” I suspect the resolution is now clear. In the end, China

1.1.1 Lec 1 The Pythagorean theorem

While early Asian mathematics is not fully documented, it clearly defined the course for at least

First recorded mathematics was that of the Chinese (5000-1200 BCE) and the Egyptians (3,300 BCE). Some of the best early record were left by the people of Mesopotamia (Iraq, 1800 BCE). Thanks to Euclid’s Elements (c323 BCE) we have an historical record, tracing the progress in geometry, as defined by the Pythagorean theorem for any right triangle

\[ c^2 = a^2 + b^2, \]

having sides of lengths \((a, b, c)\) that are either positive real numbers, or more interesting, integers, such that \(c > [a, b]\) and \(a + b > c\). Integer solutions were likely found by trial and error rather than by Euclid’s formula.

If \(a, b, c\) are lengths, then \(a^2, b^2, c^2\) are each the area of a square. Equation 1.1 says that the area \(a^2\) plus the area \(b^2\) equals the area \(c^2\). Today a simple way to prove this is to compute the magnitude of the complex number \(c = a + bj\), which forces the right angle

\[ |c|^2 = (a + bj)(a - bj) = a^2 + b^2. \] (1.2)

However, complex arithmetic was not an option for the Greek mathematicians, since complex numbers and algebra had yet to be discovered.

Almost 700 years after Euclid’s Elements, the Library of Alexandria was destroyed by fire (391 CE), taking with it much of the accumulated Greek knowledge. Thus one of the best technical records is Euclid’s Elements, along with some sparse mathematics due to Archimedes (c300 BCE) on geometrical series, computing the volume of a sphere, and the area of the parabola, and elementary hydrostatics. Additionally, a copy of a book by Diophantus Arithmetic was discovered by Bombelli (c1572) in the Vatican library (Stillwell, 2010, p. 51). This book became the inspirational resource for Galileo, Descartes, Fermat and Newton.

1.1.2 Pythagorean Triplets

Well before Pythagoras, the Babylonians had tables of Pythagorean triplets (PTs), integers \([a, b, c]\) that obey Eq. 1.1. An example is \([3, 4, 5]\). A stone tablet (Plimpton-322) dating back to 1800 BCE (Fig. 1.9, p. 47) was found with integers for \([a, c]\). Given such sets of two numbers, which determined a third positive integer \(b\) such that \(b = \sqrt{c^2 - a^2}\), this table is more than convincing that the Babylonians were well aware of PTs, but less convincing that they had access to Euclid’s formula (Eq. 1.9 p. 46).

It seems likely that Euclid’s Elements was largely the source of the fruitful 6th century era due to the Greek mathematician Diophantus (Fig. 1.2), who developed the concept of discrete mathematics, now known as Diophantine analysis.

The work of Diophantus was followed by a rich mathematical era, with the discovery of 1) negative numbers, quadratic equation (Brahmagupta, 7th CE), 2) algebra (al-Khwārizmī, 9th CE),
1.1. EARLY SCIENCE AND MATHEMATICS

Chronological history pre 16th century

200th BCE Chinese (Primes; quadratic equation; Euclidean algorithm (GCD))
180th BCE Babylonia (Mesopotamia/Iraq) (quadratic equation)
6th BCE Pythagoras (Thales) and the Pythagorean “tribe”
4th BCE Euclid (quadratic equation); Archimedes
3rd CE Diophantus c250CE;
4th CE Alexandria Library destroyed 391CE;
6th CE Brahmagupta (negative numbers; quadratic equation) 598-670CE
10th CE al-Khwārizmī (algebra) 830CE; Hasan Ibn al-Haytham (Alhazen) 965-1040CE
12th CE Bhaskara (calculus) 1114-1183; Marco Polo 1254-1324
15th Leonardo da Vinci 1452-1519; Michelangelo 1475-1564; Copernicus 1473-1543;
16th Tartaglia (cubic eqs); Bombelli 1526-1572; Galileo Galilei 1564-1642

Time Line

![Mathematical time-line between 1500 BCE and 1650 CE. The western renaissance is considered to have occurred between the 14-17 centuries. However the Asian (Chinese) “renaissance” was likely well before the 1 century (i.e., 1500 BCE?). There is significant evidence that a Chinese ‘treasure ship’ visited Italy in 1434, initiating the Italian renaissance (Menzies, 2008). This was not the first encounter between the Italians and the Chinese, as documented in ‘The travels of Marco Polo’ (c1300 CE).](fig:TimeLineBCE)

and 3) complex arithmetic (Bombelli, 15th CE). This period overlapped with the European middle (aka, dark) ages. Presumably European intellectuals did not stop thinking during these many centuries, but what happened in Europe is presently unclear given the available records.9

1.1.3 What is mathematics?

Mathematics is a language, not so different from other languages. Today’s mathematics is a written language with an emphasis on symbols and glyphs, biased toward Greek letters. The etymology of these symbols would be interesting to study. Each symbol is dynamically assigned a meaning, appropriate for the problem being described. These symbols are then assembled to make sentences. It is similar to Chinese in that the spoken and written version are different across dialects. In fact, like Chinese, the sentences may be read out loud in the language (dialect) of your choice, while the mathematical sentence (like Chinese) is universal.

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9It might be interesting to search the archives of the monasteries, where the records were kept, to figure out what happened during this strange time.
CHAPTER 1. INTRODUCTION

Figure 1.3: Above: Jakob (1655-1705) and Johann (1667-1748) Bernoulli; Below: Leonhard Euler (1707) and Jean le Rond d’Alembert (1717-1783). Euler was blind in his right eye, hence the left portrait view. The figure numbers are from Stillwell (2010).

Figure 10.4: Leonhard Euler

Figure 13.10: Portrait of Jakob Bernoulli by Nicholas Bernoulli

Figure 13.11: Johann Bernoulli
Math (i.e., the syntax) is a language: It seems strange when people complain that they “can’t learn math,”
but they claim to be good at languages. Math is a language, with the symbols taken from various languages, with a bias toward Greek, obviously due to the popularity of Euclid’s Elements. Learning languages is an advanced social skill. However the social outcomes are very different between learning a romance language and math. Learning a new language is fun because it opens doors to other cultures.

Math is different due to the rigor of the rules of the language, along with the way it is taught (e.g., not as a language). A third difference between math and the romance languages is that math evolved from physics, with important technical applications. This was the concept behind the Pythagorean school, a band of followers called the Pythagoreans.

A further problem is that pre-high-school students confuse arithmetic with math. The two topics are very different, and students need to understand this. One does not need to be good at arithmetic to be good at math (but it doesn’t hurt).

There are many rules that must be mastered. These rules are defined by algebra. For example the sentence \( a = b \) means that the number \( a \) has the same value as the number \( b \). The sentence is spoken as “\( a \) equals \( b \).” The numbers are nouns and the equal sign says they are equivalent, playing the role of a verb, or action symbol. Following the rules of algebra, this sentence may be rewritten as \( a - b = 0 \). Here the symbols for minus and equal indicate two types of actions.

Sentences can become arbitrarily complex, such as the definition of the integral of a function, or a differential equation. But in each case, the mathematical sentence is written down, may be read out loud, has a well defined meaning, and may be manipulated into equivalent forms following the rules of algebra and calculus. This language of mathematics is powerful, with deep consequences, known as theorems.

The writer of an equation should always translate (explicitly summarize the meaning of the expression), so the reader will not miss the main point. This is simply a matter of clear writing.

Language may be thought of as mathematics (turning this idea on its head). To properly write correct English it is necessary to understand the construction of the sentence. It is important to identify the subject, verb, object, and various types of modifying phrases. If you wish to read about this, look up the distinction between the words that and which, which make a nice example of this concept. Most of us work directly with what we think “sounds right,” but if you’re learning English as a second language, it is very helpful to understand the rules, which are arguably easier to master than the foreign phones (i.e., speech sounds).

1.1.4 Early physics as mathematics

The role of mathematics is to summarize algorithms (i.e., sets of rules). As outlined in Fig. 1.2, it was clear to Pythagoras (and many others before him), that there was an important relationship between mathematics and the physical world. Pythagoras may have been one of the first to capitalize on this relationship, using science and mathematics to design and make things. This was the beginnings of technology as we know it, coming from the relationship between physics and math, impacting map making, tools, implements of war (the wheel, gunpowder), art (music), sound, water transport, sanitation, secure communication, food, . . . , etc.

Why is Eq. 1.1 called a theorem, and what exactly needs to be proved? We do not need to prove that \( (a, b, c) \) obey this relationship, since this is a condition that is observed. We do not need

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10“It looks like Greek to me.”
11https://en.oxforddictionaries.com/usage/that-or-which
12It seems likely that the Chinese and Egyptians also did this, but that is more difficult to document.
to prove that $a^2$ is the area of a square, as this is the definition of the area of a square. What needs to be proved is that this relation only holds if the angle between the two shorter sides is $90^\circ$, when the sum of the areas $a^2$ and $b^2$ equals area $c^2$. It follows that lengths $a$ and $b$ are less than length $c$, which is less than $a + b$ (i.e., $0 < a \leq b < c < a + b$).

To appreciate the significance of this development it is helpful to trace the record back to before the time of the Greeks. The Pythagorean theorem (Eq. 1.1) did not begin with Euclid or Pythagoras. Rather Euclid and Pythagoras appreciated the importance of these ideas and documented them.

In the end the Pythagoreans were destroyed by fear. This may be the fate of mixing technology and politics:

“Whether the complete rule of number (integers) is wise remains to be seen. It is said that when the Pythagoreans tried to extend their influence into politics they met with popular resistance. Pythagoras fled, but he was murdered in nearby Mesopotamia in 497 BCE.”

–Stillwell (2010, p. 16)

### Chronological history 17-18\textsuperscript{th} centuries

![Exploded time-line of the two centuries between the 16\textsuperscript{th} and 18\textsuperscript{th} CE. Given the time line, it seems likely that Bombelli’s discovery of Diophantus’ book on “Arithmetic” in the Vatican library, triggered many of the ideas presented by Galileo, Descartes and Fermat, followed by others (i.e., Newton). Bombelli’s discovery might be considered as a magic moment in mathematics. The vertical red lines connect mentor-student relationships. Figure 1.2 (p. 19) gives the time-line from 1500BCE to 1650CE. Figure 1.12 (p. 54) presents the 17-20 CE (Newton–Einstein) view from 1640–1950, and Fig. 1.21 (p. 98) outlines the full (Bombelli–Einstein, 1525-1925) period.[fig:TimeLine16_18]

#### 1.1.5 The birth of modern mathematics

Modern mathematics (what we know today) was born in the 15-16\textsuperscript{th} century, in the hands of Leonardo da Vinci, Bombelli, Galileo, Descartes, Fermat, and many others (Stillwell, 2010). Many of these early master were, like the Pythagoreans, secretive to the extreme about how they solved problems. This soon changed due to Galileo, Mersenne, Descartes and Newton, causing mathematics to blossom.

**The amazing Bernoulli family** The first individual that seems to have openly recognized the importance of mathematics, to actually teach it, was Jacob Bernoulli (Fig. 1.3). Jacob worked on what is now viewed as the standard package of analytic “circular” (i.e., periodic) functions: $\sin(x)$, $\cos(x)$, $\exp(x)$, $\log(x)$.\(^{13}\) Eventually the full details were developed (for real variables) by Euler (Sections 1.3.1 p. 64 and 3.1.1, p. 178).

\(^{13}\)The log and tan functions are related by $\tan^{-1}(z) = -\frac{i}{2} \ln(\frac{1-iz}{1+iz})$. 
From Fig. 1.4 we see that Jacob was contemporary to Descartes, Fermat, Newton, and Euler. Thus, it seems likely that he was strongly influenced by Newton, who in turn was influenced by Descartes.\footnote{https://en.wikipedia.org/wiki/Early_life_of_Isaac_Newton} Vi`te and Wallis (Stillwell, 2010, p. 175). With the closure of Cambridge University due to the plague of 1665, Newton returned home to Woolsthorpe-by-Colsterworth (95 [mi] north of London), to worked by himself, for over a year.

Jacob Bernoulli, like all successful mathematicians of the day, was largely self taught. Yet Jacob was in a new category of mathematicians, because he was an effective teacher. Jacob taught his sibling Johann, who then taught his sibling Daniel. But most importantly, Johann taught Leonhard Euler (Figs. 1.3, 1.21), the most prolific (thus influential) of all mathematicians. This resulted in an explosion of new ideas and understanding. It is most significant that all four mathematicians published their methods and findings. Much later, Jacob studied with students of Descartes\footnote{It seems clear that Descartes was also a teacher.} (Stillwell, 2010, p. 268-9).

Euler went far beyond all the Bernoulli family, Jacob, Johann and Daniel, (Stillwell, 2010, p. 315). A special strength of Euler was the degree to which he published. First he would master a topic, and then he would publish. His papers continued to appear long after his death (Calinger, 2015).

Another individual of that time of special note, who also published extensively, was d’Alembert (Figs. 1.3, 1.21). Some of the most important tools were first proposed by d’Alembert. Unfortunately, and perhaps somewhat unfairly, his rigor was criticized by Euler, and later by Gauss (Stillwell, 2010).

Once the tools were being openly published, mathematics grew exponentially.\footnote{There are at least three useful exponential scales: Factors of 2, factors of $e \approx 2.7$, and factors of 10. The decibel uses factors of 2 ($6 \text{ [dB]}$) and factors of 10 ($20 \text{ [dB]}$). Information theory uses powers of 2 (1 [bit]), 4 (2 [bits]). Circuit theory uses in factors of $e \approx 3.7$ ($11.4 \text{ [dB]}$).} It was one of the most creative times in mathematics. Figure 1.21 shows the list of the many famous names, and their relative time-line. To aid in understanding the time line, note that Leonhard Euler was a contemporary of Benjamin Franklin, and James Clerk Maxwell of Abraham Lincoln.\footnote{Lincoln traveled through Mahomet IL (where I live) on his way to the Urbana Court house.}

### 1.1.6 Three Streams from the Pythagorean theorem

From the outset of his presentation, Stillwell (2010, p. 1) defines “three great streams of mathematical thought: Numbers, Geometry and Infinity,” that flow from the Pythagorean theorem, as summarized in Table 1.1. The Pythagorean theorem is the spring from which flow the three streams of all mathematics. This is a useful concept, based on reasoning not as obvious as one might think. Many factors are in play here. One of these was the strongly held opinion of Pythagoras that all mathematics should be based on integers. The rest are tied up in the long, necessarily complex history of mathematics, as best summarized by the fundamental theorems (Table 1.2, p. 35), which are each discussed in detail in a relevant chapter.

Stillwell’s concept of three streams following from the Pythagorean theorem is the organizing principle behind the this book, organized by chapter:

1. **Introduction** (Chapter 1) A detailed overview of the fundamentals and the three streams are presented in Sections

   1.2 Number systems
CHAPTER 1. INTRODUCTION

Table 1.1: Three streams followed from Pythagorean theorem: Number Systems (Stream 1), Geometry (Stream 2) and Infinity (Stream 3).

- The Pythagorean Theorem is the mathematical spring which bore the three streams.
- Several centuries per stream:
  1) **Numbers:**
     - 6th BCE: \( \mathbb{N} \) counting numbers, \( \mathbb{Q} \) (Rationals), \( \mathbb{P} \) Primes
     - 5th BCE: \( \mathbb{Z} \) Common Integers, \( \mathbb{I} \) Irrationals
     - 7th CE: zero \( \in \mathbb{Z} \)
  2) **Geometry:** (e.g., lines, circles, spheres, toroids, …)
     - 17th CE: Composition of polynomials (Descartes, Fermat)
     - Euclid’s Geometry + algebra \( \Rightarrow \) Analytic Geometry
     - 18th CE: Fundamental Theorem of Algebra
  3) **Infinity:** (\( \infty \rightarrow \) Sets)
     - 17-18th CE: Taylor series, Functions, Calculus (Newton)
     - 19th CE: \( \mathbb{R} \) Real, \( \mathbb{C} \) Complex
     - 20th CE: Set theory

1.3 Algebraic equations
1.4 Scalar Calculus
1.5 Vector Calculus

Chapter 1 is intended to be a self contained education on basic pre-college mathematics. Chapter 2-5 are dedicated to advanced topics. If you’re a novice, stick to Chapter 1. As of Version Version 0.86.08, Chapters 2-5 are still under development.

2. **Number Systems** (Chapter 2: Stream 1) Fundamentals of number systems, starting with prime numbers, through complex numbers, vectors and matrices.

3. **Algebraic Equations** (Chapter 3: Stream 2) Algebra and its development, as we know it today. The theory of real and complex equations and functions of real and complex variables. Complex impedance \( Z(s) \) of complex frequency \( s = \sigma + \omega j \) is covered with some care, developing the topic which is needed for engineering mathematics.

4. **Scalar Calculus** (Chapter 4: Stream 3a) Ordinary differential equations. Integral theorems. Acoustics.

1.2 Stream 1: Number Systems

Number theory (discrete, i.e., integer mathematics) was a starting point for many key ideas. For example, in Euclid’s geometrical constructions the Pythagorean theorem for \([a, b, c] \in \mathbb{R}\) was accepted as true, but the emphasis in the early analysis was on integer constructions (i.e., \([a, b, c] \in \mathbb{N}\)), such as Euclid’s formula for Pythagorean triplets (Eq. 1.9, Fig. 1.8, p. 46). As we shall see, the Pythagorean theorem is a rich source of mathematical constructions, such as composition of polynomials, and solutions of Pell’s equation by eigen-vector and recursive analysis methods. Recursive difference equation solutions predate calculus, at least going back to the Chinese (c2000 BCE). These are early (pre-limit) forms of differential equations, best analyzed using an eigen-vector expansion, a powerful geometrical concept from linear algebra, of an expansion in terms of an orthogonal set of normalized (unit-length) vectors (Appendix C, p. 245).

The first use of zero and \(\infty\): It is hard to imagine that one would not appreciate the concept of zero and negative numbers when using an abacus. If five beads are moved up, and one is moved down, then four are left. Then if four more are move down, that leaves zero. Taking away (subtracting) is the opposite of giving (adding). Thus subtraction, to obtain zero beads, is no different than subtraction from zero, giving negative beads. On an abacus, subtraction is obviously the inverse of addition. To assume the Romans or the Chinese did not understand negative numbers, seems a bit absurd.

However, understanding the concept of zero and negative numbers is not the same as having a symbolic notation. The Roman number system has no such symbols. The first recorded use of a symbol for zero is said to be by Brahmagupta\(^{18}\) in 628 CE.\(^{19,20}\) It does not take much imagination to go from counting numbers \(\mathbb{N}\) to the set of all integers \(\mathbb{Z}\), including zero, but apparently it takes 600 years to functionally develop a terminology, that represents these ideas. Likely this is more about politics and government rule, than mathematics. Defining the rules of subtraction required the creation of algebra (c830 CE, Fig. 1.2).

The concept that caused much more difficulty was \(\infty\). Until Riemann’s thesis in 1851 it was not clear if \(\infty\) was a number, many numbers, or even definable. Today we know.

1.2.1 Lec 2: The Taxonomy of Numbers: \(\mathbb{N}, \mathbb{P}, \mathbb{Z}, \mathbb{Q}, \mathbb{F}, \mathbb{I}, \mathbb{R}, \mathbb{C}\)

Once symbols for zero and negative numbers were accepted, progress was made. To fully understand numbers, a transparent notation was required. First one must differentiate between the different classes (genus) of numbers, providing a notation that defines each of these classes, along with their relationships. It is logical to start with the most basic counting numbers, which we indicate with the double-bold symbol \(\mathbb{N}\). All the double-bold symbols and set-theory symbols, such as \(\{\}, \cup, \cap, \in, \mathbb{Z}, \perp\) etc. are summarized in Appendix A on mathematical notation, for quick access.

Counting numbers \(\mathbb{N}\): These are known as the “natural numbers” \(\mathbb{N} = 1, 2, 3, \cdots, \infty\), denoted by the double-bold symbol \(\mathbb{N}\). For increased clarity we shall refer to the natural numbers as counting numbers, to clarify that natural means integer. The mathematical sentence “\(2 \in \mathbb{N}\)” is read as 2 is a member of the set of counting numbers. The word set means the sharing of a specific property.

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\(^{19}\)The fall of the Roman Empire has been established as Sept. 4, 476 CE.

CHAPTER 1. INTRODUCTION

Primes \( \mathbb{P} \): A number is prime \((\pi_n \in \mathbb{P})\) if its only factors are 1 and itself.\(^{21}\) Since \(1 = 1 \cdot 1\), \(1 \not\in \mathbb{P}\), as it is seen to violate this basic definition of a prime. Prime numbers \(\mathbb{P}\) are a subset of the counting numbers \((\mathbb{P} \subset \mathbb{N})\). A somewhat amazing fact, well known to the earliest mathematicians, is that every integer may be written as a unique product of primes. A second key idea is that the density of primes \(\rho_\pi(N) \sim 1/\log(N)\) \((\rho_\pi(N)\) is inversely proportional to the log of \(N\) \((\text{Eq. 2.1, p. 164})\), an observation first observed by Gauss (Goldstein, 1973). In fact, there is a prime between every integer \(N\) and \(2N\).

Exercise: Write out the first 10 (or 20) integers in factored form. Sol: \(1, 2, 3, 2^2, 5, 2 \cdot 3, 7, 2^3, 3^2, 2 \cdot 5, 11, 3 \cdot 2^2, 13, 2 \cdot 7, 3 \cdot 5, 2^4, 17, 2 \cdot 3^2, \cdots\).

Exercise: Write the integers from 2-20 in terms of \(\pi_n\). Here is a table to assist you:

<table>
<thead>
<tr>
<th>(n)</th>
<th>(\pi_n)</th>
<th>2</th>
<th>3</th>
<th>5</th>
<th>7</th>
<th>11</th>
<th>13</th>
<th>17</th>
<th>19</th>
<th>23</th>
<th>29</th>
<th>31</th>
</tr>
</thead>
<tbody>
<tr>
<td>1</td>
<td>(\pi_1)</td>
<td>2</td>
<td></td>
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<td></td>
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</tr>
<tr>
<td>2</td>
<td>(\pi_2)</td>
<td>3</td>
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<tr>
<td>3</td>
<td>(\pi_3)</td>
<td>5</td>
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<tr>
<td>4</td>
<td>(\pi_4)</td>
<td>7</td>
<td>11</td>
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<td>5</td>
<td>(\pi_5)</td>
<td>13</td>
<td>17</td>
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</tr>
<tr>
<td>6</td>
<td>(\pi_6)</td>
<td>23</td>
<td>29</td>
<td>31</td>
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<tr>
<td>7</td>
<td>(\pi_{11})</td>
<td>2 \cdot 3</td>
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<tr>
<td>8</td>
<td>(\pi_{13})</td>
<td></td>
<td>3 \cdot 7</td>
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<tr>
<td>9</td>
<td>(\pi_{14})</td>
<td>10</td>
<td>11</td>
<td>12</td>
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<tr>
<td>10</td>
<td>(\pi_{15})</td>
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<td>11</td>
<td>(\pi_{16})</td>
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<td>2 \cdot 5</td>
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<tr>
<td>12</td>
<td>(\pi_{17})</td>
<td></td>
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<td></td>
<td></td>
<td></td>
<td>2 \cdot 3 \cdot 5</td>
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</table>

We shall use the convenient notation \(\pi_n\) for the prime numbers, indexed by \(n \in \mathbb{N}\). The first 12 primes \((n = 1, \ldots, 12)\) are \(\pi_n = 2, 3, 5, 7, 11, 13, 17, 19, 23, 29, 31, 37\). Since, \(4 = 2^2\) and \(6 = 2 \cdot 3\) may be factored, \(4, 6 \not\in \mathbb{P}\) (read as: \(4 \text{ and } 6 \text{ are not in the set of primes}\)). Given this definition, multiples of a prime, i.e., \([2, 3, 4, 5, \ldots] \cdot \pi_k\) of any prime \(\pi_k\), cannot be prime. It follows that all primes except 2 must be odd and every integer \(N\) is unique in its factorization.

Coprimes are numbers with no common prime factors. For example, \(4 = 2 \cdot 2\) and \(6 = 2 \cdot 3\) are not coprime, as they have 2 as a common factor, whereas \(21 = 3 \cdot 7\) and \(10 = 2 \cdot 5\) are. By definition all distinct primes are coprime. The notation \(m \perp n\) indicates that \(m, n\) are coprime. The ratio of two coprimes is reduced, as it has no factors to cancel. The ratio of two numbers that are not coprime, may always be reduced to a smaller ratio, once the common factors are canceled. This is called the reduced form, or a irreducible fraction. When doing numerical work, it is always beneficial to work with the reduced form. This is especially important when working with the ratio of polynomials, to remove common roots, typically by long division, or deconvolution.

The fundamental theorem of arithmetic states that all integers may be uniquely expressed as a product of primes. The Prime Number Theorem estimates the mean density of primes over \(\mathbb{N}\). All real numbers have a natural order on the real line. Complex numbers do not have a natural order. For example, \(j > 1\) makes no sense.

Integers \(\mathbb{Z}\): These include positive and negative counting numbers and zero. Notionally we might indicate this using set notation as \(\mathbb{Z} : \{-\mathbb{N}, 0, \mathbb{N}\}\). Read this as The integers are in the set composed of the negative of the natural numbers \(-\mathbb{N}\), zero, and \(\mathbb{N}\). Note that \(\mathbb{N} \subset \mathbb{Z}\).

---

\(^{21}\)40 primes are generated by the quadratic \(P_k = k^2 - n + 40, k = 1 : 40\). But why?
Rational numbers $\mathbb{Q}$: These are defined as numbers formed from the ratio of two integers. Given two numbers $n, d \in \mathbb{Z}$, then $n/d \in \mathbb{Q}$. Since the integers $\mathbb{Z}$ include 1, it follows that integers are a subset of rational numbers ($\mathbb{Z} \subset \mathbb{Q}$). For example, the rational number $3/1 \in \mathbb{Z}$. The main utility of rational numbers is that they can efficiently approximate any number on the real line, to any precision. For example the rational approximation $\pi \approx 22/7$, has a relative error of $\approx 0.04\%$.

Fractional number $\mathbb{F}$: The fractionals $\mathbb{F}$ are defined as the ratio of two coprimes. If $n \perp d$, then $n/d \in \mathbb{F}$. Given this definition, $\mathbb{F} \perp \mathbb{Z}$, $\mathbb{F} \subset \mathbb{Q} = \mathbb{Z} \cup \mathbb{F}$. Because of the powerful approximating power of rational numbers, the fractional set $\mathbb{F}$ represent the most important (and the largest) portion of the rational numbers, dwarfing the size of the integers, another good reason for defining the two distinct subsets. For example $\pi \approx 22/7 = 11 \cdot 2/7 = 3 + 1/7$ with $22 \perp 7$, and $9/6 = 3/2 = 1 + 1/2$ with $3 \perp 2$.

Irrational numbers $\mathbb{I}$: Every real number that is not rational is irrational ($\mathbb{Q} \perp \mathbb{I}$). Irrational numbers include $\pi, e$ and the square roots of the primes ($\in \sqrt{\mathbb{P}}$). These are decimal numbers that never repeat, thus requiring infinite precision in their representation. Such numbers can not, and never will be represented on any computer, as they would require an infinite number of bits (precision).

The rationals $\mathbb{Q}$ and irrationals $\mathbb{I}$ split the reals ($\mathbb{R} = \mathbb{Q} \cup \mathbb{I}$), thus each is a subset of the reals ($\mathbb{Q} \subset \mathbb{R}$, $\mathbb{I} \subset \mathbb{R}$). This is analogous to the integers $\mathbb{Z}$ and fractionals $\mathbb{F}$, which split the rationals ($\mathbb{Q} = \mathbb{Z} \cup \mathbb{F}$), (thus each is a subset of the rationals ($\mathbb{Z} \subset \mathbb{Q}$, $\mathbb{F} \subset \mathbb{Q}$)).

Irrational numbers ($\mathbb{I}$) were famously problematic for the Pythagoreans, who incorrectly theorized that all numbers were rational. Like $\infty$, irrational numbers require mastering a new and difficult concept before they could even be defined: It was essential to understand the factorization of counting numbers into primes (i.e., the fundamental theorem of arithmetic) before the concept of irrationals could be sorted out. Irrational numbers can only be understood once limits were mastered.

As discussed in Sect. Lec. 1.2.5 (p. 42), fractionals can approximate any irrational number with arbitrary accuracy. But they are only equal in the limit, which can never be obtained. Integers are also important, but for a very different reason. All numerical computing today is done with $\mathbb{Q} = \mathbb{F} \cup \mathbb{Z}$. Indexing uses integers $\mathbb{Z}$, while the rest of computing (flow dynamics, differential equations, etc.) is done with fractionals $\mathbb{F}$ (i.e., IEEE-754). Computer scientists are trained on these topics, and computer engineers need to be at least conversant with them.

Real numbers $\mathbb{R}$: Reals are the union of rational and irrational numbers, namely $\mathbb{R} = \mathbb{I} \cup \mathbb{Q}$ ($\mathbb{R} = \mathbb{Z} \cup \mathbb{F} \cup \mathbb{I}$). Lengths in Euclidean geometry are reals. Many people assume that IEEE 754 floating point numbers (c1985) are real (i.e., $\in \mathbb{R}$). In fact they are rational ($\mathbb{Q} : \{\mathbb{F} \cup \mathbb{Z}\}$) approximations to real numbers, designed to have a very large dynamic range. There can be no machine realization of irrational numbers, since such a number would require infinite precision ($\infty$ bits). The hallmark of fractional numbers ($\mathbb{F}$) is their power in making highly accurate approximations of any real number.

Using Euclid’s compass and ruler methods, one can make line length proportionally shorter or longer, or (approximately) the same. A line may be made be twice as long, an angle bisected. However, the concept of an integer length in Euclid’s geometry was not defined.\textsuperscript{22} Nor can one

\textsuperscript{22}As best I know.
construct an imaginary or complex line, as all lines are assumed to be real.

Real numbers were first fully accepted only after set theory was developed by Cantor (1874) (Stillwell, 2010, pp. 461, 525, . . . ). At first brush, this seems amazing, given how widely accepted real numbers are today. In some sense they were accepted by the Greeks, as lengths of real lines.

The size of \( \mathbb{R} \): The real line may be divided into a hierarchical groups, sorted by size: \( |\mathbb{R}| > |\mathbb{I}| \gg |\mathbb{Q}| > |\mathbb{F}| \gg |\mathbb{Z}| > |\mathbb{N}| > |\mathbb{P}| \). The reals may be split into irrationals and rationals, which may be further split into the integers and the fractionals. Thus, all is not integer. If a triangle has two integer sides, then the hypotenuse can be irrational (\( \sqrt{2} = \sqrt{1^2 + 1^2} \)). This leads us to a fundamental question: “Are there integer solutions to Eq. 1.1?” We need not look further than the simple example \{3, 4, 5\}. In fact the example generalizes: the formula for generating an infinite number of integer solutions is called Euclid’s Formula (Section 1.2.5, p. 42).

The more important point is that the number of irrationals is much larger than any set other than the reals (i.e., complex numbers). Thus when we use computers to model physical systems, we are constantly needing to compute with irrational numbers. But this is impossible since every irrational numbers would require an infinite number of bits to represent it. Thus we must compute with rational approximations to the irrationals. This means we need to use the fractionals. In the end, we must work with the IEEE 754 floating point numbers,\(^{23}\) which are rationals. This can lead to serious computational problems (Section 1.3.1, p. 59).

Complex numbers \( \mathbb{C} \): Complex numbers are best defined as ordered pairs of real numbers.\(^{24}\) The word “complex,” as used here, does not mean that the number are complicated or difficult. The word “imaginary” does not mean the numbers disappear. Complex numbers are quite special in engineering mathematics, as roots of polynomials. The most obvious example is the quadratic formula for the roots of a \( 2^{nd} \) degree polynomial, having either rational, real or complex coefficients.

Today the common way to write a complex number is using the notation \( z = a + bj \in \mathbb{C} \), where \( a, b \in \mathbb{R} \). Here \( 1j = \sqrt{-1} \). We also define \( 1i = -1j \) to account for the two possible signs of the square root. Accordingly \( 1j^2 = 1i^2 = -1 \).

Multiplication of complex numbers follows the basic rules of real algebra, for example, the rules of multiplying two polynomials. Multiplication of two first degree polynomials (i.e., monomials) gives

\[
(a + bx)(c + dx) = ac + (ad + bc)x + bdx^2
\]

If we substitute \( 1j \) for \( x \), and use the definition \( 1j^2 = -1 \), we obtain the product of the two complex numbers

\[
(a + bj)(c + dj) = ac - bd + (ad + bc)j.
\]

Thus multiplication of complex numbers obey the accepted rules of algebra.

Polar representation: A alternative for complex multiplication is to work with polar coordinates. The polar form of complex number \( z = a + bj \) is written in terms of its magnitude \( \rho = \sqrt{a^2 + b^2} \) and angle \( \theta = \angle z = \tan^{-1}(z) = \arctan z \), as \( z = \rho e^{\theta j} \). From the definition of the complex natural log function

\[
\ln \rho e^{\theta j} = \ln \rho + \theta j,
\]


\(^{24}\)A polynomial \( a + bx \) and a 2-vector \([a, b]^T = \begin{bmatrix} a \\ b \end{bmatrix}\) are also examples of ordered pairs.
which is useful in engineering calculations.\textsuperscript{25} Note that the angle \( \angle \theta = \Im \{ \log z \} \). Clearly this is an improvement over \( \angle \theta = \arctan(z) \).

The polar representation make clear the utility of a complex number: its magnitude scales while its angle \( \Theta \) rotates. The property of scaling and rotating is what makes complex numbers useful in engineering calculations. This is especially obvious when dealing with impedances, which have complex roots with very special properties, as discussed in Sect. 1.3.4 (p. 70).

**Matrix representation:** An alternative way to represent complex numbers is in terms of 2x2 matrices. This relationship is defined by the mapping from a complex number to a 2x2 matrix

\[
\begin{align*}
a + bj & \leftrightarrow \begin{bmatrix} a & -b \\ b & a \end{bmatrix}, \\
2 & \leftrightarrow \begin{bmatrix} 2 & 0 \\ 0 & 2 \end{bmatrix}, \\
-j & \leftrightarrow \begin{bmatrix} 0 & 1 \\ -1 & 0 \end{bmatrix}, \\
e^{\theta j} & \leftrightarrow \begin{bmatrix} \sin(\theta) & -\cos(\theta) \\ \cos(\theta) & \sin(\theta) \end{bmatrix}.
\end{align*}
\] (1.3)

By taking the inverse of the 2x2 matrix one can define the ratio of one complex number by another. Until you try out this representation, it may not seem obvious, or even that it could work.

This representation proves that 1\( j \) is not necessary when defining a complex number. What 1\( j \) can do is to conceptually simplify the algebra. It is worth your time to become familiar with the matrix representation, to clarify any possible confusions you might have about multiplication and division of complex numbers. This matrix representation can save you time, heartache and messy algebra. Once you have learned how to multiply two matrices, it’s a lot simpler than doing the complex algebra. In many cases we will leave the results of our analysis in matrix form, to avoid the algebra altogether.\textsuperscript{26} Thus both representations are important. More on this topic may be found in Chapter 2.

**Exercise:** Using Matlab/Octave, verify that

\[
\frac{a + bj}{c + dj} = \frac{ab + bd + (bc - ad)j}{c^2 + d^2} \leftrightarrow \begin{bmatrix} a & -b \\ b & a \end{bmatrix} \begin{bmatrix} c & -d \\ d & c \end{bmatrix}^{-1} = \begin{bmatrix} a & -b \\ b & a \end{bmatrix} \begin{bmatrix} c & d \\ -d & c \end{bmatrix} \frac{1}{c^2 + d^2}.
\] (1.4)

**Sol:** The best way may be using numbers. Below is symbolic code, independent of numerics:

```matlab
syms a b c d A B
A=[a -b; b a];
B=[c -d; d c];
C=A*inv(B)
```

**Numerical taxonomy:**

A simplified taxonomy of numbers is given by the mathematical sentence

\[ \pi_k \in \mathbb{P} \subset \mathbb{N} \subset \mathbb{Z} \subset \mathbb{Q} \subset \mathbb{R} \subset \mathbb{C}. \]

This sentence says:

\textsuperscript{25}Chapter 2 discusses the definition of the phase, i.e., how is it computed (i.e., \( \arctan(e^{\theta j}) \), \( \arctan2(x, y) \)), and the importance of the unwrapped phase, as in the example (Table 1.5, p. 93) \( \delta(t - \tau) \leftrightarrow e^{-\tau} \).

\textsuperscript{26}Sometimes we let the computer do the final algebra, numerically, as 2x2 matrix multiplications.
1. Every prime number $\pi_k$ is in the set of primes $\mathbb{P}$,
2. which are a subset of the counting numbers $\mathbb{N}$,
3. which are a subset of all integers $\mathbb{Z} = \{-\mathbb{N}, 0, \mathbb{N}\}$,
4. which are a subset of the rationals $\mathbb{Q}$ (ratios of counting numbers $\mathbb{N}$),
5. which are a subset of the reals $\mathbb{R}$,
6. which are a subset of the complex numbers $\mathbb{C}$.

The rationals $\mathbb{Q}$ may be further decomposed into the fractionals $\mathbb{F}$ and the integers $\mathbb{Z}$

$$\mathbb{Q} = \{\mathbb{F}, \mathbb{Z}\} \Rightarrow \mathbb{F} \perp \mathbb{Z},$$

and the reals $\mathbb{R}$ into the rationals $\mathbb{Q}$ and the irrationals $\mathbb{I}$

$$\mathbb{R} = \{\mathbb{I}, \mathbb{Q}\} \Rightarrow \mathbb{I} \perp \mathbb{Q}.$$ 

This classification nicely defines all the numbers used in engineering and physics.

This may be expressed with the single compact sentence, starting with the prime numbers $\pi_k$ and ending with complex numbers $\mathbb{C}$:

$$\pi_k \in \mathbb{P} \subset \mathbb{N} \subset \mathbb{Z} \subset (\mathbb{Z} \cup \mathbb{F} = \mathbb{Q}) \subset (\mathbb{Q} \cup \mathbb{I} = \mathbb{R}) \subset \mathbb{C}.$$ 

This makes a Venn diagram that is easily visualized.

**Sources and utilities of numbers:** The roots of real polynomials $x_k$ are complex ($x_k \in \mathbb{C}$), independent of the genus of the coefficients (e.g., real integer coefficients give rise to complex roots). Each genus plays an important role in algebra, with prime numbers at the bottom (root of the tree) and complex numbers at the top. We shall explore this further in Chaps. 2 and 3.

As discussed in Appendix A (p. 229), all numbers may be viewed as complex. Namely every real number is complex if we take the imaginary part to be zero (Boas, 1987). For example, $2 \in \mathbb{P} \subset \mathbb{C}$. Likewise every purely imaginary number (e.g., $0 + 1j$) is complex with zero real part. It follows that $2j = 0 + \pi_2j \in \mathbb{P}$. Integers are a subset of reals, which are a subset of complex numbers.$^{27}$ *Gaussian integers* are complex integers ($\mathbb{Z} \subset \mathbb{R} \subset \mathbb{C}$).$^{28}$

Finally, note that complex numbers $\mathbb{C}$, much like vectors, do not have “rank-order,” meaning one complex number cannot be larger or smaller than another. It makes no sense to say that $j > 1$ or $j = 1$ (Boas, 1987). The real and imaginary parts and the magnitude and phase, have order. If time $t$ were complex, there could be no yesterday and tomorrow.$^{29}$

$^{27}$The plural *complexes* (a double /s/) seems an unacceptable word in English.

$^{28}$It follows that integers are a subset of Gaussian integers (the imaginary or real part of the Gaussian integer may be zero).

$^{29}$One can define $\xi = x + 1j \, ct$ to be complex ($x, t \in \mathbb{R}$, with $x$ in meters [m] and $t$ is in seconds [s], with the speed of light $c_0$, [m/s].
Applications of integers

The most relevant question at this point is “Why are integers important?” First we count with them, so we can keep track of “how much.” But there is much more to numbers than counting: We use integers for any application where absolute accuracy is essential, such as banking transactions (making change), the precise computing of dates (Stillwell, 2010, p. 70) and location (“I’ll meet you at 34 and Vine at noon on Jan. 1, 2034.”), building roads or buildings out of bricks (objects built from a unit size).

To navigate we need to know how to predict the tides, the location of the moon and sun, etc. Integers are important because they are precise: Once a month there is a full moon, easily recognizable. The next day its slightly less than full. If one could represent our position as integers in time and space, we would know exactly where we are at all times.

The Pythagoreans claimed that all was integer. From a practical point of view, it seems they were right. Today all computers compute floating point numbers as fractionals. However, in theory they were wrong. The difference is a matter of precision.

History of complex numbers: It is notable how long it took for complex numbers to be accepted (1851), relative to when they were first introduced by Bombelli (16th century CE). One might have thought that the solution of the quadratic, known to the Chinese, would have settled this question. It seems that complex integers (aka, Gaussian integers) were accepted before non-integral complex numbers. Apparently real numbers (\( \mathbb{R} \)) were not accepted (i.e., proved to exist, thus mathematically defined) until even later. It took the development of set theory in the late 19th century to sort out a proper definition of the real number, due to the existence of irrational numbers. It makes one wonder “How many angels can stand on the point of a pin?”

Numerical Representations of \( \mathbb{I}, \mathbb{R}, \mathbb{C} \): When doing numerical work, one must consider how we may compute within the family of reals (i.e., irrationals). There can be no irrational number representation on a computers. IEEE floating point numbers, which are the international standard of computation, are actually rational approximations. The mantissa and the exponent are each integers, having sign and magnitude. The size of each integer depends on the precision of the number being represented. An IEEE floating-point number is rational because it has a binary (integer) mantissa, multiplied by 2 raised to the power of a binary (integer) exponent. For example \( \pi \approx a2^b \) with \( a, b \in \mathbb{Z} \). In summary, IEEE floating-point numbers are not, and cannot, be irrational, since to do that would require an infinite number of bits.

True floating point numbers contain irrational numbers, which must be approximate by rational numbers. This leads to the concept of fractional representation, which requires the definition of the mantissa, base and exponent, where both the mantissa and the exponent are signed. Numerical results must not depend on the base. One could dramatically improve the resolution of the numerical representation by the use of the Fundamental theorem of Arithmetic (Section 1.2.2, page 34). For example one could factor the exponent into its primes and then represent the number as \( a2^{b_1}3^{c_1}5^{d_1}7^{e_1} \) (\( a, b, c, d, e \in \mathbb{Z} \)), etc. Such a representation would improve the resolution of the representation. But even so, the irrational numbers would be approximate. For example, base ten \(^{30}\) is natural using this representation since \( 10^n = 2^n5^n \). Thus

\[
\pi \cdot 10^5 \approx 314159.27 \ldots = 3 \cdot 2^55^5 + 1 \cdot 2^45^4 + 4 \cdot 2^35^3 + \ldots + 9 \cdot 2^05^0 + 2 \cdot 2^{-1}5^{-1} \ldots
\]

\(^{30}\)Base 10 is the natural world-wide standard simply because we have 10 fingers.
If we approximate $\pi$ by $22/7$, then according to the Matlab/Octave `dec2bin()` routine, the binary representation is

$$\pi \cdot 2^{17} \approx 131072_{10} \cdot 22/7 = 110, 0100, 1001, 0010, 0101_2,$$

where 1 and 0 are multipliers of powers of 2, which are then added together as follows

$$2^{18} + 2^{17} + 2^{14} + 2^{11} + 2^8 + 2^5 + 2^2 + 2^0.$$

In base 16 (i.e., hexadecimal) $2^{17} \cdot 22/7 = 2^{18} \cdot 8_{16}/7_{16}$.

Computers keep track of the decimal point using the exponent, which in this case is the factor $2^{17} = 131072_{10}$. The concept of the decimal point is replaced by an integer, having the desired precision, and a scale factor of any base (radix). This scale factor may be thought of as moving the decimal point to the right (larger number) or left (smaller number). The mantissa “fine-tunes” the value about a scale factor (the exponent).

Here is $x = 2^{17} \times 22/7$ using IEEE 754 double precision, as computed by an IEEE-754 floating point converter:\(^{31}\) $x = 411, 940.5625_{10}=2^{54} \times 1198372=0, 10010, 00, 110010, 010010, 010010, 010010_2 = 0x48c92492_{16}$. The commas in the binary (0,1) string are to help visualize the quasi-periodic nature of the bit-stream. The mantissa is 4, 793, 49010 and the exponent is $2^{18}$. The numbers are stored in a 32 bit format, with 1 bit for sign, 8 bits for the exponent and 23 bits for the mantissa. Perhaps a more instructive number is $x = 4, 793, 490.0 = 0, 100, 1010, 100, 100, 100, 100, 100, 100, 100, 100, 100, 100_2 = 0x4a92492_{16}$, which has a repeating binary bit pattern of $((00)_2$), broken by the scale factor $0x4a$. Another with even higher symmetry is $x = 6.344, 131, 191, 146, 9 \times 10^{-17} = 0x24, 924, 924_{16} = 00, 100, 100, 100, 100, 100, 100, 100, 100, 100, 100_2$. In this example the repeating pattern is clear in the Hex representation as a repeating $(942)_3$, as represented by the double brackets, with the subscript indicating the period, in this case, 3 digits. As before, the commas are to help with readability, and have no other meaning.

There are other important types of representations. As pairs of reals, complex numbers have similar approximate representations. An important representations of complex numbers is $e^z = \cosh(z) + j \sinh(z)$ with $z \in \mathbb{C}$, which includes the famous formula of Euler ($\theta \in \mathbb{R}$)

$$e^{j\theta} = \cos \theta + j \sin \theta \leftrightarrow \begin{bmatrix} \cos \theta & -\sin \theta \\ \sin \theta & \cos \theta \end{bmatrix}.$$

Some of these concepts can be generalized to include vectors, matrices and polynomials.

**Pythagoreans, Integers** The integer is the corner stone of the Pythagorean doctrine, so much so that it caused a fracture within the Pythagoreans when it was discovered that not all numbers are rational. The famous example is the isosceles triangle $1, 1, \sqrt{2}$, which lead to the next triangle $[1, 2, \sqrt{3}]$, etc. This is known as the Spiral of Theodorus: the short side is 1 and the long side is recursively incremented by one, using a simple compass-ruler construction.

There are right-triangles with integral lengths, the best known being $[3, 4, 5]$. Such triplets of integers $[a, b, c]$ that satisfy the Pythagorean formula (Eq. 1.1) are denoted *Pythagorean triplets*, which may be verified using *Euclid’s formula* (Eq. 1.9, p. 171).

To form triangles with perfect $90^\circ$ angles, the lengths need to satisfy Eq. 1.1. Such triangles are useful in constructing buildings or roads made from of bricks having a uniform size.

[^31]: [http://www.h-schmidt.net/FloatConverter/IEEE754.html](http://www.h-schmidt.net/FloatConverter/IEEE754.html)
Public-private key Security: An important application of prime numbers is public-private key (RSA) encryption, essential for internet security applications (e.g., online banking). To send secure messages the security (i.e., utility) of the internet is dependent on key encryption. Most people assume this is done by a personal login and passwords. Passwords are simply not secure, for many reasons. A proper method depends on factoring integers formed from products of primes having thousands of bits. The security is based on the relative ease in multiplying large primes, but the virtual impossibility of factoring them.

When a computation is easy in one direction, but its inverse is impossible, it is called a trapdoor function. We shall explore the reasons for this in Chapter 2. If everyone were to switch from passwords to public key encryption, the internet would be much more secure.

Puzzles: A third application of integers are imaginative problems that use integers. An example is the classic Chinese Four stone problem: “Find the weight of four stones that can be used with a scale to weigh any object (e.g., salt, gold) between 0, 1, 2, \ldots, 40 [gm].” As with the other problems, the answer is not as interesting as the method, since the problem may be easily recast into a related one. This type of problem can be found in airline magazines as entertain on a long flight. The solution to this problem is best cast as a linear algebra problem, with integer solutions. Again, once you know the trick, it is “easy.”

1.2.2 Lec 3: The role of physics in mathematics

Bells, chimes and eigen-modes: Integers naturally arose in art, music and science. Examples include the relations between musical notes, the natural eigen-modes (tones) of strings and other musical instruments. These relations were so common and well studied, it appeared that to understand the physical world (aka, the Universe), one needed to understand integers. This was a seductive view, but not actually correct. As will be discussed in Sections 1.3.1 (p. 52) and 3.1.1 (p. 177), it is best to view the relationship between acoustics, music and mathematics as historical, since these topics played such an important role in the development of mathematics. Also interesting is the role that integers play in quantum mechanics, also based on eigen-modes, but in this case, those of atoms. eigen-modes follow from solutions of the wave equation, which has natural delay due to d’Alembert’s solution, along with reflecting boundary conditions (Eq. 1.141, p. 142), resulting in roots of the characteristic equation.

Engineers are so accustomed to working with real (or complex) numbers, the distinction between real (i.e., irrational) and fractional numbers are rarely acknowledged. Integers on the other hand arise in many contexts. One cannot master programming computers without understanding integer, hexadecimal, octal, and binary representations, since all numbers in a computer are represented in numerical computations in terms of rationals ($\mathbb{Q} = \mathbb{Z} \cup \mathbb{F}$).

As discussed in Section 1.2.1 (p. 25), the primary reason integers are so important is their absolute precision. Every integer $n \in \mathbb{Z}$ is unique, and has the indexing property, which is

---

32It would seem that public key encryption could work by having two numbers with a common prime, and then by using Euclidean algorithm, that GCD could be worked out. One of the integers could be the public key and the second could be the private key. Given the difficulty of factoring the numbers into their primes, and ease of finding the GCD using Euclidean algorithm, a practical scheme may be possible. One problem with this idea is that one may store all the known primes.

33When ever someone tells you something is “easy,” you should immediately appreciate that it is very hard, but there is a concept, that once you learn, the difficulty evaporates.

34Check out the history of $1729 = 1^3 + 12^3 = 9^3 + 10^3$. 

---
essential for making lists that are ordered, so that one can quickly look things up. The alphabet also has this property (e.g., a book’s index). Other than for hexadecimal numbers, which for notional reasons use the alphabet, other than the base, letters (base 26), are equivalent to integers (base 10) (i.e., they are just as precise). A sequence of letters is equivalent to a sequence of integers.

Because of the integer’s absolute precision, the digital computer overtook the analog computer, once it was practical to make logic circuits that were fast. The first digital computer was thought to be the Eniac at the University of Pennsylvania, but it turned out that the code-breaking effort in Bletchley Park, England, under the guidance of Alan Turing, created the first digital computer (The Colossus) to break the WWII German “Enigma” code. Due to the high secrecy of this war effort, the credit was only acknowledged in the 1970s when the project was finally declassified.

There is zero possibility of analog computing displacing digital computing, due to the importance of precision (and speed). But even with binary representation, there is a non-zero probability of error, for example on a hard drive, due to physical noise. To deal with this, error correcting codes have been developed, to reduce the error by several orders of magnitude. Today this is a science, and billions of dollars are invested to increase the density of bits per area, to increasingly larger factors. A few years ago the terabyte drive was unheard of; today it is the standard. In a few years petabyte drives will certainly become available. It is hard to comprehend how these will be used by individuals, but they are essential for on-line (cloud) computing.

The role of mathematics in physics

Modern mathematics is build on a hierarchical construct of fundamental theorems, as summarized in Table 1.2. The importance of such theorems cannot be overemphasized. Gauss’ and Stokes’ Laws play a major role in understanding and manipulating Maxwell’s equations. Every engineering student needs to fully appreciate the significance of these key theorems. If necessary, memorize them. But that will not do over the long run, as each and every theorem must be fully understood. Fortunately most students already know several of these theorems, but perhaps not by name. In such cases, it is a matter of mastering the vocabulary.

The theorems are naturally organized and may be thought of in terms of Stillwell’s three streams. For Stream 1 there is the Fundamental Theorem of Arithmetic and the Prime Number Theorem. For Stream 2 there is the Fundamental Theorem of Algebra and Bézout’s theorem, while for Stream 3 there are a host of theorems on calculus, ordered by their dimensionality. Some of these theorems verge on trivial (e.g., the Fundamental Theorem of Arithmetic). Others are more challenging, such as the Fundamental Theorem of Vector Calculus and Green’s theorem.

Complexity should not be confused with importance. Each of these theorems is, as stated, fundamental. Taken as a whole, they are a powerful way of summarizing mathematics.

Stream 1: Prime Number theorems:

There are two fundamental theorems about primes,

1. *The fundamental theorem of arithmetic*: This states that every integer \( n \in \mathbb{Z} \) may be uniquely factored into prime numbers. This raises the question of the meaning of *factor* (split into a product). The *product* of two integers \( m, n \in \mathbb{Z} \) is \( mn = \sum_m n = \sum_n m \).

2. *The Prime Number Theorem*: One would like to know how many primes there are. That is easy: \( |\mathbb{P}| = \infty \) (The size of the set of primes, is infinite). One way of asking this questions is *What is the average density of primes, in the limit as \( n \to \infty \)?* This question was answered,
1. Fundamental theorems of:

(a) **Number systems: Stream 1**
   - arithmetic
   - prime number

(b) **Geometry: Stream 2**
   - algebra
   - Bézout

(c) **Calculus: Stream 3**
   - Leibniz $\mathbb{R}^1$
   - complex $\mathbb{C} \subseteq \mathbb{R}^2$
   - vectors $\mathbb{R}^3, \mathbb{R}^n, \mathbb{R}^\infty$
     - Gauss’ Law (Divergence theorem)
     - Stokes’ Law (Curl theorem, or Green’s theorem)
     - Vector calculus (Helmholtz’s theorem)

2. Other key concepts:

   - Complex analytic functions (**complex roots are finally accepted!**)
     - Complex Taylor series (complex analytic functions)
     - Region of convergence (RoC) of complex analytic series (p. 62)
     - Laplace transform, and its inverse
     - Causal time $\Rightarrow$ complex frequency $s$
     - Cauchy Integral Theorem
     - Residue integration (i.e., Green’s Thm in $\mathbb{R}^2$)
   - Riemann mapping theorem (Gray, 1994; Walsh, 1973)
   - Complex Impedance (Ohm’s Law) (Kennelly, 1893)

---

for all practical purposes, by Gauss, who in his pastime, computed the first three million
primes by hand. He discovered that, to a good approximation, the primes are equally likely,
on a log scale. This is nicely summarized by the jingle, attributed to the mathematician
Pafnuty Chebyshev (Stillwell, 2010, p. 585)

Chebyshev said, and I say it again: There is always a prime between \( n \) and \( 2n \).

When the ratio (interval) of two frequencies (pitch) is \( 2 \), the relationship is called an octave. Thus
we might say, with a slight stretch of terminology, there is at least one prime per octave. Thus one
might wonder about the maximum number of primes per octave. In modern music the octave is
further divided into 12 intervals called semitones (factors), equal to the \( \sqrt[12]{2} \). The product of 12
semitones is an octave. Thus one might ask how many primes there are per semitone? In the end,
it is a question of the density of primes on a log (i.e., ratio) scale.

**Stream 2: Fundamental theorem of algebra**

This theorem states that every polynomial in \( x \) of degree \( N \)

\[
P_N(x) = \sum_{k=0}^{N} a_k x^k
\]

has at least one root. When the root is removed, the degree of the polynomial is reduced by 1. It
follows that applied recursively, a polynomial of degree \( N \) has \( N \) roots. Note there are \( N + 1 \)
coefficients (i.e., \( a_k \)).

Besides the fundamental theorem of algebra, a second important theorem is Bézout’s theorem,
which is a generalization of the fundamental theorem of algebra. It says\(^{35}\) that the composition of
two polynomials has degree equal to the product of the degrees of each polynomial. For example,
if \( P_3(x) = x^3 \) and \( P_5(x) = x^5 \), then \( P_3(P_5)(x) = (x^5)^3 = x^{15} \). It further states that when counting
the \( N \) roots of a polynomial of degree \( N \), one must include the imaginary roots, double roots and
roots at infinity, some of which may difficult to identify.

One must wonder what happens when the degree is fractional, or worse, irrational? Are these
cases covered by Bézout?

**Stream 3: Fundamental theorems of calculus**

In Sections 1.5.10 and 1.5.11 we will deal with each of the theorems for Stream 3, where we
consider the several fundamental theorems of integration, starting with Leibniz’s formula for
integration on the real line (\( \mathbb{R} \)), then progressing to complex integration in the complex plane
(\( \mathbb{C} \)) (Cauchy’s theorem), which is required for computing the inverse Laplace transform. Gauss’
and Stokes’ Laws for \( \mathbb{R}^2 \) require closed and open surfaces, respectively. One cannot manipulate
Maxwell’s equations, fluid flow, or acoustics without understanding these theorems. Any problem
that deals with the wave equation in more than one-dimension, requires an understanding of these
theorems, thus are the basis of the derivation of the Kirchhoff voltage and current laws. The \( \nabla \)
symbol is pronounced as “del” (preferred) or “nabla.”

Finally we define the four basic vector operations based on the \( \nabla \) “operator:” the gradient
\( \nabla() \), divergence \( \nabla \cdot () \) curl \( \nabla \times () \) and the Laplacian \( \nabla \cdot \nabla() = \nabla^2() \). The first three operations are

\(^{35}\)Statements of the theorem speak of intersections and constructions of curves, rather than compositions. I find this
somewhat confusing. For example, how does intersection differ from elimination, or construction from composition
(Stillwell, 2010, p. 119)?
defined in terms of integral operations on a surface in 1, 2 or 3 dimensions, by then taking the limit as that surface, and the volume contained within, goes to zero. These three differential operators are necessary if you wish to understand Maxwell’s Equations, the crown jewel of mathematical physics. Hence mathematics plays a key role in physics, as does physics in math.

Other key concepts

Besides the widely recognized fundamental theorems for the three streams, there are a number of equally important theorems that have not yet been labeled as “fundamental.”

The widely recognized Cauchy Integral Theorem is an excellent example, since it is a stepping stone to Green’s theorem and the fundamental theorem of complex calculus. In Section 1.5.7 (p. 138) we clarify the contributions of each of these special theorems.

Once these fundamental theorems of integration (Stream 3) have been mastered, the student is ready for the complex frequency domain, which takes us back to Stream 2 and the complex frequency plane \((s = \sigma + \omega j \in \mathbb{C})\). While the Fourier and Laplace transforms are taught in mathematics courses, typically few physical connections are made, accordingly the concept of complex frequency is rarely mentioned. The complex frequency domain and causality are fundamentally related (Sects. 1.4.6–1.4.8, p. 114–115), and critical for the analysis of signals and systems, and especially the concept of impedance (Sect. 1.4.3, p. 103).

Without the concept of time and frequency, one cannot develop an intuition for the Fourier and Laplace transform relationships, especially within the context of engineering and mathematical physics.

WEEK 2

1.2.3 Lec 4: Prime numbers

If someone came up to you and asked for a theory of counting numbers, I suspect you would look them in the eye with a blank stare, and start counting. It sounds like either a stupid question or a bad joke. Yet integers are rich topic, so the question is not even slightly dumb. It is somewhat amazing that even birds and bees can count. While I doubt birds and bees can recognize primes, cicadas and other insects only crawl out of the ground in multiples of prime years, (e.g., 13 or 17 year cycles). If you have ever witnessed such an event (I have), you will never forget it. Somehow they know. Finally, there is an analytic function, first introduced by Euler, based on his analysis of the sieve, now known as the Riemann zeta function \(\zeta(s)\), which is complex analytic, with its poles at the logs of the prime numbers. The exact relationship between the primes and the poles will be discussed in Sections 3.4.2 (p. 198). The properties of this function are truly amazing, even fun.\(^{37}\) It follows that primes are a subset of the counting numbers (Section 1.2.1), that the theory of numbers (and primes) is an important topic of study. Many of the questions, and some of the answers, go back to at least the time of the Chinese (Stillwell, 2010).

\(^{36}\)It is not clear what it takes to reach this more official sounding category.

\(^{37}\)The Riemann zeta function is known as the million dollar equation as there is a cash reward for a proof of the Riemann Hypothesis.
CHAPTER 1. INTRODUCTION

The importance of prime numbers

Likely the first insight into the counting numbers started with the sieve, shown in Fig. 1.5. A sieve answers the question “How can one list the prime numbers?” The answer comes from looking for irregular patterns in the counting numbers, by playing the counting numbers against themselves.

A prime is that subset of positive integers \( \mathbb{P} \subseteq \mathbb{N} \) that cannot be factored. The number 1 is not a prime, for some non-obvious reasons, but there is no pattern in it since it is always a (useless) factor of every counting number.

Sieves

A recursive sieve method for finding primes was first devised by the Greek Eratosthenes\(^{38}\) (276-194, BCE), and summarized in Fig. 1.5.

1. Write \( N - 1 \) counting number from 2 to \( N \) (List)
2. Define loop index \( k = 1 \) and a multiplier \( n \in \mathbb{N} \) denoted \( n := \{2, \cdots, N\} \).
3. The next number on the list is prime \( \pi_k \in \mathbb{P} \)
4. Remove (Cross out) all multiples \( n \cdot \pi_n \) of \( \pi_k \)
5. \( k = k + 1 \): return to step 3.

For example, starting from \( \pi_1 = 2 \) one strikes out \( 2 \cdot 2, 2 \cdot 3, 2 \cdot 4, 2 \cdot 5, \cdots, 2 \cdot \sqrt{N} \). By definition the multiples are products of the target prime (2 in our example) and every another integer \( (n \geq 2) \). In this way all the even numbers are removed in this first iteration. The next remaining integer \( 3 \) in our example is identified as the next (second) prime \( \pi_2 \). Then all the \( (N - 2)/2 \) multiples of \( \pi_2 = 3 \) are removed. The next remaining number is \( \pi_3 = 5 \), so all multiples of \( \pi_3 = 5 \) are removed (i.e., \( \cdot 5, \cdot 10, \cdot 15, \cdot 25 \) etc., \( \cdots \)). This process is repeated until all the numbers of the list have been either canceled, or identified as prime.

As the word sieve implies, this sifting process takes a heavy toll on the integers, rapidly pruning the non-primes. In four iterations of the sieve algorithm, all the primes below \( N = 50 \) are identified (in red). The final set of primes are displayed in the caption of Fig. 1.5.

Once a prime greater than \( N/2 \) has been identified, we may stop, since twice that prime is greater than \( N \), the maximum number under consideration. Once you have reached \( \sqrt{N} \) all the primes have been identified (this follows from the fact that the next prime \( \pi_n \) is multiplied by an integer \( n = 1, \cdots, N \)). Once this number \( n\pi_n > N \) the list has been exhausted, which must be \( n < \sqrt{N} \).

When using a computer, memory efficiency and speed are the main considerations. There are various schemes for making the sieve more efficient. For example the recursion \( n\pi_k = (n-1)\pi_k + \pi_k \) will speed up the process, by replacing the multiply with an addition of \( \pi_k \).

Two fundamental theorems of primes: Early theories of numbers revealed two fundamental theorems (there are many more than two), as discussed in Section 1.2.2 and 2.1.1 (p. 163). The

\(^{38}\)https://en.wikipedia.org/wiki/Sieve_of_Eratosthenes\#Euler.27s_Sieve
1. Write \( N \) integers from 2 to \( N - 1 \). Let \( k = 1 \). The first element \( \pi_1 = 2 \) is a prime. Cross out \( n \cdot \pi_n \): (e.g., \( n \cdot \pi_1 = 4, 8, 16, 32, \ldots \)).

\[
\begin{array}{cccccccccc}
2 & 3 & 5 & 7 & 11 & 13 & 17 & 19 & 23 & 29 \\
11 & 13 & 17 & 19 & 23 & 29 & 31 & 37 & 41 & 43 \\
21 & 23 & 29 & 31 & 37 & 43 & 47 & 53 & 59 & 61 \\
31 & 33 & 37 & 41 & 43 & 47 & 53 & 59 & 61 & 67 \\
41 & 43 & 47 & 53 & 59 & 61 & 67 & 71 & 73 & 79 \\
\end{array}
\]

2. Let \( k = 2 \), \( \pi_2 = 3 \). Cross out \( n\pi_k \) (9, 15, 21, 27, 33, 39, 45, \ldots):

\[
\begin{array}{cccccccccccc}
2 & 3 & 5 & 7 & 11 & 13 & 17 & 19 & 23 & 29 & 31 & 37 & 41 & 43 & 47 & 53 & 59 & 61 \\
21 & 23 & 29 & 31 & 37 & 43 & 47 & 53 & 59 & 61 & 67 & 71 & 73 & 79 & 83 & 89 & 97 \\
31 & 33 & 37 & 41 & 43 & 47 & 53 & 59 & 61 & 67 & 71 & 73 & 79 & 83 & 89 & 97 & 101 & 103 \\
\end{array}
\]

3. Let \( k = 3 \), \( \pi_3 = 5 \). Cross out \( n\pi_3 \). (Cross out 25, 35).

\[
\begin{array}{cccccccccccc}
2 & 3 & 5 & 7 & 11 & 13 & 17 & 19 & 23 & 29 & 31 & 37 & 41 & 43 & 47 & 53 & 59 & 61 \\
21 & 23 & 29 & 31 & 37 & 43 & 47 & 53 & 59 & 61 & 67 & 71 & 73 & 79 & 83 & 89 & 97 \\
31 & 33 & 37 & 41 & 43 & 47 & 53 & 59 & 61 & 67 & 71 & 73 & 79 & 83 & 89 & 97 & 101 & 103 \\
\end{array}
\]

4. Finally let \( k = 4 \), \( \pi_4 = 7 \). Remove \( n\pi_4 \): (Cross out 49). Thus there are 15 primes less than \( N = 50 \): \( \pi_k = \{2, 3, 5, 7, 11, 13, 17, 19, 23, 29, 31, 37, 41, 43, 47\} \).

Figure 1.5: *Sieve of Eratosthenes for the case of \( N = 50 \).*

first of these is the fundamental theorem of arithmetic, which says that every integer greater than 1 may be uniquely factored into a product of primes

\[
n = \prod_{k=1}^{K} \pi_k^{\beta_k},
\]

where \( k = 1, \ldots, K \) indexes the integer’s \( K \) prime factors \( \pi_k \in \mathbb{P} \). Typically prime factors appear more than once, for example \( 25 = 5^2 \). To make the notation compact we define the multiplicity \( \beta_k \) of each prime factor \( \pi_k \). For example \( 2312 = 2^3 \cdot 7^2 = \pi_1^3 \pi_7^2 \) (i.e., \( \pi_1 = 2, \beta_1 = 3; \pi_7 = 17, \beta_7 = 2 \)) and \( 2313 = 3^2 \cdot 257 = \pi_2^2 \pi_{55} \) (i.e., \( \pi_2 = 3, \beta_2 = 2; \pi_{55} = 257, \beta_{55} = 1 \)). Our demonstration of this is empirical, using the Matlab/Octave `factor(N)` routine, which factors \( N \).

What seems amazing is the unique nature of this theorem. Each counting number is uniquely represented as a product of primes. There cannot be two integers with the same factorization. Once you multiply the factors out, the result is unique (\( N \)). Note that it’s easy to multiply integers (e.g., primes), but expensive to factor them. And factoring the product of three primes is significantly more difficult than factoring two.

---

If you wish to be a Mathematician, you need to learn how to prove theorems. If you’re a Physicist, you are happy that someone else has already proved them, so that you can use the result.
CHAPTER 1. INTRODUCTION

Factoring is much more expensive than division. This is not due to the difference in the cost of division over multiplication. The division of one number by another is only slightly more expensive than multiplying two numbers, maybe by a factor of 2.\(^{40}\) Dividing the product of two primes, given one, is trivial, slightly more expensive than multiplying. Factoring the product of two primes is nearly impossible, as one needs to know what to divide by. This is an explosion in probability, measured by Shannon Entropy. Factoring means dividing by some integer and obtaining another integer with remainder zero. Thus one could factor a product of primes \(N = \pi_k \pi_l\) by doing \(M\) divisions, where \(M\) is the number of primes less than \(N\). This assumes the list of primes less than \(N\) are known.

But the utility has to do with the density of primes (the prime number theorem, i.e., Gauss’ hypothesis). If we were simply looking up a few numbers from a short list of primes, it would be easy, but the density of primes among the integers, is logarithmic (>1 per octave, Section 2.1.1, p. 165).

This take us to the Prime Number Theorem (PNT). The security problem is the reason why these two theorems are so important: 1) Every integer has a unique representation as a product of primes, and 2) the number of primes is very dense (their are a very large number of them, the density is proportional to the log of their number). Security reduces to the “needle in the haystack problem,” the cost of a search. A more formal way to measure the density is known as Shannon entropy, couched in terms of the expected value of the log-probability of events “What is the probability of finding a prime between \(N\) and \(2N\)?”\(^{41}\)

1.2.4 Lec 5: Greatest common divisor (Euclidean algorithm)

The Euclidean algorithm is a method to find the greatest common divisor (GCD) \(k\) between two integers \(n, m\), denoted \(k = \text{gcd}(n, m)\), where \(n, m, k \in \mathbb{N}\). For example 15 = gcd(30, 105) since when factored \((30, 105) = (2 \cdot 3 \cdot 5, 7 \cdot 3 \cdot 5) = 3 \cdot 5 \cdot (2, 7) = 15 \cdot (2, 7)\). The Euclidean algorithm was known to the Chinese (i.e., not discovered by Euclid) (Stillwell, 2010, p. 41).

Why is the GCD important? Computing the GCD is simple, whereas a full factoring is extremely expensive. The GCD is important, precisely because of the fundamental difficulty of factoring large integers into their primes. This utility surfaces when the two numbers are composed of very large primes. When two integers have no common factors they are said to be coprime, thus their GCD is 1. The ratio of two integers which are coprime is automatically in reduced form (they have no common factors).

For example 4/2 ∈ \(\mathbb{Q}\) is not reduced since 2=gcd(4,2). Canceling out the common factor 2, gives the reduced form 2/1=2 ∈ \(\mathbb{N}\). Thus if we wish to form the ratio of two integers, first compute the gcd, and remove it from the two numbers, to form the ratio. This assures the rational number is in its reduced form. If the GCD were 10\(^3\) digits it is obvious that the common factor must be removed to greatly simplify further computation.

An example: Take the two integers \([873, 582]\). In factored form these are \([\pi_{25} \cdot 3^2, \pi_{25} \cdot 3 \cdot 2]\). Given the factors, we see that the largest common factor is \(\pi_{25} \cdot 3 = 291\) (\(\pi_{25} = 97\)). When we take the ratio of the two numbers this common factor cancels

\[
\frac{873}{582} = \frac{\pi_{25} \cdot 3 \cdot 3}{\pi_{25} \cdot 3 \cdot 2} = \frac{3}{2} = 1.5.
\]

\(^{40}\)https://streamcomputing.eu/blog/2012-07-16/how-expensive-is-an-operation-on-a-cpu/

\(^{41}\)When I understand this better, I’ll do a better job of explaining it.
Of course if we divide 582 into 873 this we will numerically obtain the answer $1.5 \in \mathbb{F}$.

**Exercise:** Show that in Matlab $\text{rat}(873/582) = 1+1/(-2)$ gives the wrong answer. Since $\text{factor}(873) = 3 \cdot 3 \cdot 97$ and $\text{factor}(582) = 2 \cdot 3 \cdot 97$,

$$3/2 = 1 + 1/2$$ is the correct answer. But due to rounding methods, it is not 3/2. As an example, in Matlab/Octave $\text{rat}(3/2)=2+1/(-2)$. One would expect $\text{rat}(3/2)=1+1/2$.

Matlab’s rat command using rounding rather than the floor function.

**Greatest Common Divisor:** $k=\gcd(m,n)$

- **Examples** ($m, n, k \in \mathbb{Z}$):
  - $\gcd(13 \cdot 5, 11 \cdot 5) = 5$ (The common 5 is the gcd)
  - $\gcd(13 \cdot 10, 11\cdot 10) = 10$ (The gcd(130,110) = 10 = 2 \cdot 5, is not prime)
  - $\gcd(1234,1024) = 2(1234 = 2 \cdot 617, 1024 = 2^{10})$
  - $\gcd(\pi k_m, \pi k_n) = \pi k$
  - $k=\gcd(m,n)$ is the part that cancels in the fraction $m/n \in \mathbb{F}$
  - $m/\gcd(m,n) \in \mathbb{Z}$

- **Coprimes** ($m \perp n$) are numbers with no common factors: i.e., $\gcd(m,n)=1$
  - The gcd of two primes is always 1: $\gcd(13,11) = 1$, $\gcd(\pi m, \pi n) = 1$ ($m \neq n$)
  - $m = 7 \cdot 13, n = 5 \cdot 19 \Rightarrow (7 \cdot 13) \perp (5 \cdot 19)$
  - If $m \perp n$ then $\gcd(m,n) = 1$
  - If $\gcd(m,n) = 1$ then $m \perp n$

- **The GCD may be extended to polynomials:** e.g., $\gcd(ax^2 + bx + c, \alpha x^2 + \beta x + \gamma)$
  - $\gcd((x-3)(x-4), (x-3)(x-5))=(x-3)$
  - $\gcd(x^2-7x+12, 3(x^2-8x+15))=3(x-3)$
  - $\gcd(x^2-7x+12, (3x^2-24x+45))=3(x-3)$
  - $\gcd((x-2 \pi)(x-4), (x-2 \pi)(x-5))=(x-2 \pi)$ (Needs long division)

**Figure 1.6:** The Euclidean algorithm for finding the GCD of two numbers is one of the oldest algorithms in mathematics, and is highly relevant today. It is both powerful and simple. It was used by the Chinese during the Han dynasty (Stillwell, 2010, p. 70) for reducing fractions. It may be used to find pairs of integers that are coprime (their gcd must be 1), and it may be used to identify factors of polynomials by long division. It has an important sister algorithm called the continued fraction algorithm (CFA), that is so similar in concept that Gauss referred to the Euclidean algorithm as the "continued fraction algorithm" (Stillwell, 2010, p. 48).

**Euclidean algorithm:** The algorithm is best explained by a trivial example: Let the two numbers be 6, 9. At each step the smaller number (6) is subtracted from the larger (9) and the difference (the remainder) and the smaller numbers are saved. This process continues until the two resulting numbers are equal, at which point the GCD equals that final number. If we were to take one more step, the final numbers would be the gcd and zero. For our example step 1 gives 9-6=3, leaving 6 and 3. Step 2 gives 6-3=3 and 3. Since the two numbers are the same, the GCD=3. If we take one more difference we obtain (3,0). We can easily verify this result since this example is easily factored (e.g., $3 \cdot 3 \cdot 3 \cdot 2 = 3(3,2)$. It may be numerically verified using the Matlab/Octave GCD command $\gcd(6,9)$, which returns 3.
**Matrix method:** The GCD may be written a matrix recursion given the starting vector \((m_0, n_0)^T\). The recursion is then

\[
\begin{bmatrix}
m_{k+1} \\
n_{k+1}
\end{bmatrix} =
\begin{bmatrix}
1 & -1 \\
0 & 1
\end{bmatrix}
\begin{bmatrix}
m_k \\
n_k
\end{bmatrix}
\]

This recursion continues until \(m_{k+1} < n_{k+1}\), at which point \(m\) and \(n\) must be swapped. Because the operation depends on the output, this is a nonlinear recursion (Postulate P1 (Linear/nonlinear) of Section 3.5.1, p. 205).

The direct method is inefficient because in recursively subtract \(n\) many times until the resulting \(m\) is less than \(n\), as shown in Fig. 2.1 (p. 168). It also must test for \(m < n\) at each iteration, and then swap \(m\) and \(n\) once that condition is met. This recursion is repeated until \(m_{k+1} = 0\). At that stage the GCD is \(n_{k+1}\). Figure 2.1 (p. 168), along with the above matrix relation, gives insight into the Euclidean Algorithm.

Below is a simple Matlab/Octave code to find \(k=gcd(m,n)\) based on the strict definition of the EA as described by Stillwell (2010):

```matlab
function k = gcd(m,n)
    while m ~= 0
        A=m; B=n;
        m=max(A,B); n=min(A,B);  \%m>n
        m=m-n;
    end
end
```

This program looping until \(m = 0\). An much more efficient method is described in Section 2.1.3, p. 166, using the `floor()` function, which is called *division with rounding*.

### 1.2.5 Lec 6: Continued fraction algorithm

In its simplest form, the *Continued fraction algorithm* (CFA) starts from a single real decimal number \(x \in \mathbb{R}\), and recursively expands it as a fraction \(x \in \mathbb{F}\). Thus the CFA It is used in finding rational approximations to any real number. For example, \(\pi \approx 22/7\). The GCD (the Euclidean algorithm) on the other hand operates on a pair of integers \(m, n \in \mathbb{N}\) and finds their greatest common divisor \(k \in \mathbb{N}\). Thus \(m/k, n/k \in \mathbb{N}\), reducing the ratio to its irreducible form, since \(m/k \perp n/k\).

Despite this seeming large difference in the two algorithms, apparently the CFA is closely related to the Euclidean algorithm (the GCD), so closely in fact, that Gauss referred to the Euclidean algorithm as the Continued fraction algorithm (Stillwell, 2010, P. 48). At first glance it is not clear why Gauss would call the CFA the Euclidean algorithm. One must assume that Gauss had some deeper insight into this relationship. If so, that insight would be valuable to understand. \(^{42}\)

In the following we refine the description of the CFA and give examples that go beyond the simple cases of expanding numbers. The CFA of any positive number, say \(x_o \in \mathbb{R}^+\), is defined as follows:

1. Start with \(n = 0\) and a positive input target \(x_o \in \mathbb{R}^+\).
2. Define \(a_n = \text{round}(x_n)\), which rounds to the nearest integer.

\(^{42}\) The resolution of this interrelationship is still unresolved.
3. Define \( r_n = x_n - a_n \), thus \(-0.5 \leq r_n \leq 0.5\). If \( r_n = 0 \), the recursion terminates.

4. Define \( x_{n+1} = 1/r_n \) and return to step 2, with \( n = n + 1 \).

The recursion may continue to any desired accuracy, since convergence is guaranteed.

An example: Let \( x_0 \equiv \pi \approx 3.14159\ldots \). Thus \( a_0 = 3, r_0 = 0.14159, x_1 = 7.065 \approx 1/r_0 \), and \( a_1 = 7 \). If we were to stop here we would have

\[
\hat{\pi}_2 \approx 3 + \frac{1}{7 + 0.0625} \approx 3 + \frac{1}{7} = \frac{22}{7}. \tag{1.7}
\]

This approximation of \( \pi \approx 22/7 \) has a relative error of 0.04%:

\[
\frac{\frac{22}{7} - \pi}{\pi} = 4 \times 10^{-4}.
\]

For the next approximation we continue by reciprocating the remainder \( 1/0.0625 \approx 15.9966 \) which rounds to 16 giving a negative remainder of \( \approx -1/300 \), resulting in the second approximation

\[
\hat{\pi}_3 \approx 3 + 1/(7 + 1/16) = 3 + 16/(7 \cdot 16 + 1) = 3 + 16/113 = \frac{355}{113}.
\]

Note that if we had truncated 15.9966 to 15, the remainder would have been much larger, but positive, resulting in a much less accurate rational approximation. It follows that there can be a dramatic difference, depending on the rounding scheme, which must be specified.

**Rational approximation examples**

\[
\hat{\pi}_2 = \frac{22}{7} = [3., 7] \quad \approx \hat{\pi}_2 + O(1.3 \times 10^{-3})
\]

\[
\hat{\pi}_3 = \frac{355}{113} = [3., 7, 16] \quad \approx \hat{\pi}_3 + O(2.7 \times 10^{-7})
\]

\[
\hat{\pi}_4 = \frac{104348}{33215} = [3., 7, 16, -249] \quad \approx \hat{\pi}_4 + O(3.3 \times 10^{-10})
\]

Figure 1.7: The expansion of \( \pi \) to various orders, using the CFA, along with the order of the error of each rational approximation, with rounding. For example \( \hat{\pi}_2 = 22/7 \) has an absolute error (\(|22/7 - \pi|\)) of about 0.13%.

**Notation:** Writing out all the fractions can become tedious. For example, expanding \( e = 2.7183\ldots \) using the Matlab/Octave command `rat(exp(1))` gives the approximation

\[
\exp(1) = 3 + 1/(-4 + 1/(2 + 1/(5 + 1/(-2 + 1/(-7))))) - O\left(1.75 \times 10^{-6}\right),
\]

or in the compact notation, \( \hat{e}_6 \approx [3., -4, 2, 5, -2, -7] \). Note that the leading integer part may be noted by an optional decimal point or semicolon.\(^{43}\) If the process is carried further, the values of \( a_n \in \mathbb{N} \) give increasingly more accurate rational approximations.

\(^{43}\)Unfortunately Matlab/Octave does not support the bracket notation.
CHAPTER 1. INTRODUCTION

**Rounding schemes:** In Matlab/Octave there are five different rounding schemes (i.e., mappings): \( \text{round}(c), \text{fix}(c), \text{floor}(c), \text{ceil}(c), \text{roundb}(c) \) with \( c \in \mathbb{C} \). If the rounding-down (\( \text{floor}() \)) is used \( \hat{\pi}_{12} = [3, 7, 15, 1, 292, 1, 1, 2, 1, 3, 1] \), whereas true rounding to the nearest integer (\( \text{round}() \)) gives \( \hat{\pi}_8 = [3, 7, 16, -294, 3, -4, 5, -15] \). Thus \( \text{round}() \) introduces negative coefficients when a number rounds up to the nearest integer.

**Exercise:** Based on several examples, which rounding scheme is the most accurate? Explain why. **Sol:** Rounding will give a smaller remainder at each iteration, resulting in a smaller net error and thus faster convergence.

When the CFA is applied and the expansion terminates \( (r_n = 0) \), the target is rational. When the expansion does not terminate (which is not always easy to determine, as the remainder may be ill-conditioned due to small numerical rounding errors), the number is irrational. Thus the CFA has important theoretical applications regarding irrational numbers. You may explore this using Matlab’s \texttt{rats(pi)} command.

One of the useful things about the procedure, besides its being so simple, are its generalizations, one of which will be discussed in Section 2.1.3 (p. 166).

Besides these five basic rounding schemes, there are two other important \( \mathbb{R} \to \mathbb{N} \) functions (i.e., mappings): \( \text{mod}(x, y), \text{rem}(x, y) \) with \( x, y \in \mathbb{R} \), which will be needed later. The base 10 numbers may be generated from the counting numbers using \( y = \text{mod}(x, 10) \).

**Exercise:**

1. Show how to generate a base-10 real number \( y \in \mathbb{R} \) from the counting numbers \( \mathbb{N} \) using the \( m = \text{mod}(n, 10) + k 10 \) with \( n, k \in \mathbb{N} \). **Sol:** Every time \( n \) reaches 10, \( m = 0 \). Keep track of this by adding 1 to \( k \).

2. How would you generate binary numbers (base 2) using the \( \text{mod}(x, b) \) function? **Sol:** Use the same method as in the first example above, but with \( b = 2 \).

3. How would you generate hexadecimal numbers (base 16) using the \( \text{mod}(x, b) \) function? **Sol:** Use the same method as in the first example above, but with \( b = 16 \).

4. Write out the first 19 numbers in hex notation, starting from zero. **Sol:** 0, 1, 2, 3, 4, 5, 6, 7, 8, 9, A, B, C, D, E, F, 10, 11, 12, . . .

5. What is \( FF_{16} \) in decimal notation? **Sol:** Using the Matlab/octave function \texttt{hex2dec(’ff’)} is \( 255_{10} \).

**Symmetry:** A continued fraction expansion can have a high degree of symmetry. For example, the CFA of

\[
R_1 = \frac{1 + \sqrt{5}}{2} = 1 + \frac{1}{1 + \frac{1}{1 + \cdots}} = 1.618033988749895 \cdots , \tag{1.8}
\]
Here $a_n$ in the CFA is always 1 ($R_1 \equiv [1, 1, 1, \cdots]$), thus the sequence will not terminate, proving that $\sqrt{5} \in \mathbb{I}$. A related example is $R_2 \equiv \text{rat}(1 + \sqrt{2})$, which gives $R_2 = [2, 2, 2, 2, \cdots]$.

When expanding a target irrational number ($x_o \in \mathbb{I}$), and the CFA is truncated, the resulting rational fraction approximates the irrational target. For the example above, if we truncate at three coefficients ([1; 1, 1]) we obtain

$$1 + \frac{1}{1 + \frac{1}{1+0}} = 1 + 1/2 = 3/2 = 1.5 = \frac{1 + \sqrt{5}}{2} + 0.118\ldots.$$ 

Truncation after six steps gives

$$[1., 1, 1, 1, 1, 1, 1] = 13/8 \approx 1.6250 = \frac{1 + \sqrt{5}}{2} + .0070\ldots.$$ 

Because all the coefficients are 1, this example converges very slowly. When the coefficients are large (i.e., remainder small), the convergence will be faster. The expansion of $\pi$ is an example of faster convergence.

**In summary:** Every rational number $m/n \in \mathbb{F}$, with $m > n > 1$, may be uniquely expanded as a continued fraction, with coefficients $a_k$ determined using the CFA. When the target number is irrational ($x_o \in \mathbb{Q}$), the CFA does not terminate, thus each step produces a more accurate rational approximation, converging in the limit as $n \to \infty$.

Thus the CFA expansion is an algorithm that can, in theory, determine when the target is rational, but with an important caveat: one must determine if the expansion terminates. This may not be obvious. The fraction $1/3 = 0.33333\cdots$ is an example of such a target, where the CFA terminates yet the fraction repeats. It must be that

$$1/3 = 3 \times 10^{-1} + 3 \times 10^{-2} + 3 \times 10^{-3} + \cdots.$$ 

Here $3*3=9$. As a second example\(^{44}\)

$$1/7 = 0.142857142857142857142857\cdots = 142857 \times 10^{-6} + 142857 \times 10^{-12} + \cdots$$

Note that 142857 * 7 = 999999. This also works for $1/11 = 0.090909\cdots$ and $11 \times 0.090909 = 999999$. We might conclude that when the sequence of digits repeats, the sequence is predictable, thus it must be rational.

**WEEK 3**

1.2.6 Lec 7: Pythagorean triplets (Euclid’s formula)

Euclid’s formula is a method for finding three integer lengths $[a, b, c] \in \mathbb{N}$, that satisfy Eq. 1.1. It is important to ask “Which set are the lengths [a,b,c] drawn from?” There is a huge difference, both

\(^{44}\)Taking the Fourier transform of the target number, represented as a sequence, could help to identify an underlying periodic component. The number $1/7 \leftrightarrow [1, 4, 2, 8, 5, 7]_6$ has a 50 [dB] notch at $0.8\pi$ [rad] due to its 6 digit periodicity, carried to 15 digits (Matlab/Octave maximum precision), Hamming windowed, and zero padded to 1024 samples.
practical and theoretical, if they are from the real numbers \( \mathbb{R} \), or the counting numbers \( \mathbb{N} \). Given \( p, q \in \mathbb{N} \) with \( p > q \), the three lengths \([a, b, c] \in \mathbb{N}\) of Eq. 1.1 are given by

\[
a = p^2 - q^2, \quad b = 2pq, \quad c = p^2 + q^2. \quad (1.9)
\]

This result may be directly verified, since

\[
\]
or

\[
p^4 + q^4 + 2p^2q^2 = p^4 + q^4 - 2p^2q^2 + 4p^2q^2.
\]

Thus, Eq. 1.9 is easily proven, once given. Deriving Euclid’s formula is obviously more difficult.

A well known example is the right triangle depicted in Fig. 1.8, defined by the integers \([3, 4, 5] \in \mathbb{N}\), having angles \([0.54, 0.65, \pi/2]\) [rad], which satisfies Eq. 1.1 (p. 18). As quantified by Euclid’s formula (Eq. 1.9), there are an infinite number of Pythagorean triplets (PTs). Furthermore the seemingly simple triangle, having angles of \([30, 60, 90]\) \( \in \mathbb{N}\) [deg] (i.e., \([\pi/6, \pi/3, \pi/2] \in \mathbb{I}\) [rad]), has one irrational (I) length \(([1, \sqrt{3}, 2])\).

The technique for proving Euclid’s formula for PTs \([a, b, c] \in \mathbb{Q}\), derived in Fig. 2.3 (p. 171) of Section 2.2.1, is much more interesting than the PTs themselves.

The set from which the lengths \([a, b, c]\) are drawn was not missed by the Indians, Chinese, Egyptians, Mesopotamians, Greeks, etc. Any equation whose solution is based on integers is called a Diophantine equation, named after the Greek mathematician Diophantus of Alexandria (c250 CE).

A stone tablet having the numbers engraved on it, as shown in Table 1.9, was discovered in Mesopotamia, from the 19th century [BCE], and cataloged in 1922 by George Plimpton.\(^{45}\) These numbers are \(a\) and \(c\) pairs from PTs \([a,b,c]\). Given this discovery, it is clear that the Pythagoreans were following those, long before them. Recently a second similar stone, dating between 350 and 50 [BCE] has been reported, that indicates early calculus on the orbit of Jupiter’s moons.\(^{46}\)

**1.2.7 Lec 8: Pell’s Equation**

There is a venerable history for Pell’s equation

\[
x_n^2 - Ny_n^2 = (x_n - \sqrt{N}y_n)(x_n + \sqrt{N}y_n) = 1, \quad (1.10)
\]


1.2. STREAM 1: NUMBER SYSTEMS (10 LECTURES)

<table>
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<th>$c$</th>
</tr>
</thead>
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</tr>
<tr>
<td>1771</td>
<td>3229</td>
</tr>
<tr>
<td>56</td>
<td>106</td>
</tr>
</tbody>
</table>

Figure 1.3: Pairs in Plimpton 322

Figure 1.9: “Plimpton-322” is a stone tablet from 1800 BCE, displaying $a$ and $c$ values of the Pythagorean triplets $[a, b, c]$, with the property $b = \sqrt{c^2 - a^2} \in \mathbb{N}$. Several of the $c$ values are primes, but not the $a$ values. The stone is item 322 (item 3 from 1922), from the collection of George A. Plimpton. –Stillwell (2010)

with non-square $N \in \mathbb{N}$ specified and $x, y \in \mathbb{N}$ unknown. It is suspected that Pell’s equation is directly related to the Euclidean algorithm, as applied to polynomials having integer coefficients (Stillwell, 2010, 48). For example, with $N = 2$, one solution is $x = 17, y = 12 (17^2 - 2 \cdot 12^2 = 1)$.

A 2x2 matrix recursion algorithmic, likely due to the Chinese, was used by the Pythagoreans to investigate the $\sqrt{N}$

$$
\begin{bmatrix}
    x_n \\
    y_n
\end{bmatrix} = \begin{bmatrix} 1 & N \\ 1 & 1 \end{bmatrix} \begin{bmatrix} x_{n-1} \\
    y_{n-1}
\end{bmatrix}.
$$

Starting with $[x_0, y_0]^T = [1, 0]^T$, results in solutions of Pell’s equations (Stillwell, 2010, p. 44). Their approach was likely motivated by the Euclidean algorithm (GCD, p. 40), since $y_n/x_n \to \sqrt{2}$ (Stillwell, 2010, p. 37, 55).

Note that this is a composition method, of 2x2 matrices, since the output of one matrix multiply is the input to the next. They key question what is the relationship between Pell’s equation and the linear recursion? Is it that Pell’s equation may be trivially factored? There must be some simple way to prove that Eqs. 1.11 and 1.10 are equivalent, as demonstrated on Section 2.2.3 (p. 173).

**Asian solutions:** The first solution of Pell’s equation was published by Brahmagupta (c628), who independently discovered the equation (Stillwell, 2010, p. 46). Brahmagupta’s novel solution introduced a different composition method (Stillwell, 2010, p. 69), and like the Greek result, these solutions were incomplete.

Then in 1150CE, Bâskara II obtained solutions using Eq. 1.11 (Stillwell, 2010, p.69). This is the solution method we shall explore here, as summarized in Fig. 1.10.

The best way to see how this recursion results in solutions to Pell’s equation, is by example.
Initializing the recursion with the trivial solution \([x_0, y_0]^T = [1, 0]^T\), gives

\[
\begin{bmatrix}
  x_1 \\
  y_1
\end{bmatrix} = \begin{bmatrix} 1 \\
  1 \end{bmatrix} = \begin{bmatrix} 1 & 2 \\
  1 & 1 \end{bmatrix} \begin{bmatrix} 1 \\
  0 \end{bmatrix} \quad 1^2 - 2 \cdot 1^2 = -1
\]

\[
\begin{bmatrix}
  x_2 \\
  y_2
\end{bmatrix} = \begin{bmatrix} 3 \\
  2 \end{bmatrix} = \begin{bmatrix} 1 & 2 \\
  1 & 1 \end{bmatrix} \begin{bmatrix} 1 \\
  1 \end{bmatrix} \quad 3^2 - 2 \cdot 2^2 = 1
\]

\[
\begin{bmatrix}
  x_3 \\
  y_3
\end{bmatrix} = \begin{bmatrix} 7 \\
  5 \end{bmatrix} = \begin{bmatrix} 1 & 2 \\
  1 & 1 \end{bmatrix} \begin{bmatrix} 3 \\
  2 \end{bmatrix} \quad (7)^2 - 2 \cdot (5)^2 = -1
\]

\[
\begin{bmatrix}
  x_4 \\
  y_4
\end{bmatrix} = \begin{bmatrix} 17 \\
  12 \end{bmatrix} = \begin{bmatrix} 1 & 2 \\
  1 & 1 \end{bmatrix} \begin{bmatrix} 7 \\
  5 \end{bmatrix} \quad 17^2 - 2 \cdot 12^2 = 1
\]

\[
\begin{bmatrix}
  x_5 \\
  y_5
\end{bmatrix} = \begin{bmatrix} 41 \\
  29 \end{bmatrix} = \begin{bmatrix} 1 & 2 \\
  1 & 1 \end{bmatrix} \begin{bmatrix} 17 \\
  12 \end{bmatrix} \quad (41)^2 - 2 \cdot (29)^2 = -1
\]

Thus the recursion results in a modified version of Pell’s equation

\[
x_n^2 - 2y_n^2 = (-1)^n, \quad (1.12)
\]

where only even values of \(n\) are solutions. This sign change had no effect on the Pythagoreans’ goal, since they only cared about the ratio \(y_n/x_n \rightarrow \pm \sqrt{2}\).

**Modified recursion:** We can repair the recursion to fix the \((-1)^n\) problem by multiplying the 2x2 matrix by \(j = \sqrt{-1}\), which results in Pell’s equation for every step of the recursion, giving

\[
\begin{bmatrix}
  x_1 \\
  y_1
\end{bmatrix} = j \begin{bmatrix} 1 \\
  1 \end{bmatrix} = j \begin{bmatrix} 1 & 2 \\
  1 & 1 \end{bmatrix} \begin{bmatrix} 1 \\
  0 \end{bmatrix} \quad j^2 - 2 \cdot j^2 = 1
\]

\[
\begin{bmatrix}
  x_2 \\
  y_2
\end{bmatrix} = j^2 \begin{bmatrix} 3 \\
  2 \end{bmatrix} = j \begin{bmatrix} 1 & 2 \\
  1 & 1 \end{bmatrix} \begin{bmatrix} j \end{bmatrix} \quad 3^2 - 2 \cdot 2^2 = 1
\]

\[
\begin{bmatrix}
  x_3 \\
  y_3
\end{bmatrix} = j^3 \begin{bmatrix} 7 \\
  5 \end{bmatrix} = j \begin{bmatrix} 1 & 2 \\
  1 & 1 \end{bmatrix} j^2 \begin{bmatrix} 3 \\
  2 \end{bmatrix} \quad (7j)^2 - 2 \cdot (5j)^2 = 1
\]

\[
\begin{bmatrix}
  x_4 \\
  y_4
\end{bmatrix} = j^4 \begin{bmatrix} 17 \\
  12 \end{bmatrix} = j \begin{bmatrix} 1 & 2 \\
  1 & 1 \end{bmatrix} j^3 \begin{bmatrix} 7 \\
  5 \end{bmatrix} \quad 17^2 - 2 \cdot 12^2 = 1
\]

\[
\begin{bmatrix}
  x_5 \\
  y_5
\end{bmatrix} = j^5 \begin{bmatrix} 41 \\
  29 \end{bmatrix} = j \begin{bmatrix} 1 & 2 \\
  1 & 1 \end{bmatrix} j^4 \begin{bmatrix} 17 \\
  12 \end{bmatrix} \quad (41j)^2 - 2 \cdot (29j)^2 = 1
\]

Figure 1.10: This summarizes the solution of Pell’s equation for \(N = 2\) using a slightly modified matrix recursion. Note that \(x_n/y_n \rightarrow \sqrt{2}\) as \(n \rightarrow \infty\), which was what the Pythagoreans were pursuing, and that the odd solutions are pure-imaginary.

**Solution to Pell’s equation:** By multiplying the matrix by \(1j\), all the solutions to Pell’s equation are determined. This solution is shown in Fig. 1.10 (p. 48) for the case of \(N = 2\). straightforward and is easily verified using Matlab/Octave. From Fig. 1.10 we can see that every output this slightly modified matrix recursion gives solutions to Pell’s equation (Eq. 1.10). The \(1j\) factor corrects the alternation in sign, so every iteration yields a solution. For \(n = 0\) (the initial solution) \([x_0, y_0]^T = [1, 0]^T\) are solutions. This sign change had no effect on the Pythagoreans’ goal, since they only cared about the ratio \(y_n/x_n \rightarrow \pm \sqrt{2}\). These are easily computed by this recursion, and easily checked on a hand calculator (or using Matlab/Octave).
At each iteration, the ratio \( x_n/y_n \) approaches \( \sqrt{2} \) with increasing accuracy, coupling it to the CFA. The value of \( 41/29 \approx \sqrt{2} \), with a relative error of <0.03%. The solution for \( N = 3 \) is given in Appendix D.1.1 (p. 250).

**Relations to digital signal processing:** Today we recognize Eq. 1.11 as a *difference equation*, which is a pre-limit (pre Stream 3) form of differential equation. The Greek 2x2 form is an early precursor to 17th and 18th century developments in linear algebra. Thus the Greek’s recursive solution for the \( \sqrt{2} \) and BhÁškara’s (1030 CE) solution of Pell’s equation, is an early precursor to discrete-time processing, as well as to calculus.

There are similarities between Pell’s Equation and the Pythagorean theorem. As we shall see in Chapter 2, Pell’s equation is related to the geometry of a hyperbola, just as the Pythagorean equation is related to the geometry of a circle. One would assume there is a Euclidean formula for the case of Pell’s Equations, since these are all conic sections with closely related geometry.

**Eigen-analysis:** The key to the analysis of such equations is called the *eigen-analysis* method. This is summarized in Appendix C (p. 245), applied to matrix \( A \). Given a 2x2 matrix, first its eigen-values \( \lambda_{\pm} \) and then its eigen-vectors \( e_{\pm} \) are determined. Once known, the matrix \( A \) may be transformed into a diagonal matrix \( \Lambda \)

\[
\Lambda = E^{-1}AE, \tag{1.13}
\]

where \( E \) and \( \Lambda \) are defined by the eigen-value equation for \( A \)

\[
Ae_{\pm} = \lambda_{\pm}e_{\pm}, \tag{1.14}
\]

\( E = [e_+, e_-] \) are formed, and \( \Lambda \) is a diagonal matrix of sorted eigen-values \( \lambda_k \). As an example if we take

\[
A = \begin{bmatrix} 1 & 1 \\ 1 & -1 \end{bmatrix}
\]

then the eigen-values are given by \( (1 - \lambda_{\pm})(1 + \lambda_{\pm}) = -1 \), thus \( \lambda_{\pm} = \sqrt{2} \). This method of eigen-analysis is discussed in Section 2.2.2 (p. 170) and Appendix D (p. 247).

The key idea of the 2x2 matrix solution, widely used in modern engineering, can be traced back to Brahmagupta’s solution of Pell’s equation, for arbitrary \( N \). Brahmagupta’s recursion, identical to that of the Pythagoreans’ \( N = 2 \) case (Eq. 1.11), eventually lead to the concept of linear algebra, defined by the simultaneous solutions of many linear equations. The recursion by the Pythagoreans (6th BCE) predated the creation of algebra by al-KhwÁ‘rizmi (9th CE century) (Fig. 1.2) (Stillwell, 2010, p. 88).

**WEEK 4**

**1.2.8 Lec 9: Fibonacci sequence**

Another classic problem, formulated by the Chinese, was the Fibonacci sequence, generated by the relation

\[
f_{n+1} = f_n + f_{n-1}. \tag{1.15}
\]
Here the next number $f_{n+1} \in \mathbb{N}$ is the sum of the previous two. If we start from $[0, 1]$, this difference equation leads to the Fibonacci sequence $f_n = [0, 1, 1, 2, 3, 5, 8, 13, 21, 34, \ldots]$. Alternatively, if we define $y_{n+1} = x_n$, then Eq. 1.15 may be compactly represented by a 2x2 matrix equation recursion

$$
\begin{bmatrix}
x \\
y
\end{bmatrix}_{n+1} = \begin{bmatrix} 1 & 1 \\ 1 & 0 \end{bmatrix} \begin{bmatrix} x \\
y
\end{bmatrix}_n.
$$

(1.16)

We have moved the index outside the vector to save typing.

The correspondence of Eqs. 1.15 and 1.16 is easily verified. Starting with $[x_n, y_n]^T = [0, 1]^T$ we obtain for the first few steps

$$
\begin{bmatrix}
1 \\
0
\end{bmatrix}_1 = \begin{bmatrix} 1 & 1 \\ 1 & 0 \end{bmatrix} \begin{bmatrix} 0 \\
1
\end{bmatrix}, \quad \begin{bmatrix}
1 \\
1
\end{bmatrix}_2 = \begin{bmatrix} 1 & 1 \\ 1 & 0 \end{bmatrix} \begin{bmatrix} 1 \\
0
\end{bmatrix}, \quad \begin{bmatrix}
2 \\
1
\end{bmatrix}_3 = \begin{bmatrix} 1 & 1 \\ 1 & 0 \end{bmatrix} \begin{bmatrix} 1 \\
1
\end{bmatrix}, \quad \begin{bmatrix}
3 \\
2
\end{bmatrix}_4 = \begin{bmatrix} 1 & 1 \\ 1 & 0 \end{bmatrix} \begin{bmatrix} 1 \\
2
\end{bmatrix}, \ldots
$$

From the above $x_n = [0, 1, 1, 2, 3, 5, \ldots]$ is the Fibonacci sequence since the next $x_n$ is the sum of the previous two and the next $y_n$ is $x_n$.

![Fibonacci Spiral](https://en.wikipedia.org/wiki/Golden_spiral)

**Figure 1.11:** This construction is called the Fibonacci spiral. Note how it is constructed out of squares having areas given by the square of the Fibonacci numbers. In this way, the spiral is smooth and the radius increases as the Fibonacci numbers (e.g., $8=3+5$, $13 = 5+8$, etc.). (Adapted from https://en.wikipedia.org/wiki/Golden_spiral.)

**The mean-Fibonacci sequence:** Suppose that the Fibonacci sequence recursion is replace by the mean of the last two values, namely let

$$
f_{n+1} = \frac{f_n + f_{n-1}}{2}.
$$

(1.17)

This seems like a small change. But how does the solution differ? Two answer this question it is helpful to look at the corresponding 2x2 matrix.

**Exercise:** Find the 2x2 matrix corresponding to Eq. 1.17. **Sol:** In this case we find

$$
\begin{bmatrix}
x \\
y
\end{bmatrix}_{n+1} = \frac{1}{2} \begin{bmatrix} 1 & 1 \\ 2 & 0 \end{bmatrix} \begin{bmatrix} x \\
y
\end{bmatrix}_n.
$$

(1.18)

The eigen-values of this matrix are the roots of the binomial equation $-\lambda (1 - \lambda/2) - 1/2 = 0$ which has eigen-values $[1, -1/2]$. Thus $[x_n, y_n] = [1, 1]^T$ and $[x_n, y_n] = [1, 1]^T$ are each solutions.
Exercise: Starting from \([x_n, y_n]^T = [1, 0]^T\) compute the first 5 values of \([x_n, y_n]^T\). \textbf{Sol:} Here is a Matlab/Octave code for computing \(x_n\):

```matlab
x(1:2,1)=[1;0];
A=[1 1;2 0]/2;
for k=1:10; x(k+1)=A*x(:,k); end
```

which gives the rational \((x_n \in \mathbb{Q})\) sequence: 1, 1/2, 3/4, 5/8, 11/2^4, 21/2^5, 43/2^6, 85/2^7, 171/2^8, 341/2^9, 683/2^{10}, \cdots.

Exercise: Show that the solution to Eq. 1.17 is bounded, unlike that of the Fibonacci sequence, which diverges. Explain what is going on. \textbf{Sol:} Because the next value is the mean of the last two, the sequence is bounded. To see this one needs to compute the eigen-values of the matrix of Eq. 1.17.

Exercise: Use the formula for the generalized diagonalization of a matrix to find the general solution of the mean-Fibonacci sequence. \textbf{Sol:} The eigen-values are given by the roots of

\[ 0 = -\lambda(1/2 - \lambda) - 1/2 = (\lambda - 1/4)^2 - 9/16 \]

which are \([1, -1/2]\). A recursive equation such as the above is called a matrix composition.

By studying the eigen-values of Eq. 1.18 one finds that the steady state solution approaches 1. Namely \(f_n \rightarrow 1 = (f_{n-1} + f_{n-2})/2\) is the solution, as \(n \rightarrow \infty\). Namely the average of the last two values must approach 1 for large \(n\).

Summary: The GCD (Euclidean algorithm), Pell’s equation and the Fibonacci sequence may all be written as compositions of 2x2 matrices. Thus Pell’s equation and the Fibonacci sequence are special cases of 2x2 matrix composition

\[
\begin{bmatrix}
x \\
y
\end{bmatrix}_{n+1} = \begin{bmatrix}
a & b \\
c & d
\end{bmatrix} \begin{bmatrix}
x \\
y
\end{bmatrix}_n.
\]

This is an important and common thread of these early mathematical findings. It will turn out that this form plays a special role in physics, mathematics and engineering because such equations may be solved using the eigen-analysis method. The first several thousands of years of mathematical trial and error research, set the stage for this breakthrough, but it took a long time to fully appreciate the unity of these several contributions.

2) logic and truth tables. Perhaps put 1) in convolution section.

1.2.9 \textbf{Lec 10: Exam I (In class)
CHAPTER 1. INTRODUCTION

1.3 Algebraic Equations: Stream 2

The era from 1600-1850 (Fig. 1.21, p. 98) produced a stream of fundamental theorems. A few of the individuals who played a notable role in this development, in chronological (birth) order, include Galileo, Mersenne, Newton, d’Alembert, Fermat, Huygens, Descartes and Helmholtz. These individuals were some of the first to develop the basic ideas, in various forms, that were then later reworked into the proofs, that today we recognize as The fundamental theorems of mathematics.

1.3.1 Lec 11 The physics behind nonlinear Algebra (Euclidean geometry)

Following Stillwell’s history of mathematics, Stream 2 is geometry, which led to the merging of Euclid’s geometrical methods and the 9th century development of algebra by al-Khwarizmi (830 CE). This integration of ideas lead Descartes and Fermat to develop analytic geometry. While not entirely a unique and novel idea, it was late in coming, given what was known at that time.

The mathematics up to the time of the Greeks, documented and formalized by Euclid, served students of mathematics for more than two thousand years. Algebra and geometry were, at first, independent lines of thought. When merged, the focus returned to the Pythagorean theorem, generalized as analytic conic sections rather than as geometry in Euclid’s Elements. With the introduction of Algebra, numbers, rather than lines, could be used to represent a geometrical length. Thus the appreciation for geometry grew, given the addition of the rigorous analysis using numbers.

Physics inspires algebraic mathematics: The Chinese used music, art, and navigation to drive mathematics. Unfortunately much of our knowledge has been handed down either as artifacts, such as musical bells and tools, or mathematical relationships documented, but not created by, scholars such as Euclid, Archimedes, Diophantus, and perhaps Brahmagupta. With the invention of algebra by al-Khwārizmī (830CE) this paradigm did not shift, rather it expanded. During the 16th and 17th century, it had becoming clear that differential equations (DEs), such as the wave equation, can characterize a law of nature, at a single point in space and time. This principle was not obvious. A desire to understand motions of objects and planets precipitated many new discoveries. This period is organized in Fig. 1.4. Galileo investigated gravity and invented the telescope. For example the law of gravity (first formulated by Galileo to explain the dropping to two objects of different masses) must obey conservation of energy. Kepler investigated the motion of the planets. While Kepler was the first to appreciate that the planets were described by ellipses, it seems he under-appreciate the significance of this finding, and continued with his epicycle model of the planets. Newton (c1687) went on to show that there must be a gravitational potential between two masses \(m_1, m_2\), of the form

\[
\phi_g(r(t)) \propto \frac{m_1 m_2}{r(t)},
\]

where \(r = |x_1 - x_2|\) is the Euclidean distance between the two point masses at locations \(x_1\) and \(x_2\). Using algebra and his calculus, Newton formalized the equation of gravity, forces and motion (Newton’s three laws) and showed that Kepler’s discovery of planetary elliptical motion naturally follows from these laws. With the discovery of Uranus (1781) “Kepler’s theory was ruined.” (Stillwell, 2010, p. 23).

Once Newton proposed the basic laws of gravity, he proceed to calculate, for the first time, the speed of sound. This required some form of the pressure wave equation

\[
\frac{\partial^2}{\partial x^2} \phi(x, t) = \frac{1}{c_o^2} \frac{\partial^2}{\partial t^2} \phi(x, t),
\]
a key equation in mathematical physics. The speed of sound is

\[ c_o = \sqrt{\frac{\gamma P_o}{\rho_o}} = 343, \quad [\text{m/s}] \]

which is a function of the density \( \rho_o = 1.12 \, [\text{kg/m}^3] \) and the dynamic stiffness \( \eta P_o \) of air.\(^{47}\)

If we substitute for the pressure

\[ p(x, t) = e^{2\pi(f(t \pm kx))}, \quad (1.21) \]

where \( t \) is time and \( x \) is position, we find that \( k = 2\pi/\lambda = 2\pi f/c_o \), because \( f\lambda = c_o \).

This classic relation \( \lambda f = c \) is deceivingly simple, thus confusing, because \( k = 2\pi/\lambda \) becomes complex (has both real and imaginary parts) in dispersive media (e.g., acoustic waves in tubes when losses are considered (Kirchhoff, 1868); electron waves in silicon crystals (Brillouin, 1953)). In these more general cases, \( k(f) = 2\pi f/c \) must be replaced with a complex analytic function \( \kappa(s) \) of \( s \), i.e.,

\[ p(x, t) = P_o(s)e^{st e^{\pm \kappa(s)x}}, \quad (1.22) \]

such that the wave number \( \kappa(s) \) is a complex analytic function of the Laplace frequency \( s \). This is because electron “waves” in a dispersive medium (e.g., a semi-conductor, such as silicon), are “filtered” in their magnitude and phase. This filter acts as a transmission line, or a complex “wave-filter,” forcing the wavelength to be a complex function of frequency. This view is elegantly explained by Brillouin (1953, Chap. 1), in an historical context. Acoustics provides a rich source of examples (Morse, 1948; Beranek, 1954; Beranek and Mellow, 2012).

Newton’s *Principia* was finally published in 1687, and the general solution to Newton’s wave equation [i.e., \( p(x, t) = G(t \pm x/c) \)], where \( G \) is any function, was first published 60 years later by d’Alembert (c1747), which showed that for sounds of a single frequency, the wavelength \( \lambda \) and frequency \( f \) were related by

\[ f\lambda = c. \quad (1.23) \]

Today d’Alembert’s analytic wave solution can be written as Eq. 1.21 with a real wave number \( k = 1/\lambda \, [\text{m}^{-1}] \). This formulation led to the frequency domain concept of Fourier analysis, based on the linearity (i.e., superposition) property of the wave equation (Postulate P2: Lec. 1.3.13, p. 95).

While Newton’s value for the speed of sound in air \( c_o \) was incorrect by the thermodynamic constant \( \sqrt{\eta} \), a problem that would take more than two hundred years to resolve, his success was important because it quantified the physics behind the speed of sound, and demonstrated that momentum \( mv \), not mass \( m \), was transported by the wave. His concept was correct, and his formulation using algebra and calculus represented a milestone in science. If the speed of sound in-vacuo is zero. In air, assuming no visco-elastic losses, it is constant (i.e., \( c_o = \sqrt{\eta \rho_o/P_o} \)). When including losses the wave number becomes a complex function of frequency, leading to Eq. 1.22. In periodic structures, again the wave number becomes complex due to diffraction, as commonly observed in optics (e.g., diffraction gratings) and acoustics. Thus Eq. 1.23 only holds for the most simple cases, but in general it must be considered as a complex analytic function of \( s \), as \( \kappa(s) \) in Eq. 1.22.

\(^{47}\eta = C_p/C_v = 1.4 \) is the ratio of two thermodynamic constants and \( P_o = 10^5 \, [\text{Pa}] \) is the barometric pressure of air.
The corresponding discovery for the formula for the speed of light was made 174 years after Principia, by Maxwell (c1861). Maxwell’s formulation also required great ingenuity, as it was necessary to hypothesize an experimentally unmeasured term in his equations, to get the mathematics to correctly predict the speed of light.

**Chronological history post 17th century**

17th Newton 1642-1727, Bernoulli, Johann 1667-1748
18th Bernoulli, Daniel, Cauchy 1789-1857, Euler 1707-83, d’Alembert 1717-83, Gauss 1777-1855
19th Kirchhoff 1824-87, Helmholtz 1821-1894, Riemann 1826-1866, Maxwell 1831-1879, Rayleigh 1842-1919, Heaviside 1850-1925, Poincare 1854-1912,
20th Sommerfeld 1686-1951, Einstein 1879-1955, Brillouin 1889-1969…

**Time Line**

<table>
<thead>
<tr>
<th>1640</th>
<th>1700</th>
<th>1750</th>
<th>1800</th>
<th>1850</th>
<th>1900</th>
<th>1950</th>
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<tr>
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Figure 1.12: Time-line of the three centuries from the 18th to 20th CE. This was one of the most productive of all times, perhaps starting with the deep work of von Helmholtz, educated an experienced as a military surgeon, who mastered classical music, acoustics, physiology, vision, hearing (Helmholtz, 1863b), and, most important of all, mathematics. Kirchhoff frequently expanded on Helmholtz’s contributions. It is reported that Lord Rayleigh learned German so he could read Helmholtz’s great works. The history during this time is complex. For example, Lord Kelvin wrote a letter to Stokes, suggesting that Stokes try to prove what is today known as “Stokes theorem.” As a result, Stokes posted a reward (the Smith Prize), searching for a prove of “Lord Kelvin’s theorem,” which was finally proved by Hankel (1839-73) (https://en.wikipedia.org/wiki/Hermann_Hankel). Many new concepts were being proved and appreciated over this productive period. In 1863-65, Maxwell published his famous equations, followed by a reformating in modern vector notation by Heaviside, Gibbs and Hertz. The vertical red lines connect mentor-student relationships. This figure should put to rest the idea that ones best work is done in the early years. Many of these scientists were fully productive to the end of old age. Those that were not, died early, due to poor health or accidents. Figure 1.4 (p. 22) (Newton–Gauss) gives a closer look at the 15-18 CE, and Fig. 1.21 (p.98) (Bombelli–Einstein) for the full view from 16-20 CE. [fig:TimeLine19CE]

**The first Algebra:**

Prior to the invention of algebra, people worked out problems as sentences using an obtuse description of the problem (Stillwell, 2010, p. 93). Algebra solved this problem, resulting in a compact language of mathematics, where numbers are represented as abstract symbols (e.g., $x$ and $\alpha$). The problem to be solved could be formulated in terms of sums of powers of smaller terms, the most common being powers of some independent variable (i.e., time or frequency). Today we call such
an expression a polynomial of degree $n$

\[ P_N(z) \equiv z^n + a_{n-1}z^{n-1} + \cdots + a_0z^0 = \sum_{k=0}^{n} a_k z^k = \prod_{k=0}^{n} (z - z_k). \] (1.24)

Here we have let $a_n = 1$. The coefficient $a_n$ cannot be zero, or the polynomial would not be of degree $n$. A solution is to force $a_n = 1$, since this simplifies the expression, and does not change the roots.

The key question is “What values of $z = z_k$ result in $P_N(z_k) = 0$.” In other words, what are the roots $z_k$ of the polynomial? The answer to this question consumed thousands of years, with intense efforts by many aspiring mathematicians. In the earliest attempts, it was a competition to evaluate mathematical acumen. Results were held as a secret to the death bed. It would be fair to view this effort as an obsession. Today the roots of any polynomial may be found by numerical methods, to very high accuracy. The limits to finding roots is limited by the numerical limits of the representation, namely by IEEE-754 (p. 32). There are also a number of important theorems.

Of particular interest is composing a circle with a line, when the line does not touch the circle, and finding the roots. There was no solution to this problem using geometry. This question is addressed in the assignments.

**Finding roots of polynomials**

The problem of factoring polynomials has a history more than a millennium in the making. While the quadratic (degree $N = 2$) was solved by the time of the Babylonians (i.e., the earliest recorded history of mathematics), the cubic solution was finally published by Cardano in 1545. The same year, Cardano’s student solved the quartic ($N = 4$). In 1826 (281 years later) it was proved that the quintic ($N = 5$) could not be factored by analytic methods.

As a concrete example we begin with trivial but important case of the quadratic

\[ P_2(x) = ax^2 + bx + c. \] (1.25)

The roots are those values of $x$ such that $P_2(x_k) = 0$. One of the first results (recorded by the Babylonians, c2000 BCE) was the factoring of this equation by completing the square (Stillwell, 2010, p. 93). One may isolate $x$ by rewriting Eq. 1.25 as

\[ \frac{1}{a} P_2(x) = (x + b/2a)^2 - (b/2a)^2 + c/a. \] (1.26)

This is easily verified by expanding the squared term and canceling $(b/2a)^2$

\[ \frac{1}{a} P_2(x) = [x^2 + (b/a)x + (b/2a)^2] - (b/2a)^2 + c/a. \]

Setting Eq. 1.26 to zero and solving for the two roots $x_{\pm}$, gives the quadratic formula\(^{48}\)

\[ x_{\pm} = \frac{-b \pm \sqrt{b^2 - 4ac}}{2a}. \] (1.27)

If $b^2 - ac > 0$, the two roots are real ($x_{\pm} \in \mathbb{R}$). If it is exactly zero, the roots are identical. Otherwise, they are a complex pair.

---

\(^{48}\)By direct substitution demonstrate that Eq. 1.27 is the solution of Eq. 1.25.
Since \( a \neq 0 \), the polynomial may be divided by \( a \), removing it from the equation. Thus we may set \( a = 1 \) with no loss of generality. In this special, but common case, the roots are given by

\[
x_{\pm} = -\frac{b}{2} \pm \sqrt{\frac{c}{a} - \left(\frac{b}{2}\right)^2} \approx -\frac{b}{2} \pm \sqrt{\frac{c}{a} - \left(\frac{b}{2}\right)^2}.
\]

In the important and common case that the coefficients are real and positive, and \( b \) is small (i.e., \( c > \left(\frac{b}{2}\right)^2 \)).

No insight is gained by memorizing the quadratic formula (Eq. 1.27). On the other hand, an important concept is gained by learning Eq. 1.26, which can be very helpful when doing analysis.

Importantly, the term \( \frac{b}{2a} \) has significance \( \left. \frac{P_N}{\sqrt{a}} \right|_{s=s_0} = \frac{c}{a} - \left(\frac{b}{2a}\right)^2 \), the sign of which determines if the nature of the roots (real vs. complex).

In third grade I learned the trick 50

\[
9 \cdot n = (n - 1) \cdot 10 + (10 - n).
\]  

With this simple rule I did not need to depend on my memory for the 9 times tables. By expanding the above, one can see why it works:

\[
9 \cdot n = n \cdot 10 - 10 + 10 - n = n(10 - 1).
\]

Learning an algorithm is much more powerful than memorization of the 9 times tables. How one thinks about a problem can have great impact.

**Newton's method for finding roots of** \( P_N(s) \) Newton is well known for an approximate but efficient method to find the roots of a polynomial. Consider the Taylor series of \( P_N(s) \in \mathbb{C} \) for \( s \in \mathbb{C} \)

\[
P_N(s) = c_N(s - s_0)^N + c_{N-1}(s - s_0)^{N-1} + \cdots + c_0(s - s_0)^0,
\]

where, from Taylor's formula for the coefficients,

\[
c_n = \frac{1}{n!} \left. \frac{d^n}{ds^n} P_N(s) \right|_{s=s_0}.
\]

Assuming \( (s - s_0) \ll 1 \), we need only consider the largest term \( (s - s_0) \gg (s - s_0)^n \), and truncate the series for \( n > 1 \), resulting in the linear approximation

\[
P_N(s) \approx (s - s_0) \left. \frac{d}{ds} P_N(s) \right|_{s=s_0} + P_N(s - s_0)
\]

\[
= (s - s_0) P_N'(s_0) + P_N(s_0),
\]

where \( P_N'(s) \) is shorthand for \( \frac{dP_N(s)}{ds} \). Setting this linearized expression to zero

\[
(s - s_0) P_N'(s_0) + P_N(s_0) = 0
\]

---

50This is the case for mechanical and electrical circuits having small damping. Physically \( b > 0 \) is the damping coefficient and \( \sqrt{c} > 0 \) is the resonant frequency. One may then simplify the form as \( x^2 + 2bx + c^2 = (x + b + \sqrt{c})(x + b - \sqrt{c}) \).

50E.G.: \( 9 \cdot 7 = (7 - 1) \cdot 10 + (10 - 7) = 60 + 3 \) and \( 9 \cdot 3 = (3 - 1) \cdot 10 + (9 - 3) = 20 + 7 \). As a check, note that the two terms \((n - 1)\) and \((10 - n)\), add to 9.
results in the linear approximation for the root around the expansion point \( s_0 \), providing the key formula behind Newton’s famous root-finding method

\[
s = s_0 - \frac{P_N(s_0)}{P_N'(s_0)}.
\]

This formula for the root near the evaluation point \( s_0 \) becomes increasingly more accurate as \( s \to s_0 \), since \( \lim_{s \to s_0} P_N(s - s_0) = 0 \), namely \( s_0 \) is a root.

Newton’s iteration: Newton proposed using the iteration\(^{51}\)

\[
x_{n+1} = x_n - \frac{P_N(x_n)}{P_N'(x_n)}
\]

to find a root \( P_N(x_0) \). Here \( x_n \in \mathbb{R} \) and \( P_N'(x) = dP_N(x)/dx \). Newton believed that imaginary roots and numbers had no meaning (p. 99). This relation may be explored as a graph, which puts Newton’s method in the realm of analytic geometry. The function \( P_N'(x) \) is the slope of the polynomial \( P_N(x) \) at \( x_n \). The value of \( x_n \in \mathbb{R} \) is the estimate of the root after \( n \) iterations, with \( x_0 \) is the initial guess.

Example: When the polynomial is \( P_2(x) = 1 - x^2 \), so \( P_2'(x) = -2x \) Newton’s iteration becomes

\[
x_{n+1} = x_n + \frac{1 - x_n^2}{2x_n}.
\]

To start the iteration (\( n = 0 \)) we need an initial guess for \( x_0 \), which is a “best guess” of where the root will be. If we let \( x_0 = 1/2 \), then

\[
x_1 = x_0 - \frac{1 - x_0^2}{2x_0} = x_0 + \frac{1}{2} (x_0 - 1/x_0).
\]

1. Let \( P_2(x) = 1 - x^2 \), and \( x_0 = 1/2 \). Draw a graph describing the first step of the iteration.  
   **Sol:** Start with an \((x, y)\) coordinate system and put points at \( x_0 = (1/2, 0) \) and the vertex of \( P_2(x) \), i.e., \((0, 1) \) \((P_2(0) = 1) \). Then draw \( 1 - x^2 \), along with a line from \( x_0 \) to \( x_1 \).

2. Calculate \( x_1 \) and \( x_2 \). What number is the algorithm approaching? Is it a root of \( P_2 \)?  
   **Sol:** First we must find \( P_2'(x) = -2x \). Thus the equation we will iterate is

\[
x_{n+1} = x_n + \frac{1 - x_n^2}{2x_n} = \frac{x_n^2 + 1}{2x_n}.
\]

By hand

\[
x_0 = 1/2
\]

\[
x_1 = \frac{(1/2)^2 + 1}{2(1/2)} = \frac{1}{4} + 1 = 5/4 = 1.25
\]

\[
x_2 = \frac{(5/4)^2 + 1}{2(5/4)} = \frac{25/16 + 1}{10/4} = \frac{41}{40} = 1.025
\]

These answers are approaching \( x = 1 \), the positive (real) root of \( P_2(x) \).

---

\(^{51}\)https://en.wikipedia.org/wiki/Newton’s_method

\(^{52}\)When complex starting points are introduced, the solution always converges (.//M/Newtoncubic.m).
3. Write a Matlab script to check your answer for part (a). \textbf{Sol:}

\begin{verbatim}
x=1/2;
for n = 1:3
    x = x+(1-x*x)/(2*x);
end
\end{verbatim}

(a) For \( n = 4 \), what is the absolute difference between the root and the estimate, \(|x_r - x_4|\)?
\textbf{Sol:} 4.6E-8 (very small!)

(b) What happens if \( x_0 = -1/2 \)? \textbf{Sol:} You converge on the negative root, \( x = -1 \).

4. Does Newton’s method work for \( P_2(x) = 1 + x^2 \)? Why?\(^{53}\) \textbf{Hint:} What are the roots in this case? \textbf{Sol:} In this case \( P'_2(x) = +2x \) thus the iteration gives

\[ x_{n+1} = x_n - \frac{1 + x_n^2}{2x_n}. \]

In this case the roots are \( x_{\pm} = \pm 1 \), namely purely imaginary. Obviously \textit{Newton’s method fails}, because there is no way for the answer to become complex. \textit{Real in, real out}.

5. What if you let \( x_0 = (1 + j)/2 \) for the case of \( P_2(x) = 1 + x^2 \)? \textbf{Sol:} By starting with a complex initial value, we fix the \textit{Real in = Real out} problem.

\textbf{Fractal diagrams:} When the initial guess is real, but the nearest root is complex, Newton’s iteration fails, as discussed in the exercises above. This failure to converge gives rise to fractal patterns, called the \textit{Mandelbrot set}.$^{54}$

6. For any polynomial \( P_N(s) \in \mathbb{R} \) with \( s \in \mathbb{R} \) of degree \( n \), the derivative with respect to \( s \) is uniquely defined, since every polynomial is complex analytic everywhere in \( s \). When \( P_N(s) \) is restricted to having simple poles, we christen the logarithmic derivative of \( P_N(s) \)

\[ Y_N(s) = \frac{d}{ds} \ln P_N(s) = \frac{P'_N(s)}{P_N(s)} \tag{1.29} \]

as the \textit{Newton admittance} (NA), and its reciprocal \([Z(s) = 1/Y(s)]\), the \textit{Newton impedance}. The NA has a number of important properties:

(a) \( Y_N(s) \) has simple poles \( s_p \) and zeros \( s_z \).
(b) The roots of \( P_N(s) \) (i.e., \( P_N(s_z) = 0 \)) are the poles of \( Y_N(s) \).
(c) The zeros of \( Y_N(s) \) (i.e., \( P'_N(s_z) = 0 \)) are the zeros of \( P_N(s) \).
(d) \( Y_N(s) \) is analytic everywhere other than its poles.

\(^{53}\)https://en.wikipedia.org/wiki/Newton’s_method#Complex_functions
\(^{54}\)https://en.wikipedia.org/wiki/Mandelbrot_set
1.3. STREAM 2: ALGEBRAIC EQUATIONS (12 LECTURES)

(e) The Newton impedance is analytic everywhere other than its poles.

(f) Since the zeros of \( P_N(s) \) are simple (no second order poles), it is intuitively obvious that the zeros of \( Y_N(s) \) always lie close to the line connecting the two poles. One may easily demonstrate the truth of the statement numerically, and has been quantified by the Gauss-Lucas theorem which specifies the relationship between the roots of a polynomial and those of its derivative. Rolle’s Theorem is related and much earlier related construction, for real valued functions.

(g) Every Brune admittance takes the form of a Newton admittance (they are equivalent).

(h) Newton’s method may be expressed in terms of the NA as

\[
s_{k+1} = s_k - \epsilon_o Z_N(s).
\]

We call \( \epsilon_o \) the step size since \( Z_N(s) = (s_k - s_{k+1})/\epsilon_o |_{\epsilon_o \to 0} \) is the logarithmic derivative of \( P_N(s) \).

Working with polynomials in Matlab/Octave:

In Matlab/Octave there are five functions that work together as an important set, and you need to become familiar with these:

1. \( \text{R}=\text{root}(A) \): Vector \( A = [a_N, a_{N-1}, \ldots, a_o] \in \mathbb{C} \) are the complex coefficients of polynomial \( P_N(z) = \sum_{n=0}^{N} a_n z^n \in \mathbb{C} \), where \( N \in \mathbb{N} \) is the degree of the polynomial. It is convenient to force \( a_N = 1 \), corresponding to dividing the polynomial by this value, when it is not 1, guaranteeing it cannot be zero, as mentioned above. Further \( R \) is the vector of roots, \([z_1, z_2, \ldots, z_n] \in \mathbb{C}\) such that \( \text{polyval}(A, z_k) = 0 \).

2. \( y=\text{polyval}(A,x) \): This evaluates the polynomial defined by vector \( A \in \mathbb{C}^N \) evaluated at \( x \in \mathbb{C} \), returning vector \( y(x) \in \mathbb{C}^N \).

3. \( P=\text{poly}(R) \): This is the inverse of \( \text{root}() \), returning a vector of polynomial coefficients \( P \in \mathbb{C}^N \) of the corresponding characteristic polynomial, starting from either a vector of roots \( R \), or a matrix \( A \), for example, defined with the roots on the diagonal. The characteristic polynomial is defined as the determinant of \( |A - \lambda I| = 0 \) having roots \( R \).

Due to IEEE-754 scaling issues, this can give strange results, that are numerically correct, but only within the limits of IEEE-754 accuracy.

4. \( \text{R}=\text{polyder}(N) \): This routine takes the coefficients of a polynomial \( N \) and returns the coefficients of the derivative of \( N \). This is useful with working with Newton’s method, since each step is proportional to \( P_N(x)/P_N'(x) \).

5. \( [K, R]=\text{residue}(N, D) \): Given the ratio of two polynomials \( N, D \), \( \text{residue}(N, D) \) returns vectors \( K, R \) such that

\[
\frac{N(s)}{D(s)} = \sum_k \frac{K_k}{s - s_k}.
\]

where \( s_k \in \mathbb{C} \) are the roots of the denominator \( D \) polynomial and \( K \in \mathbb{C} \) is a vector of residues, which characterize the roots of the numerator polynomial \( N(s) \). The use of \( \text{residue}() \) will be discussed in Sect. 1.3.4 (p. 70), and in greater detail in Sect. 1.5.6 (p. 134).
6. \( C = \text{conv}(A, B) \): Vector \( C \in \mathbb{C}^{N+M-1} \) contain the polynomial coefficients of the convolution of the two vector of coefficients of polynomials \( A, B \in \mathbb{C}^N \) and \( B \in \mathbb{C}^M \). For example \( [1, 2, 1] = \text{conv}([1, 1], [1, 1]) \).

7. \([C, R] = \text{deconv}(N, D)\): Vectors \( C, N, D \in \mathbb{C}^M \). This operation uses long division of polynomials to find \( C(s) = N(s)/D(s) \) with remainder \( R(s) \), such that \( N = \text{conv}(D, C) + R \). Namely

\[
C = \frac{N}{D} \quad \text{with remainder} \quad R \quad (1.31)
\]

**Exercise:** Practice the use of Matlab’s/Octave’s related functions \text{root}(), \text{poly}(), \text{polyder}(), \text{polyval}(), \text{residue}(), \text{conv}(), \text{deconv}(), to manipulate roots, polynomials and residues.

**Sol:** Try Newton’s method for various polynomials. Use \( N = \text{poly}(R) \) to provide the coefficients of a polynomial given the roots \( R \). Then use \text{root}() to factor the resulting polynomial. Then use Newton’s method and show that the iteration converges to the nearest root.\(^{55}\)

**Analytic Series and its RoC:**

Every polynomial of degree \( N \) has coefficients related to its derivatives. There is an important distinction, between a polynomial of finite degree \( N \) and of degree \( N \to \infty \). When the degree of the polynomial is infinite, the power series

\[
P(x) = \sum_{0}^{\infty} a_n(x - x_o)^n, \quad (1.32)
\]

defines an single valued function *analytic* \( P(x) \). As for the finite degree polynomial \( P_N(x) \), the coefficients of \( P(x) \) are precisely related to its derivatives, even when the the series fails to converge. For example the divergent series

\[
e^t u(t) = \sum_{0}^{\infty} \frac{1}{n!}t^n \leftrightarrow \frac{1}{s-1}
\]

is a valid description of \( e^t u(t) \), with an unstable pole at \( s = 1 \). For values of \( |x - x_o| < 1 \), the *analytic function* \( P(x) \) has a *region of convergence* (RoC). For cases where the argument is complex \( (x \in \mathbb{C}) \), this is called the *radius of convergence* (RoC). We call the region \( |x - x_o| > 1 \) the *region of divergence* (RoD), and \( |x - x_o| = 0 \), the *singular circle*. Typically the underlying function \( P(s) \), defined by the series, has a pole on the singular circle for \( s \in \mathbb{C} \). Once may isolate such poles by moving the expansion point \( s_o \) until the RoC approaches zero.

**1.3.2 Taylor Series**

For every analytic function \( P(x) \) the coefficients \( c_n \) are always determined by derivatives of \( P(x) \), evaluated at \( x = x_o \). The *Taylor series* coefficients are defined by

\[
a_n = \frac{1}{n!} \frac{d^n}{dx^n} P(x - x_o) \bigg|_{x=x_o}. \quad (1.33)
\]

\(^{55}\)A Matlab/Octave program that does this may be downloaded from [http://jontalle.web.engr.illinois.edu/uploads/493/M/NewtonJPD.m](http://jontalle.web.engr.illinois.edu/uploads/493/M/NewtonJPD.m).
The Taylor formula Eq. 1.32 states how to uniquely define the coefficients \( a_n \). Without the Taylor series formula, we would have no way of determining \( a_n \). The proof of the Taylor formula is transparent, simply by taking successive derivative of Eq. 1.32, and then evaluating the result at the expansion point. If \( P(x) \) is analytic then this procedure will always work. If \( P(x) \) fails to have a derivative of any order, then the function is not analytic and Eq. 1.32 does not valid for \( P(x) \). For example if \( P(x) \) has a pole at \( x_o \) then it is not analytic at that point.

**Exercise:** Verify that \( a_o \) and \( a_1 \) of Eq. 1.32 follow from Eq. 1.33. **Sol:** To obtain \( a_o \), for \( n = 0 \), there is no derivative (\( d^0/dx^0 \) indicates no derivative is taken), so we must simply evaluate \( P(x-x_o) = a_0 + a_1(x-x_o) + \cdots \) at \( x = x_o \), leaving \( a_o \). To find \( a_1 \) we take one derivative which results in \( P'(x) = a_1 + 2a_2(x-x_0))\cdots \). Evaluating this at \( x = x_0 \) leaves \( a_1 \). Each time we take a derivative we reduce the degree of the terms of the series by 1, leaving the next constant term.

The Taylor series plays an important role in mathematics, as the coefficients of the series uniquely determine any analytic function via its derivatives. The implications, and limitations of the power series representation are very specific. **First**, if the series fails to converge (i.e., outside the RoC), it is essentially meaningless. This is in conflict with the example of \( e^{at}u(t) \), which has a divergent series, that agrees with the unstable function. I’m not sure how to justify this conflict, other than to point out that \( t \in \mathbb{R} \), thus the series expansion of the diverging exponential is real-analytic, not complex analytic. It does have a Laplace transform, with a pole at \( s = 1 \), in agreement with its unstable nature. **Second**, the analytic function must be single valued. This follows from the fact that each term in Eq. 1.32 is single valued. **Third**, analytic functions are very “smooth,” since the may be differentiated an \( \infty \) number of times, and the series still converges. There can be no jumps or kinks in such functions.

But these properties are both the curse and the blessing of the analytic function. On the positive side analytic functions are the ideal starting point for solving differential equations, which is exactly how they were used by Newton and others. Analytic functions are “smooth” in that they are infinitely differentiable, with coefficients given by Eq. 1.33. They are single valued, so there can be no ambiguity in their interpretation.

Two well known analytic functions are the **geometric series**

\[
\frac{1}{1-x} = 1 + x + x^2 + x^3 + \ldots = \sum_{n=0}^{\infty} x^n \tag{1.34}
\]

and exponential series

\[
e^x = 1 + x + \frac{1}{2} x^2 + \frac{1}{3 \cdot 2} x^3 + \frac{1}{4 \cdot 3 \cdot 2} x^4 + \ldots = \sum_{n=0}^{\infty} \frac{1}{n!} x^n. \tag{1.35}
\]

A third important family of functions is those that define Impedance. Impedance functions are special because they must have a positive real part, so that they obey conservation of energy. A physical impedance cannot have a negative resistance (the real part), or it would act like a source of power, which violates conservation of energy. Most impedances are in the class of Brune impedances, defined by the ratio of two polynomials, of degrees \( M \) and \( N \)

\[
Z_{\text{Brune}} = \frac{P_N(s)}{P_M(s)} = \frac{s^N + a_1 S^{N-1} \ldots a_0}{s^M + b_1 S^{M-1} \ldots b_0}, \tag{1.36}
\]

where \( M = N \pm 1 \) (i.e., \( N = M \pm 1 \)). This fraction of polynomials is historically known as the “Pade approximation,” but more specifically is a Brune “pole-zero” impedance. The key propriety
of the Brune impedance is that the real part of the impedance is non-negative (positive or zero) in
the right-hand $s$ plane

$$\Re Z(s) = \Re [R(\sigma, \omega) + jX(\sigma, \omega)] = R(\sigma, \omega) \geq 0 \quad \text{for} \quad \Re s = \sigma \geq 0. \quad (1.37)$$

Since $s = \sigma + \omega j$, the right-hand plane complex frequency ($s$ plane) corresponds to $\Re s = \sigma \geq 0$. This condition is called the class of positive-real functions, and the above condition, and is called the Brune condition, which is frequently written in the abbreviated form as $\Re Z(\Re s \geq 0) \geq 0$.

As a result of this positive-real constraint on impedance functions, the subset of Brune impedances (those given by Eq. 1.36) must be complex analytic in the entire right-hand $s$ plane. This is a powerful constraint that places strict limitations on the locations of both the poles and the zeros of every Brune impedance.

**Exercise:** Verify that the coefficients of the above functions are given by Eq. 1.33.

**Exercise:** Find the RoC of the following by application of Eq. 1.33.

1. $w(x) = \frac{1}{1-xj}$. **Sol:** From a straightforward expansion we know the coefficients are

$$\frac{1}{1-xj} = 1 + xj + (xj)^2 + (xj)^3 \cdots = 1 + xj - x^2 + jx^3 \cdots .$$

Working this out using Eq. 1.33 by brute force is more work:

$$c_0 = w|_{x=0} = 1; \quad c_1 = \frac{dw}{dx}|_{x=0} = -j; \quad c_2 = \frac{d^2w}{dx^2}|_{x=0} = -2; \quad c_3 = \frac{d^3w}{dx^3}|_{x=0} = -6j;$$

However if we take derivatives of the series expansion it is much easier, and one can even figure out the term for $c_n$:

$$c_0 = 1; \quad c_1 = \sum (jx)^n |_{x=0} = j; \quad c_2 = \sum (jx)^n |_{x=0} = 2(j)^2; \quad c_3 = \sum (jx)^n |_{x=0} = 2 \cdot 3(j)^3 = -6j; \quad \cdots ,$$

$$c_n = j^n n!.$$

2. $w(x) = e^{xj}$ **Sol:** $c_n = \frac{1}{n!} j^n$.

**Region of convergence:** Determining the RoC for a given analytic function is quite important, and may not always be obvious. In general the RoC is a circle having a radius, centered on the expansion point, out to the nearest pole. Thus when the expansion point is moved, the RoC changes, since the location of the pole is fixed.

Example: For the geometric series (Eq. 1.34), the expansion point is $x_o = 0$, and the RoC is $|x| < 1$, since $1/(1-x)$ has a pole at $x = 1$. We may move the expansion point by a linear transformation. For example, by replacing $x$ with $z + 3$. Then the series becomes $1/((z+3) - 1) = 1/(z + 2)$, so the RoC becomes 2, because in the $z$ plane, the pole has moved to $-2$. 
1.3. STREAM 2: ALGEBRAIC EQUATIONS (12 LECTURES)

Example: A second important example is the function \( \frac{1}{x^2 + 1} \), which has the same RoC as the geometric series, since it may be expressed in terms of its residue expansion (aka, partial fraction expansion)

\[
\frac{1}{x^2 + 1} = \frac{1}{(x + 1)(x - 1)} = \frac{1}{2j} \left( \frac{1}{x - 1} - \frac{1}{x + 1} \right).
\]

Each term has an RoC of \(|x| < |1_j| = 1\). The amplitude of each pole is called the residue, defined in Section 1.4.5 Eq. 1.96, p. 113. The residue for the pole at \(1_j\) is \(1/2j\).

In summary, the function \(1/(x^2 + 1)\) is the sum of two geometric series, with poles at \(\pm 1_j\) which are not as obvious because the roots are complex, and conjugate. Once factored, it becomes clear what is going on.

**Exercise:** Verify the above expression is correct, and show that the residues are \(\pm 1/2j\). **Sol:** Cross-multiply and cancel, \(x\) cancels out and we are left with 1, as required.

**Exercise:** Find the residue of \(\frac{d}{dz} z^\pi\). **Sol:** Taking the derivative gives \(z^{\pi-1}\) which has a pole at \(z = 0\). Applying the formula for the residue (Eq. 1.96, p. 113) we find

\[
e^{-1} = \lim_{z \to 0} zz^{\pi-1} = \lim z^\pi = 0.
\]

Thus the residue is zero.

The exponential series converges for every finite value of \(x \in \mathbb{R}\) (the RoC is the entire real line), thus the exponential is called an entire function.

**Analytic functions:**

Any function that has a Taylor series expansion is called an analytic function. Within the RoC, the series expansion defines a single valued function. Polynomials, \(1/(1 - x)\) and \(e^x\) are examples of analytic functions that are real functions of their real argument \(x\). This is not the entire story. Because analytic functions are easily manipulated term by term, they may be used to find solutions of differential equations, since the derivatives of a series are uniquely determined within the RoC, due to Eq. 1.33.

Every analytic function has a corresponding differential equation, that is determined by the coefficients \(a_k\) of the analytic power series. An example is the exponential, which has the property that it is the eigen-function of the derivative operation

\[
\frac{d}{dx} e^{ax} = ae^{ax},
\]

which may be verified using Eq. 1.35. This relationship is a common definition of the exponential function, which is a very special, because it is the eigen-function of the derivative.

The complex analytic power series (i.e, complex analytic functions) may also be integrated, term by term, since

\[
\int^x f(x)dx = \sum \frac{a_k}{k+1} x^{k+1}. \tag{1.38}
\]

Newton took full advantage of this property of the analytic function and used the analytic series (Taylor series) to solve analytic problems, especially for working out integrals, allowing him to solve differential equations. To fully understand the theory of differential equations, one must master single valued analytic functions and their analytic power series.
**Single- vs. multi-valued functions:** Polynomials, and their \( \infty \)-degree extensions (analytic functions) are single valued: for each \( x \) there is a single value of \( P_N(x) \). The roles of the domain and codomain may be swapped, to obtain an inverse function, which is typically quite different in its properties compared to the function. For example \( y(x) = x^2 + 1 \) has the inverse \( x(y) = \pm \sqrt{y - 1} \), which is double valued, and complex when \( y < 1 \). Periodic functions, such as \( y(x) = \sin(x) \) are even more “exotic,” since \( x(y) = \arcsin(y) = \sin^{-1}(x) \) has an \( \infty \) number of \( x(y) \) values for each \( y \). This problem was first addressed in Riemann’s 1851 PhD thesis, while working with Gauss.

**Exercise:** Let \( y(x) = \sin(x) \). Then \( dy/dx = \cos(x) \). Show that \( dx/dy = -1/\sqrt{1 - x^2} \). Hint: \( x(y) = \cos^{-1}(y) = \arccos(y) \). **Sol:** See the *See implicit function theorem.* (D’Angelo, 2017, p. 104) Add solution.

**Exercise:** Let \( y(x) = \sin(x) \). Then \( dy/dx = \cos(x) \). Show that \( dx/dy = -j/\sqrt{1 + x^2} \). **Sol:** Add solution.

**Exercise:** Find the Taylor series coefficients of \( y = \sin(x) \) and \( x = \sin^{-1}(y) \). **Sol:** Add solution.

**Complex analytic functions:** When the argument of an analytic function \( F(x) \) is complex, that is, \( x \in \mathbb{R} \) is replaced by \( s = \sigma + \omega j \in \mathbb{C} \) (recall that \( \mathbb{R} \subset \mathbb{C} \))

\[
F(s) = \sum_{n=0}^{\infty} c_n (s - s_o)^n,
\]

with \( c_n \in \mathbb{C} \), that function is said to be a complex analytic.

For example, when the argument of the exponential becomes complex, it is periodic on the \( \omega \) axis, since

\[
e^{st} = e^{(\sigma + \omega j)t} = e^{\sigma t} e^{\omega j t} = e^{\sigma t} [\cos(\omega t) + j \sin(\omega t)].
\]

Taking the real part gives

\[
\Re\{e^{st}\} = e^{\sigma t} \cos(\omega t),
\]

and \( \Im\{e^{st}\} = e^{\sigma t} \sin(\omega t) \). Once the argument is allowed to be complex, it becomes obvious that the exponential and circular functions are fundamentally related. This exposes the family of entire circular functions [i.e., \( e^s, \sin(s), \cos(s), \tan(s), \cosh(s), \sinh(s) \)] and their inverses [\( \ln(s), \arcsin(s), \arccos(s), \arctan(s), \cosh^{-1}(s), \sinh^{-1}(s) \)], first fully elucidated by Euler (c1750) (Stillwell, 2010, p. 315). Note that because a function, such as \( \sin(\omega t) \), is periodic, its inverse must be multi-valued. What is needed is some systematic way to account for this multi-valued property. That methodology was provided by Riemann 100 years later, in this 1851 PhD Thesis, supervised by Gauss, in the final years of Gauss’ life.

Given a complex analytic function of a complex variable, one must resort to the extended complex plane, Riemann sheets and branch cuts, as discussed in Section 1.3.9 (p. 82). The extended complex plane is a tool that extends the domain of complex analytic to include the point at infinity. This topic is critically important in engineering mathematics, and will be discussed in length in Sections 1.3.9-1.3.12 (pp. 82-91).
Definition of the Taylor series of a complex analytic function: However there is a fundamental problem, since we cannot formally define the Taylor series for the coefficients $c_k$, since we have not defined $dF(s)/ds$, the derivative with respect to the complex variable $s \in \mathbb{C}$. Thus simply substituting $s$ for $x$ in an analytic function is leaving a major hole in our understanding of the complex analytic function.

To gain a feeling of the nature of the problem, we make take derivatives of a function with respect to various variables. For example,

$$\frac{d}{dt} e^{st} = se^{st}.$$ 

Also

$$e^{\omega t} \frac{d}{d\sigma} e^{\sigma t} = \sigma e^{st}.$$ 

and

$$e^{\sigma t} \frac{d}{d\omega} e^{\omega t} = \omega je^{st}.$$ 

are straightforward.

It was the work of Cauchy (1814) (Fig. 1.12), who uncovered much deeper relationships within complex analytic functions (Sect. 1.3.10, p. 85) by defining differentiation and integration in the complex plane, leading to several fundamental theorems of complex calculus, including the Fundamental theorem of complex integration, and Cauchy’s formula. We shall explore this in and several fundamental theorems in Sect. 1.4.1 (p. 100).

There seems to be some disagreement as to the status of multi-valued functions: Are they functions, or is a function strictly single valued? If so, then we are missing out on a host of interesting possibilities, including all the inverses of nearly every complex analytic function. For example, the inverse of a complex analytic function is a complex analytic function (e.g., $e^s$ and $\log(s)$).

Impact on Physics: It seems likely, if not obvious, that the success of Newton was his ability to describe physics by the use of mathematics. He was inventing new mathematics at the same time as he was explaining new physics. The same might be said for Galileo. It seems likely that Newton was extending the successful techniques and results of Galileo (Galileo, 1638). Galileo died on Jan 8, 1642, and Newton was born Jan 4, 1643, just short of one year later. Certainly Newton was well aware of Galileo’s great success, and naturally would have been influenced by them.

The application of complex analytic functions to physics was dramatic, as may be seen in the six volumes on physics by Arnold Sommerfeld (1868-1951), and from the productivity of his many (36) students (e.g., Debye, Lenz, Ewald, Pauli, Guillemin, Bethe, Heisenberg and Seebach, to name a few), notable coworkers (i.e., Leon Brillouin) and others (i.e., John Bardeen), upon whom Sommerfeld had a strong influence. Sommerfeld is known for having many students who were awarded the Nobel Prize in Physics, yet he was not (the prize is not awarded in mathematics). Sommerfeld brought mathematical physics (the merging of physical and experimental principles with mathematics) to a new level with the use of complex integration of analytic functions to solve otherwise difficult problems, thus following the lead of Newton who used real integration of Taylor series to solve differential equations (Brillouin, 1960, Ch. 3 by Sommerfeld, A.).

https://www.aip.org/history-programs/niels-bohr-library/oral-histories/4661-1
1.3.3 Lec 12: Polynomial root classification by convolution

Following the exploration of algebraic relationships by Fermat and Descartes, the first theorem was being formulated by d’Alembert. The idea behind this theorem is that every polynomial of degree $N$ (Eq. 1.24) has at least one root. This may be written as the product of the root and a second polynomial of degree of $N - 1$. By the recursive application of this concept, it is clear that every polynomial of degree $N$ has $N$ roots. Today this result is known as the fundamental theorem of algebra:

$$\text{Every polynomial equation } P(z) = 0 \text{ has a solution in the complex numbers.}$$

As Descartes observed, a solution $z = a$ implies that $P(z)$ has a factor $z - a$. The quotient

$$Q(z) = \frac{P(z)}{z - a} = \frac{P(z)}{a} \left[ 1 + \frac{z}{a} + \left(\frac{z}{a}\right)^2 + \left(\frac{z}{a}\right)^3 + \cdots \right]$$

(1.40)

is then a polynomial of one lower degree. ... We can go on to factorize $P(z)$ into $n$ linear factors.


The ultimate expression of this theorem is given by Eq. 1.24 (p. 55), which indirectly states that an $n^{th}$ degree polynomial has $n$ roots. We shall use the term degree when speaking of polynomials and the term order when speaking of differential equations. A general rule is that order applies to the time domain and degree to the frequency domain, since the Laplace transform of a differential equation, having constant coefficients, of order $N$, is a polynomial of degree $N$ in Laplace frequency $s$.

**Exercise:** Explore expressing Eq. 1.40 in terms of real 2x2 matrices, as described in Section 1.2.1, p. 29. **Sol:** Consider the representation of polynomials as a special case of products of 2x2 complex numbers, as discussed in Eq. 1.3, p. 29.

Today this theorem is so widely accepted we fail to appreciate it. Certainly about the time you learned the quadratic formula, you were prepared to understand the concept of polynomials having roots. The simple quadratic case may be extended a higher degree polynomial. The Matlab/Octave command `roots([1,a2,a1,a0])` provides the roots $[s1,s2,s3]$ of the cubic equation, defined by the coefficient vector $[1,a2,a1,a0]$. The command `poly([s1,s2,s3])` returns the coefficient vector. I don’t know the largest degree that can be accurately factored by Matlab/Octave, but I’m sure its well over $N = 10^3$. Today, finding the roots numerically is a solved problem.

**Factorization versus convolution:** The best way to gain insight into the polynomial factorization problem is through the inverse operation, multiplication of monomials. Given the roots $x_k$, there is a simple algorithm for computing the coefficients $a_k$ of $P_N(x)$ for any $n$, no matter how large. This method is called convolution. Convolution is said to be a trap-door since it is easy, while the inverse, factoring (deconvolution), is hard, and analytically intractable for degree $N \geq 5$ (Stillwell, 2010, p. 102).
Convolution of monomials

As outlined by Eq. 1.24, a polynomial has two descriptions, first as a series with coefficients $a_n$, and second in terms of its roots $x_r$. The question is “What is the relationship between the coefficients and the roots?” The simple answer is that they are related by convolution.

Let us start with the quadratic

$$(x + a)(x + b) = x^2 + (a + b)x + ab,$$  

(1.41)

where in vector notation $[-a, -b]$ are the roots and $[1, a + b, ab]$ are the coefficients.

To see how the result generalizes, we may work out the coefficients for the cubic ($N = 3$). Multiplying the following three factors gives (p. 180)

$$(x-1)(x-2)(x-3) = (x^2 - 3x + 2)(x-3) = x(x^2 - 3x + 2) - 3(x^2 - 3x + 2) = x^3 - 6x^2 + 11x - 6.$$

When the roots are $[1, 2, 3]$ the coefficients of the polynomial are $[1, -6, 11, -6]$. To verify, substitute the roots into the polynomial, and show that they give zero. For example $r_1 = 1$ is a root since $P_3(1) = 1 - 6 + 11 - 6 = 0$.

As the degree increases, the algebra becomes more difficult. Imagine trying to work out the coefficients for $N = 100$. What is needed is an simple way of finding the coefficients from the roots. Fortunately, convolution keeps track of the book-keeping, by formalizing the procedure.

**Convolution of two vectors:** To get the coefficients by convolution, write the roots as two vectors $[1, a]$ and $[1, b]$. To find the coefficients we must convolve the root vectors, indicated by $[1, a] * [1, b]$, where $*$ denotes convolution. Convolution is a recursive operation. The convolution of $[1, a] * [1, b]$ is done as follows: reverse one of the two monomials, padding unused elements with zeros. Next slide one monomial against the other, forming the local dot product (element-wise multiply and add):

$$
\begin{array}{cccccccc}
a & 1 & 0 & 0 & a & 1 & 0 & 0 & a & 1 & 0 & 0 & a & 1 \\
0 & 0 & 1 & b & 0 & 1 & b & 0 & 1 & b & 0 & 1 & b & 0 & 0 \\
\end{array}
$$

resulting in coefficients $[\cdots, 0, 0, 1, a + b, ab, 0, 0, \cdots]$.

By reversing one of the polynomials, and then taking successive dot products, all the terms in the sum of the dot product correspond to the same power of $x$. This explains why convolution of the coefficients gives the same answer as the product of the polynomials.

As seen by the above example, the position of the first monomial coefficients are reversed, and then slid across the second set of coefficients, the dot-product is computed, and the result placed in the output vector. Outside the range shown, all the elements are zero. In summary,

$$
[1, -1] * [1, -2] = [1, -1 - 2, 2] = [1, -3, 2].
$$

In general

$$
[a, b] * [c, d] = [ac, bc + ad, bd],
$$

Convolving a third term $[1, -3]$ with $[1, -3, 2]$ gives (p. 180)

$$
[1, -3] * [1, -3, 2] = [1, -3 - 3, 9 + 2, -6] = [1, -6, 11, -6],
$$

which is identical to the cubic example, found by the algebraic method.
By convolving one monomial factor at a time, the overlap is always two elements, thus it is never necessary to compute more than two multiplies and an add for each output coefficient. This greatly simplifies the operations (i.e., they are easily done in your head). Thus the final result is more likely to be correct. Comparing this to the algebraic method, convolution has the clear advantage.

Exercise:

1. Given the coefficient vector \([\alpha, \beta, \gamma] = [1, a + b, ab]\), find the formula for the roots \([a, b]\).

   Sol: To solve this one must complete the square of the polynomial (p. 55):
   \[
x^2 + \beta x + \gamma = 0
   \]
   \[
   (x + \beta/2)^2 = (\beta/2)^2 - \gamma
   \]
   \[
x + \beta/2 = \pm \sqrt{(\beta/2)^2 - \gamma}
   \]
   \[
x = -\beta/2 \pm \sqrt{(\beta/2)^2 - \gamma}.
   \]

   In terms of the coefficient vector (Eq. 1.41) \([1, a + b, ab]\), \(\beta = a + b, \gamma = ab\).

   \[
x_{\pm} = -\frac{a + b}{2} \pm \frac{(a + b)^2}{2} - ab.
   \] (1.42)

While it is not obvious that \(x_{\pm} = [a, b]\), it must be true. One may prove this several ways. The most obvious is working backwards from what we did above. However this gives no insight (try it), other than the fact that we did the algebra correctly. Let’s try something else:

   \[
x_{\pm} + \frac{a + b}{2} = \pm \sqrt{\left(\frac{a + b}{2}\right)^2 - ab}
   \]
   \[
   = \pm \sqrt{\frac{a^2 + b^2}{4} + ab - ab}
   \]
   \[
   = \pm \frac{1}{2} \sqrt{(a + b)(a - b)}
   \]

This does not seem to show the roots satisfy the starting equation.

Instead let’s try the product of the roots

   \[
x_+ \cdot x_- = \left(-\frac{a + b}{2} + \sqrt{\left(\frac{a + b}{2}\right)^2 - ab}\right) \left(-\frac{a + b}{2} - \sqrt{\left(\frac{a + b}{2}\right)^2 - ab}\right)
   \]
   \[
   = \left(\frac{a + b}{2}\right)^2 - \left(\sqrt{\left(\frac{a + b}{2}\right)^2 - ab}\right)^2
   \]
   \[
   = \left(\frac{a + b}{2}\right)^2 - \left(\frac{a + b}{2}\right)^2 + ab
   \]
   \[
   = \left(\frac{a + b}{2}\right)^2 + ab.
   \]
We have simultaneously solved two nonlinear equations that relate the roots \([a, b]\), and the coefficients of the polynomial \([1, \beta, \gamma]\):

\[
\begin{align*}
\beta &= (a + b)/2 \\
\gamma &= ab
\end{align*}
\]

where \(\beta\) mean and \(\gamma\) is the product of the roots.

2. What are the three nonlinear equations that one would need to solve to find the roots of a cubic? \textbf{Sol:} From our formula for the convolution of three monomials we may find the nonlinear “deconvolution” relations between the roots\(^{57}\) \([-a, -b, -c]\) and the cubic’s coefficients \([1, \alpha, \beta, \gamma]\)

\[
(x + a) \ast (x + b) \ast (x + c) = (x + c) \ast (x^2 + (a + b)x + ab)
\]

\[
= x \cdot (x^2 + (a + b)x + ab) + c \cdot (x^2 + (a + b)x + ab)
\]

\[
= x^3 + (a + b + c)x^2 + (ab + ac + cb)x + abc
\]

\[
= [1, a + b + c, ab + ac + cb, abc].
\]

It follows that the nonlinear equations must be

\[
\begin{align*}
\alpha &= a + b + c \\
\beta &= ab + ac + bc \\
\gamma &= abc.
\end{align*}
\]

Clearly these are solve by the classic cubic solution which appears to be a deconvolution problem, also know as long division of polynomials.

**The product of monomial** \(P_1(x)\) **with a polynomial** \(P_N(x)\) **gives** \(P_{N+1}(x)\): This statement is another way of stating the fundamental theorem of algebra. Each time we convolve a monomial with a polynomial of degree \(N\), we obtain a polynomial of degree \(N + 1\). The convolution of two monomials results in a quadratic (degree 2 polynomial). The convolution of three monomials give a cubic (degree 3). In general, the degree \(k\), of the product of two polynomials, of degree \(n, m\), is the sum of the degrees \((k = n + m)\). For example, if the degrees are each 5 \((n = m = 5)\), then the resulting degree is 10.

In summary, the product of two polynomials of degree \(m, n\) having \(m\) and \(n\) roots, give a polynomial of degree \(m + n\) having \(m + n\) roots. This is an analysis process, of merging polynomials, by coefficient convolution. Multiplying polynomials is a merging process, into a single polynomial.

Note that the degree of a polynomial is one less than the length of the vector of coefficients. The coefficient on the lead term should always be set to 1 since it cannot be zero, resulting in an illogical result. Always normalize the lead term of \(P_N(x)\) to 1 (i.e., \(a_N = 1\)). This has no effect on the roots.

While you already know this theorem from high school algebra class, it is important to explicitly state the fundamental theorem of algebra.

\(^{57}\)By working with the negative roots we may avoid an unnecessary and messy alternating sign problem.
Composition of polynomials: Convolution is not the only important operation between two polynomials. Another is composition, which may be defined for two functions \( f(z), g(z) \). Then the composition \( c(z) = f(z) \circ g(z) = f(g(z)) \). As a specific example, suppose \( f(z) = 1 + z + z^2 \) and \( g(z) = e^{2z}. \) With these definitions
\[
f(z) \circ g(z) = 1 + e^{2z} + (e^{2z})^2 = 1 + e^{2z} + e^{4z}.
\]
Note that \( f(z) \circ g(z) \neq g(z) \circ f(z). \)

Exercise: Find \( g(z) \circ f(z). \) Sol: \( e^{2f(z)} = e^{2(1+z+z^2)} = e^{2e^{(1+z+z^2)}} = e^{3e^{z}e^{z^2}}. \)

1.3.4 Lec 13 Residue expansions of rational functions (e.g., impedance)

As discussed in Section 1.3.1, p. 59, there are 5 important Matlab/Octave routines that are closely related: \( \text{root()}, \text{polyval()}, \text{poly()}, \text{residue()}, \text{conv}(). \) The operations defined by \( \text{residue}() \) and \( \text{conv}() \) take two polynomials (i.e., vectors of the coefficients of the polynomials) as input, making them different from the first three, which take a single polynomial (vector of coefficients) as their input.

When lines and planes are defined, the equations are said to be linear in the independent variables. In keeping with this definition of linear, we say that the equations are non-linear when the equations have degree greater than 1 in the independent variables. The term bilinear has a special meaning, in that both the domain and codomain are linearly related by lines (or planes). As an example, an impedance is defined in frequency as the ratio of the voltage over the current
\[
Z(s) = \frac{V(\omega)}{I(\omega)} = sL_o + R_o + \sum_{k=0}^{K} \frac{K_k}{s - s_k},
\]
where \( Z(s) \) is the impedance and \( V \) and \( I \) are the voltage and current at radian frequency \( \omega. \)

Such an impedance is typically specified as a rational or bilinear function, namely the ratio of two polynomials, \( P_N(s) = N(s) = [a_N, a_{N-1}, \cdots, a_0] \) and \( P_K(s) = D(s) = [b_K, b_{K-1}, \cdots, b_0] \) of degrees \( N, K \in \mathbb{N}, \) as functions of complex Laplace frequency \( s = \sigma + j\omega, \) having simple roots. Most impedances are rational functions since they may be written as \( D(s)V = N(s)I. \) Since \( D(s) \) and \( N(s) \) are both polynomials in \( s, \) rational functions are also called bilinear forms, which comes from a corresponding scalar differential equation, of the form
\[
\sum_{k=0}^{K} b_k \frac{d^k}{dt^k} i(t) = \sum_{n=0}^{N} a_n \frac{d^n}{dt^n} v(t) \quad \leftrightarrow \quad I(\omega) \sum_{k=0}^{K} b_k s^k = V(\omega) \sum_{n=0}^{N} a_n s^n.
\]
This equation, as well as 1.43, follow from the Laplace transform (See Section 1.3.12, p. 91) of the differential equation (on left), by forming the impedance \( Z(s) = V/I = A(s)/B(s). \) This form of the differential equation follows from Kirchhoff’s voltage and current laws (KCL, KVL) or from Newton’s laws (for the case of mechanics).

The physical properties of an impedance: Based on d’Alembert’s observation that the solution to the wave equation is the sum of forward and backward traveling waves, the impedance may be rewritten in terms of forward and backward traveling waves
\[
Z(s) = \frac{V}{I} = \frac{V^+ + V^-}{I^+ - I^-} = r_o \frac{1 + \Gamma(s)}{1 - \Gamma(s)},
\]
(1.45)
where \( r_o = P^+/I^+ \) is called the surge impedance of the transmission line (e.g., wire) connected to the load impedance \( Z(s) \), and \( \Gamma(s) = P^-/P_+ = I^-/I^+ \) is the reflection coefficient corresponding to \( Z(s) \). Like \( Z(s) \), \( \Gamma(s) \) is causal and complex analytic. Note that the impedance and the reflectance function must both be complex analytic, since they are connected by the bilinear (aka, Möbius) transformation, which assures the mutual complex analytic properties.

Due to the bilinear transformation, the physical properties of \( Z(s) \) and \( \Gamma(s) \) are very different. Specifically, the real part of the load impedance must be non-negative \((\Re\{Z(\omega j)\} \geq 0)\), if and only if \(|\Gamma(s)| \leq 1\). In the time domain, the impedance \( z(t) \leftrightarrow Z(s) \) must have a value of \( r_o \) at \( t = 0 \). Correspondingly, the time domain reflectance \( \gamma(t) \leftrightarrow \Gamma(s) \) must be zero at \( t = 0 \).

This is the basis of conservation of energy, which may be traced back to the properties of the reflectance \( \Gamma(s) \).

**Exercise:** Show that if the \( \Re\{Z(s)\} \geq 0 \) then \(|\Gamma(s)| \leq 1\). **Sol:** Their two equivalent proofs, both of which start from the relation between \( Z(s) \) and \( \Gamma(s) \). Taking the real part of Eq. 1.45, which must be \( \geq 0 \), we find

\[
\Re\{Z(s)\} = \frac{r_o}{2} \left[ \frac{1 + \Gamma(s)}{1 - \Gamma(s)} + \frac{1 + \Gamma^*(s)}{1 - \Gamma^*(s)} \right] = r_o \frac{1 - |\Gamma(s)|^2}{1 + |\Gamma(s)|^2} \geq 0.
\]

Thus \(|\Gamma| \leq 1\).

### 1.3.5 **Lec 14: Introduction to Analytic Geometry**

Analytic geometry is the natural consequence of Euclid’s Geometry (which deals with conical geometry (e.g., points, lines, triangles, circles, spheres, ellipses, cones, etc., in two and three physical dimensions), merged with algebra (which deals with simultaneous equations, roots of polynomials, analytic functions, and ultimately, solutions of differential equations). The combination of Euclid’s (323 BCE) geometry and al-Khwarizmi’s (830 CE) algebra provides a new powerful tool, analytic geometry.

There are many important relationships between Euclidean geometry and 16th century algebra. Important similarities include vectors, their Pythagorean lengths \([a, b, c]\)

\[
c = \sqrt{(x_2 - x_1)^2 + (y_2 - y_1)^2},
\]

\(a = x_2 - x_1\) and \(b = y_2 - y_1\), and the angles. Euclid’s geometry had no concept of coordinates, thus of vectors. One of the main differences is the ability of analytic geometry is that one may compute with numbers. A detailed comparison is given in Table 1.3.

There are several new concepts that come with the development of analytic geometry:

1. Composition of functions: If \( y = f(x) \) and \( z = g(y) \) then the composition of functions \( f \) and \( g \) is denoted \( z(x) = g \circ f(x) = g(f(x)) \).

2. Elimination: Given two functions \( f(x, y) \) and \( g(x, y) \), elimination removes either \( x \) or \( y \). This procedure, called Gaussian Elimination, was known to the Chinese. This is not always possible.
3. **Intersection**: While one may speak of the intersection of two lines to give a point, or two planes to give a line, the term *intersection* is also an important but very different concept in set theory. This is a special case of elimination when the functions $f(x, y), g(x, y)$ are linear in their arguments.

4. **Vectors**: Euclidean geometry provides the new concept of a vector, as a line with length and orientation (i.e., direction). Analytic geometry defines a vector, in any number of dimensions, as an ordered set of points $x \in \mathbb{C}$.

5. **Analytic geometry** extends the ideas of geometry with the introduction of the product of two vectors, the scalar (dot) product $f \cdot g$ and the vector (cross) product $f \times g$ (Fig. 1.13).

<table>
<thead>
<tr>
<th>Table 1.3: Comparison between Euclidean geometry and analytic geometry. I am uncertain about the items in the third column.</th>
</tr>
</thead>
<tbody>
<tr>
<td><strong>Euclidean geometry</strong></td>
</tr>
<tr>
<td>Proof, Vector, Length, Point, Direction, $\leq 3$ dimensional, Intersection, Conic Section, Dot product, Square Roots: on real line lengths.</td>
</tr>
</tbody>
</table>

What algebra added to geometry was the ability to compute with complex numbers. For example, the length of a line (Eq. 1.46) was measured in Geometry with a compass: numbers played no role. Once algebra was available, the line’s Euclidean length could be computed from the coordinates of the two ends, defined by the 3-vector

$$e = x\hat{x} + y\hat{y} + z\hat{z} = [x, y, z]^T,$$

which represents a point at $(x, y, z) \in \mathbb{R}^3 \subset \mathbb{C}^3$ in three dimensions, having direction, from the origin $(0, 0, 0)$ to $(x, y, z)$. An alternative matrix notation is $e = [x, y, z]^T$, a column vector of three numbers. These two notations are different ways of representing exactly the same thing. View them as equivalent notations.

By defining the vector, analytic geometry allows Euclidean geometry to become quantitative, beyond the physical drawing of an object (e.g., a sphere, triangle or line). With analytic geometry we have the Euclidean concept of a vector, a line having a magnitude (length) and direction, but analytic defined in terms of physical coordinates (i.e., numbers). The difference between two vectors defines a third vector, a concept already present in Euclidean geometry. For the first time, complex numbers were allowed into geometry (but rarely used until Cauchy and Riemann).

As shown in Fig. 1.13, there are two types of products, the 1) scalar $A \cdot B$ and 2) vector $A \times B$ products.
1.3. STREAM 2: ALGEBRAIC EQUATIONS (12 LECTURES)

Scalar product of two vectors: When using algebra, many concepts in geometry are made precise. There are many examples of how algebra extends Euclidean geometry, the most basic being the scalar product (aka dot product) between two vectors

\[ \mathbf{x} \cdot \mathbf{\kappa} = (x\hat{x} + y\hat{y} + z\hat{z}) \cdot (\alpha\hat{x} + \beta\hat{y} + \gamma\hat{z}), \quad \in \mathbb{C} \]

\[ = \alpha x + \beta y + \gamma z. \]

In matrix notation the scalar product is written as

\[ \mathbf{x} \cdot \mathbf{\kappa} = \begin{bmatrix} x \\ y \\ z \end{bmatrix}^T \begin{bmatrix} \alpha \\ \beta \\ \gamma \end{bmatrix} = [x, y, z] \begin{bmatrix} \alpha \\ \beta \\ \gamma \end{bmatrix} = \alpha x + \beta y + \gamma z. \quad (1.47) \]

The dot product takes the character of \( \kappa \). For example, if \( \kappa(s) \in \mathbb{C} \) is a function of complex frequency \( s \), then the dot product is complex. If \( \zeta \in \mathbb{R} \) is real, then the dot product is real.

Norm (length) of a vector: The norm of a vector

\[ \|\mathbf{e}\| \equiv + \sqrt{\mathbf{e} \cdot \mathbf{e}} \geq 0. \]

is defined as the positive square root of the scalar product of the vector with itself. This is a generalization of the length, in any number of dimensions, forcing the sign of the square-root to be non-negative. The length is a concept of Euclidean geometry, and it must always be positive and real. A complex length is not physically meaningful. A zero-length vector is a point. More generally, the Euclidean length of a line is given as the norm of the difference between two real vectors \( \mathbf{e}_1, \mathbf{e}_2 \in \mathbb{R} \)

\[ \|\mathbf{e}_1 - \mathbf{e}_2\|^2 = (\mathbf{e}_1 - \mathbf{e}_2) \cdot (\mathbf{e}_1 - \mathbf{e}_2) \]

\[ = (x_1 - x_2)^2 + (y_1 - y_2)^2 + (z_1 - z_2)^2 \geq 0. \quad (1.48a) \]

\[ = (x_1 - x_2)^2 + (y_1 - y_2)^2 + (z_1 - z_2)^2 \geq 0. \quad (1.48b) \]

From this formula we see that the norm of the difference of two vectors is simply a compact expression for the Euclidean length.

Integral definition of the norm: To now we have only considered a vector to be a set of elements. What if the elements are continuous, such as \( f(t) \). This can also be viewed as a vector with coefficients labeled by \( t \in \mathbb{R} \) rather than by \( n \in \mathbb{N} \). In this case the norm must be defined in terms of the scalar product of an integral over \( t \), rather than an index over \( t \)

\[ f(t) \cdot g(t) = \int f(t)g(t)dt. \quad (1.49) \]

This view allows a significant generalization of the concept of a vector space, and norms between vectors. This will turn out to be a useful, thus important, extension of the concept of a vector, and its norm.

Pythagorean theorem and the Schwarz inequality: Regarding Fig. 1.13, suppose we compute the difference between vector \( \mathbf{A} \in \mathbb{R} \) and \( \alpha \mathbf{B} \in \mathbb{R} \) as \( L = \|\mathbf{A} - \alpha \mathbf{B}\| \in \mathbb{R} \), where \( \alpha \in \mathbb{R} \) is a scalar that modifies the length of \( \mathbf{B} \). We seek the value of \( \alpha \), which we denote as \( \alpha^* \), that minimizes the length of \( L \). From simple geometrical considerations, \( L(\alpha) \) will be minimum when
the difference vector is perpendicular to $B$, as shown in the figure by the dashed line from the tip of $A \perp B$.

To show this algebraically we write out the expression for $L(\alpha)$ and take the derivative with respect to $\alpha$, and set it to zero, which gives the formula for $\alpha^*$. The argument does not change, but the algebra greatly simplifies, if we normalize $A, B$ to be unit vectors $a = A/||A||$ and $b = B/||B||$, which have norm $= 1$.

$$L^2 = (a - \alpha b) \cdot (a - \alpha b) = 1 - 2\alpha a \cdot b + \alpha^2. \quad (1.50)$$

Thus the length is shortest ($L = L_s$, as shown in Fig. 1.13) when

$$\frac{d}{d\alpha}L^2 = -2a \cdot b + 2\alpha^* = 0.$$ 

Solving for $\alpha^*$ we find $\alpha^* = a \cdot b$. Since $L_s > 0 \ (a \neq b)$, Eq. 1.50 becomes

$$1 - 2|a \cdot b|^2 + |a \cdot b|^2 = 1 - |a \cdot b|^2 > 0.$$ 

In conclusion $\cos \theta \equiv |a \cdot b| < 1$. In terms of $A, B$ this is $|A \cdot B| < ||A|| ||B|| \cos \theta$, as shown next to $B$ in Fig. 1.13. Thus the scalar product between two vectors is their direction cosine. Furthermore since this forms a right triangle, the Pythagorean theorem must hold. The triangle inequality says that the lengths of the two sides must be greater than the hypotenuse. Note that $\Theta \in \mathbb{R} \not\in \mathbb{C}$.

This derivation is an abbreviated version of a related discussion in Section 3.2.1 (p. 186).

![Figure 1.13: Vectors $A, B, C$ are used to define the scalar product $A \cdot B$, vector (cross) product $A \times B$ and triple product $C \cdot (A \times B)$. The two vector products complement each other, one proportional to the $\sin \theta$ of the angle $\theta$ between them and the other to the $\cos \theta$. It follows that $||A \cdot B||^2 + ||A \times B||^2 = ||A||^2 ||B||^2$. The scalar product computes the projection of one vector on the other. The vector product $A \times B$ computes the area of the trapezoid formed by the two vectors, while the triple product $C \cdot (A \times B)$ defines the volume of the formed parallelepiped (i.e., prism). When all the angles are $90^\circ$, the volume becomes a cuboid.](image)

Vector ($\times$) product of two vectors: As shown in Fig. 1.13, the vector product (aka, cross-product) $a \times b$ is the second type of product between two vectors. The vector product defines
1.3. STREAM 2: ALGEBRAIC EQUATIONS (12 LECTURES)

A vector, perpendicular (⊥) to the plane of the two vectors being multiplied. The formula for computing the cross product is

\[ \mathbf{a} \times \mathbf{b} = (a_1 \mathbf{x} + a_2 \mathbf{y} + a_3 \mathbf{z}) \times (b_1 \mathbf{x} + b_2 \mathbf{y} + b_3 \mathbf{z}) = \begin{vmatrix} \mathbf{x} & \mathbf{y} & \mathbf{z} \\ a_1 & a_2 & a_3 \\ b_1 & b_2 & b_3 \end{vmatrix}. \]

For example, if the two vectors are in \( \mathbf{x} \) and \( \mathbf{y} \), then the cross-product is \( \mathbf{z} \). It is strictly in \( \mathbf{z} \) if the two vectors are perpendicular to each other (i.e., \( \mathbf{z} = \mathbf{x} \times \mathbf{y} = -\mathbf{y} \times \mathbf{x} \)). The vector product of a vector with itself (or the difference between two vectors) is zero. For example \( \mathbf{x} \times \mathbf{x} = \mathbf{y} \times \mathbf{y} = \mathbf{z} \times \mathbf{z} = 0 \). Typically \( \mathbf{a}, \mathbf{b} \in \mathbb{R} \). If they are complex, the definition must be modified to be consistent with the physics.

The scalar product of a third vector \( \mathbf{c} \) with the vector product \( \mathbf{a} \times \mathbf{b} \), with \( \mathbf{a}, \mathbf{b}, \mathbf{c} \in \mathbb{R} \),

\[ \mathbf{c} \cdot (\mathbf{a} \times \mathbf{b}) = \begin{vmatrix} c_1 & c_2 & c_3 \\ a_1 & a_2 & a_3 \\ b_1 & b_2 & b_3 \end{vmatrix} (\in \mathbb{R}^3), \]

is called the triple product, which represents the volume of a parallelepiped.

**Impact of Analytic Geometry:** The most obvious impact of analytic geometry, was a detailed analysis of the conic sections, using algebra, rather than drawings via a compass and ruler. An important example is the composition of the line and circle, a venerable construction, presumably going back to before Diophantus (250CE). Once algebra was invented, the composition could be done using formulas. With this analysis came complex numbers.

The first two mathematicians to appreciate this mixture of Euclid’s geometry and the new algebra were Fermat and Descartes (Stillwell, 2010, p. 111-115); soon Newton contributed to this effort, by the addition of physics (calculations in acoustics, orbits of the planets, and the theory of gravity and light Stillwell (2010, p. 115-117)), significant concepts for 1687.

Given these new methods, many new solutions to problems emerged. The complex roots of polynomials continued to appear, without any obvious physical meaning. Complex numbers seem to have been viewed as more of an inconvenience than a problem. Newton’s solution to this dilemma was to simply ignore the imaginary cases (Stillwell, 2010, p. 119).

**Development of Analytic Geometry**

**Intersection and Gaussian elimination:** The first “algebra” (al-jabr) is credited to al-Khwarizmi (830 CE). Its invention advanced the theory of polynomial equations in one variable, Taylor series, and composition versus intersections of curves. The solution of the quadratic equation had been worked out thousands of year earlier, but with algebra a general solution could be defined. The Chinese had found the way to solve several equations in several unknowns, for example, finding the values of the intersection of two circles. With the invention of algebra by al-Khwarizmi, a powerful tool became available to solve the difficult problems.

**Composition and Elimination** In algebra there are two contrasting operations on functions: composition and Elimination.
Composition: Composition is the merging of functions, by feeding one into the other. If the two functions are \( f, g \) then their composition is indicated by \( f \circ g \), meaning the function \( y = f(x) \) is substituted into the function \( z = g(y) \), giving \( z = g(f(x)) \).

Composition is not limited to linear equations, even though that is where it is most frequently applied. To compose two functions, one must substitute one equation into the other. That requires solving for that substitution variable, which is not always possible in the case of nonlinear equations. However many tricks are available that may work around this restriction. For example if one equation is in \( x^2 \) and the other in \( x^3 \) or \( \sqrt{x} \), it may be possible to multiply the first by \( x \) or square the second. The point is that one of the variables must be isolated so that when it is substituted into the other equations, the variable is removed from the mix.

Examples: Let \( y = f(x) = x^2 - 2 \) and \( z = g(y) = y + 1 \). Then

\[
g \circ f = g(f(x)) = (x^2 - 2) + 1 = x^2 - 1.
\]

In general composition does not commute (i.e., \( f \circ g \neq g \circ f \)), as is easily demonstrated. Swapping the order of composition for our example gives

\[
f \circ g = f(g(y)) = z^2 - 2 = (y + 1)^2 - 2 = y^2 + 2y - 1.
\]

Intersection: Complimentary to composition is intersection (i.e., decomposition) (Stillwell, 2010, pp. 119,149). For example, the intersection of two lines is defined as the point where they meet. This is not to be confused with finding roots. A polynomial of degree \( N \) has \( N \) roots, but the points where two polynomials intersect has nothing to do with the roots of the polynomials. The intersection is a function (equation) of lower degree, implemented with Gaussian elimination.

Intersection of two lines: Unless they are parallel, two lines meet at a point. In terms of linear algebra this may be written as 2 linear equations\(^{58}\) (on the left), along with the intersection point \([x_1, x_2]^T\), given by the inverse of the 2x2 set of equations (on the right)

\[
\begin{bmatrix}
a & b \\
c & d
\end{bmatrix}
\begin{bmatrix}
x_1 \\
x_2
\end{bmatrix}
=
\begin{bmatrix}
y_1 \\
y_2
\end{bmatrix}
\Rightarrow
\begin{bmatrix}
x_1 \\
x_2
\end{bmatrix} = \frac{1}{\Delta}
\begin{bmatrix}
d & -b \\
-c & a
\end{bmatrix}
\begin{bmatrix}
y_1 \\
y_2
\end{bmatrix}.
\]

By substituting the expression for the intersection point \([x_1, x_2]^T\) into the original equation, we see that it satisfies the equations. Thus the equation on the right is the solution to the equation on the left.

Note the structure of the inverse: 1) The diagonal values \((a, d)\) are swapped, 2) the off-diagonal values \((b, c)\) are negated and 3) the 2x2 matrix is divided by the determinant \( \Delta = ad - bc \). If \( \Delta = 0 \), there is no solution. When the determinant is zero (\( \Delta = 0 \)), the slopes of the two lines

\[
slope = \frac{dx_2}{dx_1} = \frac{b}{a} = \frac{d}{c}.
\]

are equal, thus the lines are parallel. Only if the slopes differ can there be a unique solution.

\(^{58}\)When writing the equation \(Ax = y\) in matrix format, the two equations are \(ax_1 + bx_2 = y_1\) and \(dx_1 + cx_2 = y_2\) with unknowns \((x_1, x_2)\), whereas in the original equations \(ay + bx = c\) and \(dy + cx = f\), they were \(y, x\). Thus in matrix format, the names are changed. The first time you see this scrambling of variables, it can be confusing.
Exercise: Show that the equation on the right is the solution of the equation on the left. Sol: By a direct substitution of the right equation into the left equation

\[
\begin{bmatrix}
a & b \\
c & d
\end{bmatrix} \cdot \frac{1}{\Delta} \begin{bmatrix}
d & -b \\
-c & a
\end{bmatrix} \begin{bmatrix}
y_1 \\
y_2
\end{bmatrix} = \frac{1}{\Delta} \begin{bmatrix}
ad - bc & -ab + ab \\
-cd + ab & ad - bc
\end{bmatrix} \begin{bmatrix}
\Delta \\
0
\end{bmatrix},
\]

which gives the identity matrix.

Algebra will give the solution when geometry cannot. When the two curves fail to intersect on the real plane, the solution still exists, but is complex valued. In such cases, geometry, which only considers the real solutions, fails. For example, when the coefficients \([a, b, c, d]\) are complex, the solution exists, but the determinant can be complex. Thus algebra is much more general than geometric. Geometry fails when the solution has a complex intersection.

A system of linear \(Ax = y\) equations has many interpretations, and one should not be biased by the notation. As engineers we are trained to view \(x\) as the input and \(y\) as the output, in which case then \(y = Ax\) seems natural, much like the functional relation \(y = f(x)\). But what does the linear relation \(x = Ay\) mean, when \(x\) is the input? The obvious answer is that \(y = A^{-1}x\). But when working with systems of equations, there are many uses of equations, and we need to become more flexible in the our interpretation. For example \(y = A^2x\) has a useful meaning, and in fact we saw this type of relationship when working with Pell’s equation (p. 46) and the Fibonacci sequence (p. 49). As another example consider

\[
\begin{bmatrix}
z_1 \\
z_2
\end{bmatrix} = \begin{bmatrix}
a_{1x} & a_{1y} \\
a_{2x} & a_{2y}
\end{bmatrix} \begin{bmatrix} x \\ y \end{bmatrix}
\]

which is reminiscent of a three-dimensional surface \(z = f(x, y)\). We shall find that such generalizations are much more than a curiosity.

1.3.6 Lec 15 Gaussian Elimination

The method for finding the intersection of equations is based on the recursive elimination of all the variables but one. This method, known as Gaussian elimination (Appendix A.4, p. 235), works across a broad range of cases, but may be defined as a systematic algorithm when the equations are linear in the variables.\(^{59}\) Rarely do we even attempt to solve problems in several variables of degree greater than 1. But Gaussian eliminations may still work in such cases (Stillwell, 2010, p. 90).

In Appendix B.2 (p. 242) the inverse of a 2x2 linear system of equations is derived. Even for a 2x2 case, the general solution requires a great deal of algebra. Working out a numeric example of Gaussian elimination is more instructive. For example, suppose we wish to find the intersection of the two equations

\[
\begin{align*}
x - y &= 3 \\
2x + y &= 2.
\end{align*}
\]

This 2x2 system of equations is so simple that you may immediately visualize the solution: By adding the two equations, \(y\) is eliminated, leaving \(3x = 5\). But doing it this way takes advantage of

the specific example, and we need a method for larger systems of equations. We need a generalized (algorithmic) approach. This general approach is called Gaussian elimination.

Start by writing the equations in matrix format (note this is not in the standard form $Ax = y$)

$$\begin{bmatrix} 1 & -1 \\ 2 & 1 \end{bmatrix} \begin{bmatrix} x \\ y \end{bmatrix} = \begin{bmatrix} 3 \\ 2 \end{bmatrix}. \tag{1.54}$$

Next, eliminate the lower left term ($2x$) using a scaled version of the upper left term ($x$). Specifically, multiply the first equation by -2, add it to the second equation, replacing the second equation with the result. This gives

$$\begin{bmatrix} 1 & -1 \\ 0 & 3 \end{bmatrix} \begin{bmatrix} x \\ y \end{bmatrix} = \begin{bmatrix} 3 \\ 2 - 3 \cdot 2 \end{bmatrix} = \begin{bmatrix} 3 \\ -4 \end{bmatrix}. \tag{1.55}$$

Note that the top equation did not change. Once the matrix is “upper triangular” (zero below the diagonal) you have the solution. Starting from the bottom equation, $y = -4/3$. Then the upper equation then gives $x - (-4/3) = 3$, or $x = 3 - 4/3 = 5/3$.

In principle, Gaussian elimination is easy, but if you make a calculation mistake along the way, it is very difficult to find the error. The method requires a lot of mental labor, with a high probability of making a mistake. Thus you do not want to apply this method every time. For example suppose the elements are complex numbers, or polynomials in some other variable such as frequency. Once the coefficients become more complicated, the seeming trivial problem becomes highly error prone. There is a much better way, that is easily verified, which puts all the numerics at the end, in a single step.

The above operations may be automated by finding a carefully chosen upper-diagonalization matrix $U$ that does the same operation. For example let

$$U = \begin{bmatrix} 1 & 0 \\ -2 & 1 \end{bmatrix}. \tag{1.56}$$

Multiplying Eq. 1.54 by $U$ we find

$$\begin{bmatrix} 1 & 0 \\ -2 & 1 \end{bmatrix} \begin{bmatrix} 1 & -1 \\ 2 & 1 \end{bmatrix} \begin{bmatrix} x \\ y \end{bmatrix} = \begin{bmatrix} 1 & -1 \\ 0 & 3 \end{bmatrix} \begin{bmatrix} x \\ y \end{bmatrix} = \begin{bmatrix} 3 \\ -4 \end{bmatrix}, \tag{1.57}$$

we obtain Eq. 1.55. At this point we can either back-substitute and obtain the solution, as we did above, or find a matrix $L$ that finishes the job, by removing elements above the diagonal.

**Exercise:** Show that $\det(U) = \det(UA) = 1$. **Sol:** Since

$$\det \begin{bmatrix} a & b \\ c & d \end{bmatrix} = ad - bc,$$

and $\det(UA) = \det U \cdot \det(A)$, and $\det(U) = 1 - 0 = 1$, the result follows naturally. Common notation is to let $\det(A) = |A|$.

In Appendix B.2 the inverse of a general 2x2 matrix is summarized in terms of three steps: 1) swap the diagonal elements, 2) reverse the signs of the off-diagonal elements and 3) divide by the determinant $\Delta = ab - cd$. Specifically

$$\begin{bmatrix} a & b \\ c & d \end{bmatrix}^{-1} = \frac{1}{\Delta} \begin{bmatrix} d & -b \\ -c & a \end{bmatrix}. \tag{1.58}$$
There are very few things that you must memorize, but the inverse of a 2x2 is one of them. It needs to be in your tool-bag of tricks, like the quadratic formula.

While it is difficult to compute the inverse matrix from scratch (Appendix B), it takes only a few seconds to verify it (steps 1 and 2)

\[
\begin{bmatrix}
    a & b \\
    c & d \\
\end{bmatrix}
\begin{bmatrix}
    d & -b \\
    -c & a \\
\end{bmatrix}
= 
\begin{bmatrix}
    \Delta & 0 \\
    0 & \Delta \\
\end{bmatrix}.
\]

Thus dividing by the determinant gives the 2x2 identity matrix. A good strategy, when you don’t trust your memory, is to write down the inverse as best you can, and then verify.

Using the 2x2 matrix inverse on our example (Eq. 1.54), we find

\[
\begin{bmatrix}
    x \\
    y \\
\end{bmatrix}
= 
\frac{1}{1 + 2} 
\begin{bmatrix}
    1 & 1 \\
    -2 & 1 \\
\end{bmatrix}
\begin{bmatrix}
    3 \\
    2 \\
\end{bmatrix}
= 
\frac{1}{3} 
\begin{bmatrix}
    5 \\
    -6 + 2 \\
\end{bmatrix}
= 
\begin{bmatrix}
    5/3 \\
    -4/3 \\
\end{bmatrix}.
\]

If you use this method, you will rarely (never) make a mistake, and the solution is easily verified. Either you can check the numbers in the inverse, as was done in Eq. 1.59, or you can substitute the solution back into the original equation.

1.3.7 Lec 16: Transmission (ABCD) matrix composition method

Matrix composition: Matrix multiplication represents a composition of 2x2 matrices, because the input to the second matrix is the output of the first (this follows from the definition of composition: \(f(x) \circ g(x) = f(g(x))\)). Thus the ABCD matrix is also known as the transmission matrix method, or occasionally the chain matrix. The general expression for an transmission matrix \(T(s)\) is

\[
\begin{bmatrix}
    V_1 \\
    I_1 \\
\end{bmatrix}
= 
\begin{bmatrix}
    A(s) & B(s) \\
    C(s) & D(s) \\
\end{bmatrix}
\begin{bmatrix}
    V_2 \\
    -I_2 \\
\end{bmatrix}.
\]

The four coefficients \(A(s), B(s), C(s), D(s)\) are all complex functions of the Laplace frequency \(s = \sigma + j\omega\). The derivation is repeated with more detail in Section 3.3.2 (p. 191).

It is a standard convention to always define the current into the node, but since the input current (on the left) is the same as the output current on the right (\(I_2\)), hence the negative sign on \(I_2\), to meet the convention of current into every node. When transmission matrices are cascaded, all the signs then match.

We have already used 2x2 matrix composition in representing complex numbers (p. 29), and for computing the \(gcd(m,n)\) of \(m, n \in \mathbb{N}\) (p. 40), Pell’s equation (p. 46) and the Fibonacci sequence (p. 49).
Definitions of $A, B, C, D$: The definitions of the four functions of Eq. 1.61 are easily read off of the equation, as

$$A(s) = \frac{V_1}{V_2} \bigg|_{I_2=0} , \quad B(s) = -\frac{V_1}{I_2} \bigg|_{V_2=0} , \quad C(s) = \frac{I_1}{V_2} \bigg|_{I_2=0} , \quad D(s) = -\frac{I_1}{I_2} \bigg|_{V_2=0}. \quad (1.62)$$

These definitions follow trivially from Eq. 1.61. These relations have general names. $A, D$ are called voltage and current transfer functions, since they are the ratio of an output over an input, whereas $B, C$ are known as the transfer impedance and transfer conductance. Each express an output (port 2) in terms of an input at port 1.

Exercise: Explain why $C$ is given as above. Sol: Writing out the lower equation $I_1 = CV_2 - DI_2$. Setting $I_2 = 0$, we obtain the equation for $C$.

Thévenin parameters of a source: A very important concept in circuit theory is that of the Thévenin parameters (p. 79), the open-circuit voltage and the short-circuit current, the ratio of which define the Thévenin impedance. The open circuit voltage is defined as the voltage when the current is zero. In mechanics this is called the isometric force, the maximum force given zero velocity.

It trivially follows that the Thévenin voltage (open circuit voltage) is

$$V_{\text{Thév}} = \frac{1}{C} \equiv \frac{V_2}{I_1} \bigg|_{I_2=0}.$$

For example, the voltage measured across a battery having no load is its Thévenin voltage.

We may find the impedance $Z_1 = \frac{V_1}{I_1} \bigg|_{I_2=0}$ that would be seen looking into port 1, with nothing connected to port 2 ($I_2 = 0$) by taking the ratio of the upper and lower equations

$$Z_1 = \frac{AV_2 - BI_2}{CV_2 - DI_2} \bigg|_{I_2=0} = \frac{A}{C} = \frac{A}{C} V_{\text{Thév}}.$$

Using this relation we may find a second expression for the Thévenin voltage in terms of the a source voltage $V_1$ rather than in terms of the source current.

$$V_{\text{Thév}} = \frac{Z_1}{A} \equiv \frac{Z_1}{A} \frac{V_2}{V_1} \bigg|_{I_2=0}.$$

The choice of the appropriate expression depends on the physics of the actual source.

Properties of the transmission matrix: The transmission matrix is always constructed as the product of elemental matrices of the form

$$\begin{bmatrix} 1 & Z(s) \\ 0 & 1 \end{bmatrix} \quad \text{or} \quad \begin{bmatrix} 1 & 0 \\ Y(s) & 1 \end{bmatrix}.$$ 

Since the determinant of each of these matrices is 1, the determinant, the product of many elemental matrices being 1, must be one. Thus for the case of reciprocal system (P6) (p. 96)

$$\det \begin{bmatrix} A(s) & B(s) \\ C(s) & D(s) \end{bmatrix} = 1.$$ 

Anti reciprocal system may be synthesized by the use of a gyrator, and in these cases $\Delta_T = -1$. 

1.3.8 Impedance matrix

With a bit of algebra, one may find the impedance matrix in terms of $A, B, C, D$ (Van Valkenburg, 1964a, p. 310)

$$
\begin{bmatrix}
V_1 \\
V_2
\end{bmatrix} =
\begin{bmatrix}
z_{11} & z_{12} \\
z_{21} & z_{22}
\end{bmatrix}
\begin{bmatrix}
I_1 \\
I_2
\end{bmatrix} = \frac{1}{C} \begin{bmatrix} A & \Delta_T \\ 1 & D \end{bmatrix} \begin{bmatrix} I_1 \\
I_2
\end{bmatrix}.
$$

(1.63)

For reciprocal systems (P6, p. 96) $z_{12} = z_{21}$ since $\Delta_T = 1$. For anti-reciprocal systems, such as dynamic loudspeakers and microphones (Kim and Allen, 2013), $z_{21} = -z_{12} = 1/C$, since $\Delta_T = -1$.

Impedance is a very general concept, closely tied to the definition of power $\mathcal{P}(t)$ (and energy). Power is defined as the product of the effort (force) and the flow (current). As described in Fig. 1.15, these concepts are very general, applying to mechanics, electrical circuits, acoustics, thermal circuits, or any other case where conservation of energy applies. Two basic variables are defined, generalized force and generalized flow, also called conjugate variables. The product of the conjugate variables is the power, and the ratio is the impedance. For example for the case of voltage and current,

$$
\mathcal{P}(t) \equiv \int^t v(t)i(t)dt,
\quad Z(s) \equiv \frac{V(\omega)}{I(\omega)}.
$$

Ohm’s Law In general, impedance is defined as the ratio of a force over a flow. For electrical circuits, the voltage is the ‘force’ and the current is the ‘flow.’ Ohm’s law states that the voltage across and the current through a circuit element are related by the impedance of that element (which is typically a function of the Laplace frequency $s = \sigma + \omega j$). For resistors, the voltage over the current is called the resistance, and is a constant (e.g. the simplest case, $V/I = R$). For inductors and capacitors, the impedance depends on the Laplace frequency $s$ [e.g. $V/I = Z(s)$].

<table>
<thead>
<tr>
<th>Case</th>
<th>Force</th>
<th>Flow</th>
<th>Impedance</th>
<th>units</th>
</tr>
</thead>
<tbody>
<tr>
<td>Electrical</td>
<td>voltage (V)</td>
<td>current (I)</td>
<td>$Z = V/I$</td>
<td>Ohms [Ω]</td>
</tr>
<tr>
<td>Mechanics</td>
<td>force (F)</td>
<td>velocity (U)</td>
<td>$Z = F/U$</td>
<td>Mechanical Ohms [Ω]</td>
</tr>
<tr>
<td>Acoustics</td>
<td>pressure (P)</td>
<td>particle velocity (V)</td>
<td>$Z = P/V$</td>
<td>Acoustic Ohms [Ω]</td>
</tr>
<tr>
<td>Thermal</td>
<td>temperature (T)</td>
<td>heat-flux (J)</td>
<td>$Z = T/J$</td>
<td>Thermal Ohms [Ω]</td>
</tr>
</tbody>
</table>

Figure 1.15: Impedance is defined as the ratio of a force over a flow, a concept also holds in mechanics and acoustics. In mechanics, the ‘force’ is equal to the mechanical force on an element (e.g. a mass, dash-pot, or spring), and the ‘flow’ is the velocity. In acoustics, the ‘force’ is pressure, and the ‘flow’ is the volume velocity or particle velocity of air molecules.

As discussed in Fig. 1.15, the impedance concept also holds for mechanics and acoustics. In mechanics, the ‘force’ is equal to the mechanical force on an element (e.g. a mass, dash-pot, or spring), and the ‘flow’ is the velocity. In acoustics, the ‘force’ is pressure, and the ‘flow’ is the volume velocity or particle velocity of air molecules.

In this section we shall derive the method of linear composition of systems, known by several names as the $ABCD$ Transmission matrix method, or in the mathematical literature as the Möbius (bilinear) transformation. Using the method of matrix composition, a linear system of 2x2 matrices can represent a large and important family of networks. By the application of Ohm’s law to the circuit shown in Fig. 1.14, we can model a cascade of such cells.
Example of the use of the ABCD matrix composition: In Fig. 1.14 we see the network is composed of a series inductor (mass) having an impedance $Z_l = sL$, and a shunt capacitor (compliance) having an impedance $Z_c = 1/sC$, where $s \in \mathbb{C}$. As determined by Ohm’s Law, each impedance is describe by a linear relation between the current and the voltage. For the inductive impedance, applying Ohm’s law, we find

$$V_1 - V_2 = Z_l I_1$$

where $Z_l$ is the complex impedance of the inductor. For the capacitive impedance, applying Ohm’s law we find

$$V_2 = (I_1 + I_2) Z_c,$$

where $Z_c$ is the complex impedance of the capacitor.

Each of these linear impedance relations may be written in matrix form. The series inductor equation gives (note $I_1 = -I_2$)

$$\begin{bmatrix} V_1 \\ I_1 \end{bmatrix} = \begin{bmatrix} 1 & Z_l \\ 0 & 1 \end{bmatrix} \begin{bmatrix} V_2 \\ -I_2 \end{bmatrix},$$

(1.64)

while the shunt capacitor equation yields (note $V_1 = V_2$)

$$\begin{bmatrix} V_2 \\ I_1 \end{bmatrix} = \begin{bmatrix} 1 & 0 \\ Y_c & 1 \end{bmatrix} \begin{bmatrix} V_2 \\ -I_2 \end{bmatrix},$$

(1.65)

where $Y_c = 1/Z_c$ is called the admittance.

When the second matrix equation for the shunt admittance (Eq. 1.65) is substituted into the series impedance equation (Eq. 1.64), we find the ABCD matrix $(T_1 \circ T_2)$, for the cell is simply the product of two matrices

$$\begin{bmatrix} V_1 \\ I_1 \end{bmatrix} = \begin{bmatrix} 1 & Z_l \\ 0 & 1 \end{bmatrix} \begin{bmatrix} 1 + Z_l Y_c & Z_l \\ Y_c & 1 \end{bmatrix} \begin{bmatrix} V_2 \\ -I_2 \end{bmatrix}.$$  

(1.66)

Note that the determinant of the matrix $\Delta = AD - BC = 1$.

Thus $A(s) = 1 + Z_l Y_c = 1 + s^2 LC$, $B(s) = Z_l$, $C(s) = Y_c$ and $D = 1$. This equation characterizes every possible relation between the input and output voltage and current of the cell.

For example, the ratio of the output to input voltage with the output unloaded ($I_2 = 0$), known as the voltage divider relation may be found from the upper equation with $I_2 = 0$. Writing this out gives

$$\left. \frac{V_2}{V_1} \right|_{I_2=0} = \frac{1}{A(s)} = \frac{1}{1 + Z_l Y_c} = \frac{Z_c}{Z_c + Z_l}.$$  

To derive the formula for the current divider equation, use the lower equation and set $V_2 = 0$. This out gives

$$\left. -\frac{I_2}{I_1} \right|_{V_2=0} = \frac{1}{D} = 1.$$

1.3.9 Lec 17: Riemann Sphere: 3\textsuperscript{d} extension of chord and tangent method

Once algebra was formulated c830 CE, mathematicians were able to expand beyond the limits placed on it by geometry on the real plane, and the verbose descriptions of each problem in prose (Stillwell, 2010, p. 93). The geometry of Euclid’s Elements had paved the way, but after 2000 years, the addition of the language of algebra changed everything. The analytic function was a
1.3. STREAM 2: ALGEBRAIC EQUATIONS (12 LECTURES)

key development, heavily used by both Newton and Euler. Also the investigations of Cauchy made important headway with his work on complex variables. Of special note was integration and differentiation in the complex plane of complex analytic functions, which is the topic of stream 3.

It was Riemann, working with Gauss in the final years of Gauss’ life, who made the breakthrough, with the concept of the extended complex plane. This concept was based on the composition of a line with the sphere, similar to the derivation of Euclid’s formula for Pythagorean triplets (Fig. 2.3, p. 171). While the importance of the extended complex plane was unforeseen, it changed analytic mathematics forever, along with the physics it supported. It unified and thus simplified many important integrals, to the extreme. This idea is captured by the fundamental theorem of complex integral calculus (Table 1.8 p. 149) and 1.4, p. 98.

![Figure 1.16:](image_url)

The idea is outlined in Fig. 1.16. On the left is a circle and a line. The difference between this case and the derivation of the Pythagorean Triples is, that the line starts at the north pole, and ends on the real \( x \in \mathbb{R} \) axis at point \( x \). At point \( x' \), the line cuts through the circle. Thus the mapping from \( x \) to \( x' \) takes every point on \( \mathbb{R} \) to a point on the circle. For example, the point \( x = 0 \) maps to the south pole (not indicated). To express \( x' \) in terms of \( x \) one must compose the line and the circle, similar to the composition used in Fig. 2.3 (p. 171). The points on the circle, indicated here by \( x' \), require a traditional polar coordinate system, having a unit radius and an angle defined between the radius and a vertical line passing through the north pole. When \( x \to \infty \) the point \( x' \to N \), known as the point at infinity. But this idea goes much further, as shown on the right half of Fig. 1.16.

Here the real tangent line is replaced by the a tangent complex plane \( z \in \mathbb{C} \), and the complex puncture point \( z' \in \mathbb{C} \), in this case on the complex sphere, called the extended complex plane. This is a natural extension of the chord/tangent method on the left, but with significant consequences. The main difference between the complex plane \( z \) and the extended complex plane, other than the coordinate system, is what happens at the north pole. The point at \( |z| = \infty \) is not defined on the plane, whereas on the sphere, the point at the north pole is simply another point, like every other point on the sphere.

[^60]: “Gauss did lecture to Riemann but he was only giving elementary courses and there is no evidence that at this time he recognized Riemann’s genius.” [http://www-groups.dcs.st-and.ac.uk/˜history/Biographies/Riemann.html](http://www-groups.dcs.st-and.ac.uk/˜history/Biographies/Riemann.html) “In 1849 he [Riemann] returned to Göttingen and his Ph.D. thesis, supervised by Gauss, was submitted in 1851.” [http://www-groups.dcs.st-and.ac.uk/˜history/Biographies/Riemann.html](http://www-groups.dcs.st-and.ac.uk/˜history/Biographies/Riemann.html)
Open vs. closed sets: Mathematically the plane is said to be an open set, since the limit \( z \to \infty \) is not defined, whereas on the sphere, the point \( z' \) is a member of a closed set, since the north pole is defined. The distinction between an open and closed set is important, because the closed set allows the function to be complex-analytic at the north pole, which it cannot be on the plane (since the point at infinity is not defined).

The \( z \) plane may be replaced with another plane, say the \( w = F(z) \in \mathbb{C} \) plane, where \( w \) is some function \( F \) of \( z \in \mathbb{C} \). For the moment we shall limit ourselves to complex analytic functions of \( z \), namely \( w = F(z) = u(x, y) + v(x, y)j = \sum_{n=0}^{\infty} c_n z^n \).

In summary, given a point \( z = x + yj \) on the open complex plane, we map it to \( w \in \mathbb{C} \), the complex \( w = u + vj \) plane, and from there to the closed extended complex plane \( w'(z) \). The point of doing this is that it allows us to allow the function \( w'(z) \) to be analytic at the north pole, meaning it can have a convergent Taylor series at the point at infinity \( z \to \infty \). Since we have not yet defined \( dw(z)/dz \), the concept of a complex Taylor series remains undefined.

Möbius bilinear transformation

In mathematics the Möbius transformation has special importance because it is linear in its action. In the engineering literature this transformation is known as the bilinear transformation. Since we are engineers we shall stick with the engineering terminology. But if you wish to read about this on the internet, be sure to also search for the mathematical term, which may be better supported.

When a point on the complex plane \( z = x + yj \) is composed with the bilinear transform \((a, b, c, d \in \mathbb{C})\), the result is \( w(z) = u(x, y) + v(x, y)j \)

\[
w = \frac{az + b}{cz + d} \tag{1.67}
\]

the transformation from \( z \to w \) is a cascade of four independent compositions

1. translation \( (w = z + b; a = 1, b \in \mathbb{C}, c = 0, d = 1) \),
2. scaling \( (w = |a|z; a \in \mathbb{R}, b = 0, c = 0, d = 1) \)
3. rotation \( (w = \frac{a}{|a|}z; a \in \mathbb{C}, b = 0, c = 0, d = |a|) \) and
4. inversion \( (w = \frac{1}{z}; a = 0, b = 1, c = 1, d = 0) \).

Each of these transformations are a special case of Eq. 1.67, with the inversion the most complicated. A wonderful video showing the effect of the bilinear (Möbius) transformation on the plane is available that I highly recommend: https://www.youtube.com/watch?v=0zlFIsUNh04

When the extended plane (Riemann sphere) is analytic at \( z = \infty \), one may take the derivatives there, defining the Taylor series with the expansion point at \( \infty \). When the bilinear transformation rotates the Riemann sphere, the point at infinity is translated to a finite point on the complex plane, revealing the analytic nature at infinity. A second way to transform the point at infinity by the bilinear transformation \( \zeta = 1/z \), mapping a zero (or pole) at \( z = \infty \) to a pole (or zero) at \( \zeta = 0 \). Thus this construction of the Riemann sphere and the Möbius (bilinear) transformation allow us to understand the point at infinity, and treat it like any other point. If you felt that you never understood the meaning of the point at \( \infty \) (likely), this should help.
1.3.10 Lec 18: Complex analytic mappings (Domain-coloring)

One of the most difficult aspects of complex functions of a complex variable is understanding the mapping from \( z = x + yj \) to \( w(z) = u + vj \). For example, \( w(z) = \sin(x) \) is trivial when \( z = x + yj \) is real (i.e., \( y = 0 \)), because \( \sin(x) \) is real. Likewise for the case where \( x = 0 \), where \( \sin(yj) = e^{-y} - e^y - 2/e^y \) is purely imaginary. But the general case, \( w(z) = \sin(z) \in \mathbb{C} \)

\[
\sin(zj) = \sin(xj - y) = j\sinh(z).
\]

is not easily visualized. Thus when \( u(x, y) \) and \( v(x, y) \) are not well known functions, \( w(z) \) can be much more difficult to visualize.

Fortunately with computer software today, this problem can be solved by adding color to the chart. A Matlab/Octave script \( \texttt{zviz.m} \) has been used to make the charts shown here.\(^{61}\) This tool is also known as Domain-coloring.\(^{62}\) Rather than plotting \( u(x, y) \) and \( v(x, y) \) separately, domain-coloring allows us to display the entire function on one chart. Note that for this visualization we see the complex polar form of \( w(s) = |w|e^{jw} \), rather than as the four dimensional Cartesian graph \( w(x + yj) = u(x, y) + v(x, y)j \).

Visualizing complex functions: The mapping from \( z = x + iy \) to \( w(z) = u(x, y) + iv(x, y) \) is a \( 2 \cdot 2 = 4 \) dimensional graph. This is difficult to visualize, because for each point in the domain \( z \), we would like to represent both the magnitude and phase (or real and imaginary parts) of \( w(z) \). A good way to visualize these mappings is to use color (hue) to represent the phase and intensity (dark to light) to represent the magnitude. The Matlab program \( \texttt{zviz.m} \) has been provided to do this (see Lecture 17 on the class website).

To use the program in Matlab/Octave, use the syntax \( \texttt{zviz <function of z>} \) (for example, type \( \texttt{zviz z^2} \)). Note the period between \( z \) and \( z^2 \).

\(^{61}\)http://jontalle.web.engr.illinois.edu/uploads/298/zviz.zip

\(^{62}\)This is also called ‘domain coloring’: https://en.wikipedia.org/wiki/Domain_coloring
Figure 1.18 shows a colorized plot of red is direction. The visible zeros of $w$ axis by $1$ $z$ has a zero (dark) at $w = 1$, since there $\log(1) = 0$ (the imaginary part is zero). Note the branch cut from $x = 0$ to $x = -\infty$. On branches other than the one shown, there is are no zeros, since the phase ($\angle z = 2\pi n$) is not zero.

will render a ‘domain coloring’ (aka colorized) version of the function. Examples you can render with `zviz` are given in the comments at the top of the `zviz.m` program. A good example for testing is `zviz z-sqrt(j)`, which should show a dark spot (a zero) at $(1+1j)/\sqrt{2} = 0.707(1+1j)$.

Example: Figure 1.18 shows a colorized plot of $w(z) = \sin(\pi(z - i)/2)$ resulting from the Matlab/Octave command `zviz sin(pi*(z-i)/2)`. The abscissa (horizontal axis) is the real $x$ axis and the ordinate (vertical axis) is the complex $iy$ axis. The graph is offset along the ordinate axis by $1i$, since the argument $z - i$ causes a shift of the sine function by $1$ in the positive imaginary direction. The visible zeros of $w(z)$ appear as dark regions at $(-2,1), (0,1), (2,1)$. As a function of $x$, $w(x+1j)$ oscillates between red (phase is zero degrees), meaning the function is positive and real, and sea-green (phase is $180^\circ$), meaning the function is negative and real.

Along the vertical axis, the function is either a $\cosh(y)$ or $\sinh(y)$, depending on $x$. The intensity becomes lighter as $|w|$ increases, and white as $w \to \infty$. the intensity becomes darker as $|w|$ decreases, and black as $w \to 0$.

Mathematicians typically use the more abstract (i.e., non–physical) notation $w(z)$, where $w = u + vi$ and $z = x + yi$. Engineers think in terms of a physical complex impedance $Z(s) = R(s) + X(s)j$, having resistance $R(s)$ and reactance $X(s)$ [Ohms], as function of the complex Laplace radian frequency $s = \sigma + \omega j$ [rad], as used, for example, with the Laplace transform (Sect. 1.3.12, p. 91). In Fig. 1.17 we use both notations, with $Z(s) = s$ on the left and $w(z) = z - \sqrt{j}$ on the right, where we show this color code as a 2x2 dimensional domain-coloring graph. Intensity (dark to light) represents the magnitude of the function, while hue (color) represents the phase, where (see Fig. 1.17) red is $0^\circ$, sea-green is $90^\circ$, blue-green is $135^\circ$, blue is $180^\circ$, and violet is $-90^\circ$ (or $270^\circ$). The function $w = s$ has a dark spot (a zero) at $s = 0$, and becomes brighter away from the origin. On the right is $w(z) = z - \sqrt{j}$, which shifts the zero to $z = \sqrt{j}$. Thus domain–coloring gives the full picture of the complex analytic function mappings $w(x, y) = u(x, y) + v(x, y)j$ in colorized polar coordinates.

Two additional examples are given in Fig. 1.19 to help you interpret the two complex mappings $w = e^s$ (left) and its inverse $s = \ln(w)$. The exponential is relatively easy to understand because...
\[ w(s) = |e^{\sigma}e^{i\omega}| = e^{\sigma}. \] The red region is where \( \omega \approx 0 \) in which case \( w \approx e^{\sigma}. \) As \( \sigma \) becomes large and negative, \( w \to 0, \) thus the entire field becomes dark on the left. The field is becoming light on the right where \( w = e^{\sigma} \to \infty. \) If we let \( \sigma = 0 \) and look along the \( \omega \) axis, we see that the function is changing phase, sea-green (90°) at the top and violet (-90°) at the bottom.

In the right panel note the zero for \( \ln(w) = \ln|w| + \omega j \) at \( w = 1. \) The root of the log function is \( \log(w_r) = 0 \) is \( w_r = 1, \phi = 0, \) since \( \log(1) = 0. \) More generally, the log of \( w = |w|e^{\phi j} \) is \( s = \ln|w| + \phi j. \) Thus \( s(w) \) can be zero only when the angle of \( w \) is zero.

The \( \ln(w) \) function has a branch cut along the \( \phi = 180^\circ \) axis. As one crosses over the cut, the phase goes above 180°, and the plane changes to the next sheet of the log function. The only sheet with a zero is the principle value, as shown. All others, the log function is either increasing or decreasing monotonically, and there is no zero, as seen on sheet 0 (the one showing in Fig. 1.19).

### 1.3.11 Lec 19: Signals: Fourier transforms

Two basic transformations in engineering mathematics are the Fourier and the Laplace transforms, which deal with time–frequency analysis.

The Fourier transform takes a time domain signal \( f(t) \in \mathbb{R} \) and transforms it to the frequency domain \( \omega \in \mathbb{R}, \) where it is complex \( (F(\omega) \in \mathbb{C}). \) For the Fourier transform, both the time \(-\infty < t < \infty \) and frequency \( -\infty < \omega < \infty \) are strictly real. The relationship, between \( f(t) \) and its transform \( F(\omega), \) is indicated by the double arrow symbol

\[ f(t) \leftrightarrow F(\omega). \]

Since the FT obeys superposition, it is possible to define a FT of a complex time function \( f(t) \in \mathbb{C}, t \in \mathbb{R}. \) This is useful in certain engineering applications (i.e, Hilbert envelope, Hilbert transforms).

The Laplace transform takes a real signals \( f(t) \in \mathbb{R}, \) as a function of real time \( t \in \mathbb{R}, \) that are causal, i.e., strictly zero for negative time \( (f(t) = 0 \text{ for } t < 0), \) and transforms them into complex functions \( F(s) \in \mathbb{C} \) of complex frequency \( s = \sigma + \omega j. \) As for the Fourier transform, there is the notation \( f(t) \leftrightarrow F(s). \)

When a signal is zero for negative time \( f(t < 0) = 0, \) it is said to be causal, and the resulting transform \( F(s) \) is then complex analytic over significant regions of the \( s \) plane. For a function of time to be causal, time must be real \( (t \in \mathbb{R}), \) since if it were complex, it would lose the order property (thus it could not be causal). It is helpful to emphasize the causal nature of \( f(t)u(t) \) by indicating the causal nature, using the Heaviside step function \( u(t). \)

Restrictions on a function (e.g., real, causal, periodic, positive real part, etc.) are called a symmetry property. There are many forms of symmetry (Section 1.3.13, p. 95). The concept of symmetry is very general and widely used in both mathematics and physics, where it is more generally known as Group theory. We shall restrict ourselves to only a few very basic symmetries (Section 3.5.1, p. 205).

**Notation:** It is common in engineering and Physics books and papers to use the case of the variable to indicate the type of argument. For example a time function \( f(t), \) where \( t \) has units of seconds \([s]\), is in lower case, whereas its Fourier transform, a function of frequency, having units of either Hertz \([Hz]\) or radians \([rad]\) is written using upper case \( F(\omega). \) This helps the reader parse the type of variable under consideration. This is a helpful piece of notation, but not entirely in agreement with notation used in Mathematics.
Exercise: Consider the $\mathcal{F}T$ as a scalar (dot) product (Eq. 1.47, p. 73) between “vectors” $f(t)$ and $e^{-j\omega_k t}$

$$f(t) \cdot e^{-j\omega_k t} = \frac{1}{T_o} \int_0^{T_o} f(t)e^{-j\omega_k t} dt,$$

where $\omega_0 = 2\pi/T_o$ and $\omega_k = k\omega_0$. If $f(t) = e^{j\omega_n t}$ what is the value of the scalar product?

**Sol:** Evaluating the scalar product we find

$$e^{j\omega_n t} \cdot e^{-j\omega_k t} = \frac{1}{T_o} \int_0^{T_o} e^{j\omega_n t} e^{-j\omega_k t} dt = \frac{1}{T_o} \int_0^{T_o} e^{2\pi j(n-k)t/T_o} dt = \begin{cases} 1 & n = k \\ 0 & n \neq k \end{cases}.$$

The two signals (vectors) are orthogonal.

Periodic signals: Besides these two basic types of time–frequency transforms, there are several variants that depend on the symmetry in time and frequency. For example, when the time signal is sampled (discrete in time), the frequency response becomes periodic, leading to the Discrete-time Fourier transform (DTFT). When a time response is periodic, the frequency response is sampled (discrete in frequency), leading to the Fourier Series. These two symmetries may be simply characterized as periodic in time $\Rightarrow$ discrete in frequency, and periodic in frequency $\Rightarrow$ discrete in time. When a function is both discrete in time and frequency, it is necessarily periodic in time and frequency, leading to the Discrete Fourier Transform (DFT). The DFT is typically computed with an algorithm called the Fast Fourier Transform (FFT), which can dramatically speed up the calculation when the data is a power of 2 in length.

A very important symmetry is when functions that are causal (in time) and periodic (in frequency). The best known example is the class of signals that have z transforms, which are causal (one-sided in time) discrete-time signals. The harmonic series (Eq. 1.34, p. 61) is the z-transform of the discrete-time step function, thus, by symmetry, analytic within the RoC in the complex frequency ($z$) domain.

The double brackets on $f((t))_{T_o}$ indicates that $f(t)$ is periodic in $t$ with period $T_o$, i.e., $f(t) = f(t + kT_o)$ for all $k \in \mathbb{N}$. Averaging over on period and dividing by the $T_o$ gives the average value of $f(t)$.

Definition of the Fourier transform: The forward transform takes $f(t)$ to $F(\omega)$ while the inverse transform takes $F(\omega)$ to $f(t)$. The tilde symbol indicates that in general recovered inverse transform signal can be slightly different from $f(t)$. We give examples of this in Table 1.4.

$$F(\omega) = \int_{-\infty}^{\infty} f(t)e^{-j\omega t} dt$$

$$f(t) = \frac{1}{2\pi} \int_{-\infty}^{\infty} F(\omega)e^{j\omega t} d\omega$$

Properties of Fourier Transforms:

1. Both time $t$ and frequency $\omega$ are real.

2. For the forward transform (time to frequency), the sign of the exponential is negative.
Table 1.4: The following table provides a brief table of simple Fourier Transforms. Note \( a > 0 \in \mathbb{R} \) has units [rad/s]. To flag this necessary condition, we use \(|a|\) to assure this condition will be met. The other constant \( T_o \in \mathbb{R} [s] \) has no restrictions, other than being real. Complex constants may not appear as the argument to a delta function, since complex numbers do not have the order property.

<table>
<thead>
<tr>
<th>( f(t) \leftrightarrow F(\omega) )</th>
<th>Name</th>
</tr>
</thead>
<tbody>
<tr>
<td>( \tilde{\delta}(t) \leftrightarrow 1(\omega) \equiv 1 \forall \omega )</td>
<td>Dirac</td>
</tr>
<tr>
<td>( 1(t) \equiv 1 \forall t \leftrightarrow 2\pi \delta(\omega) )</td>
<td>Dirac</td>
</tr>
<tr>
<td>( \text{sgn}(t) = \frac{t}{</td>
<td>t</td>
</tr>
<tr>
<td>( \tilde{u}(t) = \frac{1(t) + \text{sgn}(t)}{2} \leftrightarrow \pi \delta(\omega) + \frac{1}{j\omega} \equiv \tilde{U}(\omega) )</td>
<td>step</td>
</tr>
<tr>
<td>( \tilde{\delta}(t - T_o) \leftrightarrow e^{-j\omega T_o} )</td>
<td>delay</td>
</tr>
<tr>
<td>( \tilde{\delta}(t - T_o) * f(t) \leftrightarrow F(\omega)e^{-j\omega T_o} )</td>
<td>delay</td>
</tr>
<tr>
<td>( \tilde{u}(t)e^{-</td>
<td>a</td>
</tr>
<tr>
<td>( \text{rec}(t) = \frac{1}{T_o} [\tilde{u}(t) - \tilde{u}(t - T_o)] \leftrightarrow \frac{1}{T_o} \left(1 - e^{-j\omega T_o}\right) )</td>
<td>pulse</td>
</tr>
<tr>
<td>( \tilde{u}(t) * \tilde{u}(t) \leftrightarrow \delta^2(\omega) )</td>
<td>Not defined</td>
</tr>
</tbody>
</table>

FT Properties

\[
\frac{d}{dt}v(t) \leftrightarrow j\omega V(\omega) \quad \text{deriv}
\]
\[
f(t) \ast g(t) \leftrightarrow F(\omega)G(\omega) \quad \text{conv}
\]
\[
f(t)g(t) \leftrightarrow \frac{1}{2\pi} F(\omega) \ast G(\omega) \quad \text{conv}
\]
\[
f(at) \leftrightarrow \frac{1}{a} F\left(\frac{\omega}{a}\right) \quad \text{scaling}
\]
3. The limits on the integrals in both the forward and reverse FTs are \([-\infty, \infty]\).

4. When taking the inverse Fourier transform, the scale factor of \(1/2\pi\) is required to cancel the \(2\pi\) in the differential \(d\omega\).

5. The Fourier step function may be defined by the use of superposition of 1 and \(\text{sgn}(t) = t/|t|\) as

\[
\tilde{u}(t) \equiv \frac{1 + \text{sgn}(t)}{2} = \begin{cases} 
1 & \text{if } t > 0 \\
1/2 & t = 0 \\
0 & \text{if } t < 0 
\end{cases}
\]

Taking the FT of a delayed step function

\[
\tilde{u}(t - T_o) \leftrightarrow \frac{1}{2} \int_{-\infty}^{\infty} [1 - \text{sgn}(t - T_o)] e^{-j\omega t} dt = \pi \tilde{\delta}(\omega) + \frac{e^{-j\omega T_o}}{j\omega}
\]

Thus the FT of the step function has the term \(\pi \tilde{\delta}(\omega)\) due to the 1 in the definition of the Fourier step. This term introduces a serious flaw with the FT of the step function: While it appears to be causal, it is not.

6. The convolution \(\tilde{u}(t) \ast \tilde{u}(t)\) is not defined because both 1 \(\ast\) 1 and \(\tilde{\delta}^2(\omega)\) do not exist (and cannot be defined).

7. The inverse FT has convergence problems whenever there is a discontinuity in the time response. This we indicate with a hat over the reconstructed time response. The error between the target time function and the reconstructed is zero in the root-mean sense, but not pointwise.

Specifically, at the discontinuity point for the Fourier step function \((t = 0)\), \(\tilde{u}(t) \neq u(t)\), yet \(\int |\tilde{u}(t) - u(t)|^2 dt = 0\). At the point of the discontinuity the reconstructed function displays Gibbs ringing (it oscillates around the step, hence does not converge at the jump).\(^{63}\) The \(\mathcal{LT}\) does not exhibit Gibbs ringing.

8. The FT is not always analytic in \(\omega\), as in this example of the step function. The step function cannot be expanded in a Taylor series about \(\omega = 0\), because \(\tilde{\delta}(\omega)\) is not analytic in \(\omega\).

9. The Fourier \(\delta\) function is denoted \(\tilde{\delta}(t)\), to differentiate it from the Laplace delta function \(\delta(t)\). They differ because the step functions differ, due to the convergence problem described above.

10. One may define

\[
\tilde{u}(t) = \int_{-\infty}^{t} \tilde{\delta}(t) dt,
\]

and define the somewhat questionable notation

\[
\tilde{\delta}(t) = \frac{d}{dt} \tilde{u}(t),
\]

since the Fourier step function is not analytic.

\(^{63}\)https://en.wikipedia.org/wiki/Gibbs_phenomenon
11. The rec\( (t) \) function is defined as
\[
\text{rec}(t) = \frac{\tilde{u}(t) - \tilde{u}(t - T_o)}{T_o} = \begin{cases} 
0 & \text{if } t > 0 \\
1/T_o & 0 < t < T_o \\
0 & \text{if } t < 0 
\end{cases}.
\]
It follows that \( \tilde{\delta}(t) = \lim_{T_o \to 0} \). Like \( \tilde{\delta}(t) \), the rec\( (t) \) has unit area.

1.3.12 Lec 20: Systems: Laplace transforms

When dealing with engineering problems it is convenient to separate the signals we use from the systems that process them. We do this by treating signals, such as a music signal, differently from a system, such as a filter. In general signals may start and end at any time. The concept of causality has no mathematical meaning in signal space. Systems, on the other hand, obey very rigid rules (to assure that they remain physical). These physical restrictions are described in terms of the Network Postulates, which are first discussed in Sect. 1.3.13, and then in greater detail in Sect. 3.5.1.

Definition of the Laplace transform: The forward and inverse Laplace transforms are
\[
\begin{align*}
F(s) &= \int_{0^-}^{\infty} f(t)e^{-st}dt \\
f(t) &= \frac{1}{2\pi j} \int_{\sigma_o - \infty j}^{\sigma_o + \infty j} F(s)e^{st}ds
\end{align*}
\] (1.70)
\[
F(s) \leftrightarrow f(t) \\
f(t) \leftrightarrow F(s)
\] (1.71)

1. Time \( t \in \mathbb{R} [s] \) and Laplace frequency [rad] are defined as \( s = \sigma + \omega j \in \mathbb{C} \).

2. Given a Laplace transform (LT) pair \( f(t) \leftrightarrow F(s) \), in the engineering literature, the time domain is always lower case \([f(t)]\) and causal (i.e., \( f(t < 0) = 0 \)) and the frequency domain is upper-case [e.g. \( F(s) \)]. Maxwell’s venerable equations are the unfortunate exception to this otherwise universal rule.

3. When taking the forward transform \( (t \to s) \), the sign of the exponential is negative. This is necessary to assure that the integral converges when the integrand \( f(t) \to \infty \) as \( t \to \infty \). For example, if \( f(t) = e^t u(t) \) (i.e., without the negative \( \sigma \) exponent), the integral does not converge.

4. The target time function \( f(t < 0) = 0 \) (i.e., it must be causal). The time limits are \( 0^- < t < \infty \). Thus the integral must start from slightly below \( t = 0 \) to integrate over a delta functions at \( t = 0 \). For example if \( f(t) = \delta(t) \), the integral must include both sides of the impulse. If you wish to include non-causal functions such as \( \delta(t + 1) \) it is necessary to extend the lower time limit. In such cases simply set the lower limit of the integral to \(-\infty\), and let the integrand \( (f(t)) \) determine the limits.

5. The limits on the integrals of the forward transform are \( t: (0^-, \infty) \in \mathbb{R} \), and the reverse LTs are \([\sigma_o - \infty j, \sigma_o + \infty j] \in \mathbb{C}\). These limits will be further discussed in Section 1.4.9 (p. 116).

6. When taking the inverse Laplace transform, the normalization factor of \( 1/2\pi j \) is required to cancel the \( 2\pi j \) in the differential \( ds \) of the integral.
7. The frequency for the LT must be complex, and in general \( F(s) \) is complex analytic for \( \sigma > \sigma_o \). It follows that the real and imaginary parts of \( F(s) \) are related. Given \( \Re\{F(s)\} \) it is possible to find \( \Im\{F(s)\} \) (Boas, 1987). More on this in Section 1.3.12 (p. 91).

8. To take the inverse Laplace transform, we must learn how to integrate in the complex \( s \) plane. This will be explained in Sections 1.4.5-1.4.9 (p. 112-116).

9. The Laplace step function is defined as

\[
\begin{array}{l}
u(t) = \int_{-\infty}^{t} \delta(t) dt = \begin{cases} 
1 & \text{if } t > 0 \\
\text{NaN} & \text{if } t = 0 \\
0 & \text{if } t < 0 
\end{cases}.
\end{array}
\]

Alternatively one could define \( \delta(t) = du(t)/dt \).

10. It is easily shown that \( u(t) \leftrightarrow 1/s \) by direct integration

\[
F(s) = \int_0^\infty u(t) e^{-st} dt = \left. \frac{e^{-st}}{s} \right|_0^\infty = \frac{1}{s}.
\]

With the LT step \( (u(t)) \) there is no Gibbs ringing effect.

11. In many physical applications, the Laplace transform takes the form of a ratio of two polynomials. In such case the roots of the numerator polynomial are call the zeros while the roots of the denominator polynomial are called the poles. For example the LT of \( u(t) \leftrightarrow 1/s \) has a pole at \( s = 0 \), which represents integration, since

\[
u(t) \star f(t) = \int_{-\infty}^{t} f(\tau) d\tau \leftrightarrow \frac{F(s)}{s}.
\]

12. The LT is quite different from the FT in terms of its analytic properties. For example, the step function \( u(t) \leftrightarrow 1/s \) is complex analytic everywhere, except at \( s = 0 \). The FT of \( 1 \leftrightarrow 2\pi \delta(\omega) \) is not analytic anywhere.

13. Dilated step function \( (a \in \mathbb{R}) \)

\[
u(at) \leftrightarrow \int_{-\infty}^{\infty} u(at) e^{-st} dt = \frac{1}{a} \int_{-\infty}^{\infty} u(\tau) e^{-\left(s/a\right)\tau} e^{-s/t} \tau = \frac{a}{|a|} \frac{1}{s} = \pm \frac{1}{s},
\]

where we have made the change of variables \( \tau = at \). The only effect that \( a \) has on \( u(at) \) is the sign of \( t \), since \( u(t) = u(2t) \). However \( u(-t) \neq u(t) \), since \( u(t) \cdot u(-t) = 0 \), and \( u(t) + u(-t) = 1 \), except at \( t = 0 \), where it is not defined.

Once complex integration in the complex plane has been defined (Section 1.4.2, p. 101), we can justify the definition of the inverse LT (Eq. 1.70).

Example: Here \( s = \sigma + j\omega \) is the complex Laplace frequency in [radians] and \( t \) is time in [seconds].
1.3. STREAM 2: ALGEBRAIC EQUATIONS (12 LECTURES)

Table 1.5: The following table provides a brief table of Laplace Transforms of \( f(t) \), \( \delta(t) \), \( u(t) \), \( \text{rect}(t), T_o \), \( p, c, e \in \mathbb{R} \) and \( F(s) \), \( G(s) \), \( s, a \in \mathbb{C} \). Given a Laplace transform (LT) pair \( f(t) \leftrightarrow F(s) \), the frequency domain will always be upper-case [e.g. \( F(s) \)] and the time domain lower case \( [f(t)] \) and causal (i.e., \( f(t < 0) = 0 \)). An extended table of transforms is given in Table 1.6 on page 95.

<table>
<thead>
<tr>
<th>( f(t) \leftrightarrow F(s) )</th>
<th>( t \in \mathbb{R}; \ s, F(s) \in \mathbb{C} )</th>
<th>Name</th>
</tr>
</thead>
<tbody>
<tr>
<td>( \delta(t) \leftrightarrow 1 )</td>
<td></td>
<td>Dirac</td>
</tr>
<tr>
<td>( \delta(</td>
<td>a</td>
<td>t) \leftrightarrow \frac{1}{</td>
</tr>
<tr>
<td>( \delta(t - T_o) \leftrightarrow e^{-st_o} )</td>
<td></td>
<td>delayed Dirac</td>
</tr>
<tr>
<td>( \delta(t - T_o) \star f(t) \leftrightarrow F(s)e^{-st_o} )</td>
<td></td>
<td>–</td>
</tr>
<tr>
<td>( \sum_{n=0}^{\infty} \delta(t - nT_o) = \frac{1}{1 - \delta(t - T_o)} \leftrightarrow \frac{1}{1 - e^{-st_o}} = \sum_{n=0}^{\infty} e^{-snT_o} )</td>
<td>one-sided impulse train</td>
<td></td>
</tr>
<tr>
<td>( u(t) \leftrightarrow \frac{1}{s} )</td>
<td></td>
<td>step</td>
</tr>
<tr>
<td>( u(-t) \leftrightarrow -\frac{1}{s} )</td>
<td></td>
<td>anti-causal step</td>
</tr>
<tr>
<td>( u(at) \leftrightarrow \frac{a}{s} )</td>
<td>( a \neq 0 \in \mathbb{R} )</td>
<td>dilated or reversed step</td>
</tr>
<tr>
<td>( e^{at}u(t) \leftrightarrow \frac{1}{s + a} )</td>
<td>( a &gt; 0 \in \mathbb{R} )</td>
<td>damped step</td>
</tr>
<tr>
<td>( e^ {-at}u(t) \leftrightarrow \frac{1}{s + a} )</td>
<td>( a \in \mathbb{C} )</td>
<td>modulated step</td>
</tr>
<tr>
<td>( \cos(at)u(t) \leftrightarrow \frac{1}{2} \left( \frac{1}{s - a} + \frac{1}{s + a} \right) )</td>
<td>( a \in \mathbb{R} )</td>
<td>( \cos )</td>
</tr>
<tr>
<td>( \sin(at)u(t) \leftrightarrow \frac{1}{2j} \left( \frac{1}{s - a} - \frac{1}{s + a} \right) )</td>
<td>( a \in \mathbb{C} )</td>
<td>“damped” sin</td>
</tr>
<tr>
<td>( u(t - T_o) \leftrightarrow \frac{1}{s} e^{-st_o} )</td>
<td>( T_0 &gt; 0 \in \mathbb{R} )</td>
<td>time delay</td>
</tr>
<tr>
<td>( \text{rect}(t) = \frac{1}{T_o} [u(t) - u(t - T_o)] \leftrightarrow \frac{1}{sT_o} \left( 1 - e^{-st_o} \right) )</td>
<td>rect-pulse</td>
<td></td>
</tr>
</tbody>
</table>

\[
\begin{align*}
  u(t) \star u(t) &= tu(t) \leftrightarrow 1/s^2 \\
  u(t) \star u(t) \star u(t) &= \frac{1}{2} t^2 u(t) \leftrightarrow 1/s^3 \\
  t^p u(t) &= \frac{\Gamma(p + 1)}{s^{p+1}} \\
  p > 0 &\in \mathbb{R}
\end{align*}
\]
As discussed in Section 1.4.8 (p. 115), we must use the Cauchy Residue Theorem (CRT), requiring closure of the contour $\mathcal{C}$ at $\omega j \rightarrow \pm j\infty$

$$\oint_{\mathcal{C}} = \int_{\sigma_0 + j\infty} + \int_{\infty}$$

where the path represented by ‘$\subset \infty$’ is a semicircle of infinite radius. For a causal, ‘stable’ (e.g. doesn’t “blow up” in time) signal, all of the poles of $F(s)$ must be inside of the Laplace contour, in the left-half $s$-plane.

Hooke’s Law for a spring states that the force $f(t)$ is proportional to the displacement $x(t)$, i.e., $f(t) = Kx(t)$. The formula for a dash-pot is $f(t) = Rv(t)$, and Newton’s famous formula for mass is $f(t) = d[Mv(t)]/dt$, which for constant $M$ is $f(t) = Mdv/dt$.

The equation of motion for the mechanical oscillator in Fig. 1.20 is given by Newton’s second law; the sum of the forces must balance to zero

$$M \frac{d^2 x(t)}{dt^2} + R \frac{dx(t)}{dt} + Kx(t) = f(t). \quad (1.72)$$

These three constants, the mass $M$, resistance $R$ and stiffness $K$ are all real and positive. The dynamical variables are the driving force $f(t) \leftrightarrow F(s)$, the position of the mass $x(t) \leftrightarrow X(s)$ and its velocity $v(t) \leftrightarrow V(s)$, with $v(t) = dx(t)/dt \leftrightarrow sX(s)$.

Newton’s second law (c1650) is the mechanical equivalent of Kirchhoff’s (c1850) voltage law (KCL), which states that the sum of the voltages around a loop must be zero. The gradient of the voltage results in a force on a charge (i.e., $E = qE$).

Equation 1.72 may be re-expressed in terms of impedances, the ratio of the force to velocity, once it is transformed into the Laplace frequency domain.

The key idea that every impedance must be complex analytic and $\geq 0$ for $\sigma > 0$ was first proposed by Otto Brune in his PhD at MIT, supervised by a student of Arnold Sommerfeld, Ernst Guillianman, an MIT ECE professor, who played a major role in the development of circuit theory. Brune’s primary (non-MIT) advisor was Cauer, who was also well trained in 19th century German mathematics.

Summary: While the definitions of the FT ($\mathcal{F}$) and LT ($\mathcal{L}$) transforms appear superficially similar, they are not. The key difference is that the time response of the Laplace transform is causal, leading to a complex analytic frequency responses. The frequency response of the Fourier transform is real, thus typically not analytic. These are not superficial differences. The concept of

\[64\text{It must be noted that Prof. ‘Mac’ Van Valkenburg from the University of IL, was arguably more influential in circuit theory, during the same period. Mac’s book are certainly more accessible, but perhaps less widely cited.}\]
symmetry is helpful in understanding the many different types of time-frequency transforms. Two fundamental types of symmetry are causality and periodicity.

The Fourier transform $\mathcal{F}$ characterizes the steady-state response while the Laplace transform $\mathcal{L}$ characterizes both the transient and steady-state response. Given a system response $H(s) \leftrightarrow h(t)$ with input $x(t)$, the output is

$$y(t) = h(t) * x(t) \leftrightarrow Y(\omega) = H(s) \bigg|_{s=j\omega} X(\omega).$$

Table 1.6: Functional relationships between Laplace Transforms.

<table>
<thead>
<tr>
<th>$\mathcal{L}$ functional properties</th>
</tr>
</thead>
<tbody>
<tr>
<td>$f(t) * g(t) = \int_{t=0}^{t} f(t-\tau)g(\tau)d\tau \leftrightarrow F(s)G(s)$</td>
</tr>
<tr>
<td>$u(t) * f(t) = \int_{t=0}^{t} f(t)dt \leftrightarrow \frac{F(s)}{s}$</td>
</tr>
<tr>
<td>$f(at)u(at) \leftrightarrow \frac{1}{a}F\left(\frac{s}{a}\right)$</td>
</tr>
<tr>
<td>$f(t)e^{-at}u(t) \leftrightarrow F(s+a)$</td>
</tr>
<tr>
<td>$f(t-T)e^{-a(t-T)}u(t-T) \leftrightarrow e^{-sT}F(s+a)$</td>
</tr>
<tr>
<td>$f(-t)u(-t) \leftrightarrow F(-s)$</td>
</tr>
<tr>
<td>$f(-t)e^{-at}u(-t) \leftrightarrow F(a-s)$</td>
</tr>
<tr>
<td>$\frac{d}{dt}f(t) = \delta'(t) * f(t) \leftrightarrow sF(s)$</td>
</tr>
<tr>
<td>$\sin(t)u(t) \leftrightarrow \tan^{-1}(1/s)$</td>
</tr>
</tbody>
</table>

1.3.13 Lec 21: Ten network postulates

Solutions of differential equations, such as the wave equation, are conveniently described in terms of mathematical properties, which we present here in terms of 10 network postulates:

(P1) causality (non-causal/acausal): Causal systems respond when acted upon. Virtually all physical systems obey causality. An example of a causal system is an integrator, which has a response of a step function. Filters are also examples of causal systems. Signals represent acausal responses. They do not have a clear beginning or end, such as the sound of the wind or traffic noise.
LINEARITY (nonlinear): Linear systems obey superposition. If two signals \( x(t) \) and \( y(t) \) are the inputs to a linear system, producing outputs \( x'(t) \) and \( y'(t) \), then if the inputs are presented together as \( ax(t) + by(t) \), with weights \( a, b \in \mathbb{R} \), then the output will be \( ax'(t) + by'(t) \). If either \( a, b \) are zero, that signal is removed from the output.

Nonlinear system mix two inputs, thereby producing other signals not present in the input. For example, if the inputs to a nonlinear system are two sine waves, the output will contain distortion components, having frequencies not present at the input. An example of a nonlinear system is one that multiplies the two inputs. A second is a diode, which rectifies a signal, letting current flow only in one direction. Most physical systems have some degree of nonlinear response, but this is not always desired. Other systems are designed to be nonlinear, such as the diode example.

PASSIVE (active): An active system has a power source, such as a battery while a passive system has no power source. While you may consider a transistor amplifier to be active, it is only so when connected to a power source.

REAL (complex) time response: Typically systems are “real in, real out.” They do not naturally have complex responses (real and imaginary parts). While a Fourier transform takes real inputs and produces complex outputs, this is not an example of a complex time response. P4 is a characterization of the input signal, not its Fourier transform.

TIME-INVARIANT (time varying): For a system to be a time varying system its properties must depend on when the signal starts and stops. If the output, relative to the input, is independent of the starting time, then the system is time-invariant (static).

RECIPROCAL (non- or anti-reciprocal): In many ways this is the most difficult property to understand. It is best characterized by the ABCD matrix. If \( B = C \) the system is said to be reciprocal. If \( B = -C \) it is said to be anti-reciprocal. The loudspeaker is anti-reciprocal, which is why it is modeled by the gyrator rather than a transformer. All non-reciprocal systems are modeled by such gyrator, which swap the force and flow variables.

REVERSIBILITY (non-reversible): If the system can be flipped, between input and output, the it is said to be reversible. In terms of the ABCD parameters, if \( A = D \) is is reversible.

SPACE-INVARIANT (space-variant): If a system operates independently as a function of where it physically is in space, then it is space-invariant. When the parameters that characterize the system depend on position, it is space-variant.

DETERMINISTIC (random): Predictability depends on the light cone. Given the wave equation, along with the boundary conditions, the solution may be deterministic, or not, depending on the extent of the “light cone.” Consider the wave with a wave propagating out into uncharted territory. When the wave hits an object, the reflection will return a wave that might considered not be predicted. This is an example of the boundary condition being the radiation impedance, which is not known in advance. One must do the experiment to do solve the problem, and the radiation impedance is not known in advance. To determine it, one must do the experiment. In this example, that determination could take light years. Thus is, I believe, an example of a “random” system. But the system is deterministic only if the radiation impedance is known accurately, in advance. To define the impedance requires that it be time
invariant (P5). One cannot look beyond the light-cone, which depends on the system to be time invariant.

(P10) quasi-static (aka, uni-modal assumption): Quasi-statics follows from systems that have dimensions that are small compared to the local wavelength. This is a very general assumption, that must be false when the frequency is raised and the wavelength becomes short. Thus this is also known as the long-wavelength approximation. It is a very powerful tool, frequently used in modeling 1 dimensional systems, such as transmission lines, horns and one-dimensional systems. Quasi-statics is normally stated as \( \lambda < ka \), where \( k = \frac{c_o}{f} \) and \( a \) is the minimum dimension of the system.

In acoustics this is restrictive because the speed of sound is only \( c_o = 345 \text{ [m/s]} \), as compared to the speed of light which is \( c_o = 0.3 \times 10^8 \text{ [m/s]} \). Given the wave speed the wavelength is \( \lambda = \frac{c_o}{f} \), we see that the wavelength is directly proportional to the wave speed, and inversely proportional to frequency.

Postulate P10 is closely related to the Feynman (1970c, Ch. 12-7) titled The “underlying unity” of nature, well worth a read once you have understood the meaning of quasi-statics. He asks the question “Why do we need to treat the fields as smooth?” My answer is so that we may model them analytic functions. In some ways this is a mathematical approximation which critically depends on quasi-statics, which in turn depends on the frequency being low enough that the wavelength is much greater than the distance between molecules, air particles, electrons, etc.

This raises the question “Are Maxwell’s equations a band-limited approximation to reality? Personally I have no idea what the answer is. This is a question waiting to be answered (once asked).

Each postulate has two (or more) categories. For example for (P2) a system is either linear or non-linear and for (P1) is either causal, non-causal or acausal. P6 and P9 only apply to 2-port networks (those having an input and an output). The others can apply to both a 2– or 1–port networks (e.g., an impedance is a 1-port).

Related forms of these postulates had been circulating in the literature for many years, widely accepted in the network theory literature (Van Valkenburg, 1964a,b; Ramo et al., 1965). The first six of these were formally introduced Carlin and Giordano (1964), and (P7-P9) were added by Kim et al. (2016).

Are these postulates independent of each other? Clearly they are not. It would be helpful to reduce this set to the minimum set, but not here, or now.

1.3.14 Lec 22: Exam II (Evening Exam)
1.4 Stream 3a: Scalar (i.e., Ordinary) Differential Equations

Stream 3 is $\infty$, a concept which typically means unbounded (immeasurably large), but in the case of calculus, $\infty$ means infinitesimal (immeasurably small), since taking a limit requires small numbers. Taking a limit means you may never reaching the target, a concept that the Greeks called Zeno’s paradox (Stillwell, 2010, p. 76).

When speaking of the class of ordinary (versus vector) differential equations, the term scalar is preferable, since the term “ordinary” is vague, if not a meaningless label. There are a special subset of fundamental theorems for scalar calculus, all of which are about integration, as summarized in Table 1.8 (p. 149), starting with Leibniz’s theorem. These will be discussed below, and more extensively in Lec. 1.4.1 (p. 100) and Chapters 4 (p. 213) and 5 (p. 213).

Following the integral theorems on scalar calculus, are those on vector calculus, without which there could be no understanding of Maxwell’s equations. Of these, the fundamental theorem of complex calculus (aka, Helmholtz decomposition), Gauss’s Law and Stokes theorem, form the cornerstone of modern vector field analysis. These theorems allow one to connect the differential (point) and macroscopic (integral) relationships. For example, Maxwell’s equations may be written as either vector differential equations, as shown by Heaviside (along with Gibbs and Hertz), or in integral form. It is helpful to place these two forms side-by-side, to fully appreciate their significance. To understand the differential (microscopic) view, one must understand the integral (macroscopic) view. These are presented in Sections 1.5.10 (p. 144) and Fig. 1.31 (p. 148).

Chronological history post 16th century

16th: Bombelli 1526-1572

17th: Galileo 1564-1642, Kepler 1571-1630, Newton 1642-1727 Principia 1687; Mersenne; Huygen; Pascal; Fermat, Descartes (analytic geometry); Bernoullis Jakob, Johann & son Daniel

18th: Euler 1707-1783 Student of Johann Bernoulli; d’Alembert 1717-1783 Kirchhoff, Lagrange; Laplace; Gauss 1777-1855


20th: Hilbert; Einstein; …

Time Line

|1525 |1600 |1700 |1800 |1875 |1925

Bombelli Descartes Huygen

Newton Johann Bernoulli Euler

Mersenne Daniel Bernoulli Rayleigh

Galileo Jacob Bernoulli Cauchy

Fermat d’Alembert Helmholtz

 Mozart US Civil War

Figure 1.21: Time-line of the four centuries from the 16th and 20th CE. Given the time line, it is likely that Bombelli’s discovery of Diophantus’ book on “Arithmetic” in the Vatican library, triggered many of the ideas presented by Descartes and Fermat, followed by others (i.e., Newton). Bombelli’s discovery might be considered as a magic moment in mathematics. The vertical red lines connect mentor-student relationships. Figure 1.4 (p. 22) for a closer look at the 15-18c, and Fig. 1.12 (p.54) for a closer look at the 18-20c. 65 https://en.wikipedia.org/wiki/History_of_Maxwell\%27s_equations
The beginning of modern mathematics

The birth of mathematics as we know it today, occurred during the 16th to 18th centuries, perhaps starting with Galileo, Descartes, Fermat, Newton, the Bernoulli family, and Euler, as outlined in Fig. 1.21 (p. 98). Galileo was formidable, due to his fame, fortune, and his “successful” stance against the powerful Catholic establishment. His creativity in scientific circles was certainly well known due to his many skills and accomplishments. Fermat and Descartes were at the forefront of merging algebra and geometry, but while Fermat kept meticulous notebooks, he did not publish what he knew, and tended to be secretive.

Regarding the development of calculus, much was yet to be done by Newton and Leibniz, using term by term integration of functions, based on a Taylor series representation. This was a powerful technique, but as stated, incomplete, because the series is only valid for single-valued functions, within the RoC. But more importantly, Newton (and others) failed to recognize the powerful generalization to complex analytic functions. The first breakthrough was Newton’s publication of Principia (1687), and a second was Riemann (1851), with his thesis, advised by Gauss.

Following Newton’s lead, the secretive and introverted behavior of the typical mathematician dramatically changed with the Bernoulli family (Fig. 1.21, p. 98). The oldest brother, Jacob, taught his much younger brother Johann, who then taught his son Daniel, and Johann’s star pupil, Euler. Euler first mastered all the tools and then published, with a prolific voracity previously unknown.

Euler and the circular functions: The first major task was to understand the family of analytic circular functions, $e^x, \sin(x), \cos(x)$, and $\log(x)$, a task begun by the Bernoulli family, but mastered by Euler. Euler sought relations between these many functions, some of which may not be thought of as being related, such as the log and sin functions. The connection that may “easily” be made is through their Taylor series (Eq. 1.33, p. 60). By the manipulation of the analytic series representations, the relationship between $e^x$, and the $\sin(x)$ and $\cos(x)$ was precisely captured by

$$e^{xj} = \cos(x) + j\sin(x),$$

and its analytic inverse (Greenberg, 1988, p. 1135)

$$\tan^{-1}(z) = \frac{1}{2j} \ln \left( \frac{1-jz}{1+jz} \right).$$  \hspace{1cm} (1.73)

The solution is in Section 4.1.1 (p. 213).

These not only relate the exponential function to the circular functions, but also contains $j = \sqrt{-1}$. While every high school student has learned Euler’s relation, it is doubtful (i.e., rare) that they fully appreciate the significance of complex analytic functions (Eq. 1.39, p. 64). While Euler found these relationships using real power series (i.e., analytic functions), neither he, nor those who followed, seemed to fully appreciated the importance of complex analytic, and its role. For example, Newton famously ignored imaginary numbers, and called them imaginary in a disparaging (pejorative) way. Given Newton’s prominence, that certainly must have attenuated interest in complex algebra, even though it had been previously quantified by Bombelli in 1525, likely based on his serendipitous finding of Diophantus’ book Arithmetic in the Vatican library.

While Euler was fluent with $j = \sqrt{-1}$, he did not consider functions to be complex analytic. That concept was first explored by Cauchy, almost a century later. The missing link to the concept of complex analytic is the definition of the derivative with respect to the complex argument (i.e., $dF(s)/ds$ with $s = \sigma + \omega j$), without which the complex analytic Taylor coefficients may not be defined.
Lec 23: Fundamental theorems of calculus

History of the fundamental theorem of calculus: It some sense, the story of calculus begins with the fundamental theorem of calculus (FTC), also known generically as Leibniz’s formula. The simplest integral is the length of a line $L = \int_0^x dx$. If we label a point on a line as $x = 0$ and wish to measure the distance to any other point $x$, we form the line integral between the two points. If the line is straight, this integral is simply the Euclidean length given by the difference between the two ends (Eq. 1.48, p. 73).

If $F(x)$ describes a height above the line, then the area under $F(x)$ is

$$f(x) - f(0) = \int_{x=0}^x F(\chi) d\chi,$$

where $\chi$ is a dummy variable of integration. Thus the area under $F(x)$ ($F(x), x \in \mathbb{R}$) only depends on the difference of the area evaluated at the end points. This makes intuitive sense, by viewing $f(x)$ as the area of a graph of $F(x)$.

This property, of the area as an integral over an interval, only depending on the end points, has important consequences in physics in terms of conservation of energy, making generalizations important.

Fundamental theorems of real calculus:

If $F(x)$ is analytic (Eq. 1.32, p. 60), then

$$F(x) = \frac{d}{dx} f(x),$$

which is known as the Fundamental theorem of (real) calculus (FTC). Thus Eq. 1.75 may be viewed as an anti-derivative, or exact differential. This is easily shown by evaluating

$$\frac{d}{dx} f(x) = \lim_{\delta \to 0} \frac{f(x + \delta) - f(x)}{\delta} = F(x),$$

starting from Eq. 1.74. If $F(x)$ is not analytic then the limit may not exist, so this seems like a necessary condition. There are many important variations on this very basic theorem (see Sect. 1.4). For example, the limits could depend on time. Also when taking Fourier and Laplace transforms, the integrand depends on both time and frequency via an exponential “kernel” function $e^{\pm st}$.

Case of a complex path The Fundamental theorem of complex calculus (FTCC) states that for any complex analytic function $F(s) \in \mathbb{C}$ with $s = \sigma + \omega j \in \mathbb{C}$

$$f(s) - f(s_0) = \int_{s_0}^{s} F(\zeta) d\zeta.$$

(Greenberg, 1988, p. 1197). If we compare this to Eq. 1.75, they differ in that the path of the integral is complex, meaning that the integral is over $s \in \mathbb{C}$, rather than a real integral over $x \in \mathbb{R}$. The Fundamental theorems of complex calculus (FTCC) states that the integral only depends on the end points, since

$$F(s) = \frac{d}{ds} f(s).$$

Comparing Eq. 1.75 (FTC) with Eq. 1.77 (FTCC), it would appear that this can only be true if $F(s) \in \mathbb{C}$ is complex analytic, meaning it must have a Taylor series in powers of $s \in \mathbb{C}$. 
Complex analytic functions: The definition of a complex analytic function $F(s)$ of $s \in \mathbb{C}$ is that the function may be expanded in a Taylor series (Eq. 1.39, p. 64) about an expansion point $s_o \in \mathbb{C}$. This definition follows the same logic as the FTC. Thus we need a definition for the coefficients $c_n \in \mathbb{C}$, which most naturally follow from Taylor’s formula

$$c_n = \frac{1}{n!} \left. \frac{d^n}{ds^n} F(s) \right|_{s=s_o}.$$  \hspace{1cm} (1.78)

The requirement that $F(s)$ have a Taylor series naturally follows by taking derivatives with respect to $s$ at $s_o$. The problem is that both integration and differentiation of functions of complex Laplace frequency $s = \sigma + \omega j \in \mathbb{C}$ have not yet been defined.

Thus the question is: “What does it mean to take the derivative of a function $F(s) \in \mathbb{C}$, $s = \sigma + \omega j \in \mathbb{C}$, with respect to $s$, where $s$ defines a plane rather than a real line?” We learned how to form the derivative on the real line. Can the same derivative concept be extended to the complex plane?

The answer is affirmative. The question may be resolved by applying the rules of the real derivative when defining the derivative in the complex plane. However for the complex case, there is an issue, regarding direction. Given any analytic function $F(s)$, is the partial derivative with respect to $\sigma$ different from the partial derivative with respect to $\omega j$? For complex analytic functions, the FTCC states that the integral is independent of the path in the $s$ plane. Based on the chain rule, the derivative must also be independent of direction at $s_o$. This directly follows from the FTCC. If the integral of a function of a complex variable is to be independent of the path, the derivative of a function with respect to a complex variable must be independent of the direction. This follows from Taylor’s formula, Eq. 1.78 for the coefficients of the complex analytic formula. The FTCC defines the area as an integral over a real differential ($dx \in \mathbb{R}$), while the FTCC relates an integral over a complex function $F(s) \in \mathbb{C}$, along a complex interval (i.e., path) ($ds \in \mathbb{C}$). For the FTCC the area under the curve only depends on the end points of the anti-derivative $f(x)$. But what is the meaning of an “area” along a complex path? The Cauchy-Riemann conditions provide the answer.

### 1.4.2 Lec 24: Cauchy-Riemann conditions

For the integral of $Z(s) = R(\sigma, \omega) + X(\sigma, \omega)j$ to be independent of the path, the derivative of $Z(s)$ must be independent of the direction of the derivative. As we show next, this leads to a pair of equations known as the Cauchy-Riemann conditions. This is an important generalization of Eq. 1.1, p. 18 which goes from real integration ($x \in \mathbb{R}$) to complex integration ($s \in \mathbb{C}$), based on lengths, thus on area.

To define

$$\frac{d}{ds} Z(s) = \frac{d}{ds} [R(\sigma, \omega) + jX(\sigma, \omega)]$$

take partial derivatives of $Z(s)$ with respect to $\sigma$ and $j\omega$, and equate them:

$$\frac{\partial Z}{\partial \sigma} = \frac{\partial R}{\partial \sigma} + j \frac{\partial X}{\partial \sigma} \equiv \frac{\partial Z}{j\partial \omega} = \frac{\partial R}{j\partial \omega} + j \frac{\partial X}{\partial \omega}.$$  \hspace{1cm} (1.79)

This says that a horizontal derivative, with respect to $\sigma$, is equivalent to a vertical derivative, with respect to $j\omega$. Taking the real and imaginary parts gives the Cauchy Riemann conditions

$$\text{CR-1: } \frac{\partial R(\sigma, \omega)}{\partial \sigma} = j \frac{\partial X(\sigma, \omega)}{\partial \omega} \quad \text{and} \quad \text{CR-2: } \frac{\partial R(\sigma, \omega)}{\partial \omega} = -j \frac{\partial X(\sigma, \omega)}{\partial \sigma}.$$  \hspace{1cm} (1.79)
The $j$ cancels in CR-1, but introduces a $j^2 = -1$ in CR-2. These equations are known as the Cauchy-Riemann (CR) conditions. They may also be written in polar coordinates ($s = re^{j\theta}$) as

$$\frac{\partial R}{\partial r} = \frac{1}{r} \frac{\partial X}{\partial \theta} \quad \text{and} \quad \frac{\partial X}{\partial r} = -\frac{1}{r} \frac{\partial R}{\partial \theta}.$$  

If you are wondering what would happen if we took a derivative at 45 degrees, then we only need to multiply the function by $e^{j\pi/4}$. But that will not change the derivative. Thus we may take the derivative in any direction by multiplying by $e^{j\theta}$, and the CR conditions will not change.

The CR conditions are necessary conditions that the integral of $Z(s)$, and thus its derivative, be independent of the path, expressed in terms of conditions on the real and imaginary parts of $Z$. This is a very strong condition on $Z(s)$, which follows assuming that $Z(s)$ may be approximated by a Taylor series in $s$

$$Z(s) = Z_0 + Z_1 s + \frac{1}{2} Z_2 s^2 + \cdots,$$  

(1.80)

where $Z_n \in \mathbb{C}$ are complex constants given by the Taylor series formula (Eq. 1.78, p. 101).

**Complex analytic functions:** Every function that may be expressed as a Taylor series about a point $s$ is said to be complex analytic at that point. This series, which is single valued, is said to converge within a radius of convergence (RoC). This highly restrictive condition has significant physical consequences. As an important example, every impedance function $Z(s)$ obeys the CR conditions over large regions of the $s$ plane, including the entire right half plane (RHP) ($\sigma > 0$). This condition is summarized by the Brune condition $\Re\{Z(\sigma > 0)\} \geq 0$ (Section 1.4.3, Eq. 1.88, p. 105). When the CR condition is generalized to volume integrals, it is called Green’s theorem, used heavily in the solution of boundary value problems in engineering and physics. (Kusse and Westwig, 2010) The last chapter of this course shall depend heavily on these concepts.

We may merge these equations into a pair of second order equations by taking a second round of partials. Specifically, eliminating the real part $R(\sigma, \omega)$ of Eq. 1.79 gives

$$\frac{\partial^2 R(\sigma, \omega)}{\partial \sigma \partial \omega} = \frac{\partial^2 X(\sigma, \omega)}{\partial^2 \omega} = -\frac{\partial^2 X(\sigma, \omega)}{\partial \sigma},$$  

(1.81)

which may be compactly written as $\nabla^2 X(\sigma, \omega) = 0$. Eliminating the imaginary part gives

$$\frac{\partial^2 X(\sigma, \omega)}{\partial \omega \partial \sigma} = \frac{\partial^2 R(\sigma, \omega)}{\partial^2 \sigma} = -\frac{\partial^2 R(\sigma, \omega)}{\partial \omega},$$  

(1.82)

which may be written as $\nabla^2 R(\sigma, \omega) = 0$.

In summary, for a function $Z(s)$ to be complex analytic, the derivative $dZ/ds$ must be independent of direction (path), which requires that the real and imaginary parts of the function obey Laplace’s equation, i.e.,

$$\text{CR-3: } \nabla^2 R(\sigma, \omega) = 0 \quad \text{and} \quad \text{CR-4: } \nabla^2 X(\sigma, \omega) = 0.$$  

(1.83)

The CR equations are easier to work with because they are first order, but the physical intuition is best understood by noting fact 1) the derivative of a complex analytic function is independent of its direction, and fact 2) the real and imaginary parts of the function each obey Laplace’s equation. Such relationships are known as harmonic functions. 66

66 When the function is the ratio of two polynomials, as in the cases of the Brune impedance, they are also related to Möbius transformations, also known as bi-harmonic operators.
As we shall see in the next few lectures, complex analytic functions must be smooth since every analytic function may be differentiated an infinite number of times, within the RoC. The magnitude must attain its maximum and minimum on the boundary. For example, when you stretch a rubber sheet over a jagged frame, the height of the rubber sheet obeys Laplace’s equation. Nowhere can the height of the sheet rise above or below its value at the boundary.

Harmonic functions define Conservative fields, which means that energy (like a volume or area) is conserved. The work done in moving a mass from \( a \) to \( b \) in such a field is conserved. If you return the mass from \( b \) back to \( a \), the energy is retrieved, and zero net work has been done.

1.4.3 Lec 25: Complex Analytic functions and Brune Impedance

It is rarely stated that the variable that we are integrating over, either \( x \) (space) or \( t \) (time), is real (\( x,t \in \mathbb{R} \)), since that fact is implicit, due to the physical nature of the formulation of the integral. But this intuition must be refined once complex numbers are included with \( s \in \mathbb{C} \), where \( s = \sigma + j \omega \).

The fact that time and space are real variables is more than an assumption, rather it is a requirement, due to the order property. Real numbers have order. For example, if \( t = 0 \) is now (the present), then \( t < 0 \) is the past and \( t > 0 \) is the future. The order property of time and space allows one to order these along a real axis. If the axis were complex, as in frequency \( s \), the order property is invalid. It follows that if we require order, time and space must be real (\( t,x \in \mathbb{R} \)). Interestingly, it was shown by d’Alembert (1747) that time and space are related by the pure delay due to the wave speed \( c_o \). To obtain a solution to the governing wave equation, that d’Alembert first proposed for sound waves, \( x,t \in \mathbb{R}^3 \) may be combined as functions of

\[
\zeta_\pm = t \pm x/c_o,
\]

where \( c \) [m/s] is the phase velocity of the waves. The d’Alembert solution to the wave equation, describing waves on a string under tension, is

\[
u(x,t) = F(t - x/c_o) + G(t + x/c_o), \quad (1.84)
\]

which describes the transverse velocity (or displacement) of two independent waves \( F(\zeta_-), G(\zeta_+) \) on the string, which represent forward and backward traveling waves. For example, starting with a string at rest, if one displaces the left end, at \( x = 0 \), by a step function \( u(t) \), then that step displacement will propagate to the right as \( u(t - x/c_o) \), arriving at location \( x_o \) [m], at time \( x_o/c_o \) [s]. Before this time, the string will not move to the right of the wave-front, at \( x_o \) [m], and after \( t_o \) [s] it will have displacement 1. Since the wave equation obeys superposition (postulate P1, p. 95), it follows that the “plane-wave” eigen-functions of the wave equation for \( x,k \in \mathbb{R}^3 \) are given by

\[
\psi_\pm(x,t) = e^{st \pm jk \cdot x} \quad (1.85)
\]

where \( s = \sigma + j \omega \) and \( |k| = 2\pi/\lambda \) is the wave number, \( \lambda = |\lambda| \) is the wavelength, and \( |k| = 2\pi f / c_o \).

When propagation losses are considered, we must replace \( jk \) with a complex analytic wave number \( \kappa(s) = k_r(s) + jk(s) \), which is denoted the dispersion relation. Forms of loss include viscosity and radiation. Important examples include acoustic plane waves, electromagnetic wave propagation, antenna theory, and the more complex case of 3D electron wave propagating in crystals (i.e., silicon), where electrons and EM waves are in a state of equilibrium. Perhaps electrons
and photons are simply different EM states, where \( \kappa(x, s) \) describes the crystal’s dispersion relations as functions of both frequency and directions, famously known as Brillouin zones. Dispersion is a property of the medium such that the wave velocity is a function of frequency and possibly direction, as in silicon.\(^{67}\) Several highly informative discussions on the history of this topic may be found in Brillouin (1953).

**Complex impedance functions**

Conservation of energy (or power) is a corner stone of modern physics. It may have first been under consideration by Galileo Galilei (1564-1642). Today the question is not if it is true, the questions is, why. Specifically, what is the physics behind conservation of energy? Surprisingly, the answer is straightforward, based on its definition, and the properties of impedance. Recall that the power is the product of the force and flow, and impedance is their ratio.

The power is given by the product of two variables, sometimes called *conjugate* variables, the force and the flow. In electrical terms, these are voltage (force) \((v(t) \leftrightarrow V(\omega))\) and current (flow) \((i(t) \leftrightarrow I(\omega))\), thus the electrical power at any instant of time is

\[
P(t) = v(t)i(t). \tag{1.86}
\]

The total energy \( w(t) \) is the integral of the power, since \( P(t) = dw/dt \). Thus if we start with all the elements at rest (no currents and no voltages, then the energy as a function of time is always positive

\[
w(t) = \int_0^t P(t)dt \geq 0, \tag{1.87}
\]

and is simply the total energy applied to the network (Van Valkenburg, 1964a, p. 376). Since the voltage and current are related by either an impedance or an admittance, conservation of energy depends on the property of impedance. From Ohm’s law and P1 (every impedance is causal)

\[
v(t) = z(t) \ast i(t) = \int_{\tau=0}^t z(\tau)i(t - \tau)d\tau \leftrightarrow Z(s)I(s)
\]

From the expression for the energy, in terms of Ohm’s law

\[
w(t) = u(t) \ast (v(t)i(t)) = u(t) \ast (i(t) \ast z(t) \ast i(t))
\]

\[
= \int_{t=0}^t i(t)\left(\int_{\tau=0}^t z(\tau)i(t - \tau)d\tau\right)dt
\]

\[
= \int_{\tau=0}^t z(\tau)\left(\int_{t=0}^t i(t)i(t - \tau)dt\right)d\tau
\]

\[
\leq \int_{\tau=0}^t z(\tau)|w|^2(\tau)d\tau
\]

\[
\leftrightarrow \frac{1}{s}Z(s)|I(\omega)|^2 \geq 0.
\]

The step from time to frequency follows from the fact that

\[
|w|^2(\tau) = i(t) \ast i(t) = \int_{\tau=0}^t i(\tau)i(t - \tau) \rightarrow |I(\tau, \omega)|^2 > 0
\]

\(^{67}\)In case you missed it, what I’m suggesting is that photons and electrons are the same thing. The difference is the medium, which determines the dispersion relation.
always has a positive Fourier transform for every possible \( i(t) \)

**Example:** Let \( i(t) = \delta(t) \). Then \( |w|^2(\tau) = i(t) \ast i(t) = \delta(\tau) \). Thus

\[
w(t) = \int_{\tau=0}^{t} z(\tau)|w|^2(\tau)d\tau = \int_{\tau=0}^{t} z(\tau)\delta(\tau)d\tau = \int_{0}^{t} z(\tau)d\tau.
\]

The Brune impedance always has the form \( \hat{z}(t) = r_o\delta(t) + \zeta(t) \). The surge impedance is defined as

\[
r_o = \int_{0}^{\infty} z(t)dt.
\]

The integral of the reactive part (\( \zeta(t) \)) is always zero, since the reactive part cannot not store energy.

Perhaps easier to visualize is when working in the frequency domain where the total energy, equal to the integral of the real part of the power, is

\[
\frac{1}{s} \Re{VI} = \frac{1}{2s}(V^*I + VI^*) = \frac{1}{2s}(Z^*I^*I + ZII^*) = \frac{1}{s} \Re{Z(s)}|I|^2 \geq 0.
\]

Formally this is related to a positive definite operator where the positive resistance forces the definiteness, which is sandwiched between the current.

In conclusion conservation of energy is totally dependent on the properties of the impedance. Thus of the most important and obvious applications of complex functions of a complex variable is the impedance function. This seems to be the ultimate example of the FTC, applied to \( z(t) \), in the name of conservation of energy.

**Poles and zeros of PR functions must be first order:** We conjecture that this proof also requires that the poles and the zeros of the impedance function be simple (only first order). Second order poles would have a reactive “secular” responses of the form \( h(t) = t\sin(\omega_k t + \phi)u(t) \), and these terms would not average to zero, depending on the phase, as is required of an impedance. As a result, only single order poles would be possible. Furthermore, when the impedance is the ratio of two polynomials, where the lower order polynomial is the derivative of the higher order one, then the poles and zeros must alternate. This is a well known property of the Brune impedance, that has never been adequately explained except for very special cases (Van Valkenburg, 1964b, p. 104-107). I suspect that no one has ever reported an impedance having second order poles and zeros as that would be rare impedance. Network analysis books never report 2d order poles and zeros in their impedance functions. Nor has there ever been any guidance as to where the poles and zeros might lie in the LHP. This would resolve this long-standing mystery.

**Every impedance must obey conservation of energy (P3):** According to Postulate P3 Section 1.3.13 (p. 95), a system is passive if it does not contain a power source. Drawing power from an impedance violates conservation of energy. This propriety is also called positive real, which is defined as (Brune, 1931a,b)

\[
\Re{Z(s \geq 0)} \geq 0,
\]

namely the real part of every impedance must be non-negative for \( \sigma \geq 0 \). When this condition holds, one cannot draw more power than stored in the impedance. A second condition requires that the impedance has simple poles. If there were a pole in the region \( \sigma > 0 \), then the first condition

\[68\]Secular terms result from second order poles since \( u(t) \ast u(t) = tu(t) \).
would not be met. Therefore there can only be simple poles (degree of 1) in the region $\sigma \leq 0$. The region $\sigma \leq 0$ is called the left half s plane (LHP), and the complementary region, $\sigma > 0$, is called the right half s plane (RHP). The condition on the simple poles is sufficient, but not necessary, as $Z(s) = 1/\sqrt{s}$ is a physical impedance, but is not a first order pole. The impedance function $Z(s) = R(\sigma, \omega) + jX(\sigma, \omega)$ has resistance $R$ and reactance $X$ as a function of complex frequency $s = \sigma + j\omega$. The function $z(t) \leftrightarrow Z(s)$ are defined by a Laplace transform pair. From the causality postulate (P1) of Sections 1.3.13 and 3.5.1 (p. 205), $z(t < 0) = 0$.

As an example, a resistor $R_o$ in series with an capacitor $C_o$ has an impedance given by (Table 1.5, p. 93)

$$Z(s) = R_o + 1/sC_o \leftrightarrow R_o \delta(t) + \frac{1}{C_o} u(t) = z(t), \quad (1.89)$$

with constants $R_o, C_o \in \mathbb{R} > 0$. In mechanics an impedance composed of a dash-pot (damper) and a spring have the same form. A resonant system has an inductor, resistor and a capacitor, with an impedance given by (Table 1.6, p. 95)

$$Z(s) = \frac{sC_o}{1 + sC_o R_o + s^2 C_o M_o} \leftrightarrow C_o \frac{d}{dt} \left( e^{s+t} + e^{-s-t} \right) = z(t) \quad (1.90)$$

is a second degree polynomial with two complex resonant frequencies $s_{\pm}$. When $R_o > 0$ these roots are in the left half s plane, with $z(t) \leftrightarrow Z(s)$.

Systems (networks) containing many elements, and transmission lines can be much more complicated, yet still have a simple frequency domain representation. This is the key to understanding how these physical systems work, as will be described below.

**Complex analytic functions:** To solve a differential equation, or integrate a function, Newton used the Taylor series to integrate one term at a time. However he only used real functions of a real variable, due to the fundamental lack of appreciation of the complex analytic series. This same method is how one finds solutions to scalar differential equations today, but using an approach that makes the solution method less obvious. Rather than working directly with the Taylor series, today we use the complex exponential, since the complex exponential is an eigen-function of the derivative

$$\frac{d}{dt} e^{st} = se^{st}.$$ 

Since $e^{st}$ may be expressed as a Taylor series, having coefficients $c_n = 1/n!$, in some real sense the modern approach is a compact way of doing what Newton did. Thus every linear constant coefficient differential equation in time may be simply transformed into a polynomial in complex Laplace frequency $s$, by looking for solutions of the form $A(s)e^{st}$, transforming the differential equation into a polynomial $A(s)$ in complex frequency. For example

$$\frac{d}{dt} f(t) + af(t) \leftrightarrow (s + a)F(s).$$

The root of $A(s_r) = s_r + a = 0$ is the eigen-value of the differential equation. A powerful tool for understanding the solutions of differential equations, both scalar and vector, is to work in the Laplace frequency domain. The Taylor series has been replaced by $e^{st}$, transforming Newton’s real Taylor series into the complex exponential eigen-function. In some sense, these are the same method, since

$$e^{st} = \sum_{n=0}^{\infty} \frac{(st)^n}{n!}. \quad (1.91)$$
Taking the derivative with respect to time gives

\[ \frac{d}{dt} e^{st} = se^{st} = s \sum_{n=0}^{\infty} \frac{(st)^n}{n!}, \]  

(1.92)

which is also complex analytic. Thus if the series for \( F(s) \) is valid (i.e., it converges), then its derivative is also valid. This was a very powerful concept, exploited by Newton for real functions of a real variable, and later by Cauchy and Riemann for complex functions of a complex variable. The key here is “Where does the series fail to converge?” in which case, the entire representation fails. This is the main message behind the FTCC (Eq. 1.77).

The FTCC (Eq. 1.77) is formally the same as the FTC (Eq. 1.75) (Leibniz formula), the key (and significant) difference being that the argument of the integrand \( s \in \mathbb{C} \). Thus this integration is a line integral in the complex plane. One would naturally assume that the value of the integral depends on the path of integration.

But, according to FTCC, it does not. In fact they are clearly distinguishable from the FTC. And the reasoning is the same. If \( F(s) = df(s)/ds \) is complex analytic (i.e., has a power series \( f(s) = \sum c_k s^k \), with \( f(s), c_k, s \in \mathbb{C} \), then it may be integrated, and the integral does not depend on the path. At first brush, this is sort of amazing. The key is that \( F(s) \) and \( f(s) \) must be complex analytic, which means they are differentiable. This all follows from the Taylor series formula Eq. 1.78 (p. 101) for the coefficients of the complex analytic series. For Eq. 1.77 to hold, the derivatives must be independent of the direction, as discussed in Section 1.4.2. The concept of a complex analytic function therefore has eminent consequences, in the form of several key theorems on complex integration discovered by Cauchy (c1820).

The use of the complex Taylor series formula in the complex plane, generalizes the functions they describe, with unpredictable consequences, as nicely shown by the domain coloring diagrams presented in Section 1.3.10 (p. 85). The tool of complex integration was first exploited in physics by Sommerfeld (1952), to explain the onset transients in waves, as explained in detail in Brillouin (1960, Chap. 3). Up to 1910, when Sommerfeld first published his results using complex analytic signals and saddle point integration in the complex plane, there was a poor understanding of the implications of the causal wave-front. It would be reasonable to say that his insights changed our understanding of wave propagation, for both light and sound. Sadly this insight has not been fully appreciated, even to this day. If you question this summary, please read Brillouin (1960, Chap. 1).

The full power of the complex analytic function was first appreciated by Bernard Riemann (1826-1866), in his University of Göttingen PhD Thesis of 1851, under the tutelage of Carl Friedrich Gauss (1777-1855), and drawing heavily on the work of Cauchy.

The key definition of a complex analytic function is that it has a Taylor series representation over a region of the complex frequency plane \( s = \sigma + j\omega \), that converges in a region of convergence (RoC) about the expansion point, with a radius determined by the nearest pole of the function. A further surprising feature of all analytic functions is that within the RoC, the inverse of that function also has a complex analytic expansion. Thus given \( w(s) \), one may also determine \( s(w) \) to any desired accuracy, critically depending on the RoC.

### 1.4.4 Lec 26: Multi-valued functions

In the field of mathematics there seems to have been a tug-of-war regarding the basic definition of the concept of function. The accepted definition today seems to be a single-valued mapping from the domain to the codomain (or range). This makes the discussion of multi-valued functions
somewhat tedious. In 1851 Riemann (working with Gauss) seems to have resolved this problem for the natural set of multi-valued functions, by introducing the concept of the *branch-cut* and *sheets*.

Two simple examples multi-valued functions are the circle \( z^2 = x^2 + y^2 \) and \( w = \log(z) \). For example, assuming \( z \) is the radius of the circle, when solving for \( y(x) \) gives the multi-valued function
\[
y(x) = \pm \sqrt{z^2 - x^2}.
\]
If we accept the modern definition of a function, then \( y(x) \) is not a function, nor even two functions. For example what if \( x > z \)? Or worse, what if \( z = 2j \) with \( |x| < 1 \). Riemann’s solution resolves all these difficulties (as best I know).

To proceed, we need definitions and classifications of the various types of complex singularities:

1. Poles of order 1 are called *simple poles*. Their amplitude called the *residue* (e.g. \( \alpha/s \) has residue \( \alpha \)). Simple poles are special (Eq. 1.95, p. 112)\(^{69}\) and play a key role in mathematical physics. Consider the function \( y(s) = \sqrt[2]{1-s^\alpha} \) with \( \alpha \in \mathbb{Z}, \mathbb{F}, \mathbb{R} \) and \( \mathbb{C} \).
2. When the numerator and denominator of a rational function have a common root (i.e., factor), that root is said to be *removable*.
3. A singularity that is not 1) removable, a 2) pole or 3) branch point, is called *essential*.
4. When the first derivative of a function \( Z(s) \) has a simple pole at \( s_o \), then \( s_o \) is said to be a *branch point* of \( Z(s) \) (e.g., \( d\ln(s^\alpha)/ds = \alpha/s \)). However the converse does not necessarily hold.
5. A complex function which is analytic, except for isolated poles, is called *metamorphic*.\(^{70}\) Metamorphic functions can have any number of poles, even an infinite number.

More complex typologies are being researched today, and progress that is expected to accelerate due to modern computing technology.\(^{71}\) It is helpful to identify the physical meaning of these more complex surfaces, to guide us in their interpretation and possible applications.\(^{72}\)

**Branch cuts:** Up to this point we have only considered poles of order \( k \), of the form \( 1/s^k \), with \( k \in \mathbb{Z} \). The concept of a *branch cut* allows one to manipulate (and visualize) multi-valued functions, for which \( k \in \mathbb{F} \). This is done by breaking each region into a single valued *sheets*. The concepts of the branch cut, the sheets, and the *extended plane*, were first devised by Riemann, working with Gauss (1777-1855), and first described in his thesis of 1851. Of course it was these three mathematical and geometrical constructions that provide the deep insight to complex analytic functions, supplementing the important earlier work of Cauchy (1789-1857) on the calculus of complex analytic functions. For an intuitive discussion of Riemann sheets and branch cuts, see Boas (1987, Section 29, pp. 221-225).

**Square root function:** The branch cut is a line that separates the various single valued parts of a multi-valued function. For example, in Fig. 1.22 we see the single valued function \( w(z) = s^2 \) (left), and on the right, its inverse, the double-valued mapping of \( s(w) = \pm \sqrt{w} \).

\(^{69}\)https://en.wikipedia.org/wiki/Pole_(complex_analysis)

\(^{70}\)https://en.wikipedia.org/wiki/Meromorphic_function

\(^{71}\)https://www.maths.ox.ac.uk/about-us/departmental-art/theory

\(^{72}\)https://www.quantamagazine.org/secret-link-uncovered-between-pure-math-and-physics-20171201
1.4. STREAM 3A: SCALAR CALCULUS (11 LECTURES)

Here we see the mapping for the square root function $w(z) = \pm \sqrt{z}$ which has two single-valued sheets, corresponding to the two signs of the square root. The lower sheet is $+\sqrt{z}$, and the upper sheet is $-\sqrt{z}$. The location of the branch cut may be moved by rotating the $z$ coordinate system. For example, $w(z) = \pm j\sqrt{z}$ and $w(z) = \pm \sqrt{-z}$ have a different branch cuts, as may be easily verified using the Matlab/Octave commands `j*zviz(z)` and `zviz(-z)`. A function is analytic on the branch cut, since the cut may be moved. If a Taylor series is formed on the branch cut, it will describe the function on the two different sheets. Thus the complex analytic series (i.e., the Taylor formula, Eq. 1.78) does not depend on the location of a branch cut, as it only describes the function uniquely (as a single valued function), valid in its local region of convergence. This figure has been taken from Stillwell (2010, p. 303). A more comprehensive view presented in the text. The branch cut lies in the domain ($x \in \mathbb{C}$), not in the codomain $w(x) \in \mathbb{C}$ (Kusse and Westwig, 2010). This becomes clear by studying how `zviz.m` works, and understanding its output. An example that shows this is Fig. 1.23, where the axes are $s$ and $\omega$, with the branch cut along the negative $\sigma$ axis ($\theta = \pi$).

Figure 1.22: Here we see the mapping for the square root function $w(z) = \pm \sqrt{z}$ which has two single-valued sheets, corresponding to the two signs of the square root. The lower sheet is $+\sqrt{z}$, and the upper sheet is $-\sqrt{z}$. The location of the branch cut may be moved by rotating the $z$ coordinate system. For example, $w(z) = \pm j\sqrt{z}$ and $w(z) = \pm \sqrt{-z}$ have a different branch cuts, as may be easily verified using the Matlab/Octave commands `j*zviz(z)` and `zviz(-z)`. A function is analytic on the branch cut, since the cut may be moved. If a Taylor series is formed on the branch cut, it will describe the function on the two different sheets. Thus the complex analytic series (i.e., the Taylor formula, Eq. 1.78) does not depend on the location of a branch cut, as it only describes the function uniquely (as a single valued function), valid in its local region of convergence. This figure has been taken from Stillwell (2010, p. 303). A more comprehensive view presented in the text. The branch cut lies in the domain ($x \in \mathbb{C}$), not in the codomain $w(x) \in \mathbb{C}$ (Kusse and Westwig, 2010). This becomes clear by studying how `zviz.m` works, and understanding its output. An example that shows this is Fig. 1.23, where the axes are $s$ and $\omega$, with the branch cut along the negative $\sigma$ axis ($\theta = \pi$).

The multi-valued nature of the square root is best understood by working with the function in polar coordinates. Let

$$s_k = re^{\theta j}e^{2\pi k j},$$

where $k$ is the sheet-index, and

$$w = \rho e^{\psi j} = \sqrt{r}e^{\theta j/2}e^{\pi k j},$$

where $r, \rho \in \mathbb{R}$ are the magnitudes and $\theta, \psi \in \mathbb{R}$ are the angles. The domain-coloring program `zviz.m` assumes that the angles go from $-\pi < \theta < \pi$, with $\theta = 0$ being a light red and $\pm \pi$ a blue color. This angle to color map is shown in the left panel of Fig. 1.23. The first Riemann sheet ($k = 0$) is define for $-\pi < \theta < \pi$. The second sheet ($k = 1$) picks up at $\theta = \pi$ and continues on to $\pi + 2\pi = 3\pi$. The first sheet maps the angle of $w$ (i.e., $\phi = \angle w = \theta/2$) from $-\pi/2 < \phi < \pi/2$ ($w = \sqrt{r}e^{\theta j/2}$). This corresponds to $u = \Re\{w(s)\} > 0$. The second sheet maps $\pi/2 < \psi < 3\pi/2$ (i.e., 90° to 270°), which is $\Re\{w\} = u < 0$. In summary, twice around the $s$ plane is once around the $w(s)$ plane, because the angle is half due to the $\sqrt{s}$. This then describes the multi-valued nature of the square root function.

This concept of analytic inverses becomes important only when the function is multi-valued. For example, if $w(s) = s^2$ then $s(w) = \pm \sqrt{w}$ is multivalued. Riemann dealt with such extensions with the concept of a branch-cut with multiple sheets, labeled by a sheet number. Each sheet
Figure 1.23: Colorized plots of two sheets of \( w(s) = \pm \sqrt{s} \) with the branch cut at \( \theta = \pi \). **Left:** Color map reference plane \((z = s)\). **Center:** \( e^{\pi j} \sqrt{s} = -\sqrt{s} \). This sheet was generated by rotating \( w \) by \(180^\circ\). This should be compared to the right panel of Fig. 1.24 which shows \(-\sqrt{-s}\), and to the upper sheet of Fig. 1.22. Note how the panel on the right of Fig. 1.24 matches the right half of \( s \) (purple = -90\(^\circ\), yellow/green = +90\(^\circ\)) while the middle panel above comes from the left side of \( s \) (green to purple). The center sheet is green at -180\(^\circ\), and purple at +180\(^\circ\), which matches the right sheet at \( \pm 180^\circ \) respectively. (i.e., \( e^{\pi j} \sqrt{s} \)). **Right:** the lower sheet of Fig. 1.22.

describes an analytic function (Taylor series), that converges within some RoC, having a radius out to the nearest pole. This Riemann’s branch cut and sheets explicitly deal with the need to define unique single valued inverses of multi-valued functions. Since the square root function has two overlapping regions, corresponding to the \pm due to the radical, there must be two connected regions, sort of like mathematical Siamese-twins, distinct, yet the same.

Branch cuts emanate from **branch points**, singularities that can have fractional order, for example \(1/\sqrt{s}\), and terminate at either another singularity having the same fractional order, which may be at \( \infty \). For example, suppose that in the neighborhood of the pole, at \( s_o \) the function is

\[
 f(s) = \frac{w(s)}{(s - s_o)^k},
\]

where \( w, s, K, k \in \mathbb{C} \). When \( k = 1 \), \( s_o = \sigma_o + \omega_o j \) is the pole location in the \( s \) plane and \( w(s_o) \) is the residue. When \( k \) is a complex (or real) constant, it defines the **order** of the pole.\(^73\)

When \( k \in \mathbb{F} \) there must be a branch cut, of order \( k \). For example, if \( k = 1/2 \), the singularity (branch cut) is of order 1/2, and there are two Riemann sheets, as shown in Fig. 1.22. An important example is the Bessel function

\[
 \delta(t) + \frac{1}{t} J_1(t) u(t) \leftrightarrow \sqrt{s^2 + 1},
\]

which is related to the solution to the wave equation in two-dimensional cylindrical coordinates (Table 1.5, p. 93). Bessel functions are the solutions to guided acoustic waves in round pipes, or surface waves on the earth (seismic waves), or waves on the surface of a pond.

It is important to understand that the function is analytic on the branch cut, but not at the branch point. One is free to move the branch cut (at will). It does not need to be on a line, it could be cut in almost any connected manner, such as a spiral. The only rule is that it must start and stop at the matching branch points, or at \( \infty \), which must have the same order.

There are a limited number of possibilities for the order, \( k \in \mathbb{Z} \) or \( \in \mathbb{F} \). If the order is drawn from \( \mathbb{R} \notin \mathbb{F} \), the pole cannot not have a residue. According to the definition of the residue, \( k \in \mathbb{F} \) will not give a residue. But there remains open the possibility of generalizing the concept of the Riemann Integral Theorem, to include \( k \in \mathbb{F} \). This seems unlikely however.

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\(^73\)We shall refer to the **order** of a pole and the **degree** of a differential equation.
1.4. STREAM 3A: SCALAR CALCULUS (11 LECTURES)

If the singularity had an irrational order ($k \in \mathbb{I}$), the branch cut has the same “irrational order.” Accordingly there would be an infinite number of Riemann sheets, as in the case of the log function. An example is $k = \pi$, for which

$$F(s) = \frac{1}{s\pi} = e^{-\log(s\pi)} = e^{-\pi \log(s)} = e^{-\pi \log(\rho)} e^{-\pi \theta j},$$

where the domain is expressed in polar coordinates $s = \rho e^{\theta j}$. When $k \in \mathbb{I}$ is may be arbitrarily close to 1 (e.g., two very close primes, such as 881 and 883, primes 152 and 153),\textsuperscript{74} the branch cut could be very subtle (it could even be unnoticed), but it would have a significant impact on the nature of the function, and of course, on the inverse Laplace transform.

**Multivalued functions:** The two basic functions we review, to answer the questions about multi-valued functions and branch cuts, are $w(s) = s^2$ and $w(s) = e^s$, along with their inverse functions $w(s) = \sqrt{s}$ and $w(s) = \log(s)$. For uniformity we shall refer to the complex abscissa ($s = \sigma + \omega j$) and the complex ordinate $w(s) = u + v j$. When the complex abscissa and domain are swapped, by taking the inverse of a function, multi-valued functions are a common consequence. For example, $f(t) = \sin(t)$ is single valued, and analytic in $t$, thus has a Taylor series. The inverse function $t(f)$ is not so fortunate as it is multivalued.

The modern terminology is the *domain* and *range*, or alternatively the *co-domain*.\textsuperscript{75}

**Log function:** Next we discuss the multi-valued nature of the log function. In this case there are an infinite number of Riemann sheets, not well captured by Fig. 1.19 (p. 86), which only displays the principal sheet. However if we look at the formula for the log function, the nature is easily discerned. The abscissa $s$ may be defined as multi-valued since

$$s_k = \tau e^{2\pi k j} e^{\theta_j}.$$

\textsuperscript{74}since there are no even primes, the minimum difference is 2. Out of $10^6$ primes 5 have a spacing of 80, and the distribution is linear on a log scale.)

\textsuperscript{75}The best way to create confusion is the rename things. The confusion grows geometrically with each renaming. I suspect that everyone who cares knows the terms abscissa and ordinate, and some fraction know the equivalent terms *domain* and *range.*
Here we have extended the angle of $s$ by $2\pi k$, where $k$ is the sheet index $\in \mathbb{Z}$. Taking the log

$$\log(s) = \log(r) + (\theta + 2\pi k).$$

When $k = 0$ we have the principal value sheet, which is zero when $s = 1$. For any other value of $k$ $w(s) \neq 0$, even when $r = 1$, since the angle is not zero, except for the $k = 0$ sheet.

1.4.5 **Lec 27: Three Cauchy Integral Theorems**

Cauchy’s theorems for integration in the complex plane

There are three basic definitions related to Cauchy’s integral formula. They are closely related, and can greatly simplify integration in the complex plane.

1. **Cauchy’s (Integral) Theorem:**

$$\oint_C F(s)ds = 0, \quad (1.93)$$

if and only if $F(s)$ is complex-analytic inside of a simple closed curve $C$ (Boas, 1987, p. 45), (Stillwell, 2010, p. 319). The FTCC (Eq. 1.77) says that the integral only depends on the end points if $F(s)$ is complex-analytic. By closing the path (contour $C$) the end points are the same, thus the integral must be zero, as long as $F(s)$ is complex analytic.

2. **Cauchy’s Integral Formula:**

$$\frac{1}{2\pi j} \oint_B \frac{F(s)}{s - s_o} ds = \begin{cases} F(s_o), s_o \in C \text{ (inside)} \\ 0, s_o \notin C \text{ (outside)}. \end{cases} \quad (1.94)$$

Here $F(s)$ is required to be analytic everywhere within (and on) the contour $C$ (Greenberg, 1988, p. 1200), (Boas, 1987, p. 51), (Stillwell, 2010, p. 220). The value $F(s_o) \in \mathbb{C}$ is the residue of the pole $s_o$ of $F(s)/(s - s_o)$.

3. **The (Cauchy) Residue Theorem** (Greenberg, 1988, p. 1241), (Boas, 1987, p. 73)

$$\oint_C F(s)ds = 2\pi j \sum_{k=1}^{K} c_k = 2\pi j \sum_{k=1}^{K} \oint \frac{F(s)}{s - s_k} ds, \quad (1.95)$$

where the residues $c_k \in \mathbb{C}$, corresponding to the $k$th poles of $f(s)$, enclosed by the contour $C$. By the use of Cauchy’s integral formula, the last form of the residue theorem is equivalent to the residue form.\(^\text{76}\)

**How to calculate the residue:** The case of first order poles, while special, is important, since the Brune impedance only allows simple poles and zeros, increasing the utility of this special case. The residues for simple poles are $F(s_k)$, which is complex analytic in the neighborhood of the pole, but not at the pole.

Consider the function $f(s) = F(s)/(s - s_k)$, where we have factored $f(s)$ to isolate the first-order pole at $s = s_k$, with $F(s)$ analytic at $s_k$. Then the residue of the poles at $c_k = F(s_k)$. This\(^\text{76}\)This theorem is the same as a 2D version of Stokes thm (citations).
coefficient is computed by removing the singularity, by placing a zero at the pole frequency, and
taking the limit as \( s \to s_k \), namely

\[
c_k = \lim_{s \to s_k} [(s - s_k)F(s)]
\]  

(1.96)


When the pole is an \( N^{th} \) order, the procedure is much more complicated, and requires taking
\( N - 1 \) derivatives of \( f(s) \), followed by the limit process (Greenberg, 1988, p. 1242). Higher order
poles are rarely encountered, thus it is good to know that this formula exists, but perhaps it is not
worth the effort to memorize it.

**Summary and examples:** These three theorems, all attributed to Cauchy, collectively are related
to the fundamental theorems of calculus. Because the names of the three theorems are so similar,
they are easily confused.

1. In general it makes no sense to integrate through a pole, thus the poles (or other singularities)
must not lie on \( C \).

2. The Cauchy integral theorem (Eq. 1.93), follows trivially from the fundamental theorem of
complex calculus (Eq. 1.77, p. 100), since if the integral is independent of the path, and the
path returns to the starting point, the closed integral must be zero. Thus Eq. 1.93 holds when
\( F(s) \) is complex analytic within \( C \).

3. Since the real and imaginary parts of every complex analytic function obey Laplace’s equa-
tion (Eq. 1.83, p. 102), it follows that every closed integral over a Laplace field, i.e., one
defined by Laplace’s equation, must be zero. **In fact this is the property of a conservative system**, corresponding to many physical systems. If a closed box has fixed potentials on the
walls, with any distribution what so ever, and a point charge (i.e, an electron) is placed in
the box, then a force equal to \( F = qE \) is required to move that charge, thus work is done. However if the point is returned to its starting location, the net work done is zero.

4. Work is done in charging a capacitor, and energy is stored. However when the capacitor is
discharged, all of the energy is returned to the load.

5. Soap bubbles and rubber sheets on a wire frame, obey Laplace’s equation.

6. These are all cases where the fields are Laplacian, thus closed line integrals must be zero. Laplacian fields are commonly observed, because they are so basic.

7. We have presented the impedance as the primary example of a complex analytic function. Physically, every impedance has an associated stored energy, and every system having stored energy has an associated impedance. This impedance is usually defined in the frequency \( s \) domain, as a force over a flow (i.e., voltage over current). The power \( P(t) \) is defined as the
force times the flow and the energy \( E(t) \) as the time integral of the power

\[
E(t) = \int_{-\infty}^{t} P(t) dt,
\]

(1.97)

which is similar to Eq. 1.74 (p. 100) [see Section 3.2.1, Eq. 3.10 (p. 187)]. In summary, impedance and power and energy are all fundamentally related.
1.4.6 Lec 28: Cauchy Integral Formula & Residue Theorem

The Cauchy integral formula (Eq. 1.94) is an important extension of the Cauchy integral theorem (Eq. 1.93) in that a pole has been explicitly injected into the integrand at \( s = s_0 \). If the pole location is outside of the curve \( C \), the result of the integral is zero, in keeping with Eq. 1.93. However, when the pole is inside of \( C \), the integrand is no longer complex analytic, and a new result follows. By a manipulation of the contour in Eq. 1.94, the pole can be isolated with a circle around the pole, and then taking the limit, the radius goes to zero.

For the related Cauchy residue theorem (Eq. 1.95) the same result holds, except it is assumed that there are \( K \) simple poles in the function \( F(s) \). This requires the repeated application of Eq. 1.94, \( K \) times, so it represents a minor extension of Eq. 1.94. The function \( F(s) \) may be written as \( f(s)/P_K(s) \), where \( f(s) \) is analytic in \( C \) and \( P_K(s) \) is a polynomial of degree \( K \), with all of its roots \( s_k \in \mathbb{C} \).

Non-integral degree singularities: The key point is that this theorem applies when \( n \in \mathbb{I} \), including fractionals \( n \in \mathbb{F} \), but for these cases the residue is always zero, since by definition the residue is the amplitude of the \( 1/s \) term (Boas, 1987, p. 73).

Examples:

1. The function \( 1/\sqrt{s} \) has a zero residue (Hint: apply the definition of the residue Eq. 1.96).
2. When \( n \in \mathbb{F} \), the residue is, by definition, zero.
3. When \( n \in \mathbb{I} \), the residue is, by definition, zero.
4. When \( n = 1 \), the residue is non-zero.

This point is equally important when defining the inverse Laplace transform. When integrating over \( \omega \in \mathbb{R} \), the value passes through all possible exponents, including all rational and irrational numbers. The only value of \( \omega \) that has a residue, are those at the poles of the integrand.

1.4.7 Lec 29: Inverse Laplace transform & Cauchy residue theorem

The inverse Laplace transform Eq. 1.71 transforms a function of complex frequency \( F(s) \) and returns a causal function of time \( f(t) \)

\[
f(t) \leftrightarrow F(s),
\]

where \( f(t) = 0 \) for \( t < 0 \). Examples are provided in Table 1.5 (p. 93). We next discuss the details of finding the inverse transform by use of the Cauchy residue theorem, and how the causal requirement \( f(t < 0) = 0 \) comes about.

The integrand of the inverse transform is \( F(s)e^{st} \) and the limits of integration are \( -\sigma_0 \pm \omega j \). To find the inverse we must close the curve, at infinity, and show that the integral at \( \omega j \to \infty \). There are two ways to close these limits, to the right \( \sigma > 0 \) (RHP), and to the left \( \sigma < 0 \) (LHP), but there needs to be some logical reason for this choice. That logic is the sign of \( t \). For the integral to converge the term \( e^{st} \) must go to zero as \( \omega \to \infty \). In terms of the real and imaginary parts of \( s = \sigma + \omega j \), the exponential may be rewritten as \( e^{\sigma t}e^{\omega j} \). Note that both \( t \) and \( \omega \) go to \( \infty \). So it is the interaction between these two limits that determines how we pick the closure, RHP vs. LHP.
1.4. STREAM 3A: SCALAR CALCULUS (11 LECTURES)

Case for causality \((t < 0)\): Let us first consider negative time, including \(t \to -\infty\). If we were to close \(C\) in the left half plane \((\sigma < 0)\), then the product \(\sigma t\) is positive \((\sigma < 0, t < 0, \text{ thus } \sigma t > 0)\). In this case as \(\omega \to \infty\), the closure integral \(|s| \to \infty\) will diverge. Thus we may not close in the LHP for negative time. If we close in the RHP \(\sigma > 0\) then the product \(\sigma t < 0\) and \(e^{st}\) will go to zero as \(\omega \to \infty\). This then justifies closing the contour, allowing for the use the Cauchy theorems.

If \(F(s)\) is analytic in the RHP, the FTCC applies, and the resulting \(f(t)\) must be zero, and the inverse Laplace transform must be causal. This argument holds for any \(F(s)\) that is analytic in the RHP \((\sigma > 0)\).

Case of unstable poles: An important but subtle point arises: If \(F(s)\) has a pole in the RHP, then the above argument still applies if we pick \(\sigma_o\) to be to the right of the RHP pole. This means that the inverse transform may still be applied to unstable poles (those in the RHP). This explains the need for the \(\sigma_o\) in the limits. If \(F(s)\) has no RHP poles, then \(\sigma_o = 0\) is adequate, and this factor may be ignored.

Case for zero time \((t = 0)\): When time is zero, the integral does not, in general, converge, leaving \(f(t)\) undefined. This is most clear in the case of the step function \(u(t) \leftrightarrow 1/s\), where the integral may not be closed, because the convergence factor \(e^{st} = 1\) is lost for \(t = 0\).

1.4.8 Lec 30: Inverse Laplace transform \((t > 0)\)

Case of \(t > 0\): Next we investigate the convergence of the integral for positive time \(t > 0\). In this case we must close the integral in the LHP \((\sigma < 0)\) for convergence, so that \(st < 0\) \((\sigma \leq 0 \text{ and } t > 0)\). When there are poles on the \(\omega_j = 0\) axis, \(\sigma_o > 0\) assures convergence by keeping the on-axis poles inside the contour. At this point the Cauchy residue theorem (Eq. 1.95) is relevant. If we restrict ourselves to simple poles (as required for a Brune impedance), the residue theorem may be directly applied.

The most simple example is the step function, for which \(F(s) = 1/s\) and thus

\[
\begin{align*}
u(t) &= \int_{\text{LHP}} \frac{e^{st}}{s} \, ds \leftrightarrow \frac{1}{s},
\end{align*}
\]

which is a direct application of the Cauchy Residue theorem, Eq. 1.95 (p. 112). The forward transform of \(u(t)\) is straightforward, as discussed in Section 1.3.12 (p. 91). This is true of most if not all of the elementary forward Laplace transforms. In these cases, causality is built into the integral by the limits, so is not a result, as it must be in the inverse transform. An interesting problem is proving that \(u(t)\) is not defined at \(t = 0\).

The inverse Laplace transform of \(F(s) = 1/(s + 1)\) has a residue of 1 at \(s = -1\), thus that is the only contribution to the integral. More demanding cases are Laplace transform pairs

\[
\begin{align*}
\frac{1}{\sqrt{t}} u(t) &\leftrightarrow \sqrt{\frac{\pi}{s}} \quad \text{and} \quad J_o(t)u(t) \leftrightarrow \frac{1}{s^2 + 1} = \frac{1}{(s + j)(s - j)}.
\end{align*}
\]

Many of these are easily proved in the forward direction, but are much more difficult in the inverse direction, due to the properties at \(t = 0\), unless of course, the residue theorem (Eq. 1.95, p. 112) is invoked. The last \(\mathcal{L}\mathcal{T}\)-pair helps us understand the basic nature of the Bessel functions \(J_0(z)\), and \(H_0^{(1)}(z^2)\), with a branch cut along the negative axis (see Fig. 3.3, p. 197).
CHAPTER 1. INTRODUCTION

Some open questions: Without the use of the CRT (Eq. 1.95) it is difficult to see how to evaluate the inverse Laplace transform of \(1/s\) directly. For example, how does one show that the above integral is zero for negative time (or that it is 1 for positive time)? The CRT neatly solves this difficult problem, by the convergence of the integral for negative and positive time. Clearly the continuity of the integral at \(\omega \to \infty\) plays an important role. Perhaps the Riemann sphere plays a role in this, that has not yet been explored.

1.4.9 Lec 31: Properties of the LT (e.g., Linearity, convolution, time-shift, modulation, etc.)

As shown in the table of Laplace transforms, there are integral (i.e., integration, not integer) relationships, or properties, that are helpful to identify. The first of these is a definition not a property:

\[
f(t) \leftrightarrow F(s).
\]

When taking the LT, the time response is given in lower case (e.g., \(f(t)\)) and the frequency domain transform is denoted in upper case (e.g., \(F(s)\)). It is required, but not always explicitly specified, that \(f(t < 0) = 0\), that is, the time function must be causal (P1: Section 1.3.13).

Linearity: A key property so basic that it almost is forgotten, is the linearity property of the LT. These properties are summarized as P2 of Section 1.3.13, 96).

Convolution property: One of the most basic and useful properties is that the product of two LTs in frequency, results in convolution in time

\[
f(t) \ast g(t) = \int_0^t f(\tau)g(t - \tau)d\tau \leftrightarrow F(s)G(s),
\]

where we use the \(\ast\) operator as a shorthand for the convolution of two time functions.

\[
\text{Put this notional property in Appendix A.}
\]
A key application of convolution is filtering, which takes on many forms. The most basic filter is the moving average, the moving sum of data samples, normalized by the number of samples. Such a filter has very poor performance. It also introduces a delay of half the length of the average, which may, or may not constitute a problem, depending on the application. Another important example is a low-pass filter, that removes high frequency noise, or a notch filter that removes line-noise (i.e., 60 [Hz] in the US, and its 2d and 3d harmonics, 120 and 180 [Hz]). Such noise is typically a result of poor grounding and ground loops. It is better to solve the problem at its root, than to remove it with a notch filter. Still, filters are very important in engineering.

By taking the LT of the convolution we can derive this relationship

\[
\int_0^\infty [f(t) \ast g(t)] e^{-st} dt = \int_0^\infty \left[ \int_0^t f(\tau) g(t - \tau) d\tau \right] e^{-st} dt
\]

\[
= \int_0^t f(\tau) \left( \int_0^\infty g(t - \tau) e^{-st} dt \right) d\tau
\]

\[
= \int_0^t f(\tau) \left( e^{-s\tau} \int_0^\infty g(t') e^{-st'} dt' \right) d\tau
\]

\[
= G(s) \int_0^t f(\tau) e^{-s\tau} d\tau
\]

\[
= G(s) F(s)
\]

We first encountered this relationship in Section 1.3.4 (p. 70)) in the context of multiplying polynomials, which was the same as convolving their coefficients. Hopefully the parallel is obvious. In the case of polynomials, the convolution was discrete in the coefficients, and here it is continuous in time. But the relationships are the same.

**Time-shift propriety:** When a function is time-shifted by time \(T_o\), the LT is modified by \(e^{sT_o}\), leading to the propriety

\[
f(t - T_o) \leftrightarrow e^{-sT_o} F(s).
\]

This is easily shown by applying the definition of the LT to a delayed time function.

**Time derivative:** The key to the eigen-function analysis provided by the LT, is the transformation of a time derivative on a time function, that is

\[
\frac{d}{dt} f(t) \leftrightarrow sF(s).
\]

Here \(s\) is the eigen-value corresponding to the time derivative of \(e^{st}\). Given the definition of the derivative of \(e^{st}\) with respect to time, this definition seems trivial. Yet that definition was not obvious to Euler. It needed to be extended to the space of complex analytic function \(e^{st}\), which did not happen until at least Riemann (1851).

**Given a differential equation of order \(K\), the LT results in a polynomial in \(s\), of degree \(K\).**

It follows that this LT property is the corner-stone of why the LT is so important to scalar differential equations, as it was to the early analysis of Pell’s equation and the Fibonacci sequence, as presented in earlier chapters. This property was first uncovered by Euler. It is not clear if he fully appreciated its significance, but by the time of his death, it certainly would have been clear to him. Who first coined the term *eigen-value* and *eigen-function*? The word *eigen* is a German word meaning *of one*. It seem likely that this term became popular somewhere between the 19th and 20th century.
Initial and final value theorems: There are much more subtle relations between \( f(t) \) and \( F(s) \) that characterize \( f(0^+) \) and \( f(t \to \infty) \). While these properties can be very important in certain application, they are beyond the scope of the present treatment. These relate to so-called initial value theorems. If the system under investigation has potential energy at \( t = 0 \), then the voltage (velocity) need not be zero for negative time. An example is a charged capacitor or a moving mass. These are important situations, but better explored in a more in depth treatment.

1.4.10 Lec 32: ’Lecture:’ Review for Exam III (Evening Exam)
1.5. STREAM 3B: VECTOR CALCULUS (10 LECTURES)

1.5 Vector Calculus (Stream 3b)

1.5.1 Lec 33 Properties of Fields and potentials

Before we can define the vector operations $\nabla()$, $\nabla\cdot()$, $\nabla\times()$, $\nabla^2()$, we must define the objects they operate on, scalar\textsuperscript{78} and vector\textsuperscript{79} fields. The word field has two very different meanings, a mathematical definition, which defines an algebraic structure,\textsuperscript{81} and a physical one, discussed next. Differentiable functions, defined here as

Scalar fields: We use the term scalar field interchangeably with analytic in a connected region of the spatial vector $x = [x, y, z]^T \in \mathbb{R}^3$, denoted $\mathcal{V}$. In mathematics, functions that are piece-wise differentiable are called smooth, which is quite different from analytic. Since an analytic function may be written as a power series, it is both single valued and infinitely differentiable. A smooth function has at least one or more derivatives, but need not be single valued. For example the function $tu(t)$ is smooth, and has one derivative $d/dt (tu(t)) = u(t)$, but does not have a second derivative at $t = 0$. Thus $tu(t)$ is not analytic at $t = 0$. The function $\pm tu(t)$ is differentiable

$$\frac{d}{dt} [\pm tu(t)] = \begin{cases} +u(t) & \text{for } t > 0 \\ -u(t) & \text{for } t < 0 \end{cases}$$

but double-valued for $t > 0$, and not defined at $t = 0$, thus not analytic. More interesting examples of multivalued smooth functions are easily constructed. We shall abandon the word smooth in favor of the stronger concept of analytic.

Initially we simplify the field by limiting the definition to an analytic surface $S(x)$, as shown in Fig. 1.27, having height $z(x, y) \in \mathbb{R}$, as a function of $x, y \in \mathbb{R}^2$ (a plane)

$$z(x, y, t) = \phi(x, y, t),$$

where $z(x, y, t)$ describes a surface that is analytic in $x$. We must allow the field to optionally be a single valued function of time $t \in \mathbb{R}$, since that is the nature of the solutions of the equations we wish to solve. Examples will be given below.

For example, picture of a smooth single-valued mountain (Fig. 1.27), having isolines (lines on a surface with constant slope).\textsuperscript{82}

Vector fields: A vector field (Appendix A, p. 232) is composed of three scalar fields. For example, the electric field used in Maxwell’s equations $\mathbf{E}(x, t) = [E_x, E_y, E_z]^T$ [V/m] has three components, each of which is a scalar field. When the magnetic flux vector $\mathbf{B}(x)$ is static (P5, p. 96), the voltage $\phi(x)$ [V] uniquely defines $\mathbf{E}(x, t)$, as the gradient of the voltage (potential)

$$\mathbf{E}(x, t) = -\nabla \phi(x, t). \quad [V/m]$$

(1.98)

Note that the electric strength $\mathbf{E}$ [V/m] is not a force, rather the electric force on a charge $q$ is $\mathbf{F} = q\mathbf{E}$.

\textsuperscript{78}https://en.wikipedia.org/wiki/Scalar_field
\textsuperscript{79}https://en.wikipedia.org/wiki/Vector_field
\textsuperscript{80}https://www.grc.nasa.gov/www/k-12/airplane/vectors.html
\textsuperscript{81}https://en.wikipedia.org/wiki/Field_(mathematics)
\textsuperscript{82}https://en.wikipedia.org/wiki/Euler_method
CHAPTER 1. INTRODUCTION

This figure shows an open surface $S(x)$, formed by bifurcating the closed volume $V$ with a plane (not shown), creating the boundary $B$. Only the upper half of the bifurcated volume is shown. A unit vector $\hat{n}$ is defined as perpendicular ($\perp$) to a second plane, tangent to $S$, any given point on $S$.

Figure 1.27: Figure defining the analytic open surface $S$, having boundary $B$.

Scalar potentials: The above derivation showed the importance of a potential in defining a vector field (Eq. 1.98). The difference between a potential and a scalar field is that potentials have units, thus have a physical meaning. Scalar potentials (i.e., voltage $\phi(x, t) [\text{V}]$, temperature $T(x, t) [\text{°C}]$ and pressure $\rho(x, t) [\text{Pascals}]$) are examples of important scalar fields. All potentials are composed of scalar fields, but not all scalar fields are potentials. For example $E_y(x, t) = \hat{y} \cdot E(x, t) [\text{V/m}]$, the $\hat{y}$ component of $E$, is not a potential.

Example: The voltage

$$\phi(x, t) = e^{\mp \kappa \cdot x} u(t) \leftrightarrow \frac{1}{s} e^{\mp \kappa \cdot x} \quad \text{[Volts]}$$

(1.99)

is an important case since it is one of the two d’Alembert solutions (Eq. 1.84, p. 103) of the wave equation (Eq. 1.21, p. 53), as well as an eigen-function of the gradient operator $\nabla$. From the definition of the scalar (dot) product of two vectors (Fig. 1.13, p. 74),

$$\kappa \cdot x = \kappa_x x + \kappa_y y + \kappa_z z = ||\kappa|| \cdot ||x|| \cos \theta_{\kappa x},$$

where $||\kappa|| = \sqrt{\kappa_x^2 + \kappa_y^2 + \kappa_z^2}$ and $||x|| = \sqrt{x^2 + y^2 + z^2}$ are the lengths of vectors $\kappa$, and $x$ and $\theta_{\kappa x}$ is the angle between them. As before $s = \sigma + j \omega$ is the Laplace frequency.

To keep things simple let $\kappa = [\kappa_x, 0, 0]^T$ so that $\kappa \cdot x = \kappa_x x \hat{x}$. We shall soon see that $||\kappa|| = 2\pi / \lambda$, follows from the basic relationship between a wave’s radian frequency $\omega = 2\pi f$ and its wavelength $\lambda$

$$\omega \lambda = c_0.$$  

(1.100)

As frequency increases, the wavelength becomes shorter. This key relationship may have been first research by Galileo (c.1564), followed by (c.1627) Mersenne\(^3\) (Fig. 1.4, p. 22).

Example: The step function $u(t) \leftrightarrow 1/s$ is not analytic at $t = 0$ in time. In the Laplace frequency domain, $1/s$ is not complex analytic at $s = 0$, due to the pole. By definition, $\phi(x, t)$ is analytic in $x$ on $S$.

Vector potentials: Vector potentials, like scalar potentials, are vector fields with a physical meaning (have physical units). Thus they are more complicated than scalar potentials because they are composed of three scalar fields, rather than one. But in every other way the are just as

\(^3\)http://www-history.mcs.st-and.ac.uk/Biographies/Mersenne.html

"In the early 1620s, Mersenne listed Galileo among the innovators in natural philosophy whose views should be rejected. However, by the early 1630s, less than a decade later, Mersenne had become one of Galileo's most ardent supporters."

1.5. STREAM 3B: VECTOR CALCULUS (10 LECTURES)

important, because of the Fundamental theorem of vector calculus or simply Helmholtz’ decomposition theorem.

If you found it useful set up (define) your problem using voltage, and then take the gradient to find \( \mathbf{E}(x, t) \), the the same will hold when using the vector potential. When operating on a scalar potential we use a gradient, whereas for the vector potential, we must operate with the curl. In Eq. 1.98 we assumed that the magnetic flux vector \( \mathbf{B}(x) \) was static, thus \( \mathbf{E}(x, t) \) is the gradient of the time dependent voltage \( \phi(x, t) \). However when the magnetic field is dynamic (not static), Eq. 1.98 is not valid due to magnetic induction: A voltage induced into a loop of wire, proportional to the time varying flux cutting across that loop of wire. This is known as Ampere’s law. In the static case the induced voltage is zero. Lets explore just how useful the potentials can be in terms of quantifying Maxwell’s equations. When the magnetic field is time varying, Eq. 1.98 must be extended to include both the scalar \( \phi(x, t) \) and vector potentials \( \mathbf{A}(x, t) \) (Sommerfeld, 1952, p. 146)

\[
\mathbf{E}(x, t) = -\nabla\phi(x, t) - \frac{\partial \mathbf{A}}{\partial t}, \quad \text{[V/m]} \tag{1.101}
\]

(Feynman, 1970b, p. 18-10) and the magnetic flux \( \mathbf{B}(x, t) \) is

\[
\mathbf{B}(x, t) = \nabla \times \mathbf{A}(x, t) \quad \text{[W/m]} \tag{1.102}
\]

(Feynman, 1970b, p. 18-9). Thus the electric field strength includes both the voltage \( \phi(x, t) \) and magnetic flux \( \mathbf{A}(x, t) \) components, while the magnetic field strength only depends on the magnetic potential.

To better understand the difference between scalar and vector potentials, I recommend that you read the first chapters of Feynman (1970b, Chapter 1.1), which runs a close parallel to these pages. The two discussions complement each other and have a similar level of difficulty and presentation style.

1.5.2 Lec 34: Gradient \( \nabla \), divergence \( \nabla \cdot \), Curl \( \nabla \times \), and Laplacian \( \nabla^2 \)

There are three key vector differential operators that are required for understanding linear PDEs, such as the wave equation and the diffusion equation. All of these begin with the \( \nabla \) operator

\[
\nabla = \hat{x} \frac{\partial}{\partial x} + \hat{y} \frac{\partial}{\partial y} + \hat{z} \frac{\partial}{\partial z}.
\]

The official name of this operator is nabla. It has three basic uses: 1) the gradient of a scalar field, the 2) divergence of a vector field, and 3) the curl of a vector field. If properly noted, the shorthand notation \( \nabla \phi(x) = (\hat{x} \partial_x + \hat{y} \partial_y + \hat{z} \partial_z) \phi(x) \) is convenient.

**Basic differential vector operator definitions:** The basic definitions of each of the vector operators are summarized in Table 1.7.

**Examples:**

1. The gradient transforms a scalar field into vector field. In \( \mathbb{R}^3 \) the gradient of a scalar field \( \nabla \phi(x) \) is defined as

\[
\nabla \phi(x) = \left( \frac{\partial \phi}{\partial x} \hat{x} + \frac{\partial \phi}{\partial y} \hat{y} + \frac{\partial \phi}{\partial z} \hat{z} \right) \phi(x), = \frac{\partial \phi}{\partial x} \hat{x} + \frac{\partial \phi}{\partial y} \hat{y} + \frac{\partial \phi}{\partial z} \hat{z}. \tag{1.103}
\]
Table 1.7: The three vector operators manipulate scalar and vector fields, as indicated here. The gradient converts scalar fields into vector fields. The divergence eats vector fields and outputs scalar fields. Finally the curl maps vector fields into vector fields. It is helpful to have a name for second order operators (e.g., DoG, GoD: mnemonics defined in Sect. 1.5.10, p. 150).

<table>
<thead>
<tr>
<th>Name</th>
<th>Input</th>
<th>Output</th>
<th>Operator</th>
<th>Mnemonic</th>
</tr>
</thead>
<tbody>
<tr>
<td>Gradient</td>
<td>Scalar</td>
<td>Vector</td>
<td>(\nabla())</td>
<td>Grad</td>
</tr>
<tr>
<td>Divergence</td>
<td>Vector</td>
<td>Scalar</td>
<td>(\nabla \cdot ())</td>
<td>Div</td>
</tr>
<tr>
<td>Curl</td>
<td>Vector</td>
<td>Vector</td>
<td>(\nabla \times ())</td>
<td>Curl</td>
</tr>
<tr>
<td>Laplacian</td>
<td>Scalar</td>
<td>Scalar</td>
<td>(\nabla \cdot \nabla = \nabla^2())</td>
<td>DoG</td>
</tr>
<tr>
<td>Vector Laplacian</td>
<td>Vector</td>
<td>Vector</td>
<td>(\nabla \nabla = \nabla^2())</td>
<td>GoD</td>
</tr>
</tbody>
</table>

The gradient may be factored into a unit vector \(\hat{n}\), as defined in Fig. 1.27, defining the direction of the gradient, and the gradient’s length \(||\nabla()||\), defined in terms of the norm. Thus the gradient of \(\phi(x)\) may be written in “polar coordinates” as \(\nabla \Phi(x) = ||\nabla(\Phi)|| \hat{n}\), which is used to define \(\hat{n}\):

\[
\hat{n} = \frac{\nabla(\Phi(x))}{||\nabla(\Phi)||}.
\]

Important examples of the use of the gradient include the electric field vector \(E(x) = -\nabla \phi(x) \text{[Volts/m]}\), which is the gradient of a voltage [V], and the force density \(f(x) = -\nabla \phi(x) \text{[N/m]}\), which is the gradient of a pressure [Pa].

2. The divergence of a vector field results in a scalar field. For example, the divergence of the electric field flux vector \(D(x) \text{[Col/m}^2\]\) equals the scalar field charge density \(\rho(x) \text{[Col/m}^3\]\):

\[
\nabla \cdot D(x) \equiv \left(\frac{\partial}{\partial x} \hat{x} + \frac{\partial}{\partial y} \hat{y} + \frac{\partial}{\partial z} \hat{z}\right) \cdot D(x) = \frac{\partial D_x}{\partial x} + \frac{\partial D_y}{\partial y} + \frac{\partial D_z}{\partial z} = \rho(x).
\] (1.104)

This is analogous to the scalar (dot) product between two vectors.

When working with guided waves (narrow tubes of flux), when the diameter is small compared with the wavelength (P10, p. 97), the divergence is

\[
\nabla \cdot D(x) = \nabla_r D_r = \frac{1}{A(r)} \frac{\partial}{\partial r} A(r) D_r(r),
\] (1.105)

where \(r\) is the distance down the horn (range variable), \(A(r)\) is the area of the iso-response surface, as a function of range \(r\), and, \(D_r(r)\) is the radial component of vector \(D\), as a function of range \(r\). For example, in spherical coordinates the area \(A(r) = A_o r^2\) is proportional to the square of the range. This expression reduces to the radial component of the divergence of \(D(x)\) in spherical coordinates. In cylindrical coordinates \(A(r) = A_o r\), and in rectangular coordinates the area \(A = \pi r_o^2\) is independent of the range \(r\).

A general and detailed derivation of these cases is derived in Section 5.1.5, p. 218.

3. The curl transforms a vector field into a vector field. For example the curl of the magnetic intensity \(H(x) \text{[Amp/m}^2\]\) vector is equal to the vector current density \(C(x) \text{[Amps/m}^2\]\):

\[
\nabla \times H(x) \equiv \begin{vmatrix} \hat{x} & \hat{y} & \hat{z} \\ \frac{\partial}{\partial x} & \frac{\partial}{\partial y} & \frac{\partial}{\partial z} \\ H_x & H_y & H_z \end{vmatrix} = C(x).
\] (1.106)
The notation $|\cdot|$ indicates the determinant (Appendix A, p. 229), $\partial_x$ is shorthand for $\partial/\partial x$ and $H = [H_x, H_y, H_z]^T$.

4. The Laplacian $(\nabla^2 \equiv \nabla \cdot \nabla)$ is
\[
\frac{\partial^2}{\partial x^2} + \frac{\partial^2}{\partial y^2} + \frac{\partial^2}{\partial z^2},
\]
(1.107)
takes a scalar field onto a scalar field.

Example: Starting from a scalar field, the gradient produces a vector, which is then operated on by the divergence to take the output of the gradient back to a scalar field. One of the classic cases is the Laplacian of a voltage field. The gradient of the scalar voltage $\Phi(x)$ [V] results in the electric field vector $E$
\[
E(x) = [E_x(x), E_y(x), E_z(x)]^T = -\nabla \Phi(x), \quad [V/m]
\]
which in free space is proportional to the electric flux $D = \varepsilon_0 E$ [C/m²], the divergence of which gives the charge density $\rho(x)$ [C/m³]. Here $\varepsilon_0$ [F/m] is the vacuum permittivity, which is $\approx 8.8542 \times 10^{-12}$ [F/m].

Example: The simplest example of a scalar field is the voltage between two very large (think $\infty$) conducting parallel planes, or plates (large so that we may ignore the edge effects). In this case the voltage varies linearly (the voltage is complex analytic) between the two plates. For example
\[
V(x, y, z) = V_o(1 - x)
\]
is a scalar field. At $x = 0$ the voltage is $V_o$ and at $x = 1$ the voltage is zero. Between 0 and 1 the voltage varies linearly. Thus $V(x, y, z)$ defines a scalar field.

If the same setup were used but the two plates were $1 \times 1$ [cm²], with a 1 [mm] air gap, there will be a small “fringe" effect at the edges that would (slightly) modify the ideal fields. This effect can be made small by changing the air gap to area ratio, so that the ends do not significantly impact the capacitor’s value.

Example: A second classic example is an acoustic pressure field $\rho(x, t)$ [Pa], which defines a vector force density $f(x, t) = -\nabla \rho(x, t)$ [N/m²] (Eq. 1.114, p. 131). When this force density [N/m²] is integrated over an area, the net radial force [N] is
\[
F_r = -\int_S \nabla \rho(x)||dS||. \quad [N]
\]
(1.109)

An inflated balloon with a static internal pressure of 3 [atm], in an ambient pressure of 1 [atm], forms a sphere due to the elastic nature of the rubber, which acts as a stretched spring under tension. The net force on the surface of the balloon is its area times the pressure drop of 2 atm across the surface. Thus the static pressure is
\[
\rho(x) = 3u(r_o - r) + 1, \quad [Pa]
\]
where $u(r)$ is a step function of the radius $r = ||x|| > 0$, centered at the center of the balloon, having radius $r_o$.

Taking the gradient gives the negative\(^\text{84}\) of the radial force density (i.e., perpendicular to the surface of the balloon)
\[
-f_r(r) = \nabla \rho(x) = \frac{\partial}{\partial r} 3u(r_o - r) + 1 = -2\delta(r_o - r). \quad [Pa]
\]
\(^\text{84}\)The force is pointing out, stretching the balloon.
This describes a static pressure that is 3 atmospheres inside the balloon, and 1 atmosphere \([\text{At}](1 \text{ At} = 10^5 \text{ Pa})\) outside. Note that the net positive force density is the negative of the gradient of the static pressure.

Taking the divergence of this radial force gives the Laplacian of the scalar pressure field

\[
\nabla^2 \rho(x) = \nabla \cdot \nabla \rho(x) = -\nabla \cdot f(x).
\]

From Appendix 5.2 (Eq. 5.17 p. 220) the Laplacian in spherical coordinates simplifies (Eq. 5.19, p. 220)

\[
\nabla^2 \rho(x) = \frac{1}{r^2} \frac{\partial}{\partial r} r^2 \frac{\partial}{\partial r} \rho(x) = \frac{1}{r} \frac{\partial^2}{\partial r^2} r \rho(x).
\]

Thus the net force \(f(x) = [F_r, 0, 0]^T\), (Eq. 1.109), in spherical coordinates, has a radial component \(F_r\), and angular components of zero. This force is well approximated by a delta function across the thin sheet of stretched rubber.

**Exercise:** Show that Eq. 1.99 is an eigen-function of the Gradient operator \(\nabla\). Taking the gradient of \(\phi(x,t)\) gives

\[
\nabla e^{\kappa x} u(t) = (\nabla \kappa \cdot x) e^{\kappa x} u(t) = \kappa e^{\kappa x} u(t),
\]

or in terms of \(\phi(x,t)\)

\[
\nabla \phi(x,t) = \kappa \phi(x,t) \leftrightarrow \frac{\kappa}{s} e^{\kappa x}.
\]

Thus \(\phi(x,t)\) is an eigen-function of \(\nabla\), having the vector eigen-value \(\kappa\). As before \(\nabla \phi\) is proportional to the current, since \(\phi\) is a voltage, and the ratio, the eigen-value, is an impedance.

**Exercise:** Compute \(\hat{n}\) for \(\phi(x,t)\) (Eq. 1.99). \(\hat{n} = \kappa/||\kappa||\). This represents a unit vector in the direction of the current.

**Exercise:** If the sign of \(\kappa\) is negative, what are the eigen-vectors and eigen-values of \(\nabla \phi(x,t)\)?

\[
\nabla e^{-\kappa x} u(t) = -\kappa \cdot \nabla (x) e^{-\kappa x} u(t) = -\kappa e^{-\kappa x} u(t).
\]

Nothing changes other than the sign of \(\kappa\). Physically this means the wave is traveling in the opposite direction, corresponding to the forward and retrograde d’Alembert waves.

Prior to this section, we have only considered the Taylor series in one variable, such as for polynomials \(P_n(x), x \in \mathbb{R}\) (Sect. 1.3.1, Eq.1.24 p. 55) and \(P_n(s), s \in \mathbb{C}\) (Sect. 1.4.2, Eq.1.39 p. 64). The generalization from real to complex analytic functions led to the Laplace Transform, an the hosts of integration theorems (FTCC, Cauchy I, II, III). What is in store when we generalize from one spatial variable (\(\mathbb{R}\)) to three (\(\mathbb{R}^3\))?

**Exercise:** If \(E(x,t) = E_x \hat{x}\), express \(E(x,t)\) in terms of the voltage potential \(\phi(x,t)\) [V].

The electric field strength may be found from the voltage as \(E(x,t) = -\nabla \phi(x,t) = -\hat{x} \partial \phi(x,t)/\partial x\) [V/m].
Exercise: Find the velocity $v(t)$ of an electron in a field $E$. From Newton’s 2d law, $-qE = m_e \dot{v}(t) \,[\text{Nt}]$, where $m_e$ is the mass of the electron. Thus we must solve this first order differential equation to find $v(t)$. This is easily done in the frequency domain $v(t) \leftrightarrow V(\omega)$.

Role of Potentials: Note that the scalar fields (e.g., temperature, pressure, voltage) are all scalar potentials, summarized in Fig. 1.15 (p. 81). In each case the gradient of the potential results in a vector field, just as in the electric case above (Eq. 1.98).

It is important to understand the physical meaning of the gradient of a potential, which is typically a generalized force (Electric field, acoustic force density, temperature flux), which in turn generates a flow (current, velocity, heat flux). The ratio of the gradient over the flow determines the impedance.

1. Example 1: The voltage drop across a resistor causes a current to flow, as described by Ohm’s Law. Taking the difference in voltage between two points is a crude form of gradient when the frequency $f \,[\text{Hz}]$ is low, such that the wavelength is much larger than the distance between the two points. This is the essence of the quasi-static approximation P10 (97).

2. Example 2: The gradient of the pressure gives rise to a force density in the fluid medium (air, water, oil, etc.), which causes a flow (velocity vector) in the medium.

3. Example 3: The gradient of the temperature also causes a flow of heat, which is proportional to the thermal resistance, given Ohms law for heat (Feynman, 1970c, p. 3-7).\(^85\)

4. Example 4: Nernst’ potential: When a solution contains charged ions, it is called an electro-chemical Nernst’ potential $N(x,t)$.\(^86\) This electro-chemical field is similar to a voltage or temperature field, the gradient of which defines a virtual force on the charged ions.

Thus in the above examples there is a potential, the gradient of which is a force, which when applied to the medium (an impedance) causes a flow (flux or current) proportional to that impedance, due to the medium. This is a very general set of concepts, worthy of some thought. In every case there is a force and a flow. The product of the force and flow is a power, while the ratio an impedance, which we have learned about in Section 1.3.7 in the form of 2x2 ABCD matrices (Eq. 1.61, p. 79).

Exercise: Show that the integral of Eq. 1.98 is an anti-derivative.

---

\(^85\)https://en.wikipedia.org/wiki/Thermal_conduction#Fourier’s_law
\(^86\)https://en.wikipedia.org/wiki/Nernst_equation
The solution requires the definition of the anti-derivative, defined by the FTC (Eq. 1.75, p. 100):

\[
\phi(x, t) - \phi(x_o, t) = \int_{x_o}^{x} E(x, t) \cdot dx
\]

\[
= -\int_{x_o}^{x} \nabla \phi(x, t) \cdot dx
\]

\[
= -\int_{x_o}^{x} \left( \hat{x} \frac{\partial \phi}{\partial x} + \hat{y} \frac{\partial \phi}{\partial y} + \hat{z} \frac{\partial \phi}{\partial z} \right) \phi(x, t) \cdot dx
\]

\[
= -\int_{x_o}^{x} \left( \hat{x} \frac{\partial \phi}{\partial x} + \hat{y} \frac{\partial \phi}{\partial y} + \hat{z} \frac{\partial \phi}{\partial z} \right) \cdot (\hat{x} dx + \hat{y} dy + \hat{z} dz)
\]

\[
= -\int_{x_o}^{x} \frac{\partial \phi}{\partial x} dx - \int_{y_o}^{y} \frac{\partial \phi}{\partial y} dy - \int_{z_o}^{z} \frac{\partial \phi}{\partial z} dz
\]

\[
= -\int_{x_o}^{x} d\phi(x, t)
\]

\[
= -(\phi(x, t) - \phi(x_o, t)).
\]

is the anti-derivative. We may verify this by taking the gradient of both sides

\[
\nabla \phi(x, t) - \nabla \phi(x_o, t) = -\nabla \int_{x_o}^{x} E(x, t) \cdot dx = E(x, t).
\]

Applying the FTC (Eq. 1.75, p. 100), the anti-derivative must be \(\phi(x, t) = E_x x \hat{x} + 0 \hat{y} + 0 \hat{z}\). This very same point, made by Feynman (1970c, p. 4-1, Eq. 4.28), is worth reading.

Given that the force on a charge is proportional to the gradient of the potential, from the above Exercise showing that the integral of the gradient only depends on the end points, the work done in moving a charge only depends on the limits of the integral. This is the definition of a \textit{conservative field}. This only holds when \(E\) is determined by Eq. 1.98, and the medium of the charge has no friction (i.e., there are no other forces on the charge).

**The conservative field:** This leads to an important question: “When is a field conservative?” A field is conservative when the work done by the motion is independent of the path of motion. This seems related to the FTC, that the integral only depends on the end points, for a case where the integral represents the total work done.

A proper answer must wait for the introduction of the \textit{fundamental theorem of vector calculus}, discussed in Section 1.5.11 (Eq. 1.157, p. 152), but it is insightful to give several specific examples: Example: The force of gravity is given by the gradient of Newton’s gravitational potential (Eq. 1.19, p. 52)

\[
F = -\nabla \phi_g(r).
\]

Historically this was the first conservative field, used to explain the elliptic orbits of the planets around the sun.

Example: If a temperature field is a time-varying constant (i.e., \(T(x, t) = T_o(t)\)), there is no “heat flux,” since \(\nabla T_o(t) = 0\). When there is no heat flux (i.e., flux, or flow), there is no heat power, since the power is the product of the force times the flow.

Example: The viscosity of air is quite small, thus as a simplifying assumption, is typically assumed to be zero. However at high frequencies, where the wavelength is small (e.g., at 100 [kHz] \(\lambda = c_o / f = 345/10^5 = 3.45 [\text{mm}]\)), the viscosity is significant in sound propagation, and there is an associated power loss.\(^{87}\) If the viscosity is assumed to be significant, the field is no

\(^{87}\)https://en.wikipedia.org/wiki/Laminar_flow#Examples
1.5. STREAM 3B: VECTOR CALCULUS (10 LECTURES)

longer conservative.

Example: A voltage is a scalar potential, the gradient of which defines the electric field (Eq. 1.98), which drives a current (flow) across a resistor (or any impedance). If the impedance is infinite, the flow will be zero, leading to zero power dissipation. When the impedance is lossless, the system is conservative.

1.5.3 Partial differential equations and field evolution:

There are three main category of partial differential equations (PDEs):

1. **diffusion equations** (Eq. 1.110), which describes the evolution of the scalar temperature $T(x,t)$ (a scalar potential) and gradients of solution concentrations (i.e., ink in water) and Brownian motion.

2. **scalar wave equations** (Eq. 1.20), which describe the propagation of a pressure $\rho(x,t)$ sound field (a scalar potential),

3. **vector wave equations** (i.e., Maxwell’s equations), which describe the propagation of electric $E(x,t)$ and $D(x,t)$ and magnetic $B(x,t)$ and $H(x,t)$ vector fields.

All of these PDEs describe the evolution of either scalar fields (pressure $\rho(x,t)$ and temperature $T(x,t)$ or vector fields ($E, D, B, H$) in space, as functions of space $x$ and time $t$.

They are distinguished by the order of the derivative with respect to time. The diffusion equation is first order in time, while propagation is second order. When these equations are Laplace transformed, diffusion has a single real root, while the the wave equation has pairs of complex conjugate roots. This is the most important distinction between diffusion and propagation. The diffusion equation is thus said to be **parabolic** (first order in time, second order in space), while the wave equation is said to be **hyperbolic** (second order in time and space). Laplace’s’ equation is a special case, where the solution is static, where the solution is said to be **elliptic**. These categories have descriptive, but not significant mathematical utility.

Partial differential equations (PDEs) define the “evolution” of the field. There are two basic categories of field evolution, **diffusion** and **propagation**.

1. **Diffusion:** The simplest and easiest PDE example, easily visualized, is a static\(^{88}\) (time invariant) scalar temperature field $T(x)$ [°C]). Just like an impedance or admittance, a field has regions where it is analytic, and for the same reasons, $T(x,t)$ satisfies Laplace’s equation

$$\nabla^2 T(x,t) = 0.$$  

Since there is no current when the field is static, such systems are lossless, thus are conservative.

When $T(x,t)$ depends on time (is not static), it is described by the **diffusion equation**

$$\nabla^2 T(x,t) = \kappa \frac{\partial}{\partial t} T(x,t),$$  \hspace{1cm} (1.110)

a rule for how $T(x,t)$ evolves with time from its initial state $T(x,0)$. The constant $\kappa$ is called the **thermal conductivity** which depends on the fluid in the container, with $s\kappa$ being the admittance per unit area.

\(^{88}\)Postulate P3, p. 96.
CHAPTER 1. INTRODUCTION

Note that if \( T(x, \infty) \) reaches steady state as \( t \to \infty \), it evolves into a static state, thus \( \nabla^2 T = 0 \). This depends on what is happening at the boundaries. When the wall temperature of a container is a function of time, then so will the internal temperature continue to change, but with a delay, that depends on the thermal conductivity \( \kappa \).

Such a system is analogous to an electrical resistor-capacitor series circuit, connected to a battery. The wall temperature and the voltage on the battery represent the potential driving the system, the thermal conductivity \( \kappa \) and the electrical resistor are analogous, and the fluid heat capacity and the electrical capacitor, are being heated (charged) by the flux. In both cases Ohm’s law defines the flux.

How this happens can only be understood once the solution to the equations has been established.

2. Propagation Pressure and electromagnetic waves are described by a scalar potential (pressure) (Eq. 1.20, p. 52) and a vector potential (electromagnets) (Eq. 1.162, p. 156) resulting in scalar and vector wave equations.

All these partial differential equations, scalar and vector wave equations, and the diffusion equation, depend on the Laplacian \( \nabla^2 \), which first appeared to us as Laplace’s equation, which we first saw with the Cauchy-Riemann conditions (Eq. 1.83, p. 102).

The vector Taylor series: Next we shall expand on the concept of the one-dimensional Taylor series, a function of one variable, to \( x \in \mathbb{R}^3 \). Just as we generalized the derivative with respect to the real frequency variable \( \omega \in \mathbb{R} \) to the complex analytic frequency \( s = \sigma + \omega \in \mathbb{C} \), here we generalize the derivative with respect to \( x \in \mathbb{R} \), to the vector \( x \in \mathbb{R}^3 \).

Since the scalar field is analytic in \( x \), it is a perfect place to start. Assuming we have carefully defined the Taylor series 1.33 (p. 60) in one and two (Eq. 1.80, p. 102) variables, the Taylor series of \( f(x) \) in \( x = 0 \in \mathbb{R}^3 \) may be defined as

\[
f(x + \delta x) = f(x) + \nabla f(x) \cdot \delta x + \frac{1}{2!} \sum_k \sum_l \frac{\partial^2 f(x)}{\partial x_k \partial x_l} \delta x_k \delta x_l + \text{HOT}. \tag{1.111}
\]

From this definition it is clear that the gradient is the generalization of the second term in the 1D Taylor series expansion.

Vector field: A vector field is an ordered set of scalar fields. For example

\[
A(x) = [\phi(x), \psi(x), \theta(x)]^T
\]

is a vector field in \( \mathbb{R}^3 \) when each of the three function is differentiable (i.e., analytic). For example \( A(x) = [x, xy, xyz]^T \) is a legal vector field (the components are analytic in \( x \)).

Taking an example from Maxwell’s equations, the magnetic flux vector is given by

\[
B(x, t) = -\nabla \times A(x, t).
\]

We shall see that this is always true because the magnetic charge \( \nabla \cdot B(x, t) \) is always 0.

To verify that a field is a potential, check out the units [V, A, °C]. However a proper mathematical definition is that the potential must be an analytic function of \( x \) and \( t \), so that one may operate on it with \( \nabla () \) and \( \nabla \times () \). The divergence of a scalar field is not a legal operation.

An important discussion of vector potentials, with extensive examples, may be found in Feynman (1970c, p. 14-1 to 14.3). If you need to master vector potentials, Read Ch. 14.1 of Feynman (1970c).
In summary: For every potential: a force, proportional to the gradient of a scalar, drives a generalized flow. When the system is linear (P2, p. 96), the product of the force and flow is the power, and the ratio (force/flow) is an impedance. This impedance statement is called either Ohm’s Law, Kirchhoff’s Laws, Laplace’s Law, or Newton’s Laws. In the simplest of cases, they are all linearized (proportional) relationships, between a force and a flow. Very few impedance relationships are inherently linear over a large range of force or current, but for physically useful levels, they are typically linear (P2, p. 96).

Note that it is the difference in the potential (i.e., voltage, temperature, pressure) that is proportional to the flux. This is a major simplification of the gradient relationship, justified by the quasistatic assumption P10 (p. 97), and is the physical basis of the Fundamental theorem of algebra (p. 66), since the impedance is the ratio of the force (potential drop) over the flow (flux).

In electrical circuits it is common to define a zero potential ground point, that all voltages use as the referenced potential. This results in abstracting away (hiding) the difference in voltage. The ground is useful convention, as a simplifying rule, but it obscures the physics, and obscures the fact that the voltage is not the force. Rather the force is the voltage drop, referenced to the ground, which is defined as zero volts. More precisely, \( F = -\nabla \Phi \). It seems misleading (more precisely it is wrong) to state Ohm’s law as the voltage over the current, since Ohm’s law actually says that the voltage drop (i.e., voltage gradient) over the current, that defines the impedance. Like a voltage, the pressure is the potential, the gradient of which is a force density, which drives the flow. More on this in section 1.5.11 (p. 150), where we introduce The Fundamental theorem of vector calculus (aka Helmholtz’ decomposition theorem), which generalizes Ohm’s law to include circulation (e.g., EM magnetic) effects.

1.5.4 Lec 35: Scalar Wave Equation (Acoustics)

In this Section we discuss the general solution to the wave equation. In general the wave equation has two forms, scalar waves (acoustics) and vector waves (electromagnetics). These have an important mathematical distinction, but have a similar solution space, one scalar and the other vector. To understand the differences we start with the scalar wave equation.

Scalar waves:

There are three distinct mathematical classes of equations used to describe physical systems:

1. The scalar wave equation describes the evolution of a scalar field, such as a pressure or voltage, or equivalently, the flow. The best known examples are electrical and acoustic transmission lines. Such systems are loosely referred to as the telegraph or telephone equations, referring back to early days of their discovery. In acoustics, guided waves are called horns, such as the horn connected to the first phonographs from around the turn of the century (Webster, 1919). Thus the names can reflect their historical development, with the mathematics and the applications frequently running in close parallel.

2. The vector wave equation describes the evolution of a vector field, such as Maxwell’s electric field vector \( \mathbf{E}(x, t) \). When these fields are restricted to a one dimensional domain they are known as guided waves constrained by wave guides.

3. The third class is systems of lumped elements, such as electrical inductors, capacitors and resistors, and their mechanical counterparts, masses, springs and dash-pots. Such systems
are represented by systems of matrix equations, the size of which is equal to the number of elements in the network. These networks are not viewed as supporting waves, thus not viewed as guided systems. However there is an overlap with scalar waves and transmission lines, which may be modeled as lumped networks of 2x2 matrices (Section 1.3.7, Eq. 1.61, p. 79).

The scalar wave equation: The starting point for understanding PDEs is to explore the scalar wave equation, thus we shall limit our analysis to acoustics, the classic case of scalar waves. Acoustic wave propagation was first analyzed mathematically by Isaac Newton (electricity had yet to be discovered) in his famous book *Principia* (1687), where he was the first to calculate the speed of sound, based on conservation of mass and momentum. His early development understandably ignored viscous and thermal losses, which can be significant in a thin region called the *boundary layer*, at acoustic frequencies when the radius of the container (i.e., the horn) becomes less than the viscous boundary layer (e.g., much less than 1 mm) as first described in Helmholtz (1863b) and extended to include thermal losses by Kirchhoff (1868, 1974, English translation), as succinctly summarized by Rayleigh (1896). The experimental verification of these theoretical studies was ultimately verified by Mason (1927, 1928).

Loss-less horns include an important generalization of the solution of the 1-dimensional (1D) wave equation, in regions where the area $A(r)$ of the horn varies along the *range* axial (e.g., along the direction of wave propagation), as depicted in Fig. 1.28. Classic applications of horns include vocal tract acoustics, loudspeaker design, cochlear mechanics, and cases having wave propagation in periodic media (Brillouin, 1953).

For the 1D scalar wave equation (guided waves, aka, acoustic horns) the Laplacian is

$$\nabla^2 \varphi(r, t) \equiv \nabla \cdot \nabla \varphi(r, t) = \frac{1}{A(r)} \frac{\partial}{\partial r} A(r) \frac{\partial}{\partial r} \varphi(r, t), \quad (1.112)$$

which assumes that the horn’s diameter $d$ is less than first allowed “radial mode’s” (a half wavelength) (i.e., $d < \lambda/2$).\(^{89}\) The term on the right is an approximation that applies for thin tubes (e.g., spherical, cylindrical coordinates, etc.), with angular symmetry, where the potential $\psi(x, t) \leftrightarrow \Psi(x, s)$ is averaged over the iso-potential cross-sectional area (Appendix 5.1.5, p. 218). Such formulations, which explicitly depend on the area of the horn, are called *Webster horn equations*.

It follows that the 1D acoustic wave equation is

$$\frac{1}{A(r)} \frac{\partial}{\partial r} A(r) \frac{\partial}{\partial r} \varphi(r, t) = \frac{1}{c_o^2} \frac{\partial^2}{\partial t^2} \varphi(r, t) \leftrightarrow \frac{s^2}{c_o^2} \mathcal{P}(r, s), \quad (1.113)$$

where $\varphi(r, t) \leftrightarrow \mathcal{P}(r, s)$ is the average pressure (Hanna and Slepian, 1924; Mawardi, 1949; Morse, 1948), Olson (1947, p. 101), Pierce (1981, p. 360). Extensive experimental analysis for various types of horns (conical, exponential, parabolic) along with a review of horn theory is provided by Goldsmith and Minton (1924).

**Separable coordinate systems:** Classically PDEs are often solved by a technique called *separation of variables*, which is limited to a small number coordinate systems. By the use of the quasistatic approximation (P10, 97), the limitations of separation of variables may be avoided, at

---

\(^{89}\)This condition may be written in forms, the most common being that for a ridged walled tube $ka < 1$, where $k = 2\pi/\lambda$ and $a$ is the radius. This may be expressed in terms of the diameter as $\frac{2\pi d}{\lambda} < 1$, which is $d < \lambda/p < \lambda/2$. Thus $d < \lambda/2$ is a more precise metric, by a factor of $\pi/2 \approx 1.6$. 
1.5. STREAM 3B: VECTOR CALCULUS (10 LECTURES)

the price of a key assumption, quasi-statics (P10, p. 97). The most common examples of separable coordinate systems are rectangular, cylindrical and spherical coordinates (Morse, 1948, p. 296-7). Even a slight deviation from separable specific coordinate systems represents a major barrier to further understanding, blocking insight into the general case. These few separable coordinate systems are special cases, which have high degrees of symmetry, while the wave equation is not tied to a specific coordinate system. Quasi-statics can proved solutions over a much wider class of geometries.

When the pressure (i.e., a force potential) is not constrained to a “separable” coordinate system, the PDE is called the Sturm-Liouville equation.

Early history: The study of wave propagation begins at least as early as Huygens (ca. 1678), followed soon after (ca. 1687) by Sir Isaac Newton’s calculation of the speed of sound (Pierce, 1981). To obtain a wave, one must include two basic components, the stiffness of air, and its mass. These two equations shall be denoted (1) Newton’s 2nd Law \( F = ma \) and (2) Hooke’s Law \( F = kx \), respectively. In vector form these equations are (1) Euler’s equation (i.e., conservation of momentum density)
\[
-\nabla \varrho(x, t) = \rho_o \frac{\partial}{\partial t} \mathbf{u}(x, t) \leftrightarrow \rho_o s \mathbf{V}(x, s), \tag{1.114}
\]
assuming \( \rho_o \) is independent of time and position \( x \), and (2) the continuity equation (i.e., conservation of mass density)
\[
-\nabla \cdot \mathbf{u}(x, s) = \frac{1}{\eta_o P_o} \frac{\partial}{\partial t} \varrho(x, t) \leftrightarrow \frac{s}{\eta_o P_o} \mathbf{P}(x, s) \tag{1.115}
\]
(Pierce, 1981, page 15). Combining Eqs. 1.114 and 1.115 (removing \( \mathbf{u}(x, t) \)) results in the 3-dimensional (3D) scalar pressure wave equation
\[
\nabla^2 \varrho(x, t) = \frac{1}{c_o^2} \frac{\partial^2}{\partial t^2} \varrho(x, t) \leftrightarrow \frac{s^2}{c_o^2} \mathbf{P}(x, s), \tag{1.116}
\]
where \( c_o \) is the sound velocity.

Exercise: Show that Eqs. 1.114 and 1.115 can be reduced to Eq. 1.116. Take the divergence of Eq. 1.114 gives
\[
-\nabla \cdot \nabla \varrho(x, t) = \rho_o \frac{\partial}{\partial t} \nabla \cdot \mathbf{u}(x, t). \tag{1.117}
\]
Note that $\nabla \cdot \nabla = \nabla^2$. Next substitute Eq. 1.115 into the above relation results in the scalar wave equation Eq. 1.116, since $c_o^2 = \frac{\mu_0 \rho_o}{\rho_o}$.

**Primitive solutions** $\varrho^\pm (r, t)$

Because the wave equation (Eq. 1.116) is 2d order in time, there are two causal independent primitive solutions. Likewise there are two causal independent primitive solutions of the homogeneous (i.e., undriven) Webster horn equation, for each choice of area function $A(x)$, identified as $\varrho^\pm (x, t)$ an outbound (right-traveling) and inbound (left-traveling) wave, denoted as the Laplace transform pair

$$\varrho^\pm (r, s) \leftrightarrow \mathcal{P}^\pm (r, s) = \int_0^\infty \varrho^\pm (r, t) e^{-st} dt,$$

typically normalized so that $\mathcal{P}^\pm (r_o, s) = 1$, where $r_o$ defines the input excitation point. This construction is equivalent to a separation of time and space variables.

These two primitive solutions depend on $A(r)$, which causes local reflections with any area variations. Thus there is a tight relationship between the area change $dA(r)/dr$, internal resonances (eigen-modes), and the primitive wave solutions $\mathcal{P}^\pm (r, s)$. These relationships will be explored further in Chapter 5.

**Complex vs. real frequency:** We shall continue to maintain the distinction that functions of $\omega$ are Fourier transforms and functions of Laplace frequency $s$ correspond to Laplace transforms, which, because they are causal, are necessarily complex analytic in $s$ in the right half plane (RHP) region of convergence (ROC). This distinction is critical, since we typically describe impedance $Z(s)$ and admittance $Y(s)$, which must be causal, in terms of their poles and zeros, as complex analytic functions in $s$.

**Primitive solutions of the 1D wave equation:** As best we know, Huygen (1690) was the first to gain insight into wave propagation. While his concept, today known as “Huygens’ principle” showed a deep insight, it was seriously flawed, as it ignored the backward traveling wave (Miller, 1991). In 1747 a famous French intellectual and mathematician, Jean le Rond d’Alembert, published the general form of the solution to the 1D scalar wave equation

$$\varrho(x, t) = f(t - x/c_o) + g(t + x/c_o),$$

(1.118)

where $f(\cdot)$ and $g(\cdot)$ are quite general functions of their argument. That this is the solution may be easily shown by use of the chain rule, by taking partials with respect to $x$ and $t$. Surprisingly, this is the solution even when the functions are not differentiable. For example $u(t \mp x/c_o)$, and $\delta(t \mp x/c_o)$, are valid solutions, even though the proof is more difficult.

**Exercise:** By the use of the chain rule, prove that d’Alembert’s formula satisfies the 1D wave equation. Taking a derivatives with respect to $t$ and $r$ give

$\bullet$ $\partial_t \varrho(r, t) = -c_o f'(r - c_o t) + c_o g'(r + c_o t)$

---

90When an analytic function of complex variable $s$ includes the pole it is called a **Laurent series** in $s$. For example, the impedance of a capacitor $C$ is $Z_c(s) = 1/sC$, which is analytic in $s$ everywhere other than $s = 0$. The capacitor has a voltage time response given by the integral of the current, i.e., $v(t) = \frac{1}{C} \int i(t) dt = \frac{1}{C} u(t) \ast i(t)$, where $u(t)$ is the Heaviside step function and $\ast$ represents convolution.
• \( \partial_{rr} g(r, t) = f'(r - c_o t) + g'(r + c_o t) \),

and a second derivatives gives

• \( \partial_{tt} g(r, t) = c_o^2 f''(r - c_o t) + c_o^2 g''(r + c_o t) \)

• \( \partial_{rr} g(r, t) = f''(r - c_o t) + g''(r + c_o t) \).

From these last two equations we have the 1D wave equation

\[
\partial_{rr} g(r, t) = \frac{1}{c_o^2} \partial_{tt} g(r, t),
\]

at odds with Huygens’ 1690 “principle” (Miller, 1991).

In terms of the physics, d’Alembert’s general solution describes two arbitrary wave-forms \( f(\cdot) \), \( g(\cdot) \), traveling at a speed \( c_o \), one forward, and one reversed. Thus his solution is quite easily visualized. As previously discussed, Newton (1687) was the first to calculate the speed of sound \( c_o \), be it with an error of \( \sqrt{1.4} \). This error was not corrected for over 200 years, following the creation of thermodynamics, and the concept of an adiabatic process.

**Example:** Assuming \( f(\cdot), g(\cdot) \) are \( \delta(\cdot) \), find the Laplace transform of the solution. Using Table 1.5 (p. 93) of Laplace Transforms on Eq. 1.118 gives

\[
\hat{g}(x, t) = \delta(t - x/c_o) + \delta(t + x/c_o) \leftrightarrow e^{-sx/c_o} + e^{sx/c_o}.
\]

Note that the delay is \( T_o = \pm x/c_o \).

**3D d’Alembert waves:** The d’Alembert solution generalizes to 3D waves by changing the area function in Eq. 1.113 with the 3D area \( A(r) = A_o r \), as explained in Appendix 5.2, Eq. 5.17 (p. 220), resulting in the 3D wave equation

\[
\nabla^2 g(r, t) = \frac{1}{r} \frac{\partial^2}{\partial r^2} rg(r, t) = \frac{1}{c_o^2} \frac{\partial^2}{\partial t^2} g(r, t).
\]

Multiplying by \( r \) results in the general spherical (3D) d’Alembert wave equation solution

\[
\hat{g}(r, t) = \frac{f(t - r/c_o)}{r} + \frac{g(t + r/c_o)}{r},
\]

for arbitrary wave-forms \( f(\cdot) \) and \( g(\cdot) \). These are the primitive wave solutions for the spherical scalar wave equation.

### 1.5.5 Related partial differential equations:

Historically the wave equation was seen to be related to several other important partial differential equations, as the theory evolved, resulting in the

1. **Fourier diffusion equation**

\[
D_o \frac{\partial T(x, t)}{\partial t} = \nabla^2 T(x, t) \tag{1.120}
\]

which describes, for example, the temperature \( T(x, t) \) as proposed by Fourier in 1822, or for the diffusion of two miscible liquids (Fick, 1855) and Brownian motion (Einstein, 1905),
2. Poisson’s equation
\[ \nabla^2 \Phi(x, t) = \rho(x, t) \] (1.121)
which holds for gravitational fields, or the voltage around a charge.

3. Laplace’s equation
\[ \nabla^2 \Phi(x, t) = 0, \] (1.122)
which describes, for example, the voltage inside a closed chamber with various voltages on the walls, or the steady state temperature within a closed container, given a specified temperature distribution on the walls.

Each of these four equations has properties that may be simply explained, and visualized, in simple geometries, and all contain the Laplacian \( \nabla^2 = \nabla \cdot \nabla(). \)

The Laplacian \( \nabla^2: \) We first discussed the Laplacian as a 2D operator in Section 1.4.2 (p. 101), when we studied complex analytic functions. Then an approximation for horns (variable tubes) was presented as Eq. 4. In 3D rectangular coordinates it is defined as
\[ \nabla^2 T(x) = \left( \frac{\partial^2}{\partial x^2} + \frac{\partial^2}{\partial y^2} + \frac{\partial^2}{\partial z^2} \right) T(x). \] (1.123)
In summary, the Laplacian operator is ubiquitous in mathematical physics, starting with simple complex analytic functions (Laplace’s equation), and progressing to the Poisson’s equation, the diffusion equation, and finally the wave equation. Only the wave equation expresses delay. The diffusion equation “wave” has an instantaneous spread (the effective “wave” velocity is infinite, i.e., its not a wave).

1.5.6 Lec 36: Properties of the Generalized impedance

Up to now the impedance has been a simple quantity, defined as the force over the flow
\[ \text{flow} = \frac{\text{force}}{\text{flow}}. \]
This works well for an electrical circuit \( Z(s) = -\nabla \Phi(s)/I(s) \), or a mechanical system, such as a mass \( Z(s) = m s U(\omega)/\rho_o U(\omega) \), or heat circuit, such as a thermal conductivity \( \kappa \) \( Z_{\text{thermal}} = -T/\kappa \nabla T \). In general these are vector quantities that depend on space and time. This leads a generalized force and flow, with impedance as a matrix. But even in a 1 dimensional system we have not yet fully characterized impedance.

Impedance: Impedance plays a crucial role in the solution of differential equations, and in the theory and practice of complex variables. The concept broadly applies to electricity, mechanics, acoustics and heat (Table 1.15, p. 81). Impedance is one of the best examples of a complex analytic function, thus plays a key role in potential theory.

Theoretically, impedance plays the role of a boundary condition in the solution of differential equations, accounting for the reflectance of a wave at the boundary. It plays a critical role in conservation of energy, since if energy at the boundary (surface) must always be lost, given a passive boundary.

The first person to fully appreciate the importance of impedance at the boundary may have been Arnold Sommerfeld, who expressed this in what today is known as the radiation impedance, which is the impedance boundary condition as a point source wave radiates to \( |R| \to \infty \).
1.5. STREAM 3B: VECTOR CALCULUS (10 LECTURES)

**Brune Impedance:** As the concept of impedance evolved (it is a linear relation between a force and a flow), it was incorporated into more theories, such as electrical circuit theory and mechanics. The first persons to quantify the concept of impedance was Ohm, followed by Kirchhoff and Heaviside. Kennelly, not Heaviside, was the first to express the idea as a complex variable of a complex function of frequency (Kennelly, 1893). Perhaps Heaviside fully appreciated the concept, and has been given proper credit.

There are several important theorems here that follow from Brune’s Theorem on positive-real functions, defined as

\[ \Re\{Z(\sigma > 0)\} \geq 0, \]  

(1.124)

where \( Z(s) = R(s) + jX(s) \in \mathbb{C} \) is a Brune impedance having real part \( R(s) = \Re\{Z(s)\} \) and imaginary part \( X(s) = \Im\{Z(s)\} \), each as functions of the Laplace frequency \( s = \sigma + \omega \) (Brune, 1931a; Van Valkenburg, 1964b). Thus a Brune impedance has a positive analytic real part in the right-half plane \( \sigma > 0 \). This condition has several non-obvious important ramifications.

1. \( Z(s) \leftrightarrow z(t) \) is real and causal
2. \( Y(s) = 1/Z(s) \leftrightarrow y(t) \) is real and causal
3. The phase \( \angle Z(\omega) \) lies between \( \pm \pi/2 \) (i.e., \( Z(s) \) is minimum-phase)
4. All the poles and zeros are first order (all the poles and zeros are strictly first order)
5. If two poles approach each other, there must be a zero between them.
6. All poles and zeros lie in the left half plane LHP \( (\sigma \leq 0) \)
7. \( Z(\sigma > 0) \) is complex-analytic \([Z(s) \text{ obeys the Cauchy-Riemann conditions in the RHP } (\sigma > 0)] \]
8. All poles and zeros of \( Z(s) \) on the \( \omega \) or on the \( \sigma \) axis must alternate \([\text{Foster’s Theorem (Van Valkenburg, 1964b)}]\]
9. All Brune impedances are quasi-static (lumped-parameter, where the size of the impedance element is much smaller than the wavelength).
10. The Brune impedance is a subset of impedance, where delay has been approximated away in the long-wavelength limit. For example, transmission lines are not in the class of Brune impedances.

In general it is difficult to test for Eq. 1.88 because it must be true everywhere in the RHP \( (\sigma > 0) \). Once simple test we have access to today is \texttt{zviz.m}. A plot of any given \( Z(s) \), and its reciprocal \( Y(s) \) must be positive in the RHP, which is easily visualized with \texttt{zviz.m}.

A Brune impedance is defined as the ratio of the force \( F(\omega) \) over the flow \( U(\omega) \), and may be expressed in residue form as

\[ Z(s) = sL_o + R_o + \sum_{k=1}^{K} \frac{c_k}{s - s_k} = \frac{N(s)}{D(s)}, \]

(1.125)
known as the partial fraction expansion. The term $L_o$ represents an inductance, and $R_o$ a resistor. Coefficients $c_k$ are the residues, and $s_k$ the roots of $D(s)$. It follows that

$$D(s) = \prod_{k=1}^{K} (s - s_k) \quad \text{and} \quad c_k = \lim_{s \to s_k} (s - s_k)Z(s) = \prod_{n'=1}^{K-1} (s - s_n),$$

where the prime on index $n'$ means that $n = k$ is not included in the product. Every Brune impedance may be expanded in a partial fraction expansion of the form of Eq. 1.125, and every partial fraction expansion may be converted back into the ratio of two polynomials, using the Matlab/Octave commands $[C,P,K]=\text{residue}(N,D)$ and $[N,D]=\text{residue}(C,P,K)$. These two representations are interchangeable.

Given a 2-port network, the input impedance and the transfer function share the same poles. In fact the transfer function must be all-pole.

**Exercise:** Find the Laplace transform ($\mathcal{L}T$) of the three impedance relations in terms of the force $F(s)$ and the velocity $V(s)$, along with the electrical equivalent impedance: Each classical law is a linear relation

1. Hooke’s Law $f(t) = Kx(t)$. Sol: First $\mathcal{L}T$ Hooke’s Law and then write it in terms of force and velocity.

Taking the $\mathcal{L}T$ gives

$$F(s) = KX(s).$$

Since

$$v(t) = \frac{d}{dt}x(t) \Leftrightarrow V(s) = sX(s).$$

Thus the impedance of the spring is

$$Z_s(s) = \frac{F(s)}{V(s)} = \frac{K}{s},$$

which is analogous to the impedance of an electrical capacitor. If we specifying the compliance $C$ of a spring as $C = 1/K$, the relation looks just like the electrical case.

While Hooke’s law says the force and displacement are proportional, in terms of impedance variables force and flow,

$$f(t) = K \int_0^t v(t)dt.$$  

2. Dash-pot resistance $f(t) = Rv(t)$. Sol: From the $\mathcal{L}T$ this becomes

$$F(s) = RV(s)$$

and the impedance of the dash-pot is then

$$Z_r = R,$$

analogous to that of an electrical resistor.

---

91 Is $C(s)$ of the ABCD matrix the same as the impedance denominator $D(s)$? Verify!
Exercise: If $Z_r$ is the impedance of a resistor, find the formula time domain impedance.

Sol: $Z_r = R \leftrightarrow z_r(t) = R \delta(t)$.

3. Newton’s 2d Law for a constant mass $M$: $f(t) = M \frac{dv(t)}{dt}$. Sol: Taking the $LT$ gives

$$F(s) = s M V(s)$$

thus

$$Z_m(s) \equiv \frac{F(s)}{V(s)} = s M,$$

analogous to an electrical inductor.

4. Find the total impedance seen by the net force $f(t)$ applied to the mass $M$. Sol: Summing the forces must equal the applied force, Eq. 1.72, p. 94.

5. Take the Laplace transform ($LT$) of Eq. 1.72 (p. 94), and evaluate the total impedance $Z(s)$ of the mechanical circuit. Sol: From the properties of the $LT$ that $\frac{dx}{dt} \leftrightarrow sX(s)$, we find

$$M s^2 X(s) + R s X(s) + K X(s) = F(s).$$

In terms of velocity this is $(M s + R + K / s) V(s) = F(s)$. Thus the circuit impedance is

$$Z(s) = \frac{F}{V} = \frac{K + R s + M s^2}{s}.$$

6. What are $N(s)$ and $D(s)$ (e.g. Eq. 1.125, p. 135)? Sol: $D(s) = s$ and $N(s) = K + R s + M s^2$.

7. Assume that $M = R = K = 1$. Find the residues (e.g. Eq. 1.125, p. 135) of the admittance $Y(s) = 1 / Z(s)$, in terms of the roots $s_{\pm}$. Check your answer with the Matlab/Octave command `residue`. Sol: First find the roots of the numerator of $Z(s)$ (the denominator of $Y(s)$):

$$s_{\pm}^2 + s_{\pm} + 1 = (s_{\pm} + 1/2)^2 + 3/4 = 0,$$

which is

$$s_{\pm} = \frac{-1 \pm j\sqrt{3}}{2}.$$

Next form the partial fraction expansion of the admittance

$$Y(s) = \frac{s}{1 + s + s^2} = c_0 + \frac{c_+}{s - s_+} + \frac{c_-}{s - s_-} = \frac{s(c_+ + c_-) - (c_+ s_- + c_- s_+)}{1 + s + s^2}.$$

Comparing the two sides requires that $c_0 = 0$. We also have two equations for the residues $c_+ + c_- = 1$ and $c_+ s_- + c_- s_+ = 0$. The best way to solve this is to set up a matrix relation and take the inverse

$$\begin{bmatrix} 1 & 1 \\ s_- & s_+ \end{bmatrix} \begin{bmatrix} c_+ \\ c_- \end{bmatrix} = \begin{bmatrix} 1 \\ 0 \end{bmatrix}$$

thus:

$$\begin{bmatrix} c_+ \\ c_- \end{bmatrix} = \frac{1}{s_+ - s_-} \begin{bmatrix} s_+ & -1 \\ -s_- & 1 \end{bmatrix} \begin{bmatrix} 1 \\ 0 \end{bmatrix}.$$
which gives \( c_\pm = \pm \frac{s_\pm}{s_+ - s_-} \) The denominator is \( s_+ - s_- = j\sqrt{3} \) and the numerator is \( \pm 1 + j\sqrt{3} \). Thus 
\[
c_\pm = \pm \frac{s_\pm}{s_+ - s_-} = \frac{1}{2} \left( 1 \pm \frac{j}{\sqrt{3}} \right).
\]

As always, finding the coefficients is always the most difficult part. Using 2x2 matrix algebra can really simplify up the process, and will more likely give the correct answer.

8. By applying the CRT, find the inverse Laplace transform \((\mathcal{L}^{-1})\). Use the residue form of the expression that you derived in the previous exercise. Sol:

\[
z(t) = \frac{1}{2\pi j} \oint_{\mathcal{C}} Z(s)e^{st}ds.
\]

were \( \mathcal{C} \) is the Laplace contour which encloses the entire left-half \( s \) plane. Applying the CRT

\[
z(t) = c_+e^{s_+t} + c_-e^{s_-t}.
\]

where \( s_\pm = -1/2 \pm j\sqrt{3}/2 \) and \( c_\pm = 1/2 \pm j/(2\sqrt{3}) \).

### 1.5.7 Lec 37 The 2D Webster horn equation

Newton’s conservation of momentum Law (Eq. 1.114, p. 131)), along with conservation of mass (Eq. 1.115, p. 131), are modern versions of Newton’s starting point for calculating the speed of sound. Following the simplification of averaging the normal component of the particle velocity over the iso-pressure wave front (defining the volume velocity), the two equations may be written as a 2x2 matrix (Ramo et al., 1965) in terms of acoustic variables average pressure \( P(r,\omega) \) and volume velocity \( V(r,\omega) \), sometimes referred to as conjugate variables

\[
\frac{d}{dr} \begin{bmatrix} P(r,\omega) \\ V(r,\omega) \end{bmatrix} = - \begin{bmatrix} 0 & \mathcal{Z}(s,r) \\ \mathcal{Y}(s,r) & 0 \end{bmatrix} \begin{bmatrix} P(r,\omega) \\ V(r,\omega) \end{bmatrix}.
\]

(1.126)

This key transformation, from particle to volume velocity, is discussed in Section 5.1.5 (Eqs. 5.15,5.16, p. 218). The Fourier-transform pair of the average pressure and volume velocity are denoted \( g(r,t) \leftrightarrow P(r,\omega) \) and \( \nu(r,t) \leftrightarrow V(r,\omega) \). The complex Laplace frequency \( s = \sigma + j\omega \) is required when defining the per-unit-length impedance

\[
\mathcal{Z}(r,s) \equiv s \frac{\rho_o}{A(r)} = sM(r)
\]

(1.127)

and per-unit-length admittance

\[
\mathcal{Y}(s,r) \equiv s \frac{A(r)}{\eta_o P_o} = sC(r),
\]

(1.128)

---


93 Notation: Lower case variables (i.e., \( g(r,t),\nu(r,t) \)) denote time-domain functions while upper case letters (i.e., \( P(r,\omega),V(r,\omega),\mathcal{Z}(s,r),\mathcal{Y}(s,r) \)) indicate frequency domain Fourier (\( \omega \)) or Laplace (\( s \)) transforms, the latter being analytic in \( s = \sigma + j\omega \) for \( \sigma > 0 \). An exception to this notional rule is \( \kappa(s) \).
1.5. STREAM 3B: VECTOR CALCULUS (10 LECTURES)

to indicate that these functions are causal, and except at their poles, analytic in $s$. Here $M(r) = \rho_o/A(r)$ is the horn’s per-unit-length mass, $C(r) = A(r)/\eta_o P_o$ per-unit-length compliance, $\rho_o$ is the density of air (i.e., 1.2 [kgm/m$^3$]), $P_o$ is the atmospheric pressure (i.e., $10^5$ [N/m$^2$]), $\eta_o = c_o/c_v \approx 1.4$ (air). It is helpful to express the bulk properties of the waves in terms of $Z$ and $Y$, which is provided next. Equation 1.113 gives the full pressure wave equation when Eqs. 1.127 and 1.128 are reduced to a single equation for the pressure.

**Exercise:** Show that Eq. 1.128 is the same as the 2d order equation in the pressure results in the traditional Webster Horn equation

$$\nabla^2 P(r,\omega) \equiv \frac{1}{A(r)} \frac{\partial}{\partial r} A(r) \frac{\partial}{\partial r} P(r,\omega) = \frac{s^2}{c_0^2} P(r,\omega).$$

(1.129)

Namely the Laplacian in the range variable $r$ depends on the area function $A(r)$, which in turn depends on the geometry of the horn. For example, for a uniform horn, $A(r) = A_o$, and for a conical horn, $A(r) = A_o r^2$. These cases are specifically dealt with in the next section.

Two fundamental parameters fall out of this formulation, the speed of sound and the characteristic admittance and impedance.

**Bulk acoustic parameters:** The bulk acoustic parameters are the free-field speed of sound

$$c_o = \sqrt{\frac{\eta_o P_o}{\rho_o}} = \sqrt{\frac{\text{stiffness}}{\text{mass}}},$$

(1.130)

the specific acoustic resistance is

$$\rho_o c_o = \sqrt{\rho_o \eta_o P_o} = \sqrt{\text{stiffness} \cdot \text{mass}}.$$

(1.131)

**Characteristic admittance $Y(r)$ and impedance $Z(r)$:** An important definition is each primitive wave’s characteristic admittance $Y(r)$, defined as the square root of the ratio of $Y$ over $Z$ (Eq. 2.8, p. 169 and Fig. 2.2, p. 169),

$$Y^{\pm}(r) \equiv \frac{Y^{\pm}}{Z^{\pm}} = \sqrt{\frac{Y(r,s)}{Z(r,s)}} = \sqrt{\frac{sA(r) A(r)}{\eta_o P_o} s\rho_o} = \frac{A(r)}{\rho_o c} > 0,$$

(1.132)

which depends specifically on $A(r)$, but not on frequency $s$ (Campbell, 1903, 1910, 1922). Based on physical requirements that the admittance must be positive, only the positive square root is allowed. The characteristic impedance $Z(r) = 1/Y(r)$.

Since the horn (Eq. 1.113, p. 130) is loss less, $Y^{\pm}(r)$ must be real (and positive). If losses are introduced, $\kappa(s)$ and $Y(r,s)$ can become complex analytic functions of the Laplace frequency $s$ (Kirchhoff, 1974; Mason, 1928; Ramo et al., 1965; Pierce, 1981, p. 532-4). This more difficult case is addressed in Section 5 (p. 215).

---

94 A function $F(s)$ is said to be complex analytic in $s = \sigma + \omega j$ at point $a$ if it may be represented by a convergent Taylor series in $s$ the neighborhood of $s = a$. That is if $F(s) = \sum_{n=0}^{\infty} f^{(n)}(a)(s-a)^n$, where $f^{(n)}(a) = \frac{d^n F(s)}{ds^n} \bigg|_{s=a}$. 

See previous page
Propagation function $\kappa(s)$: The primitive solutions of the horn equation depend on the complex wave propagation function \( \kappa(s) \), defined as the square root of the product of \( \mathcal{Z} \) and \( \mathcal{Y} \) (Eq. 2.7, p. 169 and Eq. 5.35, p. 227):

\[
\kappa(s) \equiv \sqrt{\mathcal{Z}(r,s)\mathcal{Y}(r,s)} = \sqrt{\frac{s \rho_o A(r)}{\eta_o P_o}} = \frac{s}{c_o}.
\]

Note that since the area \( A(r) \) cancels in this expression, \( c_o \) is independent of the range \( r \). While horns are generally dispersive, namely the solution can have a group velocity (and thus a group delay) that depends on frequency, the wave-front speed is always constant, defined by the free-field sound velocity \( c_o \). These relations between the phase, group and wave-front velocities (delay) were first demonstrated between 1907–1914 by Brillouin (1960, English version), while working with Arnold Sommerfeld (Pleshko and Placz, 1969).

Horn input admittance: A horn’s acoustic input admittance \( Y_{in}(r,s) \) is the admittance looking into a horn (Fig. 1.28, p. 131), at any point \( r \). It is defined as the ratio of the volume velocity over the average pressure

\[
Y_{in}(r,s) \equiv \frac{\mathcal{V}(r,s)}{\mathcal{P}(r,s)},
\]

where the direction of flow \( \mathcal{V}(r,s) \) is into the admittance. \( Y_{in}(r,s) \) depends on the entire horn, and any terminating admittance, in the case of a finite length horn.

The radiation admittance \( Y_{rad}(s) \) is defined for an infinite horn, when there is no terminating load, as

\[
Y_{rad}(s) = \lim_{r \to \infty} Y_{in}(r,s).
\]

This may be viewed as the horn’s termination at \( r = \infty \).

It is a physical requirement that the real part of every physical admittance (impedance) be positive (non-negative). To guarantee this physical requirement, care must be taken when fixing the flow direction. Thus the velocity (or current) for any admittance must be defined as directed into the admittance, or else the real part will automatically be negative.

From Eq. 1.126 (p. 138) (top equation), the volume velocity is

\[
\mathcal{V}(r,\omega) = -\frac{1}{2\mathcal{Z}(r,s)} \frac{d}{dr} \mathcal{P}(r,\omega).
\]

Note that \( \mathcal{V} \) is always defined looking into the horn, to guarantee that the resistance must be positive, (i.e., \( \Re Y_{in}(s) \geq 0 \)), as required by the physics (Postulate P3, Section 3.5.1, p. 95).

1.5.8 General 1D solutions of the horn equation

In the mathematical literature, equations of this form are known as Sturm-Liouville problems, the solution of which may be obtained using integration by parts, following a multiplication by an “integration factor” \( \sigma(r) \):

\[
\frac{1}{\sigma(r)} \partial_r \left[ \sigma(r) \mathcal{P}_r \right] = \mathcal{P}_{rr} + \frac{\sigma_r(r)}{\sigma(r)} \mathcal{P}_r.
\]
1.5. STREAM 3B: VECTOR CALCULUS (10 LECTURES)

Exercise: Write out the above using standard notation (i.e., use the chain rule). Sol:

\[
\frac{1}{\sigma(r)} \frac{\partial}{\partial r} \left\{ \sigma(r) \frac{\partial}{\partial r} P(r, s) \right\} \equiv \frac{\partial^2}{\partial r^2} P(r, s) + \left\{ \frac{1}{\sigma(r)} \frac{\partial \sigma(r)}{\partial r} \right\} \frac{\partial}{\partial r} P(r, s).
\]

By a direct comparison of the Eq. 1.137 and Eq. 5.1, we see that \( \sigma_r/\sigma \equiv -Z_r/Z \). From the definition of \( Z(r, s) = \frac{s\rho_o}{A(r)} \) (Eq. 1.127) it follows that

\[
\sigma(r) \equiv A(r). \tag{1.138}
\]

Thus one may conclude that the integration factor is physically the area function. This explains why the integration factor in the Sturm-Liouville theory must be strictly positive.\(^{96}\) Thus the Webster horn equation Eq. 1.113 (Morse, 1948, p. 269) directly follows from Eq. 1.113, due to Eq. 1.138.

In summary, if one starts with the 3D wave equation, and by the application of Gauss’ law, transforms it into a “1-dimensional” (i.e., single range variable) equivalent horn equation, then a Sturm-Liouville equation (a generalized scalar differential equation) results. This may be integrated by the application of the area function as the integration factor resulting in the Webster horn equation

\[
\frac{1}{A(r)} \frac{\partial}{\partial r} \left( A(r) \frac{\partial}{\partial r} P(r, s) \right) = \frac{s^2}{c^2} P(r, s), \leftrightarrow \frac{1}{c^2} \frac{\partial^2}{\partial t^2} P(r, s),
\]

Thus it is convenient to normalize \( P_\pm(r_o, s) = 1 \). As before, separation of time and place variables explains this relationship.

1.5.9 Lec 38 Three examples of horns

The uniform horn

The 1D wave equation \([A(r) = A_o]\)

\[
\frac{d^2}{dr^2} P = \frac{s^2}{c_o^2} P.
\]

Solutions: The two primitive solutions of this equation are the two d’Alembert waves (Eq. 1.84, p. 103)

\[
g(x, t) = g^+(t - x/c) + g^-(t + x/c) \leftrightarrow P^+(r_o, s)e^{-\kappa x} + P^-(r_o, s)e^{\kappa x},
\]

where \( P^\pm(r_o, s) \) are Laplace transform pairs representing the causal forward and retrograde traveling wave pressure amplitudes. Thus the normalized primary solutions are

\[
g^+(x, t) = \delta(t - x/c_o) \leftrightarrow P^+(r_o, s)e^{-\kappa x}
\]

and

\[
g^-(x, t) = \delta(t + x/c_o) \leftrightarrow P^-(r_o, s)e^{\kappa x}.
\]

Thus it is convenient to normalize \( P^\pm(r_o, s) = 1 \). As before, separation of time and place variables explains this relationship.
CHAPTER 1. INTRODUCTION

**Characteristic admittance $Y_0$:** Since $\mathcal{V}^\pm = -\partial_x \ln \mathcal{P}^\pm$ we find

$$Y^\pm \equiv \frac{\mathcal{Y}^\pm}{\mathcal{P}^\pm} = A_o/\rho_oc \quad (1.140)$$

(Eq. 5.29, p. 223), independent of direction. The signs must be “physically chosen,” with the velocity $\mathcal{V}^\pm$ into the port, to assure that $\mathcal{Y} > 0$, for both waves.

**Applying the boundary conditions:** The general solution is compactly formulated as an ABCD matrix (i.e., Section 1.3.7, p. 79), starting from

$$\begin{bmatrix} \mathcal{P}(x) \\ \mathcal{V}(x) \end{bmatrix} = \begin{bmatrix} e^{-\kappa x} & e^{\kappa x} \\ \mathcal{Y}e^{-\kappa x} & -\mathcal{Y}e^{\kappa x} \end{bmatrix} \begin{bmatrix} \alpha \\ \beta \end{bmatrix},$$

where $\alpha, \beta$ are the relative weights on the two unknown primitive solutions that must be determined by the boundary conditions at $x = 0, L$. Note that the determinant $\Delta_L = -2\mathcal{Y}$ is independent of $x$.

Solving Eq. 1.141 for $\alpha$ and $\beta$ (with $Z = 1/\mathcal{Y}$)

$$\begin{bmatrix} \alpha \\ \beta \end{bmatrix} = \frac{-1}{2\mathcal{Y}} \begin{bmatrix} -\mathcal{Y}e^{-\kappa x} & -e^{\kappa x} \\ -\mathcal{Y}e^{-\kappa x} & e^{\kappa x} \end{bmatrix}_L \begin{bmatrix} \mathcal{P}_L \\ -\mathcal{V}_L \end{bmatrix} = \frac{1}{2} \begin{bmatrix} e^{\kappa L} & Z e^{\kappa L} \\ 1 & -Z \end{bmatrix} \begin{bmatrix} \mathcal{P}_L \\ -\mathcal{V}_L \end{bmatrix}.$$  

We may uniquely determine these two weights given the pressure and velocity at the boundary $x = L$, which is typically determined by the impedance (the load) at $x = L$. Note that the sign of $\mathcal{V}_L$ has been changed, as required by the ABCD method.

Once the weights have been determined, they may be substituted back into Eq. 1.141, to determine the pressure and velocity amplitudes at every point $0 \leq x \leq L$. Applying the boundary condition at the input, we evaluate Eq. 1.142 at $x = 0$ to obtain the final ABCD matrix

$$\begin{bmatrix} \mathcal{P}_o \\ \mathcal{V}_o \end{bmatrix} = \frac{1}{2} \begin{bmatrix} 1 & e^{-\kappa L} \\ \mathcal{Y} & -\mathcal{Y}e^{-\kappa L} \end{bmatrix} \begin{bmatrix} e^{\kappa L} & Z e^{\kappa L} \\ 1 & -Z \end{bmatrix} \begin{bmatrix} \mathcal{P}_L \\ -\mathcal{V}_L \end{bmatrix}.$$  

Multiplying these out gives the final transmission matrix

$$\begin{bmatrix} \mathcal{P}_o \\ \mathcal{V}_o \end{bmatrix} = \begin{bmatrix} \cosh(\kappa L) & Z \sinh(\kappa L) \\ \mathcal{Y} \sinh(\kappa L) & \cosh(\kappa L) \end{bmatrix} \begin{bmatrix} \mathcal{P}_L \\ -\mathcal{V}_L \end{bmatrix},$$  

with $\kappa = s/c$, $\mathcal{Y} = 1/Z = A_o/\rho_oc$ (Pipes, 1958). The two velocities are defined into their respective ports.

**Exercise:** Show that the above equation follows from Eq. 1.143. Sol: This follows from the definitions of $\cosh(x) = \frac{e^x + e^{-x}}{2}$ and $\sinh(x) = \frac{e^x - e^{-x}}{2}$.

**Impedance matrix:** Expressed Eq. 1.144 as an impedance matrix gives (algebra required)

$$\begin{bmatrix} \mathcal{P}_o \\ \mathcal{P}_L \end{bmatrix} = \frac{Z}{\sinh(\kappa L)} \begin{bmatrix} \cosh(\kappa L) & 1 \\ 1 & \cosh(\kappa L) \end{bmatrix} \begin{bmatrix} \mathcal{V}_o \\ \mathcal{V}_L \end{bmatrix},$$

96We are not aware of this observation in the existing literature. It seems likely that Morse (1948) must have been aware of this connection, however to our knowledge he did explicitly say so.
The input admittance: Given the input admittance of the horn, it is possible to determine if it is uniform, without further analysis. Namely if the horn is uniform and infinite in length, the input impedance at $x = 0$ is

$$Y_{in}(0, s) \equiv \frac{V(0, \omega)}{P(0, \omega)} = Y,$$

since $\alpha = 1$ and $\beta = 0$. That is for an infinite uniform horn, there are no reflections.

When the horn is terminated with a fixed impedance $Z_L$ at $x = L$, one may substitute pressure and velocity measurements into Eq. 1.142 to find $\alpha$ and $\beta$, and given these, one may calculate the reflectance at $x = L$ (see Eq. 5.28, 223)

$$\Gamma_L(s) \equiv \left. \frac{P_-}{P_+} \right|_{x=L} = \frac{\beta}{\alpha} = \frac{P(L, \omega) - ZV(L, \omega)}{P(L, \omega) + ZV(L, \omega)} = \frac{Z_L - Z}{Z_L + Z}$$

given sufficiently accurate measurements of the throat pressure $P(0, \omega)$, velocity $V(0, \omega)$, and the characteristic impedance of the input $Z = \rho_o c/A(0)$.

3D Conical Horn

For each horn we must find the natural normalization from the range variable to the normalize range variable $x$. For the conical horn the radius is proportional to the range variable $r$ thus

$$A(r) = 4\pi \sin^2 (\Theta/2) \, r^2 \, [m^2].$$

The angle $\Theta$ is a measure of the solid (cone) angle. When $\Theta = \pi$ we have the case of the entire sphere so the solid angle is $4\pi$ [steradians] and the area is $4\pi r^2$. The formula for the area may be simplified by defining $A_\theta \equiv 4\pi \sin^2 (\Theta/2) \, r_o^2 \, [m^2]$, resulting in the more convenient relation

$$A(r) = A_\theta \left( \frac{r}{r_o} \right)^2 \, [m^2].$$

This is a bit tricky because $A_\theta$ is not a constant since it depends on the place where the area was normalized, in this case $r_o$.

Using the conical horn area $A(r) \propto r^2$ in Eq. 1.113, p. 130 [or Eq. 1.126 (p. 138)] results in the spherical wave equation (Appendix 5.2 p. 220)

$$P_{rr}(r, \omega) + \frac{2}{r} P_r(r, \omega) = \kappa^2 P(r, \omega)$$

(1.145)

(one must remember the steradian scale factor $A_\theta$). Here $F(r) = \partial_r \ln A(r) = \frac{2}{r}$ (see Table 5.1, p. 225).

Exponential horn: The case of the exponential horn

$$P^\pm(r) = e^{-mr} e^{\mp j \sqrt{\omega^2 - \omega_c^2} \, r/c},$$

(1.146)

which is of special interest because the radiation impedance is purely reactive below the horn’s cutoff frequency ($\omega < \omega_c = mc_o$) where the reflection coefficient has magnitude 1 (no energy can radiate from an open horn below this frequency.)
Figure 1.29: Throat acoustical resistance $r_A$ and acoustical reactance $x_A$, frequency characteristics of infinite parabolic, conical, exponential, hyperbolic and cylindrical horns having a throat area of 1 square centimeter. Note how the “critical” frequency (defined here as the frequency where the reactive and real parts of the radiation impedance are equal) of the horn reduces dramatically with the type of horn. For the uniform horn, the reactive component is zero, so there is no cutoff frequency. For the parabolic horn (1), the cutoff is around 3 kHz. For the conical horn (2), the cutoff is at 0.6 [kHz]. For the exponential horn the critical frequency is around 0.18 [kHz], which is 16 times smaller than for the parabolic horn. For each horn the cross-sectional area of the parabolic, conical exponential and hyperbolic horns is defined as 100 square centimeters at a distance of 100 centimeters from the throat (Olson, 1947, p. 101).

1.5.10 Lec 39: Integral forms of $\nabla()$, $\nabla \cdot ()$ and $\nabla \times ()$

There are two definitions for each operator, the point or differential form, and the integral limit form. The integral form gives an intuitive view of what the operator does, and in the limit, converges to the differential form. These two definitions are discussed next.

The divergence and the curl each have fundamental theorems: Gauss’ Law (divergence theorem) and Stokes’ Law (curl theorem). Without the use of these very fundamental vector calculus theorems, Maxwell’s equations cannot be understood. The history of these important theorems is discussed in the caption of Fig. 1.12, p. 54.

Gradient: $\mathbf{E} = -\nabla \phi(x, y, z)$ [V/m]

As briefly summarized on page 121, the differential definition of the gradient maps $\mathbb{R}^1 \ni \nabla \mapsto \mathbb{R}^3$. For example, the electric field strength is the gradient of the voltage

$$\mathbf{E}(\mathbf{x}) = -[\partial_z, \partial_y, \partial_x] \phi(\mathbf{x}) = -\left[\frac{\partial \phi}{\partial x}, \frac{\partial \phi}{\partial y}, \frac{\partial \phi}{\partial z}\right]^T (\mathbf{x}).$$

The negative sign is optional.

The gradient is the slope of the tangent plane of the potential $\phi(\mathbf{x})$ at $\mathbf{x}$ pointing in the direction of the maximum slope. The gradient (i.e., $\mathbf{E}(\mathbf{x})$) is $\perp$ of the constant potential contours of $\phi(\mathbf{x})$ (the iso-potential lines), since derivatives along any iso-contours must be zero.

As a specific example, consider the paraboloid $z = 1 - (x^2 + y^2)$ as the potential, with iso-potential circles of constant $z$ that have radius of zero at $z = 1$, and unit radius at $z = 0$. The negative gradient

$$\mathbf{E}(\mathbf{x}) = -\nabla z(x, y) = 2(x \hat{x} + y \hat{y} + 0 \hat{z})$$

is $\perp$ to the circles of constant radius (constant $z$), thus points in the direction of the radius.

If one were free-fall skiing this surface, they would be the first one down the hill. Normally skiers try to stay close to the iso-clines (not in the direction of the gradient), so they can stay in control. If you ski an iso-cline, you must walk, since there is no pull due to gravity.
The integral definition of the gradient: The gradient may be defined in terms of the integral

\[ \nabla \varphi(x, y, z) \equiv \lim_{|S| \to 0} \left\{ \frac{\int_S \varphi(x, y, z) \hat{n} \, d|S|}{|S|} \right\}, \tag{1.147} \]

over a closed surface \( S \), having area \(|S|\) and volume \(||S||\), centered at \((x, y, z)\) (Greenberg, 1988, p. 773).\(^{97}\) Here \( \hat{n} \) is the unit vector perpendicular to the surface \( S \)

\[ \hat{n} = \frac{\nabla \varphi}{||\nabla \varphi||}. \]

The dimensions of Eq. eq:GradInt are in the units of the potential times the area, divided by the volume, as needed for a gradient (e.g., [Volts/m]).

The natural way to define the surface and volume is to place the surface on the iso-potential surfaces, forming either a cube or pill-box shaped volume. As the volume goes to zero, so must the area. One must avoid \textit{irregular volumes} such that the area is finite as the volume goes to zero (Greenberg, 1988, footnote p. 762).

A well known example is the potential

\[ \varphi(x, y, z) = \frac{Q \varepsilon_0}{\sqrt{x^2 + y^2 + z^2}} = \frac{Q}{\varepsilon_0 R} \quad \text{[Volts]} \]

around a point charge \( Q \) [SI Units of Coulombs]. The constant \( \varepsilon_0 \) is the \textit{permittivity} [Farad/m\(^2\)].

\textbf{How does this work?} To better understand what Eq. 1.147 means, consider a three-dimensional Taylor series expansion of the potential in \( \mathbf{x} \) about the limit point \( \mathbf{x}_o \) To compute the \textit{higher order terms} (HOT) one needs the \textit{Hessian} matrix\(^{98}\)

\[ \phi(x) \approx \phi(x_o) + \nabla \phi(x) \cdot (x - x_o) + \text{HOT}. \]

We could define the gradient using this relationship as

\[ \nabla \phi(x_o) = \lim_{x \to x_o} \frac{\phi(x) - \phi(x_o)}{x - x_o}. \]

For this definition to apply, \( x \) must approach \( x_o \) along \( \hat{n} \).

The natural form for the surface \(|S|\) is to lie along the iso-potential surfaces as much as possible, so that the integral is a constant (the potential) times the area. The remainder of the surface must be perpendicular to these iso-potential surfaces, in the direction of the gradient, or maximum change of the potential. The secret to the integral definition is in taking the limit. As the volume \(||S||\) shrinks to zero, the HOT terms are small, and the integral reduces to the first order term in the Taylor expansion, since the constant term integrates to zero. Such a construction is used in the proof of the Webster Horn equation (1.5.7, p. 138; Fig. 5.2, p. 220).

\(^{97}\)See further discussions on pages Greenberg (1988, pp. 778, 791, 809)

\(^{98}\)\(H_{ij} = \frac{\partial^2 \phi}{\partial x_i \partial x_j}, \) which will exist if the potential is analytic in \( \mathbf{x} \) at \( \mathbf{x}_o \).
Figure 1.30: Derivation of Gauss’s Law: The divergence of the velocity $\nabla \cdot \mathbf{u}$, within $\delta x$, shown as the filled shaded region, is integrated over the closed volume. Then the divergence theorem is applied, transforming the integral to a surface integral normal to the surface, resulting in the difference of the two volume velocities $\delta \nu = \nu(x + \delta x) - \nu(x) = [\mathbf{u}(x + \delta x) \cdot \mathbf{A}(x + \delta x) - \mathbf{u}(x) \cdot \mathbf{A}(x)]$, as the limit of $\delta x \rightarrow 0$. Because the ends are in the direction of $\mathbf{u}$, the contribution to the volume velocity from the ends is zero.

Divergence: $\nabla \cdot \mathbf{D} = \rho \text{[Col/m}^3\text{]}$

As briefly summarized on page 122, the differential definition of the gradient which maps $\mathbb{R}^3 \mapsto \nabla \mathbb{R}^1$ is

$$\nabla \cdot \mathbf{D} \equiv [\partial_x, \partial_y, \partial_z] \cdot \mathbf{D} = \left[ \frac{\partial D_x}{\partial x} + \frac{\partial D_y}{\partial y} + \frac{\partial D_z}{\partial z} \right] = \rho(x, y, z)$$

The divergence is a direct measure of the flux (flow) of vector field it operates on ($\mathbf{D}$), coming from $x$. A vector field is said to be incompressible if the divergence of that field is zero. It is therefore compressible when the divergence is non-zero, Compared to air, water is considered to be incompressible. However at very low frequencies, air can also be considered as incompressible. Thus the definition of compressible depends on the wavelength in the medium, so the terms must be used with some awareness of the circumstances.

If we restrict ourselves to the electrical case, then at this point it is helpful to think of the physics of the electron, a negatively charged particle that is a single point in space. Of course this it is nonsense to consider an electron to be a point, but given that it is so small (it is so small it is difficult to estimate its size). This size is called the Lorentz radius, and is estimated to be $2.810^{-15}$ [m]. One could summarize the Lorentz radius as follows: Here lie many unsolved problems in physics. More specifically, at dimensions of the Lorentz radius, what exactly is the structure of the electron?

Ignoring the difficulties, if one integrates the charge density of the electron over the Lorentz radius and places the total charge at a single point, then one may make a grossly over simplified model of the electron. For example the Electric Displacement (flux density) around a point charge ($\mathbf{D} = \varepsilon_o \mathbf{E}$)

$$\mathbf{D} = -\varepsilon_o \nabla \phi(R) = -Q \nabla \left\{ \frac{1}{R} \right\} = -Q \delta(R).$$

This is a formula taught in many classic texts, but one should remember how crude a model of an electron it is. But it does describe the electric flux in an easily remembered form. However, computationally, it is less nice, due to the delta function. A limitation of this model is that the electron has a magnetic dipole moment (aka, spin), which a simple point charge does not capture. When placed in a magnetic field, due to the dipole, the electron will align itself with the field.
Divergence and Gauss’ Law

Like the gradient, the divergence of a vector field may be defined as the surface integral of a *compressible* vector field, as a limit as the volume enclosed by the surface goes to zero. As for the case of the gradient, for this definition to make sense, the surface \( S \) must be a closed, defining volume \( V \). The difference is that the surface integral is over the normal component of the a vector field being operated on. Specifically (Greenberg, 1988, p. 762-763)

\[
\nabla \cdot D = \lim_{|S| \to 0} \left\{ \int_S \frac{D \cdot \hat{n} \, d|S|}{|S|} \right\} = \rho(x, y, z).
\]

As with the case of the gradient we have defined the surface as \( S \), its area as \( |S| \) and the volume within as \( ||S|| \). As the area \( |S| \) goes to zero, so does the volume \( ||S|| \). This is a necessary condition for the integral to converge to the divergence.

Here \( \hat{n} \) is a unit vector normal to the surface \( S \). The limit as the volume goes to zero, defines the total flux across the surface. Thus the surface integral is a measure of the total flux \( \perp \) to the surface. It is helpful to compare this formula with that for the gradient Eq. 1.147.

**Gauss’ Law:** The above definitions resulted in a major breakthrough in vector calculus, and the first\(^{99} \) fundamental theorem of vector calculus, *Gauss’ Law\(^{100} \)*

*The surface integral over the normal component of the flux (i.e., the total flux) is equal to the volume integral over the divergence of the flux.*

For the electrical case this is equivalent to the observation that the total flux across the surface is equal to the net charge enclosed by the surface. Since the volume integral over charge density \( \rho(x, y, z) \) is total charge enclosed \( Q_{enc} \) [Col]

\[
Q_{enc} = \iiint_{||S||} \nabla \cdot D \, ||dS|| = \iiint_{|S|} D \cdot \hat{n} \, d|S|.
\]

(1.149)

It trivially follows that when \( Q_{enc} = 0 \). That is when the surface integral over the normal component of \( D(x) \), and the volume integral the charge density \( \nabla \cdot D = \rho(x) \), are both zero.

Taking the derivative with respect to time Eq. 1.149 becomes, the current flowing out of the surface (i.e., the component normal to the surface)

\[
I = \iint_{|S|} D \cdot \hat{n} \, d|S| = \dot{Q}_{enc} = \iiint_{||S||} \dot{\rho}_{enc} \, d||S||.
\]

(1.150)

Paraphrasing Feynman (1970c, p. 13-2)

*The current leaving the closed surface \(|S|\) equals the rate of the charge leaving that volume \(||S||\), defined by that surface.*

Of course the surface must be closed to define the volume, a necessary condition for Gauss’s law. This reduces to a common sense summary that can be grasped intuitively, an example of the beauty in Feynman’s understanding.

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\(^{99}\)As best I know, the three fundamental theorems of vector calculus, Gauss’ Law, the Stokes’ Law and Helmholtz’s theorem (aka Vector decomposition theorem) are not ordered.

\(^{100}\)aka, Gauss’s Law
CHAPTER 1. INTRODUCTION

\[ \nabla \times \mathbf{H} \equiv \lim_{|S| \to 0} \left\{ \int_{S} \hat{n} \times \mathbf{H} \, d|S| \right\} \]

\[ I_{\text{enc}} = \int\int_{S} (\nabla \times \mathbf{H}) \cdot \hat{n} \, dA = \oint_{B} \mathbf{H} \cdot dl \quad [\text{Amps}] \]

**Figure 1.31:** The integral definition of the curl is similar to that of the divergence (Greenberg, 1988, p. 774), except that 1) the integration is over the plane tangent to the surface, defined as the cross product of \( \hat{n} \), the normal to the surface, crossed with a current density of \( \hat{n} \times \mathbf{H} \) [Amps/m\(^2\)], and 2) the surface is open, giving it a boundary \( B \), along the open edge. As with the divergence, which lead to Gauss’ Law, this definition leads to a second fundamental theorem of vector calculus: Stokes’ Law (aka the Curl theorem). There has been a serious problem in the past, since an open area does not have any enclosed volume \( ||dV|| \) (Greenberg, 1988, p. 814). Greenberg (1988, p. 824) try’s to resolve this, but perhaps has not resolved it. To solve it here we take the limit over the open surface \( ||dS|| \).

**Integral definition of the Curl:** \( \nabla \times \mathbf{H} = C \) [Amps]

As briefly summarized on page 122 (p. 122), the differential definition of the Curl maps \( \mathbb{R}^3 \mapsto \nabla \times \mathbb{R}^3 \).

For example, the curl of the magnetic field strength \( \mathbf{H}(x) \) is equal to the total current \( C \)

\[ \nabla \times \mathbf{H} \equiv \begin{vmatrix} \hat{x} & \hat{y} & \hat{z} \\ \partial_x & \partial_y & \partial_z \\ H_x & H_y & H_z \end{vmatrix} = C. \]

As we shall see in Sect. 1.5.12 (p. 154), the curl and the divergence are both key when writing out Maxwell’s equation four equations. Without a full understanding of these two differential operators (\( \nabla \cdot \mathbf{(} \mathbf{)} \), \( \nabla \times \mathbf{(} \mathbf{)} \)), there is no hope of understanding Maxwell’s basic result, typically viewed as the most critical elements of mathematical physics, and the starting point for Einstein’s relativity theories. Some will say that quantum mechanics fall outside the realm of MEs, but this at least open to debate, if not hotly debated.

The curl is a measure of the rotation of a vector field. If this were water, it would correspond to the angular momentum of the water, such as water going down the drain, as in a whirlpool, or with wind, a tornado. A spinning top is another excellent example, given a spinning solid body. While a top (aka gyro-scope) will fall over if not spinning, once it is spinning, it can stability stand on its pointed tip. These structure are stable due to conservation of angular momentum: Once something is spinning, it will continue to spin.

By way of example, when \( \mathbf{H} = -y\hat{x} + x\hat{y} + 0\hat{z} \) then \( \nabla \times \mathbf{H} = 2\hat{z} \), thus has a constant rotation; when \( \mathbf{H} = 0\hat{x} + 0\hat{y} + z^2\hat{z} \) then \( \nabla \times \mathbf{H} = 0 \) has a curl of zero, thus has zero rotation thus is said to be irrotational. There are rules that precisely govern when a vector field is rotational versus irrotational, and compressible versus incompressible. These classes are dictated by Helmholtz’s Theorem, the third fundamental theorem of vector calculus (Eq. 1.157, p. 152).

**Curl and Stokes Law:** As in the cases of the gradient and divergence, the curl also may be written in integral form, allowing for the physical interpretation of its meaning.

Surface integral definition of \( \nabla \times \mathbf{H} = C \) where the current \( C \) is \( \perp \) to the rotation plane of \( \mathbf{H} \). As shown in Fig. 1.31, Stokes law states that the open surface integral over the normal component of the curl of the magnetic field strength (\( \hat{n} \nabla \times \mathbf{H} \) [Amps/m\(^2\)]) is equal to the line integral \( \oint_{B} \mathbf{H} \cdot dl \).
along the boundary $\mathcal{B}$. Namely, Stokes’ Law is

$$I_{\text{enc}} = \iint_{\mathcal{S}} (\nabla \times \mathbf{H}) \cdot \hat{n} \, dA = \oint_{\mathcal{B}} \mathbf{H} \cdot d\mathbf{l} \quad \text{[Amps]},$$

(1.151)

namely

*The line integral of $\mathbf{H}$ along the open surface's boundary $\mathcal{B}$ is equal to the total current enclosed $I_{\text{enc}}$ [Amps].*

If Feynman didn’t say this, he could have.

### Table 1.8: Summary of the fundamental theorems of integral calculus, each of which deals with integration. There are at least two main theorems related to scalar calculus, and three more for vector calculus.

<table>
<thead>
<tr>
<th>Name</th>
<th>Mapping</th>
<th>p.</th>
<th>Description</th>
</tr>
</thead>
<tbody>
<tr>
<td>Leibniz (FTC)</td>
<td>$\mathbb{R}^1 \rightarrow \mathbb{R}^2$</td>
<td>100</td>
<td>Area under a real curve.</td>
</tr>
<tr>
<td>FTCC</td>
<td>$\mathbb{R}^1 \rightarrow \mathbb{R}^2$</td>
<td>100</td>
<td>Area under a complex curve.</td>
</tr>
<tr>
<td>Cauchy’s Theorem</td>
<td>$\mathbb{C}^1 \rightarrow \mathbb{C}^2$</td>
<td>112</td>
<td>Close integral over analytic region is zero</td>
</tr>
<tr>
<td>Cauchy’s Integral Formu</td>
<td>$\mathbb{C}^1 \rightarrow \mathbb{C}^2$</td>
<td>112</td>
<td>Residue integration and complex analytic functions.</td>
</tr>
<tr>
<td>Residue Theorem</td>
<td>$\mathbb{C}^1 \rightarrow \mathbb{C}^2$</td>
<td>112</td>
<td>Residue integration and complex analytic functions.</td>
</tr>
</tbody>
</table>

**Summing it up:** As mentioned earlier (Fig. 1.12, p. 54), the history of the discovery and proof of this theorem was both complex and interesting.

Since integration is a linear process (sums of smaller elements), one may tile, or tessellate the surface, breaking it up into a sum over smaller surfaces, and their boundaries, the sum over which is equal to the integral over the original boundary. This is an important concept, which leads to the proof of Stokes’ Law.

The integral formulations of Gauss’ and Stokes’ Laws use $\hat{n} \cdot \mathbf{D}$ and $\mathbf{H} \times \hat{n}$ in the integrands. The key distinction between the two laws naturally follows from the properties of the scalar ($\mathbf{A} \cdot \mathbf{B}$) and vector ($\mathbf{A} \times \mathbf{B}$) products, as discussed in Sect. 1.3.5, p. 71, and detailed in Fig. 1.13, p. 74. To fully appreciate the differences between Gauss’ and Stokes’ Laws, these two types of vector products must be mastered.

Paraphrasing Feynman (1970c, 3-12).

1. $V_2 = V_1 + \int_{1}^{2} \nabla \Phi \cdot dS$
2. $\oint \mathbf{D} \cdot \hat{n} \, d|S| = \oint \nabla \cdot \mathbf{D} \, d|S|$
3. $\oint_{\mathcal{B}} \mathbf{E} \, d\mathbf{l} = \int_{|S|} (\nabla \times \mathbf{E}) \cdot \hat{n} \, d|S|$

The first is the FTC. The second is Gauss’s Law, and the third is Stokes’ theorem.
CHAPTER 1. INTRODUCTION

1.5.11 Lec 40 Second-order operators: Terminology

Besides the above first order vector derivatives, second order combinations exist, the most common being the scalar Laplacian $\nabla \cdot \nabla () = \nabla^2 ()$ (Table 1.7, p. 122; Appendix 5.2, p. 220).

There are other important second-order combinations of $\nabla$, enough that we need a memory aid to remember them. Thus I define mnemonics DoC, DoG, CoG CoC and GoD as follows:

1. **DoG**: Divergence of the Gradient ($\nabla \cdot \nabla = \nabla^2$), i.e., Laplacian
2. **DoC**: Divergence of the Curl ($\nabla \cdot \nabla \times$),
3. **CoG**: Curl of the Gradient ($\nabla \times \nabla$),
4. **CoC**: Curl of the curl ($\nabla \times \nabla \times$), and
5. **GoD**: Gradient of the Divergence ($\nabla^2$), or the vector Laplacian.

Two of these, DoC and CoG, are special because they are always zero,

$$\nabla \times \nabla \phi = 0; \ \nabla \cdot \nabla \times A = 0,$$

a property that makes them useful in proving the Fundamental theorem of vector calculus (Helmholtz’ decomposition) (Eq. 1.157, p. 152). A third key vector identity CoC may be expanded as

$$\nabla \times \nabla \times A = \nabla (\nabla \cdot A) - \nabla^2 A,$$

(1.152)

thus defining the vector Laplacian (GoD) $\nabla^2$.

When using second order differential operators one must be careful with the order of operations, which can be subtle in a few cases. Most of this is common sense. For example, don’t operate on a scalar field with $\nabla \times$, and don’t operate on a vector field with $\nabla$. The vector Laplacian GoD must not be though of as $\nabla (\nabla \cdot A)$, rather it acts as the Laplacian on each vector component $\nabla^2 (A) = \nabla^2 A_x \hat{x} + \nabla^2 A_y \hat{y} + \nabla^2 A_z \hat{z}$.

**Helmholtz’s decomposition**

We may now restate everything defined above in terms of two types of vector fields that decompose every analytic vector field. The *irrotational fields* is define as one that is “curl free.” An *incompressible field* is one that is “diverge free.” According to Helmholtz’s decomposition, every analytic vector field may be decomposed into independent rotational and a compressible components. Another name for *Helmholtz decomposition* is the *Fundamental theorem of vector calculus* (FTVC), Gauss’ and Stokes’ Laws, along with Helmholtz’s decomposition, form the three key fundamental theorems of vector calculus. Images of Helmholtz and Kirchhoff are provided in Fig. 1.32, p. .152.

The term *magnetic solenoidal field* is one that is generated by a solenoidal coil, is an excellent approximation, uniform inside the coil. As a result, the curl of a solenoidal field is, to a good approximation, zero. I recommend you know this term (it is widely used), but suggest the preferred term *irrotational*. Strictly speaking the term “solenoidal field” only applies to a magnetic field produced by a solenoid, thus the term is specific to that case.
The decomposition of differentiable vector fields: This theorem is easily stated (and proved), but less easily appreciated (Heras, 2016). A physical description facilitates: *Every vector field may be split into two independent parts: dilation and rotation.* We have seen this same idea appear in vector algebra, where the vector and cross products of two vectors are perpendicular (Fig. 1.13, p. 74). Also think of linear and angular momentum, which are independent in that they represent different ways of absorbing and delivering kinetic energy, leading to independent degrees of freedom. Thus the idea of the linear and rotational parts being independent, is a common theme, rooted in geometry. In the same sense, a vector field may be split into a dilation and rotational parts, which are independent (but can interact under certain conditions). An object with mass can be moving along a path, and independently be rotating. The two modes of motion define different types of kinetic energy, translational (compressional) and rotational. In some real sense, Helmholtz decomposition quantifies these degrees of freedom, one DoF for translation and three DoFs for rotation. Each eigen-mode of vibration can be viewed as a DoF. The role of the FTVC is especially powerful when applied to Maxwell’s Eqs.

**FTVC, DoC and CoG**

The application of FTVC is not as obvious as it might seem when applied to Maxwell’s equations, due to the power of DoC and Cog. In Section 1.5.1 we defined the electric and magnetic field strengths in terms of the scalar and vector potentials \( \phi(x, t) \) and \( A(x, t) \), repeated here for convenience. The electric field strength is (Eq. 1.101, p. 121)

\[
E = -\nabla \phi - \frac{\partial}{\partial t} A \tag{1.101}
\]

while the magnetic field strength is (Eq. 1.102, p. 121)

\[
B = \nabla \times A \tag{1.102}
\]

(Sommerfeld, 1952, p. 146). Next we derive Maxwell’s equations from these potential representations, by the use of DoC and Cog.

Taking the curl of Eq. 1.101 recovers the equation for Maxwell’s electric intensity

\[
\nabla \times E = -\nabla \times \nabla \phi^{0} + \nabla \times \frac{\partial A}{\partial t} = -\frac{\partial B}{\partial t}. \tag{1.153}
\]

Here we have used the fact that the curl of the gradient is always zero (CoG = 0).

A similar argument may be used for Maxwell’s magnetic potential equation Eq. 1.102. Taking the divergence and using DoC=0 results in another of Maxwell’s equations

\[
\nabla \cdot \nabla \times E^{0} = -\frac{\partial \nabla \cdot B(x, t)}{\partial t} = 0, \tag{1.154}
\]

which says that \( \nabla \cdot B \) is independent of time, namely

\[
\nabla \cdot B(x) = 0. \tag{1.155}
\]

Thus starting from the equation Eq. 1.101, along with DoC=0 and CoG=0, two of Maxwell’s equations Eq. 1.153 and Eq. 1.155 result. It would appear that Eq. 1.101 is the key. This equation defines the magnetic component of the field, expressed in terms of its vector potential, in the same way as Eq. 1.98 describes \( E(x, t) \) in terms of the scalar potential (voltage). Does the same argument apply for Eq. 1.102?
Exercise: Take the divergence of Maxwell’s equation for the magnetic intensity
\[ \nabla \times H(x, t) = J(x, t) + \frac{\partial}{\partial t} D(x, t). \]
and explain what results. Sol: The divergence of the curl is always zero (DoC=0), thus
\[ \nabla \cdot \nabla \times H(x, t) = \nabla \cdot J(x, t) + \frac{\partial}{\partial t} \rho(x, t) = 0, \]
which is conservation of charge.

Helmholtz’s decomposition is expressed as the linear sum of a scalar potential \( \phi(x, y, z) \) (think voltage) and a vector potential (think magnetic vector potential). Specifically
\[ E(x, s) = -\nabla \phi(x, s) + \nabla \times A(x, s), \]
where \( \phi \) is the scalar and \( A \) is the vector potential, as a function of the Laplace frequency \( s \). Of course this decomposition is general (not limited to the electro-magnetic case). It applies to linear fluid vector fields, which includes most liquids and air. When the rotational and dilation become coupled, this relation must break down.\(^{101}\)

Figure 1.32: Left: von Helmholtz portrait taken from the English translation of his 1858 paper “On integrals of the hydrodynamic equations that correspond to Vortex motions” (in German) (Helmholtz, 1978). Right: Gustav Kirchhoff

To show how this relationship splits the vector fields \( E \) into its two parts, we need DoC and CoG, the two key vector identities that are always zero for analytic fields: the curl of the gradient (CoG)
\[ \nabla \times \nabla \phi(x) = 0, \]
and the divergence of the curl\(^{102}\) (DoC)
\[ \nabla \cdot (\nabla \times A) = 0. \]
The above identities are easily verified by working out a few specific examples, based on the definitions of the three operators, gradient, divergence and curl, or in terms of the operator’s integral

\(^{101}\)The nonlinear Navier–Stokes equations may be an example.
\(^{102}\)Helmholtz was the first person to apply mathematics in modeling the eye and the ear (Helmholtz, 1863a).
definitions, defined in Sect. 1.5.10, p. 144. The identities have a physical meaning, as stated above: every vector field may be split into its translational and rotational parts. If \( \mathbf{E} \) is the electric field [V/m], \( \phi \) is the voltage and \( \mathbf{A} \) is the induced rotational part, induced by a current. We shall explore this in our discussion of Maxwell’s equations in Sect. 1.5.12 and Chapter 5.

By applying these two identities to Helmholtz’s decomposition, we can better appreciate the theorems’ significance. It is a form of proof actually, once you have satisfied yourself that the vector identities are true. In fact, one can work backward using a physical argument, that rotational momentum (rotational energy) is independent of the translational momentum. Once these forces are made clear, the meaning of the vector operations all take on a very well defined meaning, and the mathematical constructions, centered around Helmholtz’s theorem, begins to provide some common-sense meaning. One could conclude that the physics is simply related to the geometry via the scalar and vector product.

Specifically, if we take the divergence of Eq. 1.157, and use the DoG

\[
\nabla \cdot \mathbf{E} = \nabla \cdot \{- \nabla \phi + \nabla \times \mathbf{A}\} = \nabla \cdot \nabla \phi = -\nabla^2 \phi,
\]

since the DoG zeros the vector potential \( \mathbf{A}(x, y, z) \). If instead we use the CoG

\[
\nabla \times \mathbf{E} = \nabla \times \{-\nabla \phi + \nabla \times \mathbf{A}\} = \nabla \times \nabla \times \mathbf{A} = \nabla (\nabla \cdot \mathbf{A}) - \nabla^2 \mathbf{A},
\]

since the CoG zeros the scalar field \( \phi(x, y, z) \), and followed with the used of GoD.

<table>
<thead>
<tr>
<th>Field: ( \mathbf{v}(x, t) )</th>
<th>Compressible</th>
<th>Incompressible</th>
</tr>
</thead>
<tbody>
<tr>
<td>Rotational ( \nabla \times \mathbf{v} \neq 0 )</td>
<td>( \mathbf{v} = \nabla \phi + \nabla \times \mathbf{w} ) ( \nabla^2 \mathbf{v} = \frac{1}{c^2} \ddot{\mathbf{v}} )</td>
<td>( \mathbf{v} = \nabla \times \mathbf{w} ) Lubrication theory Boundary layers</td>
</tr>
<tr>
<td>Irrotational ( \nabla \times \mathbf{v} = 0 ) Conservative</td>
<td>Acoustics ( \mathbf{v} = \nabla \psi )</td>
<td>Statics ( \nabla^2 \phi = 0 )</td>
</tr>
<tr>
<td>( \nabla^2 \ddot{\mathbf{g}}(x, t) = \frac{1}{c^2} \ddot{\mathbf{g}}(x, t) )</td>
<td>Laplace’s Eq. ((c \rightarrow \infty))</td>
<td></td>
</tr>
</tbody>
</table>

Figure 1.33: Figure showing the four possible classifications of scalar and vector potential fields, rotational/irrotational, compressible/incompressible. Rotational fields are generated by on the vector potential (e.g., \( \mathbf{A}(x, t) \)), while compressible fields are generated by a the scalar potentials (e.g., voltage \( \phi(x, t) \), (velocity \( \psi \), pressure \( \varphi(x, t) \)), temperature \( T(x, t) \)).

The four categories of linear fluid flow: The following is a summary of the four cases for fluid flow, as summarized in Fig. 1.33:

1.1 Compressible and rotational fluid (general case): \( \nabla \phi \neq 0, \nabla \times \mathbf{w} \neq 0 \). This is the case of wave propagation if a medium where viscosity can not be ignored, as in the case of acoustics close to the boundaries, where viscosity contributes losses (Batchelor, 1967).

1.2 Incompressible, but rotational fluid (Lubrication theory): \( \mathbf{v} = \nabla \times \mathbf{w} \neq 0, \nabla \cdot \mathbf{v} = 0, \nabla^2 \phi = 0 \). In this case the flow is dominated by the walls, the viscosity and heat transfer introduce shear. This is typical of lubrication theory.
2.1 Fluid compressible irrotational flow (acoustics): \( \mathbf{v} = \nabla \phi, \nabla \times \mathbf{w} = 0 \). Here losses (viscosity and thermal diffusion) are small (assumed to be zero). One may define a velocity potential \( \psi \), the gradient of which gives the air particle velocity. For an irrotational fluid (\( \nabla \times \mathbf{v} = 0 \)) (Greenberg, 1988, p. 826), thus \( \mathbf{v} = -\nabla \phi \). This is the case of the conservative field, where \( \int \mathbf{v} \cdot \hat{n} \, dR \) only depends on the end points, and \( \oint \mathbf{v} \cdot \hat{n} \, dR = 0 \). When a fluid may be treated as having no viscosity, it is typically assumed to be irrotational, since it is the viscosity that introduces the shear (Greenberg, 1988, p. 814). A fluid’s angular velocity is \( \Omega = \frac{1}{2} \nabla \times \mathbf{v} = 0 \), thus irrotational fluids have zero angular velocity (\( \Omega = 0 \)).

2.2 Incompressible and irrotational fluid (statics): \( \nabla \cdot \mathbf{v} = 0 \) and \( \nabla \times \mathbf{v} = 0 \) thus \( \mathbf{v} = \nabla \phi \) and \( \nabla^2 \phi = 0 \). An example of such a case is water in a small space at low frequencies, where the wavelength is long compared to the size of the container, the fluid may be treated as incompressible. When \( \nabla \times \mathbf{v} = 0 \), the effects of viscosity may be ignored, as it is the viscosity that creates the shear leading to rotation. This is the case of modeling the cochlea, where losses are ignored and the quasi-static limit is justified.

In summary, each of the cases is some sort of approximation that best applies in the low frequency limit. This is why it is called quasi-static, meaning low, but not zero frequency, where the wavelength is large compared with the dimensions (e.g., diameter).

1.5.12 Lec 41 Maxwell’s Equations: The unification of electricity and magnetism

Once you have mastered the three basic vector operations, the gradient, divergence and curl, you are ready to appreciate Maxwell’s equations. Like the vector operations, these equations may be written in integral or vector form. A very important difference is, that with Maxwell’s equations, we are dealing with well defined physical quantities. The scalar and vector fields take on meaning, and units. Thus to understand these important equations, one must master the names of the four fields \( \mathbf{E}, \mathbf{H}, \mathbf{B}, \mathbf{D} \).

Figure 1.34: A solenoid is a uniform coil of wire. When a current is passed through the wire, a uniform magnetic field intensity \( \mathbf{H} \) is created. From a properties point of view, this coil is indistinguishable from a permanent bar magnet, having north and south poles. Depending on the direction of the current, one end of a finite solenoidal coil is the north pole of the magnet, and the other end is the south pole. The uniform field inside the coil is called solenoidal, a confusing synonym for irrotational. (Figure from Wikipedia.)

Field strengths \( \mathbf{E}, \mathbf{H} \): As summarized in Fig. 1.35 there are two field strengths, the electric \( \mathbf{E} \), with units of [Volts/meter] and the magnetic \( \mathbf{H} \) having units of [Amps/meter]. Their the ratio \( |\mathbf{E}|/|\mathbf{H}| \) is in [ohms].

To understand the meaning of \( \mathbf{E} \), if two conducting plates are place 1 [m] apart, with 1 [V] across them, the electric field is \( \mathbf{E} = 1 \) [V/m]. If a charge (i.e., and electron) is placed in an electric field, it feels a force \( \mathbf{f} = q\mathbf{E} \), where \( q \) is the magnitude of the charge [Col].

To understand the meaning of \( \mathbf{H} \), consider the solenoid made of wire, as shown in Fig. 1.34, which carries a current of 1 [Amp]. The magnetic field \( \mathbf{H} \) inside such a solenoid is uniform and is pointed along the long axis, with a direction that depends on the polarity of the applied voltage (i.e., direction of the current in the wire).
### 1.5. STREAM 3B: VECTOR CALCULUS (10 LECTURES)

<table>
<thead>
<tr>
<th>Symbol</th>
<th>Name</th>
<th>Units</th>
<th>Maxwell’s Eq.</th>
</tr>
</thead>
<tbody>
<tr>
<td>$E$</td>
<td>EF: Electric Field strength</td>
<td>[Volts/m]</td>
<td>$\nabla \times E = -\frac{\partial}{\partial t} B$</td>
</tr>
<tr>
<td>$D = \varepsilon_o E$</td>
<td>ED: Electric Displacement (flux density)</td>
<td>[Col/m$^2$]</td>
<td>$\nabla \cdot D = \rho$</td>
</tr>
<tr>
<td>$H$</td>
<td>MF: Magnetic Field strength</td>
<td>[Amps/m]</td>
<td>$\nabla \times H = \frac{\partial}{\partial t} D$</td>
</tr>
<tr>
<td>$B = \mu_o H$</td>
<td>MI: Magnetic Induction (flux density)</td>
<td>[Webers/m$^2$]</td>
<td>$\nabla \cdot B = 0$</td>
</tr>
</tbody>
</table>

Figure 1.35: The variables of Maxwell’s equations have names (e.g., EF, MI) and units (in square brackets [SI Units]). The units are required to gain a full understanding of each of the four variable and their corresponding equation. For example, Eq. EF has units [Volts/m]. When you integrate $E$ from $x = a, b$, you obtain the voltage difference between those two points. The speed of light in vacuo is $c = 3 \times 10^8 = 1/\sqrt{\mu_o \varepsilon_o}$ [m/s], and the characteristic resistance of light $r_o = \frac{377}{\mu_o \varepsilon_o}$ [Ohms]. The dot over a vector is shorthand for the partial with respect to time (i.e., $\dot{B} = \frac{\partial B}{\partial t}$).

**Flux:** Flux is a flow, such as the mass flux of water flowing in a pipe [kg/s], driven by a force (pressure drop) across the ends of the pipe, or the heat flux in a thermal conductor, having a temperature drop across it (i.e., a window or a wall). The flux is the same as the flow, be it current, mass or heat. In Maxwell’s equations there are also two fluxes, the electric flux $D$, and the magnetic flux $B$. The flux density units for $D$ [Amps/m$^2$] (flux in [Amps]) and the magnetic flux is $B$ [Tesla] (flux in [Webers]).

**Maxwell’s equations**

As shown in Fig. 1.35, Maxwell’s equations consist of two curl equations, operating on the field strengths EF $E$ and MF $H$, and two divergence equations, operating of the field fluxes ED $D$ and MI $B$. Stokes’ Law may be applied to the curl equations and Gauss’ Law may be used on the divergence equations. This should be logically obvious.

**Example:** When a static current is flowing in a wire in the $\hat{z}$ direction, the magnetic flux is determined by Stokes theorem (Fig. 1.31). Thus just outside of the wire we have

$$I_{enc} = \oint_S (\nabla \times H) \cdot \hat{n} \ d|S| = \oint_B H \cdot d\ell. \ [\text{Amps}]$$

For this simple geometry, of the current in a wire is related to $H(x, t)$ by

$$I_{enc} = \oint_B H \cdot d\ell = H_\theta 2\pi r.$$ 

Here $H_\theta$ is perpendicular to both the radius $r$ and the direction of the current $\hat{z}$. Thus

$$H_\theta = \frac{I_{enc}}{2\pi r},$$

and we see that $H$, and thus $B = \mu_o H$ drop of as the reciprocal of the radius $r$.

**Exercise:** Explain how Stokes’ Law may be applied to $\nabla \times E = -\dot{B}$, and explain what it means. Hint: This is the identical argument given above for the current in a wire, but for the electric case.

**Sol:** Integrating the left side of equation EF over an open surface results in an voltage (emf) induced in the loop closing the boundary $B$ of the surface

$$\phi_{induced} = \oint_S (\nabla \times E) \cdot \hat{n} \ d|S| = \oint_B E \cdot d\ell \ [\text{Volts}].$$
The emf (electromagnetic force) is the same as the Thévenin source voltage induced by the rate of change of the flux. Integrating the Eq. 1.5.12 over the same open surface $S$ results in the source of the induced voltage $\phi_{\text{induced}}$, which is proportional to the rate of change of the flux [Webers]

$$\phi_{\text{induced}} = -\frac{\partial}{\partial t} \int_S \mathbf{B} \cdot \hat{n} \, dA = L\dot{\psi}, \text{ [Webers/s].}$$

where $L$ is the inductance of the wire. The area integral on the left is in [W/m$^2$] resulting in the total flux crossing normal to the surface $\psi$ [W].

If we apply Gauss’ Law to the divergence equations, we find the total flux crossing the closed surface.

**Exercise:** Apply Gauss’ Law to equation ED and explain what it means in physical terms. **Sol:** The area of the normal component of $D$ is equal to the volume integral over the charge density, thus Gauss’ Law says that the total charge within the volume $Q_{\text{enc}}$, found by integrating the charge density $\rho(x)$ over the volume $\mathcal{V}$ is equal to the normal component of the flux $D$ crossing the surface $S$

$$Q_{\text{enc}} = \iiint_{\mathcal{V}} \nabla \cdot D \, dV = \iint_{S} D \cdot \hat{n} \, dA.$$  

**Summary:** Maxwell’s four equations relate the field strengths to the flux densities. There are two types of variables, field strengths ($E, H$) and flux densities ($D, B$). There are two classes, electric ($E, D$) and magnetic ($H, B$). One might naturally view this as a 2x2 matrix, with rows being electric and magnetic strengths, and columns being electric and magnetic flux densities, defining a total of four variables:

<table>
<thead>
<tr>
<th>Strength</th>
<th>Flux</th>
</tr>
</thead>
<tbody>
<tr>
<td>Electric</td>
<td>$E$ [V/m]</td>
</tr>
<tr>
<td>Magnetic</td>
<td>$H$ [A/m]</td>
</tr>
</tbody>
</table>

Applying Stokes’ curl law to the forces induces a Thévenin voltage (emf) or Norton current source. Applying Gauss’ divergence law to the flows, gives the total charge enclosed. The magnetic charge always zero since $\nabla \cdot B = 0$, because magnetic mono-poles do not exist. However magnetic dipoles do exist, as in the example of the electron which contains a magnetic dipole.

**The wave equation:** When Maxwell’s equations are combined the vector wave equation results. There is the famous story that Maxwell introduce the displacement current $\dot{D}$ to make the equations symmetric, allowing him to predict the formula for the speed of light. The displacement current was not verified by the experiments of Faraday and Ampere.

Taking the curl of Eq. EF [Maxwell’s Electric Field strength equation for the $\nabla \times E$ (Fig. 1.35)], we obtain in the electric vector wave equation

$$\nabla \times \nabla \times E = -\nabla \times \dot{B} = -\mu_o \nabla \times \dot{H} = -\mu_o \epsilon_o \ddot{E}. \quad (1.161)$$

Using the GoD identity (Eq. 1.152) gives

$$\nabla^2 \dot{E} - \frac{1}{\epsilon_o} \nabla (\nabla \cdot D) = \mu_o \epsilon_o \ddot{E}. \quad (1.162)$$
In a charge free region \( \nabla \cdot \mathbf{D} = 0 \). Since \( \frac{1}{c^2} = \frac{\mu_o}{\epsilon_o} \), the vector wave equation results:

\[
\nabla^2 \mathbf{E}(\mathbf{x},t) = \frac{1}{c^2} \dot{\mathbf{E}}(\mathbf{x},t) \leftrightarrow \frac{s^2}{c^2} \mathbf{E}(\mathbf{x},\omega). \tag{1.163}
\]

It is important to distinguish the vector Laplacian from the scalar Laplacian. As was discussed in Table 1.7 (p. 122), Recall the d’Alembert solutions of the scalar wave equation (Eq. 1.84, p. 103)

\[
\mathbf{E}(\mathbf{x},t) = f(\mathbf{x} - ct) + g(\mathbf{x} + ct),
\]

where \( f, g \) are arbitrary vector fields.

In a like manner one may derive the wave equation in terms of \( \mathbf{H} \)

\[
\nabla^2 \mathbf{H}(\mathbf{x},t) = \frac{1}{c^2} \dot{\mathbf{H}}(\mathbf{x},t) \leftrightarrow \frac{s^2}{c^2} \mathbf{H}(\mathbf{x},\omega) \tag{1.164}
\]

This equation does not have the restriction that there is no free charge, because \( \nabla \cdot \mathbf{B} = 0 \). Thus both \( \mathbf{E}, \mathbf{H} \) obey the wave equation (thus they are locked together in space-time if we assume no free charge) (Sommerfeld, 1952).

**Electrical impedance seen by an electron:** Up to now we have only considered the Brune impedance which is a special case with no branch points or branch cuts. We can define impedance for the case of diffusion, as in the case of the diffusion of heat. There is also the diffusion of electrical and magnetic fields at the surface of a conductor, where the resistance of the conductor dominates the dielectric properties, which is called the electrical skin effect, where the conduction currents are dominated by the conductivity of the metal rather than the displacement currents. In such cases the impedance is proportional to \( \sqrt{s} \), implying that it has a branch cut. Still in this case the real part of the impedance must be positive in the right-half \( s \) plane, the required condition of all impedances, such that postulate P3 (p. 96) is satisfied.

**Example:** When we deal with Maxwell’s equations the force is defined by the Lorentz force

\[
f = q \mathbf{E} + qv \nabla \times \mathbf{B} = q \mathbf{E} + \mathbf{C} \times \mathbf{B},
\]

which is the force on a charge (e.g., electron) due to the electric \( \mathbf{E} \) and magnetic \( \mathbf{B} \) fields. The magnetic field plays a role when the charge has a velocity \( v \). When a charge is moving with velocity \( v \), it may be viewed as a current \( \mathbf{C} =qv \).

In this case the impedance in a wire, where the current is constrained, the complex impedance density is

\[
Z(s) = \sigma + s\epsilon_o, \quad [\Omega/m^2]
\]

which when integrated over an area is the impedance in Ohms (Feynman, 1970c, p. 13-1). Here \( \sigma \) is the electrical conductivity and \( \epsilon_o \) is the electrical permittivity. Since \( \sigma \gg \omega \epsilon_o \) this reduces to the resistance of the wire, per unit length.

**1.5.13 Lec 42 The Quasi-static approximation**

There are a number of assumptions and approximations that result in special cases, many of which are classic. These manipulations are all done at the differential equation level, by making assumptions that change the basic equations that are to be solved. These approximations distinct from assumptions made while solving a specific problem.

A few and important examples include
CHAPTER 1. INTRODUCTION

1. \textit{In vacuo} waves (free-space scalar wave equation)

2. Expressing the vector wave equation in terms of scalar and vector potentials

3. Quasi-statics
   (a) scalar wave equation
   (b) Kirchhoff’s low-frequency lumped approximation (LRC networks)
   (c) Transmission line equations (Telephone and Telegraph equations)

One of the very first insights into wave propagation was due to Huygens (c1640) (Fig. 1.21).

**Quasi-statics and its implications:** The term \textit{quasi-statics} (Postulate P9, p. 97) is an approximation, used to reduce a partial differential equations to a scalar (one-dimensional) equation (Sommerfeld, 1952). Quasi-statics is a way of reducing a three dimensional problem to a one-dimensional problem. So that it is not misapplied, it is important to understand the nature of this approximation, which goes to the heart of transmission line theory. The quasi-static approximation states that the wavelength $\lambda$ is greater than the dimensions of the size of the object $\Delta$ (e.g., $\lambda \gg \Delta$). The best known example, Kirchhoff’s current and voltage laws, KCL and KVL, almost follow from Maxwell’s equations given the quasi-static approximation (Ramo et al., 1965). These laws state that the sum of the currents at a node must be zero (KCL) and the sum of the voltages around a loop must be zero (KCL).

These well known laws the analogue of Newton’s laws of mechanics. The sum of the forces at a point is the analogue of the sum of the voltages. Voltage $\phi$ is the force potential, since the electric field $E = -\nabla \phi$. The sum of the currents is the analogue of the vector sum of velocities at a point is zero.

The acoustic wave equation describes how the scalar field pressure $p(x, t)$, the vector force density potential ($f(x, t) = -\nabla p(x, t)$ [N/m$^2$]), propagates in three dimensions. (The net force is the integral of the pressure gradient over an area.) If the wave propagation is restricted to a pipe (e.g., organ pipe), or to a string (e.g., an guitar or lute), the transverse directions may be ignored, due to the quasi-static approximation. What needs to be modeled by the equations is the wave propagation along the pipe (string). Thus we may approximate the restricted three-dimensional wave by a one-dimensional wave.

However if we wish to be more precise about this reduction in geometry ($\mathbb{R}^2 \rightarrow \mathbb{R}$), we need to consider the quasi-static approximation, as it makes assumptions about what is happening in the other directions, and quantifies the effect ($\lambda \gg \Delta$). Taking the case of wave propagation in a tube, say the ear canal, there is the main wave direction, down the tube. But there is also wave propagation in the transverse direction, perpendicular to the direction of propagation. As shown in Table 3.2 (p. 211), the key statement of the quasi-static approximation is that the wavelength in the transverse direction is much larger than the radius of the pipe. This is equivalent to saying that the radial wave reaches the walls and is reflected back, in a time that is small compared to the distance propagated through the pipe. Clearly the speed of sound down the pipe and in the transverse direction is the same if the medium is homogeneous (i.e., air or water). Thus the sound reaches the walls and is returned to the center line in a time that the axial wave traveled about 1 diameter along the pipe. So if the distance traveled is several diameters, the radial parts of the wave have time to come to equilibrium. So the question one must ask is, what are the conditions of such an equilibrium. The most satisfying answer to this is to look at the internal forces on the air, due to the gradients in the pressure.
The pressure $\rho(x, t)$ is a potential, thus its gradient is a force density $f(x, t) = -\nabla \rho(x, t)$. What this equation tells us is that as the pressure wave approaches that of a plane wave, the radial (transverse) forces go to zero. If the tube has a curvature, or a change in area, then there will be local forces that create radial flow. But after traveling a few diameters, these forces will come to equilibrium and the wave will return to a plane wave. The internal stress caused by a change in area must settle out very quickly. There is a very important caveat however: it is only at low frequencies that the plane wave can dominate. At frequencies such that the wavelength is very small compared to the diameter, the distance traveled between reflections is much greater than a few diameters. Fortunately the frequencies where this happens are so high that they play no role in frequencies that we care about. This effect is referred to as cross-modes which imply some sort of radial standing waves. In fact such modes exist in the ear canal, but on the eardrum where the speed of sound is much slower than that of air. Because of the slower speed, the ear drum has cross-modes, and these may be seen in the ear canal pressure. Yet they seem to have a negligible effect on our ability to hear sound with good fidelity. The point here is that the cross modes are present, but we call upon the quasi-static approximation as a justification for ignoring them, to get closer to the first-order physics.

**Quasi-statics and Quantum Mechanics**

It is important to understand the meaning of Planck’s constant $h$, which appears in the relations of both photons (light “particles”) and electrons (mass particles). If we could obtain a handle on what exactly Planck’s constant means, we might have a better understanding of quantum mechanics, and physics in general. By cataloging the dispersion relations (the relation between the wavelength $\lambda(\nu)$ and the frequency $\nu$), for between electrons and photons, this may be attainable.

Basic relations from quantum mechanics for photons and electrons include:

1. Photons (mass=0, velocity = $c$)
   
   (a) $c = \lambda \nu$: The speed of light $c$ is the product of its wavelengths $\lambda$ times its frequency $\nu$. This relationship is only for mono-chromatic (single frequency) light.
   
   (b) The speed of light is
   
   $$c = \frac{1}{\sqrt{\mu_0 \epsilon_0}} = 0.3 \times 10^6 \ [\text{m/s}]$$

   (c) The characteristic resistance of light $r_o = \sqrt{\mu_0 / \epsilon_0} = |E|/|H| = 377 \ [\text{ohms}]$ is defined as the magnitude of the ratio of the electric $E$ and magnetic $H$ field, of a plane wave in-vacuo.

   (d) $E = h\nu$: the photon energy is given by Planck’s constant $h \approx 6.623 \times 10^{-34} \ [\text{m}^2 \text{kgm/s}]$, times the frequency (or bandwidth) of the photon.

2. Electrons (mass = $m_e$, velocity $V = 0$):
   
   (a) $E_e = m_e c^2 \approx 0.91 \cdot 10^{-30} \cdot 0.3^2 \cdot 10^{12} = 8.14 \times 10^{-20} \ [\text{J}]$ is the electron rest energy (velocity $V = 0$) of every electron, of mass $m_e = 9.1 \times 10^{-31} \ [\text{kgm}]$, where $c$ is the speed of light.
(b) \( p = h/\lambda \): The momentum \( p \) of an electron is given by Planck’s constant \( h \) divided by the wavelength of an electron \( \lambda \). It follows that the bandwidth of the photon is given by

\[ \nu_e = \frac{E_e}{h} \]

and the wavelength of an electron is

\[ \lambda_e = \frac{h}{p_e}. \]

One might reason that QM obeys the quasi-static (long wavelength) approximation. If we compare the velocity of the electron \( V \) to the speed of light \( c \), then we see that

\[ c = \frac{E}{p} \gg V = \frac{E}{p} = mV^2/V \]

**Conjecture on photon energy:**

Photons are seen as quantized because they are common generator by atoms, which produce light-particles having the difference in two energy (quantum, or eigen-states) levels. The relation \( E = h\nu \) does not inherently depend on \( \nu \) being a fixed frequency. Planck’s constant \( h \) is the EM energy density over frequency, and \( E(\nu_o) \) is the integral over frequency

\[ E(\nu_o) = h \int_{\nu_o}^{\nu_o} d\nu = 2h\nu_o. \]

When the photon is generated by an atom \( \nu_o \) is quantized by the energy level difference that corresponds to the frequency (energy level difference) of the photon jump.

1.5.14 **Lec 43: Review for the Final Exam**

**Summary**

Mathematics began as a simple way of keeping track of how many things there were. But eventually physics and mathematics evolved together as tools to help us navigate our environment, not just physically around the globe, but how to solve daily problems such as food, water and waste management, understand the solar system and the stars, defend ourselves, use tools of war, etc.

Based on the historical record of the abacus, one can infer that people precisely understood the concept of counting, addition, subtraction and multiplication (recursive addition).

There is some evidence that the abacus, a simple counting tool, formalizing the addition of very large numbers, was introduced to the Chinese by the Romans, where it was used for trade.

However this working knowledge of arithmetic did not to show up in written number systems. The Roman numerals were not useful for doing calculations done on the abacus. The final answer would then be expressed in terms of the Roman number system.

According to the known written record, the number zero (null) had no written symbol until the time of Brahmagupta (628 CE). One should not assume the concept of zero was not understood simply because there was no symbol for it in the Roman Numerical system. Negative numbers and zero would be obvious when using the abacus. Numbers between the integers would be represented as *rational numbers* \( \mathbb{Q} \) since any number may be approximated with arbitrary accuracy with rations numbers.
Mathematics is the science of formalizing a repetitive method into a set of rules, and then generalizing it as much as possible. Generalizing the multiplication and division algorithm, to different types of numbers, becomes increasingly more complex as we move from integers to rational numbers, irrational numbers, real and complex numbers and ultimately, vectors and matrices. How do you multiply two vectors, or multiply and divide one matrix by another? Is it subtraction as in the case of two numbers? Multiplying and dividing polynomials (by long division) generalizes these operations even further. Linear algebra is further important generalization, fallout from the fundamental theorem of algebra, and essential for solving the generalizations of the number systems.

Many of the concepts about numbers naturally evolved from music, where the length of a string (along with its tension) determined the pitch (Stillwell, 2010, pp. 11, 16, 153, 261). Cutting the string’s length by half increased the frequency by a factor of 2. One forth of the length increases the frequency by a factor of 4. One octave is a factor of 2 and two octaves a factor of 4 while a half octave is $\sqrt{2}$. The musical scale was soon factored into rational parts. This scale almost worked, but did not generalize (sometimes known as the Pythagorean comma\(^3\)), resulting in today’s well tempered scale, which is based on 12 equal geometric steps along one octave, or 1/12 octave ($\frac{13}{12} \approx 1.05946 \approx \frac{18}{17} = 1 + \frac{1}{17}$).

But the concept of a factor was clear. Every number may be written as either a sum, or a product (i.e., a repetitive sum). This led the early mathematicians to the concept of a prime number, which is based on a unique factoring of every integer. At this same time (c5000 BCE), the solution of a second degree polynomial was understood, which lead to a generalization of factoring, since the polynomial, a sum of terms, may be written in factored form. If you think about this a bit, it is sort of an amazing idea, that needed to be discovered (Stillwell, 2010, p.). This concept lead to an important string of theorems on factoring polynomials, and how to numerically describe physical quantities. Newton was one of the first to master these tools with his proof that the orbits of the planets are ellipses, not circles. This lead him to expanding functions in terms of their derivatives and power series. Could these sums be factored? The solution to this problem led to calculus.

So mathematics, a product of the human mind, is a highly successful attempt to explain the physical world. All aspects of our lives were impacted by these tools. Mathematical knowledge is power. It allows one to think about complex problems in increasingly sophisticated ways. An equation is a mathematical sentence, expressing deep knowledge. Witnessed $E = mc^2$ and $\nabla^2 \psi = \ddot{\psi}$.

Reading List: The above concepts come straight from mathematical physics, as developed in the 17th–19th centuries. Much of this was first developed in acoustics by Helmholtz, Stokes and Rayleigh, following in Green’s footsteps, as described by Lord Rayleigh (1896). When it comes to fully appreciating Green’s theorem and reciprocity, I have found Rayleigh (1896) to be a key reference. If you wish to repeat my reading experience, start with Brillouin (1953, 1960), followed by Sommerfeld (1952); Pipes (1958). Second tier reading contains many items Morse (1948); Sommerfeld (1949); Morse and Feshbach (1953); Ramo et al. (1965); Feynman (1970a); Boas (1987). A third tier might include Helmholtz (1863a); Fry (1928); Lamb (1932); Bode (1945); Montgomery et al. (1948); Beranek (1954); Fagen (1975); Lighthill (1978); Hunt (1952); Olson (1947). It would be a mistake to ignore other massive physics writings by stalwart authors, J.C.\(^3\)https://en.wikipedia.org/wiki/Pythagorean_comma
Slater\textsuperscript{104} and Landau and Lifshitz\textsuperscript{105} and their impressive series of mathematical physics books.

You must enter at a level that allows you to understand. Successful reading of these books critically depends on what you already know. A rudimentary (high school) level of math comprehension must be mastered first. Read in the order that helps you best understand the material.

Without a proper math vocabulary, mastery is hopeless. I suspect that one semester of college math can bring you up to speed. This book is my attempt to present this level of understanding.

\textsuperscript{104}https://en.wikipedia.org/wiki/John_C._Slater
\textsuperscript{105}https://www.amazon.com/Mechanics-Third-Course-Theoretical-Physics/dp/0750628960
Chapter 2

Number Systems: Stream 1

Chapters 2-5 contain advanced material that go beyond Chapter 1. You should start with Chapter 1, and once mastered, then move to the rest of the book.

This chapter (2) is devoted to Number Systems (Stream 1), starting with the counting numbers \(\mathbb{N}\). In this chapter we delve more deeply into the details of the topics of Chapter 1, Lectures 4-9.

2.1 Week 2

In Section 1.2.3 we explore in more detail the two fundamental theorems of prime numbers, working out a sieve example, and explore the logarithmic integral \(Li(N)\) which approximates the density of primes \(\rho_k(N)\) up to prime \(N\).

The topics of Section 1.2.4 consider the practical details of computing the greatest common divisor (GCD) of two integers \(m, n\) (Matlab’s routine \(l = \text{gcd}(m, n)\)), with detailed examples and comparing the algebraic and matrix methods. Homework assignments will deal with these two methods. Finally we explore the relationship between coprimes and the GCD. In Section 1.2.5 we defined the Continued Fraction algorithm (CFA), a method for finding rational approximations to irrational numbers. The CFA and GCD are closely related, but the relation needs to be properly explained. In Section 1.2.6 we derive Euclid’s formula, the solution for the Pythagorean triplets (PT), based on Diophantus’s chord–tangent method. This method is used many times throughout the course notes, first for computing Euclid’s formula for the PTs, then for finding a related formula in Section 1.2.7 for the solutions to Pell’s equation, and finally for finding the mapping from the complex plane to the extended complex plane (the Riemann sphere).

Finally in Section 1.2.8 the general properties of the Fibonacci sequence is discussed. This equation is a special case of the second order digital resonator (well known in digital signal processing), so it has both historical and practical application for engineering. The general solution of the Fibonacci is found by taking the Z-transform and finding the roots, resulting in an eigenvalue expansion (Appendix D).

2.1.1 Lec 4 Two theorems on primes

Theorem 1: Fundamental Theorem of Arithmetic

Factoring integers: Every integer \(n \in \mathbb{N}\) has a unique factorization (Stillwell, 2010, p. 43) (Eq. 1.6, p. 39).
Cofactors: Integers 2312 and 2313 are said to be coprime, since they have no common factors. Coprimes may be identified via the greatest common divisor:

\[ \gcd(a, b) = 1 \]

using the Euclidean algorithm (Stillwell, 2010, p. 41).

**Theorem 2: Prime Number Theorem**

The primes are a random field since there is no way to predict when the next prime will occur. Thus one needs to use statistical methods to characterize their density. Based on a sample of approximately 3 million primes, Gauss showed empirically that the average total number of primes less than \( N \) is

\[
\sum_{n=1}^{N} \delta_n \sim \frac{N}{\ln N}.
\]  

These primes were obtained by manual calculations “as a pastime” in 1792-3 (Goldstein, 1973).

Define \( \delta_n = 1 \) if \( n \) is a prime, and zero otherwise.\(^1\)

It follows that the average density of primes is \( \rho \pi(N) \sim 1/\ln N \), thus

\[
\rho \pi(N) \equiv \frac{1}{N} \sum_{n=1}^{N} \delta_n \approx \frac{1}{N} \int_{2}^{N} \frac{d\xi}{\ln \xi},
\]

where \( Li(N) \) is the offset logarithmic integral (Stillwell, 2010, p. 585). The primes are distributed as \( 1/\ln(n) \) since the average total number of primes is proportional to the logarithmic integral \( Li(n) \) (Goldstein, 1973; Fine, 2007).

Here is a Matlab/Octave code that tests this formula:

```
% Computes density of primes from average density
NP=1e6; % 10^6 primes
p=primes(NP); % compute primes
delta=zeros(1,NP); delta(p)=1; % put 1 at each prime
PI=cumsum(delta); % Number of primes vs N
rho=PI./(1:NP); % estimate of the density of primes= PI(N)./N

figure(1)
semilogy(rho); % plot of density vs number of primes
title('Density of primes vs. N'); ylabel('
\rho(N)'); xlabel('N'); grid on
```

From the Prime Number Theorem it is clear that the density of primes is large (they are not scarce). As best I know there are no methods to find primes other than by the sieve method (Section 1.2.3, p. 37). If there is any good news it is that they only need to be computed once, and saved. In practical applications this may not help much, given their large number. But I suspect they could easily be saved on a modern disk (e.g., 1 TB).

Not surprisingly, playing with primes has been a popular pastime of mathematicians. Perhaps this is because those who have made inroads, providing improved understanding, have become famous.

---

\(^1\)You may view \( \delta_n \) for the first 100 numbers with the one-line Matlab/Octave command `stem(isprime(1:100))`
2.1. WEEK 2

2.1.2 RSA public-private keys

Internet security depends on a public-private key system, called RSA\(^2\) which is built on the difficulty of factoring large primes. When a forward computation is easy, such as multiplying two primes, is easy, and the inverse problem (factoring a product into two primes) is hard, it is called a trap door.

What makes RSA work is a trap door. Internet security is not as important as global warming, but its pretty important. Global warming has no “end game.” Internet security does, redesign the internet.

Let’s consider the problem of factoring two primes, given the product \(N = \pi_k \pi_l\), \(k \neq l\). Let’s further assume we have list of all the primes \(\{\pi_1, \ldots, \pi_m\}\) including \(N < \pi_m\). To factor \(N\) we must divide \(N\) by each prime less than \(N\). With intelligent sampling, on average, we would find the prime factor in \(N/2\) tries. In theory, given primes \(\pi_n\) up to \(n = N\), the density \(\rho_{\pi}(N)\) could help one search for a particular prime of known size \(N\), by estimating how many primes there are in the neighborhood of \(N\). The cost of factoring any integer \(\in \mathbb{N}\) would be related to the magnitude of \(N = \pi_k \pi_l\), where \(N\) is the size of the private key, assuming a simple product key. Worst case an exhaustive search would take \(m - 1\) divides of the form \(N/\pi_k k = 1 \cdots m - 1\). If one knew approximately where the factor is \(\in \mathbb{P}\) (i.e., \(k\) of \(\pi_k\)), and knows one of the primes, the search could be done with an exhaustive search, by addition. The number of primes less that \(N\) may be computed using Gauss’ formula for the density of primes.

So the question remains: How does internet security actually work? Unfortunately at this time I cannot give you a proper summary. But it is more complex than I have let on. The full answer requires a proper course in number theory and internet security, beyond what can be presented.\(^3\) To save you the trouble of looking up the method, it is quoted here\(^4\).

Setting up an RSA system involves choosing large prime numbers \(p\) and \(q\), computing \(n = pq\) and \(k = \phi(n)\), and finding two numbers \(e\) and \(d\) such that \(ed \equiv 1 \pmod k\). The numbers \(n\) and \(e\) (the “encryption key”) are released to the public, and \(d\) (the “decryption key”) is kept private.

A message, represented by an integer \(m\), where \(0 < m < n\), is encrypted by computing \(S = me \pmod n\). It is decrypted by computing \(t = Sd \pmod n\). Euler’s Theorem can be used to show that if \(0 < t < n\), then \(t = m\).

The security of an RSA system would be compromised if the number \(n\) could be factored or if \(\phi(n)\) could be computed without factoring \(n\).

The power of this security device is based on the relatively high density of primes \(\mathbb{P} \subset \mathbb{N}\), which is addressed by the Prime Number Theorem.

Example: The RSA public key is a string of characters representing a prime number \(\pi_{pub}\):

\[
\text{AAAAB3NzaC1yc2EAAADAgABAAAABACcysydp9uXubR/ukXMpVqK2Et3tJqJeIEMg6cm7jRa2hrhFCV1+r9Y5d9sKzdkkefHLH13VAYthE04mGChBFPr4k8jNnHaWqHbukhO7BnqCg517WdpEqH72EqPf2I2jMs+3y17b055hNZW9W65qzFCNmtJr1IUkY3xUnrBsb/b7n5sA6GnViyLup+8rSNLqW1oH89qCqHGNm0h7P5Y/b7dgyWzX/9t5MoTNeuq7jghP0J6aqUaFTXAg00SBKXYjUGB}
\]

\(^2\)https://en.wikipedia.org/wiki/RSA_(cryptosystem)
\(^3\)https://en.wikipedia.org/wiki/RSA_(cryptosystem)#Operation
\(^4\)https://en.wikipedia.org/wiki/Euler\%27s_totient_function#The_RSA_cryptosystem
The RSA private key is a second string of characters, representing the product of prime numbers $\pi_{pri}$:

\[
\text{MIIEpAIBAAKCAQEArMn6fbg17m0f7pFzKvYCTnhLBHUCxhCDiOpgu6oWto4aR4ldqf/WENanCn2Jhhexy/hiNLQMk4RPeJuAphwzepFMayzsc920MOvCvYyTrQapwpoDode1g8R6kqIAj3pyYLZpZ2Uu29OYTYTL/VukgsxQ3j2s9aDV5HFS562wfy22yDLoqBhpiSel7qyfvo3jalatb/PeEhbxjJ9i3e+Wp2yXYGFA1/beKE3xergu441Zz9CemqlJAH01kWtAEG9I8LDCox20k/J3JRBB5aymq2+3j3wa17z1ezvyd6hITIvop3jy2qbg6qNiBN1Bqvyuo1uq5zQL/g0yDHDu6EWN2QlDAAAb1ABQCFp9aMura1hcbzCv/TVdmEbnmz2k/e7Qdvy1YVWvzrzwv5wO2M6HIA02J2s9ewVCgb7i1uwxwSmj/S+U13QgFodyb815SP/wyhll1LrpZ+qpyvCrCNNWVc/t7kwBj2FP7hBn2kmmkMG0hTSL5kvrTve/8Zmkn-VRK7qWqFowRtdvufu1SCjrB1hLCLx6QsC5qCHLbhtJiY0YmNw2Bgl1fniqXAOGBvXpOp9vr/ko5y5u/hds1S99dcsn6h60G/XXEsh4bHywq2qcgcsKzixa2ar38B0USE6NjeqVLCk1i5uMyEXzqbxFWhy+kvjm80ZoFgLwks2ODeo+a/DKY10XCdHm1aBAoCGBANO1HcQA+tW0/j0/ImcgmLkTKFbm8x1o2jw4bKIfRTq8jPVnsQjwJAc6oazaC73TDR75Yed0YNg72yujmi7lmo88MN9v53sRgfeCtgC41xv7o2annFDS9ysOF8981gQ104/I2BAPwybVvQ26j2mfragB90RVgAy1pyX+9jcrPmdRaoGABANE3cdyJ3zsA1c3YvB0Fsp8kBULa2xonsQ15+cDgHe/ugr6tawunim5Sc8J0nMeyqKydLyAd6E30qjRjBRhns4szfYRhexBN2apAbw30a6/+4MRPakWvFDCDWY10oU67vyqvKu7s60nuUJFLzfms6EB1qin8f5x5DA62UYTVaOGAFJi1VYli1oFCNBS7e5X59yYKb52Ldnp1J0k1KeWgwUBmT/xxy51gst62i1JDA2vVxwkb27d7K1yG70mTmFvBoasAjtjwFEDrEYexNAA3jwkpgK/hpDB8vCqDEvoqgKmp5XwU0e8BnPc2ptd0q2jx6JZKjfkp2RUKo3AxuswFECQYB8+c0bN2X8sVpMvoo8zs1Nxs5inOpVc2mW1A9wOwGaw/FqBE8D92SK9P807P9b9c/2aRLUL5TeKrk/L07Bsk1L5d3A5YGPNvP/24Khr+rpWmt727pYkz9/1/IEKY1Jk1xwzBeD1Pn1a207ucQdU90l1eE/XA2CHz2v7Us1nB0AQRekDQCNExgYEF2o1MQ3Qwvyk9ghSVv1us2ulmgC2M8KX1inJdzQZ22M0n64YFqUs0PyU1j17IMU1UB6sFzFWFq277ykgyFwbq0VB0RAQ/kX1EdKBcrUK/k09JGqhvSnqE1hS6NYMjFnh4Wnt30hD09JQcFvKcFa1r+0E2ZS1yDqkr=>

This much longer string of character represents a second prime number $\pi_{pri}$, the private key.

If I gave you the ASCII string that represents the product of the two prime numbers $\pi_{k} \pi_{l}$, along with $\pi_{k}$, can you find $\pi_{l}$? Yes, divide one by the other, with remainder zero. If I ask you to find $\pi_{l}$ given $\pi_{k} \pi_{l}$, you would need to call the NSA, or be some sort of expert on factoring primes. Given $\pi_{k} \pi_{l}$, for some large values of $k$ and $l$, the GCD is not helpful, nor the CFA. While the factoring can be done (it is not impossible), it is a very expensive calculation.

Why not use Matlab's factor($N$) routine to find the factors? This is where cost comes in. If the primes are large enough, even the NSA cannot do it in a lifetime. The numbers used in RSA are too large for Matlab's routine to deliver an answer. In fact, even the largest computer in the world (such as the University of Illinois’ super computer (NCSA Water) cannot do this computation. But, for a carefully chose set of primes, not to large for the NCSA to factor them, they can decrypt a message, but you cannot. This is why they pay for these massive super-computers.

As best I know you need to try one prime at a time to see if the divisor gives you a zero remainder. There are no hints, such as $l \approx k$. Nothing to help but to grind your way through all the known primes. And if the primes are too large to be on the list of all known primes, then you have the additional burden of generating a list of primes long enough to contain the ones your looking for. You can be sure that the NSA (or whoever has the fastest computer) has a longer list than you do, and its getting longer every day.

What you are given is the public key (prime $\pi_{pub}$ (the shorter one), and you need to guess the private key $\pi_{pri}$ and decrypt the message I send you with the product $\pi_{pub} \pi_{pri}$.  

2.1.3 Lec 5 Greatest common divisor (GCD)

**Division with rounding method:** This method implements gcd($a$, $b$). In matrix form we have $(m, n, k \in \mathbb{N})$

\[
\begin{bmatrix} m_{k+1} \\ n_{k+1} \end{bmatrix} = \begin{bmatrix} 0 & 1 \\ 1 & -\frac{m}{n} \end{bmatrix} \begin{bmatrix} m_k \\ n_k \end{bmatrix}.
\]

(2.3)

This starts with $k = 0$, $m_0 = a$, $n_0 = b$. With this method there is no need to test if $n < m$, as it is built into the procedure. The method uses the floor function $[x]$, which finds the integer part of $x$ ([x]) rounds toward $-\infty$. Following each step we will see that the value $n_{k+1} < m_{k+1}$. 

2.1.3 Lec 5 Greatest common divisor (GCD)
The method terminates when \( n_{k+1} = 0 \) with \( \gcd(a, b) = m_{k+1} \). The previous values \( m_k, n_k \) are the solutions to Bézout’s identity \( \gcd(n, m) = 1 \), namely \( m_km_0 + n_kn_0 = 1 \), since the terminal state and the GCD of \( a, b \) is \( m - n\lfloor m/n \rfloor = 0 \), for which \( n = \gcd(a, b) \).

Below is 1-line vectorized code that is much more efficient than the direct matrix method:

\[
\begin{align*}
k &= \gcd(m, n) \quad \text{\%entry point: input} \ m, n; \ \text{output} \ k=\gcd(m, n) \\
M &= [\text{abs}(m), \text{abs}(n)]; \ \%\text{init} \ M \\
\text{while} \ M(2) \neq 0 \ \%\text{``almost work'' with irrational inputs} \\
\quad M &= [M(2) - M(1) \times \text{floor}(M(2)/M(1)); M(1)]; \ \%M = [M(1); M(2)] \ \text{with} \ M(1) < M(2) \\
\text{end}
\end{align*}
\]

With a minor extension in the test for “end,” this code can be made to work with irrational inputs (e.g., \((n\pi, m\pi)\)).

This method calculates the number of times \( n < m \) must subtract from \( m \) using the floor function. This operation is the same as the mod function. Specifically

\[
n_{k+1} = m_k - \floor{\frac{m}{n}}n_k
\]  

so that the output is the definition of the remainder of modular arithmetic. This would have been obvious to anyone using an abacus, which explains why it was discovered so early.

Note that the next value of \( m \ (M(1)) \) is always less than \( n \ (M(2)) \), and must remain greater or equal to zero. This one-line vector operation is then repeated until the remainder \( M(1) \) is 0. The gcd is then \( n \ (M(2)) \). When using irrational numbers, this still works except the error is never exactly zero, due to IEEE 754 rounding. Thus the criterion must be that the error is within some small factor times the smallest number (which in Matlab is the number \( \text{eps} = 2.220446049250313 \times 10^{-16} \), as defined in the IEEE 754 standard.)

Thus without factoring the two numbers, Eq. 2.4 recursively simply finds the gcd. Perhaps this is best seen with some examples.

The GCD is an important and venerable method, useful in engineering and mathematics, but, as best I know, is not typically taught in the traditional engineering curriculum.

**Graphical meaning of the GCD:** The Euclidean algorithm is very simple when viewed graphically. In Fig. 2.1 we show what is happening as one approaches the threshold. After reaching the threshold, the two number must be swapped, which is addressed by upper row of Eq. 2.3.

Multiplication is simply recursive addition, and finding the gcd takes advantage of this fact. Let’s take a trivial example, \((9, 6)\). Taking the difference of the larger from the smaller, and writing multiplication as sums, helps one see what is going on. Since \(6 = 3 \times 2\), this difference may be written two different ways

\[
9 - 6 = (3 + 3 + 3) - (3 + 3) = 0 + 0 + 3 = 3,
\]

or

\[
9 - 6 = (3 + 3 + 3) - (2 + 2 + 2) = 1 + 1 + 1 = 3.
\]

Written out the first way, it is 3, because subtracting \((3+3)\) from \((3+3+3)\) leaves 3. Written out in the second way, each 3 is matched with a -2, leaving 3 ones, which add to 3. Of course the two decomposition must yield the same result because \(2 \cdot 3 = 3 \cdot 2\). Thus finding the remainder of the larger number minus the smaller yields the gcd of the two numbers.

\[\text{https://en.wikipedia.org/wiki/Modulo_operation}\]
Figure 2.1: The Euclidean Algorithm recursively subtracts \( n \) from \( m \) until the remainder \( m - kn \) is either less than \( n \) or zero. Note that this is similar to \( \text{mod}(m, n) \). The GCD recursively computes \( \text{mod} \), then swaps \( m, n \) so that \( n < M \), and repeats, until it finds the GCD. For the case depicted here the value of \( k \) that renders the remainder less than \( n \) is \( k = 6 \). If one more step were taken (\( k = 7 \)) the remainder would become negative. By linear interpolation we can find that \( m - kn = 0 \) when \( k = \text{floor}(m/n) \). For this example \( k = 6.5 \), thus \( k = 6 \). Thus \( 6 = \text{floor}(m/n) < n \). This is nonlinear arithmetic, which is the key to the GCD.

Coprimes: When the gcd of two integers is 1, the only common factor is 1. This is of key importance when trying to find common factors between the two integers. When \( 1 = \gcd(m,n) \) they are said to be coprime, which can be written as \( m \perp n \). By definition, the largest common factor of coprimes is 1. But since 1 is not a prime (\( \pi_1 = 2 \)), they have no common primes. It can be shown (Stillwell, 2010, p. 41-42) that when \( a \perp b \), there exist \( m, n \in \mathbb{Z} \) such that

\[
am + bn = \gcd(a, b) = 1.
\]

Generalizations of GCD: The GCD may be generalized in several significant ways. For example what is the GCD of two polynomials? To answer this question one must factor the two polynomials to identify common roots.

2.2 Week 3

2.2.1 Lec 6 Continued Fraction Expansion (CFA)

Continued Fractions and circuit theory: One of the most powerful generalizations of the CFA seems to be the expansion of a function of a complex variable, such as the expansion of an impedance \( Z(s) \), as a function of complex frequency \( s \), as described in Fig. 2.2 and Eq. 2.5. This is especially interesting in that it leads to a physical interpretation of the impedance in terms of a transmission line (horn), a structure well known in acoustics as having a variable area \( A(x) \) as function of the range variable \( x \).

The CFA expansion is of great importance in circuit theory, where it is equivalent to an infinitely long segment of transmission line, composed of series and shunt impedance elements. Thus such a cascade network composed of 1 ohm resistors, has an input impedance of \( (1 + \sqrt{5})/2 \approx 1.6180 \) [ohms] Eq. 1.8 (p. 44).

The CFA may be extended to monomials in \( s \). For example consider the input impedance of a cascade L-C transmission line as shown in Fig. 2.2. The input impedance of this transmission line
is given by a continued fraction expansion of the form
\[ Z_{in} = sL + \frac{1}{sC + \frac{1}{sL + \frac{1}{sC + \cdots}}} =: [sL, sC, sL, sC, \cdots]. \tag{2.5} \]

In some vague way, Eq. 2.5 is reminiscent of a Taylor series expansion about \( s = 0 \), yet very different. In the limit, as the frequency goes to zero \( (s \to 0) \), the impedance of the inductors go to zero, and that of the capacitors go to \( \infty \). In physical terms, the inductors become short circuits, while the capacitors become open circuits. The precise relation may be quantified by the use of composition, described in Fig. 1.14 (p. 79). Specifically
\[
\begin{bmatrix} P_1 \\ U_1 \end{bmatrix} = \begin{bmatrix} 1 & sL \\ 0 & 1 \end{bmatrix} \begin{bmatrix} 1 & 0 \\ sC & 1 \end{bmatrix} \cdots \begin{bmatrix} 1 & 0 \\ sL & 1 \end{bmatrix} \begin{bmatrix} 1 & 0 \\ sC & 1 \end{bmatrix} \begin{bmatrix} P_2 \\ -U_2 \end{bmatrix}. \tag{2.6} \]

It seems possible that this is the CFA generalization of the Taylor series expansion, built on composition. If we were to do the algebra we would find that \( A(s), B(s), C(s), D(s) \) (i.e., Sections 1.3.7 (pp. 79, 191)) are ratios of polynomials having rational expansions as Taylor series. This seems like an important observation, that should have applications beyond the engineering literature (Campbell, 1903; Brillouin, 1953; Ramo et al., 1965). Note that Brillouin (1953) credits Campbell (1903).

Figure 2.2: This transmission line (TL) is known as a low-pass filter wave-filter (Campbell, 1922), composed of series inductors having an impedance \( Z_k = sL \) and shunt admittance \( Y_k = sC \). The characteristic resistance of the TL is \( r_0 = \sqrt{sL/sC} = \sqrt{L/C} \) and the wave velocity is \( \kappa = \sqrt{Z_0Y_0} = s\sqrt{L/C} = s/c \). Thus the wave velocity is \( c = 1/\sqrt{LC} \). The length \( \Delta \) [m] defines the physical size of each cell. For long wavelengths the wave-filter acts as a pure delay. But when the frequency increases above a cutoff frequency \( f_c > c/\Delta \), the wavelength \( \lambda = c/f \) is less than the size of a section \( \Delta \), and the system response becomes very high-order low-pass filter.

Figure 2.6 is a piece of wire having a delay determined by its velocity and the length, in units of the cells, each of length \( \Delta \). There are two basic parameters that characterize a transmission line, the characteristic resistance \( r_0 = \sqrt{sL/sC} = \sqrt{L/C} \), and wave number depends on the product of \( L \) and \( C \)
\[
\kappa = \sqrt{sL \cdot sC} = s\sqrt{L/C} = s/c_o, \quad \text{[m}^{-1}] \tag{2.7} \]
thus \( c_o = 1/\sqrt{LC} \). The delay in each cell is
\[
\tau_o = \Delta/c_o = \Delta\sqrt{LC} \quad \text{[s]}. \]

The characteristic wave resistance (aka surge resistance) is given by the ratio of \( L \) and \( C \)
\[
r_0 = \sqrt{L/C} \quad \text{[\Omega]}. \tag{2.8} \]
The total delay of the line is \( T = \tau_0 \) [s], where \( N \) is the number of sections. In that limit the waves travel as

\[
f(t - x/c_0) = e^{-\kappa x} e^{-st},
\]
as \( \Delta \to 0 \). When \( L(x) \) and \( C(x) \) depend on position, the transmission line is called a horn (Sect. 5.1.5, p.218, and described by the Webster Horn equation (Eq.1.113, p. 130).

Since the CFA has a physical representation as a transmission line, as shown in Fig. 2.2, it can be of high utility for the engineer. The theory behind this will be discussed in greater detail in Chapter 5. If you’re ready to jump ahead, read the important book Brillouin (1960) and the collected works of Campbell (1937).

### 2.2.2 Lec 7 Derivation of Euclid’s formula for Pythagorean triplets

Pythagorean triplets (PTs) have many applications in architecture and scheduling, which explains why they are important. For example, if one wished to construct a triangle made from bricks having a perfect 90° angle, then the materials need to be squared off as shown in Fig. 1.8 (p. 46). The lengths of the sides need to satisfy PTs.

**Derivation of Euclid’s formula:** Euclid’s formula (Eq. 1.9, p. 46) provides integer solutions to the Pythagorean theorem (Eq. 1.1, p. 18). The solution method, said to be due to Diophantus, is call a chord/tangent method (Stillwell, 2010, p. 48). The method composes (Section 1.3.5, p. 71) a line and a circle, where the line defines a chord within the circle (its not clear where the tangent line might go). The slope of the line (i.e., denoted \( t \)) is then taken to be rational, allowing one to determine integer solutions of the intersections points. This solution for *Pythagorean triplets* \([a, b, c]\) is known as *Euclid’s formula* (Eq. 1.9, p. 46) (Stillwell, 2010, p. 4–9, 222).

The derivation methods of Diophantus have been lost, but Fermat and Newton figured out what Diophantus must have done (Stillwell, 2010, p. 7, 49, 218). Since Diophantus worked before algebra was invented, he described all the equations in prose (Stillwell, 2010, p. 93).

**Derivation of Euclid’s formula:** The derivation is outlined in Fig. 2.3. Starting from two integers \([p > q \in \mathbb{N}]\), composing a line having a rational slope \( t = p/q > 0 \), with a circle (Stillwell, 2010, p. 6), provides the formula for the Pythagorean triplets.

The construction starts with a circle and a line, which is terminated at the point \((-1, 0)\). The slope of the line is a free parameter \( t \). By composing the circle and the line (i.e., solving for the intersection of the circle and line), the formula for the intersection point \((a, b)\) may be determined in terms of \( t \), which will then be taken as the rational slope \( t = p/q \in \mathbb{Q} \).

In Fig. 2.3 there are three panels, two labeled “Proofs.” The *Euclidean Proof* shows the angle relationships of two triangles, the first an isosceles triangle formed by the chord, having slope \( t \) and two equal sides formed from the radius of the circle, and a second right triangle having its hypotenuse as the radius of the circle and its right angle vertex at \((a, 0)\). As shown, it is this smaller right triangle that must satisfy Eq. 1.1. The inner right triangle has its hypotenuse \( c \) between the origin of the circle (O) to the point \((a, b)\). Side \( a \) forms the \( x \) axis and side \( b \) forms the \( y \) ordinate. Thus by construction Eq. 1.1 must be obeyed.

---

6Continued fraction expansions of functions are known in the circuit theory literature as a Cauer synthesis (Van Valkenburg, 1964b).
Euclid’s formula for Pythagorean triplets \([a, b, c]\)  

### Euclidean Proof:

1) \(\phi + \eta = \pi\)  
2) \(\eta + \Theta = \pi\)  
3) \(\therefore \phi = \Theta/2\)

Two lines in the plane of Pythagoras:

- Line \(b \parallel \text{X-axis}\) goes through the points \((-c, 0)\) and \((a, b)\).
- Line \(c^2 = a^2 + b^2\) is a circle.

Composing the line with the circle gives:

\[
a^2 + (t(a + c))^2 = c^2 \\
a^2 + t^2(a^2 + 2ac + c^2) = c^2 \\
(1 + t^2)a^2 + 2ct^2a + c^2(t^2 - 1) = 0
\]

This last equation is a quadratic equation in \(a\). In some sense it is not really a quadratic equation, since we know that \(a = -c\) is a root. Dividing by \(1 + t^2\)

\[
a^2 + \frac{2ct^2}{1 + t^2}a + \frac{c^2(t^2 - 1)}{1 + t^2} = 0,
\]

makes it easier to complete the square, delivering the roots:

\[
\left(a + \frac{ct^2}{1 + t^2}\right)^2 - \left(a + \frac{ct^2}{1 + t^2}\right)^2 + \frac{c^2(t^2 - 1)}{1 + t^2} = 0
\]

\[
\left(a + \frac{ct^2}{1 + t^2}\right)^2 - \frac{c^2t^4}{(1 + t^2)^2} + \frac{c^2(t^2 - 1)(t^2 + 1)}{(1 + t^2)^2} = 0
\]

\[
\left(a + \frac{ct^2}{1 + t^2}\right)^2 - \frac{c^2t^4 + c^2(t^2 - 1)}{(1 + t^2)^2} = 0
\]

\[
\left(a + \frac{ct^2}{1 + t^2}\right)^2 = \left(\frac{c}{1 + t^2}\right)^2
\]

The second to last equation simplifies (magic happens) because the known root \(a = -c\) is embedded in the result.

### Diophantus’s Proof:

1) \(c^2 = a^2 + b^2\)  
2) \(b(a) = t(a + c)\)  
3) \(\zeta(t) \equiv a + jb = \frac{1-t^2+2it}{1+t^2}\)  
4) \(\zeta = |c|e^{i\theta} = |c|\frac{1+it}{1-it} = |c|(\cos(\theta) + i \sin(\theta))\)

### Pythagorean triplets:

1) \(t = p/q \in \mathbb{Q}\)  
2) \(a = p^2 - q^2\)  
3) \(b = 2pq\)  
4) \(c = p^2 + q^2\)

### Figure 2.3: Derivation of Euclid’s formula for the Pythagorean triplets \([a, b, c]\), based on a composition of a line, having a rational slope \(t = p/q \in \mathbb{Q}\), and a circle \(c^2 = a^2 + b^2\), \([a, b, c]\) \(\in \mathbb{N}\). This analysis is attributed to Diophantus (Di-ophan-tus) (250 CE), and today such equations are called Diophantine (Di-ophan-tine) equations. PTs have applications in architecture and scheduling, and many other practical problems.
Taking the square root gives the two roots
\[
\begin{align*}
a_\pm &+ \frac{ct^2}{1+t^2} = \pm \frac{c}{1+t^2} \\
(1+t^2)a_\pm &= -ct^2 \pm c = -c(t^2 \mp 1) \\
a_\pm &= -c \frac{t^2 \mp 1}{1+t^2}.
\end{align*}
\]

The known root is \(a_+ = -c\), because when the sign is +, the numerator and denominator terms cancel.

The root we have been looking for is
\[
a_- = c \frac{1-t^2}{1+t^2},
\]

which allows us to solve for \(b_-\)
\[
\begin{align*}
b_- &= \pm \sqrt{c^2 - a_-^2} \\
&= \pm c \sqrt{1 - \left(\frac{1-t^2}{1+t^2}\right)^2} \\
&= \pm c \sqrt{(1+t^2)^2 - (1-t^2)^2} \\
&= \pm \frac{2ct}{t^2 + 1}.
\end{align*}
\]

Therefore the coordinates \((a, b)\), the intersection point of the line and circle, are
\[
(a(t), b(t)) = c \frac{[1-t^2, 2t]}{1+t^2}.
\]

To obtain the Pythagorean triplets, as given in Fig. 2.3 (p. 171) and Eq. 1.9 (p. 46), set \(t = p/q\), assuming \(p > q \in \mathbb{N}\), and simplify.

**Complex roots:** Defining the root as a complex number \(\zeta(\Theta) \equiv a + bj\) forces \(a \perp b\) (i.e., forces the right triangle) and gives us polar coordinates, as defined by the figure as the Euclidean Proof
\[
\zeta(\Theta) = |c|e^{\Theta j} = |c| (\cos(\Theta) + j \sin(\Theta)).
\]

This naturally follows since
\[
\zeta = |c|e^{\Theta(j)} = |c| \frac{1-t^2 + 2jt}{1+t^2} = |c| \frac{(1+jt)(1+jt)}{(1+tj)(1-tj)} = (q+pj)\frac{q+jp}{q-pj}.
\]

Examples of PTs include \(a = 2^2 - 1^2 = 3, b = 2 \cdot 2 \cdot 1 = 4\), and \(c = 2^2 + 1^2 = 5, 3^2 + 4^3 = 5^2\).

Defining \(p = q+N (p, q, N \in \mathbb{N})\) gives slightly better parametric representation of the answers, as the pair \((q, N)\) are a more systematic representation than \((p, q)\), because of the condition \(p > q\), so the general properties of the solutions are expressed more naturally. Note that \(b + c\) must always be a perfect square since \(b + c = (p + q)^2 = (2q + N)^2\), as first summarized by Fermat (Stillwell, 2010, p. 212).
2.2. WEEK 3

2.2.3 Lec 8 Pell’s Equation

Calculation to show that Eqs. 1.10 and 1.11 are related. Starting from Eq. 1.10

\[ x_n^2 = (x_{n-1} + Ny_{n-1})^2 = x_{n-1}^2 + 2Nx_{n-1}y_{n-1} + N^2y_{n-1}^2 \]  \tag{2.9}

and

\[ Ny_{n}^2 = N(x_{n-1} + y_{n-1})^2 = x_{n-1}^2 + 2Nx_{n-1}y_{n-1} + Ny_{n-1}^2 \]  \tag{2.10}

Subtracting removes the cross term leaving

\[ x_n^2 - Ny_{n}^2 = (1 - N)x_{n-1}^2 - N(1 - N)y_{n-1}^2 = (1 - N)(x_{n-1}^2 - Ny_{n-1}^2). \]

Using Eq. 1.12, the actual formula that follows from Eq. 1.11

\[ (-1)^n = (1 - N)(-1)^{n-1}, \]

thus (take \( n \) either even or odd) gives \( N = 2 \). Wallah.

**Eigen-value solution to Pell’s equation (\( N = 2 \))**: To provide a full understanding of what was known to the Pythagoreans about irrational numbers, it is helpful to provide the full solution to Eq. 1.11.

As shown in Fig. 1.10 (p. 48), \((x_n, y_n)\) may be written as a power series of the 2x2 matrix \( A \). The well known modern approach to find \( A^n \) is to diagonalize the matrix as detailed in Appendix C (p. 245). For the 2x2 matrix case, this is relatively simple. The final result written out in detail for the general solution \((x_n, y_n)\) (Appendix D, p. 247):

\[
\begin{bmatrix}
x_n \\
y_n
\end{bmatrix} = f^n \begin{bmatrix}
1 & 2 \\
1 & 1
\end{bmatrix} \begin{bmatrix}
\lambda^+_n \\
0
\end{bmatrix} = E \begin{bmatrix}
\lambda^+_n & 0 \\
0 & \lambda^-_n
\end{bmatrix} E^{-1} \begin{bmatrix}
1 \\
0
\end{bmatrix}. \tag{2.11}
\]

The eigen-values are \( \lambda_{\pm} = f(1 \pm \sqrt{2}) \) while the eigen-matrix (Eq. C.1) and its inverse are

\[
E = \frac{1}{\sqrt{3}} \begin{bmatrix}
\sqrt{2} & -\sqrt{2} \\
1 & 1
\end{bmatrix} = \begin{bmatrix}
0.8165 & 0.8165 \\
0.5774 & -0.5774
\end{bmatrix}, \quad E^{-1} = \frac{\sqrt{3}}{2\sqrt{2}} \begin{bmatrix}
1 & \sqrt{2} \\
1 & -\sqrt{2}
\end{bmatrix} = \begin{bmatrix}
0.6124 & 0.866 \\
0.6124 & -0.866
\end{bmatrix}
\]

The relative “weights” on the two eigen-solutions are equal, as determined by

\[
E^{-1} \begin{bmatrix}
1 \\
0
\end{bmatrix} = \frac{\sqrt{3}}{2\sqrt{2}} \begin{bmatrix}
1 & \sqrt{2} \\
1 & -\sqrt{2}
\end{bmatrix} \begin{bmatrix}
1 \\
0
\end{bmatrix} = \frac{\sqrt{3}}{2\sqrt{2}} \begin{bmatrix}
1 \\
1
\end{bmatrix}.
\]

We still need to prove that

\[
\frac{x_n}{y_n} \to \sqrt{N}.
\]

This follows intuitively from Pell’s equation, since as \((x_n, y_n) \to \infty\), the difference between \(x_n^2\) and \(2y_n^2\), the \((\pm 1)\) becomes negligible.

Given the development of linear algebra c19th century, this may be evaluated by eigen-vector diagonalization.\(^7\)

\(^7\)https://en.wikipedia.org/wiki/Transformation_matrix#Rotation
**Pell’s Equation and irrational numbers:** Since the eigen-values of Eq. 1.11 ($\lambda_{\pm} = 1 \mp \sqrt{N} \notin \mathbb{N}$), solutions to Pell’s equation raised the possibility that all numbers are not rational. This discovery of irrational numbers forced the jarring realization that the Pythagorean dogma “all is integer” was wrong. The significance of irrational numbers was far from understood.

**Derivation of Euclid’s formula for the Pell equation**

In this section we derive Euclid formula for the Pell’s equation, following the same reasoning as for the case of the Pythagorean triplets Eq. 1.9, following the logic described in Fig. 2.3.

**Proof I:**

1) $x^2 - Ny^2 = 1$
2) $y = t(x - 1)$
3) $(x, y) = \left(\frac{Nt^2 + 1, 2t}{Nt^2 - 1}\right)$
4) $t = p/q$

Choose $q, n, N \in \mathbb{N}, p = q + n, r = Np^2 - q^2$

$x = (Np^2 + q^2)/r, y = 2pq/r, \tan(\theta) = 2qp/(Np^2 - q^2)$

**Demo:** EvalPellEq.m

**Figure 2.4:** Derivation of general solution to Pell’s equation, using a chord/tangent composition.

The problem is to find integer solutions for $x, y \in \mathbb{N}$ to Pell’s equation (Eq. 1.10) for any $N \in \mathbb{N}$. The derivation is depicted in Fig. 2.4. Following the chord/tangent method due to Diophantus, we must compose a line (dashed) with a hyperbola, as shown in Fig. 2.4. The line must have a rational slope $t \in \mathbb{Q}$ and must pass through the trivial root $(1, 0)$. The insert “Proof I” gives (1) Pell’s equation, (2) the equation of the dashed line, the intersection point (3) $(x, y)$ and (4) the equation for the rational slope $t (p, q \in \mathbb{Q})$. The step by step solution is given in Fig. 2.4.

The final solution, in terms of integers $[p, q], p > q$ is

$$x = \frac{Np^2 + q^2}{Np^2 - q^2}, \quad y = \frac{2pq}{Np^2 - q^2}. \quad (2.12)$$

Written as a complex numbers: $z = x + iy = (Np^2 + q^2 - j2qp)/(Np^2 - q^2)$

**WEEK 4**
2.3 Week 4

2.3.1 Lec 9 Eigen-Analysis of Fibonacci sequence

The Fibonacci sequence is famous in number theory. It is said that the sequence commonly appears in physical systems. Fibonacci numbers are related to the “golden ratio” \((1 + \sqrt{5})/2\).

But from a mathematical point of view, the Fibonacci sequence does not seem special. It is generated by a linear recursion relationship, where the next number is the sum of the previous two:

\[ x_{n+1} = x_n + x_{n-1}. \tag{2.13} \]

The term linear means that the principle of superposition holds [P1 (linear/nonlinear) of Section 1.3.13, p. 95] and 3.5.1, p. 205]. To understand the meaning of this we need to explore the z-transform, the discrete-time version of the Laplace transform. We will return to this in Chapter 4 (p. 213).

A related linear recurrence relation is to define the next output \(x_{n+1}\) be the average of the previous two:

\[ x_{n+1} = \frac{x_n + x_{n-1}}{2}. \]

In some ways this relationship, which has no famous name, is more useful than the famous Fibonacci recursion, since it perfectly removes oscillations of the form \((-1)^n\) (it is a 2-sample moving average, a trivial form of low-pass filter). Unlike the Fibonacci sequence, which diverges, it seems that biology prefers divergent sequences.

Equation 2.13 may be written as a 2x2 matrix relationship. If we define \(y_{n+1} = x_n\), then Eq. 2.13 is equivalent to (Eq. 1.16, p. 50)

\[
\begin{bmatrix}
  x_{n+1} \\
  y_{n+1}
\end{bmatrix} =
\begin{bmatrix}
  1 & 1 \\
  1 & 0
\end{bmatrix}
\begin{bmatrix}
  x_n \\
  y_n
\end{bmatrix}.
\tag{2.14}
\]

The first equation is \(x_{n+1} = x_n + y_n\), while the second is \(y_{n+1} = x_n\), which is the same as \(y_n = x_{n-1}\). Note that the Pell 2x2 recursion is similar in form to the Fibonacci recursion. This removes mystique from both equations.

The most general 2d order recurrence relationships is

\[ x_{n+1} = -bx_n - cx_{n-1}, \]

with constants \(b, c \in \mathbb{R}\).

In the matrix diagonalization of the Pell equation we found that the eigen-values were \(\lambda_{\pm} = 1 \mp \sqrt{N}\), and the two solutions turned out to be powers of the eigen-values. The solution to the Fibonacci recursion may similarly be expressed in terms of a matrix. These two cases may thus be reduced by the same 2x2 eigen-value solution method.

The eigen-values of the Fibonacci matrix are

\[
det \begin{bmatrix}
  1 - \lambda & 1 \\
  1 & -\lambda
\end{bmatrix} = \lambda^2 - \lambda - 1 = (\lambda - 1/2)^2 - (1/2)^2 - 1 = (\lambda - 1/2)^2 - 5/4 = 0
\]

thus \(\lambda_{\pm} = \frac{1 \pm \sqrt{5}}{2} = [1.618, -0.618]\).
General properties of the Fibonacci numbers\(^a\)

\[ x_n = x_{n-1} + x_{n-2} \]

- This is a 2-sample moving average difference equation with an unstable pole
- \( x_n = [0, 1, 1, 2, 3, 5, 8, 13, 21, 34, \cdots] \), assuming \( x_0 = 0, x_1 = 1 \):
- Analytic solution (Stillwell, 2010, p. 194):
  \[ \sqrt{5} \ x_n = (\frac{1+\sqrt{5}}{2})^n - (\frac{1-\sqrt{5}}{2})^n \to \left(\frac{1+\sqrt{5}}{2}\right)^\infty \]
  - \( \lim_{n \to \infty} \frac{x_{n+1}}{x_n} = \frac{1+\sqrt{5}}{2} \)
  - Example: \( 34/21 = 1.6190 \approx \frac{1+\sqrt{5}}{2} = 1.6180 \) 0.10% error
- Matlab’s \( \text{rat}(1 + \sqrt{5}) = 3 + 1/(4 + 1/(4 + 1/(4 + 1/(4 + 1/(4 + \cdots))))) =: [3; 4,4,\cdots] \)

\(^a\)https://en.wikipedia.org/wiki/Fibonacci_number

Figure 2.5: Properties of the Fibonacci numbers (Stillwell, 2010, p. 28).

Note that this 2x2 equation is similar to Pell’s equation, suggesting that an eigen-function expansion of Eq. 1.16 may be used to analyze the sequence, as shown in Section 1.2.8 (p. 49) (Stillwell, 2010, 192). It is also related to a $10,000 prize, that was eventually solved.\(^8\)

2.3.2 Lec 10 Exam I

Chapter 3

Algebraic Equations: Stream 2

3.1 Week 4

3.1.1 Lec 11 Algebra and geometry as physics

Before Newton could work out his basic theories, algebra needed to be merged with Euclidean geometry. The key to putting geometry and algebra together is the Pythagorean theorem (Eq. 1.1), which is both geometry and algebra. To make the identification with geometry the sides of the triangle needed to be viewed as a length. This is done by recognizing that the area of a square is the square of a length. Thus a geometric proof requires one to show that the area of the square \( A = a^2 \) plus the area of square \( B = b^2 \) must equal the area of square \( C = c^2 \). There are many such constructions that show \( A + B = C \) for the right triangle. It follows that in terms of coordinates of each vertex, the length of \( c \) is given by

\[
c = \sqrt{(x_2 - x_1)^2 + (y_2 - y_1)^2},
\]

(3.1)

with \( a = x_2 - x_1 \) and \( b = y_2 - y_1 \) (Fig. 1.8, p. 46). Thus Eq. 1.1 is both an algebraic and a geometrical statement. This is not immediately obvious.

Analytic geometry is based on coordinates of points, where the length given by Eq. 3.1. Euclidean Geometry defines lengths without specifying the coordinates (Eq. 1.1). Algebra gave a totally new view to the quantification of lengths by introducing a coordinate system. This resulted in an entire new way to work with conic sections, which were now explained in terms of equations in coordinate systems. When viewed through the lens of algebra, Eq. 1.1 is a circle having radius \( c \). Complex numbers provide an equivalent representations, since if \( z = x + iy \), the unit circle is \( z = e^{i\theta} \) and \( |z|^2 = x^2 + y^2 \). Here we explore the relationships between points, represented as coordinates, describing geometrical objects. We shall do this with simple examples from analytic geometry.

For example, in terms of the geometry, the intersection of two circles can occur at two points, and the intersection of two spheres gives a circle. These ideas may be verified using algebra, but in a very different, since the line can traverse through the circle, like a piece of thread going through the eye of a needle. In such cases the intersections are complex intersections.

For each of these problems, the lines and circles may intersect, or not, depending on how they are drawn. Yet we now know that even when they do not intersect on the sheet of paper, they still have an intersection, but the solution is complex. Finding such solutions require the use of algebra rather than geometry. These ideas were in the process of being understood, first by Fermat and Descartes, then by Newton, followed by the Bernoulli family and Euler.
CHAPTER 3. ALGEBRAIC EQUATIONS: STREAM 2

Complex analytic functions: A very delicate point, that seems to have been ignored for centuries, is that the roots of $P_n(x)$ are, in general, complex, namely $x_k \in \mathbb{C}$. It seems a mystery that complex numbers were not accepted once the quadratic equation was discovered, but they were not. Newton called complex roots imaginary, presumably in a pejorative sense. The algebra of complex numbers is first attributed to Bombelli in 1575, more than 100 years before Newton. One can only begin to imagine what Bombelli learned from Diophantus, following his discovery of Diophantus’ Arithmetica in the Vatican library (Stillwell, 2010, p. 51).

It is interesting that Newton was using power series with fractional degree. These topics will be explored in Section 3.1.1.

When the argument is complex, analytic functions takes on an entirely new character. For example Euler’s identity (1748) with $z = x + yj \in \mathbb{C}$ results in

$$e^z = e^x (\cos(y) + j\sin(y)).$$

It should be clear that the complex analytic functions results in a new type of algebra, with no further assumptions beyond allowing the argument to be complex.

Prior to 1851 most of the analysis assumed that the roots of polynomials were real ($x_k \in \mathbb{R}$), even though there was massive evidence that they were complex ($x_n \in \mathbb{C}$). This is clearly evident in Newton’s work (c1687): When he found a non-real root, he ignored it (Stillwell, 2010, pp. 115-7). Euler (c1748) first derived the Zeta function as a function of real arguments $\zeta(x)$ with $\zeta, x \in \mathbb{R}$.

Cauchy (c1814) broke this staid thinking with his analysis of complex analytic functions, but it was Riemann thesis (c1851), when working with Gauss (1777-1855), which had a several landmark breakthroughs. In this work Riemann introduced the extended complex plane, which explained the point at infinity. He also introduced Riemann sheets and Branch cuts, which finally allowed mathematics to better describe the physical world (Section 1.4.2).

The development of complex analytic functions led to many new fundamental theorems. Complex analytic functions have poles and zeros, branch cuts, Riemann sheets and can be analytic at the point at infinity. Many of these properties were first worked out by Augustin-Louis Cauchy (1789-1857), who drew heavily on the much earlier work of Euler, expanding Euler’s ideas into the complex plane (Chapter 4).

Systems of equations

We don’t need to restrict ourselves to polynomials in one variable. For example, we can work with the circle

$$y^2 + x^2 = r^2,$$

which is quadratic in $x$, $y$. Solving for roots $y(x_r) = 0$ ($y^2(x_r) = r^2 - x_r^2$) gives $(r - x_r)(r + x_r) = 0$, which simply says that when the circle crosses the $y = 0$ line at $x_r = \pm r$.

This equation may also be factored as

$$(y - x)(y + x) = r^2,$$

as is easily demonstrated by multiplying out the two monomials. This does not mean that a circle has complex roots. A root is defined by either $y(x_r) = 0$ or $x(y_r) = 0$.

Writing the conic as a 2d degree polynomial gives

$$P_2(x) = ax^2 + bx + c,$$
with \( y^2(x) = -P_2(x) \). Setting this equal to zero and completing the square (Eq. 1.26, p. 55), gives the equation for the roots

\[
\left( x \pm \frac{b}{2a} \right)^2 - \left( \frac{b}{2a} \right)^2 + \frac{c}{a} = 0,
\]

or

\[
x_\pm = -\frac{b}{2a} \pm \sqrt{\left( \frac{b}{2a} \right)^2 - \frac{c}{a}}.
\]

The polynomial in factored form is

\[
y^2 = -\left( x - \frac{b}{2a} \right)^2 + \left( \frac{b}{2a} \right)^2 - \frac{c}{a},
\]

is a conic section, and becomes a circle with \( a = 1, b = 0 \) and \( c = -r^2 \).

### 3.1.2 Lec 11a The physics of complex analytic expressions: linear vs. non-linear

The question we address here is "When do multi-variable complex analytic expressions appear in physics?" The most common example comes from the solution of the wave equation (Eq. 1.116) in three dimensions. Such cases arise in wave-guide problems, semiconductors, plasma waves, or for acoustic wave propagation in crystals (Brillouin, 1960) and the earth’s mantle (e.g., seismic waves, earthquakes, etc.). The solutions to these problems are based on the eigenfunction for the vector wave equation (see Chapter 5),

\[
P(s, \mathbf{x}) = e^{st} e^{-\kappa \cdot \mathbf{x}},
\]

where vector \( \mathbf{x} = [x \hat{x} + y \hat{y} + z \hat{z}] \in \mathbb{R}^3 \) points in the direction of the wave, unit vectors \([\hat{x}, \hat{y}, \hat{z}]\) are \( \in \mathbb{R}^3 \) and \( s = \sigma + \omega j \in \mathbb{C} \) [rad] is the Laplace frequency. The vector function \( \kappa(s) \in \mathbb{R}^3 \) is the complex vector wave number, which describes the propagation of a plane wave of angular frequency \( \omega \), in the \( x \) direction.

Just as the frequency \( s = \sigma + \omega j \) must be complex, it is important to allow \( \kappa(s) \)\(^1\) to be a complex function of \( s \). The imaginary part accounts for the delay and the real part accounts for losses, as the wave propagates. While it is common to assume there are no losses (real part of \( \kappa = 0 \)), but in reality this assumption cannot be correct. In many cases zero loss is an excellent approximation, that gives realistic answers. But it is important to start the analysis with a notation that accounts for the most general situation, so that when losses are present, the notation doesn’t need to change. With this in mind, we take the vector wave number to be complex

\[\kappa = k_r + k_j,\]

where vector expression for the lattice vector is the imaginary part of \( \kappa \)

\[
\Im \kappa = \frac{2\pi}{\lambda_x} \hat{x} + \frac{2\pi}{\lambda_y} \hat{y} + \frac{2\pi}{\lambda_z} \hat{z},
\]

\(^1\)This function has many names in the literature, such as the wave number, propagation constant. However, its neither a number nor constant. In the case of Brillouin zones, it is the electron wave dispersion function (Brillouin, 1953, Ch. 1).
is the vector wave number for three dimensional lossless plane-wave solutions \((\kappa_r = 0)\).

Equation 3.3 is linear in \(x\). If one takes the derivative of Eq. 3.2 with respect to either time or space,

\[
\frac{\partial}{\partial t} e^{st} e^{-\kappa \cdot x} = s e^{st} e^{-\kappa \cdot x}, \quad \frac{\partial}{\partial x} e^{st} e^{-k_x x} = \frac{2\pi}{\lambda_x} e^{st} e^{-k_x x}, \quad \nabla e^{st} e^{-\kappa \cdot x} = \kappa e^{st} e^{-\kappa \cdot x}
\]

we find the eigenvalue of that derivative.

The units of \(\kappa\) are reciprocal length \([m^{-1}]\) since \(\kappa \cdot x\) has units of radians. When there are losses \(\kappa_r(s) = \Re \kappa(s)\) must be a function of frequency, due to the physics behind these losses. In many important cases, such as loss-less wave propagation in semiconductors, \(\kappa \cdot x\) is a function of direction and position (Brillouin, 1960).

When the eigenfunction Eq. 3.2 is applied to the wave equation, a quadratic (degree 2) algebraic expression results, known as the dispersion relation. The three dimensional dispersion relation

\[
\left(\frac{s}{c}\right)^2 = \kappa \cdot \kappa = \left(\frac{2\pi}{\lambda_x}\right)^2 + \left(\frac{2\pi}{\lambda_y}\right)^2 + \left(\frac{2\pi}{\lambda_z}\right)^2 = k_x^2 + k_y^2 + k_z^2
\]

(3.4)

is a complex analytic algebraic relationship in four variables, frequency \(s\) and the three complex lattice wave numbers. This represents a three-dimensional generalization of the well know relation between wavelength and frequency, i.e., \(f \lambda = c\). For lossless plane waves propagating in free space, \(|\kappa(s)| = \pm |s/c|\), where the sign accounts for the direction of the plane wave.

This scalar relation \((f \lambda = c)\) was first deduced by Galileo in the 16th century and was then explored further by Mersenne a few years later. This relationship would have been important to Newton when formulating the wave equation, which he needed to estimate the speed of sound. We shall return to this in Chapters 4 and 5.

3.2 Week 5

3.2.1 Lec 12 Polynomial Factoring by deconvolution: degree > 3

Root classification for polynomials of Degree \(* = 1–4\) (p.102);
Quintic \((* = 5)\) cannot be solved: Why?
Fundamental Thm of Algebra (d’Alembert, \(\approx 1760\))

**Inversion of polynomial convolution:** As we discussed in Chapter 1, given the roots, the construction of higher degree polynomials, is greatly assisted by the convolution method. This has physical meaning, and gives insight into the problem of factoring higher order polynomials. By this method we can obtain explicit relations for the coefficients of any polynomial in terms of its roots, as shown below for the cubic case.

Extending the example of Section 1.3.3, let’s find the relations between the coefficients and the roots for the cubic. To simplify the notation, assume that the polynomial has been normalized so that the lead \(x^3\) term has coefficient 1. Then the cubic in terms of its roots \([a, b, c]\) is a convolution of three terms

\[
[1, a] \star [1, b] \star [1, c] = [1, a + b, ab] \star [1, c] = [1, a + b + c, ab + c(a + b), abc].
\]
Thus in terms of its roots, the polynomial is

\[ x^3 + (a + b + c)x^2 + (ab + ca + cb)x + abc = 0. \]

As a second example the coefficients for a \textit{quartic} gives

\[
[1, a + b + c, ab + c(a + b), abc] \ast [1, d] = \\
[1, a + b + c + d, d(a + b + c) + c(a + b) + ab, d(ab + ac + bc) + abc, abcd].
\]

thus

\[
x^4 + (a + b + c + d) x^3 \\
+ [d(a + b + c) + c(a + b) + ab] x^2 \\
+ [d(ab + ac + bc) + abc] x \\
+ abcd x^0 = 0.
\]

It is clear what is going on here. The coefficient on \(x^4\) is 1 (by construction). The coefficient for \(x^3\) is the sum over the roots. The \(x^2\) term is the sum over all possible products of pairs of roots. The \(x\) term is the sum over all triple products of the four roots. The \(x^0 = 1\) term is the product of the four roots.

In fact this is a well known, a frequently quoted result from the mathematical literature, and trivial to show given an understand of convolution. If one wants the coefficients for the quintic, it is not even necessary to use convolution, as the pattern (rule) for all the coefficients is now clear.

You can experiment with this numerically using Matlab’s convolution routine \texttt{conv(a,b)}. Once we start studying Laplace and Fourier transforms, convolution becomes critically important because

\[ f(t) \ast g(t) \leftrightarrow F(s)G(s). \]

So you didn’t need to learn how to take a Laplace transform, and then learn convolution. We have learned convolution first, independent of the Fourier and Laplace transforms.

For the case of the quadratic \((N = 2)\) we have the relations between the coefficients and the roots, found by completing the square. This required isolating \(x\) to a single term, and solving for it. We then proceeded to find the coefficients for the cubic \(N = 3\) and quartic \(N = 4\) case, after a few lines of calculation. For the quartic

\[
a_4 = 1 \\
a_3 = a + b + c + d \\
a_2 = d(a + b + c) + c(a + b) + ba \\
a_1 = d(ab + ac + bd) + abc \\
a_0 = abcd
\]

These relationships are algebraically nonlinear in the roots.

Based on the work of Galois, the system quartic \((N = 5)\), the equations are impossible to invert. Namely, given \(a_k\), one may not determine the five roots \([a, b, c, d, 3]\) analytically.
To gain some insight, let us look at the problem for $N = 2$, which of course has a closed form solution:

\[
\begin{align*}
a_2 &= 1 \\
a_1 &= a + b \\
a_0 &= ab
\end{align*}
\]

We must solve for $[a, b]$ given twice the mean, $2(a + b)/2$, and the square of the geometric mean \((\sqrt{ab})^2\). Since we already know the answer (i.e., the quadratic formula). The solution was first worked out by the Babylonians (2000 BCE) Stillwell (2010, p. 92). It is important to recognize that for physical systems, the coefficients $a_k$ are real. This requires that the roots come in conjugate pairs ($b = a^*$), thus $ab = |a|^2$ and $a + b = 2\Re a$, which makes the problem somewhat more difficult, due to the greater symmetry.

Once you have solved this problem, feel free to attempt the cubic case. Again, the answer is known, after thousands of years of searching. The solution to the cubic is given in (Stillwell, 2010, pp. 97-9), as discovered by Cardano in 1545. According to Stillwell “The solution of the cubic was the first clear advance in mathematics since the time of the Greeks.” The ability to solve this problem required algebra, and the solutions were complex numbers. The denial of complex numbers was, in my view, the main stumbling block in the progress of these solutions. For example, how can two parallel lines have a solution? Equally mystifying, how can a circle and a line, that do not intersect, have intersections? From the algebra we know that they do. This was a basic problem that needed to be overcome. This story is still alive, because the cubic solution is so difficult.

One can only begin to imagine how much more difficult the quartic is, solved by Cardano’s student Ferrair, and published by Cardano in 1545. The impossibility of the quintic was finally resolved in 1826 by Able (Stillwell, 2010, p. 102).

Finally with these challenges behind them, Analytic Geometry, relating of algebra and geometry, via coordinate systems, was born.

### 3.2.2 Lec 13 Transmission line impedance

### 3.2.3 Lec 14: Development of Hilbert Space geometry

**Composition, Intersection and Gaussian elimination:**

When the impedance load, as a function of wave penetration depth $x$ is uniform, then as the wave $P^+$ travels, then there are no reflections. In this case $\Gamma(s, x = 0) = 0$, and the impedance is equal to $r_o$. Given variations in the properties of the medium, say after a delay of $\tau = x/c$, backward propagated waves are returned to the input. This results in a change in the load impedance, both in real and imaginary parts. As long as $|\Gamma(s, x)| \leq 1$, the real part of $Z(s, x = 0 \geq 0$. This is further discussed in Sect. 1.5.7 (p. 138).

As a counter example consider the well known problem in geometry: the intersection of a plane with a cone, which leads to the conic sections: the circle, hyperbola, ellipse and parabola, along with some degenerate cases, such as the intersection of two straight lines. If we stick to such problems were first studied algebraically and Descartes (Stillwell, 2010, p. 118) and Fermat (c1637).
3-dimensional objects, we can write equations in the three variables $[x, y, z]$, and be sure that they each represent some physical geometry. For example $x^2 + y^2 + z^2 = r^2$ is a sphere of radius $r_o$.

The geometry and the algebra do not always seem to agree. Which is correct? In general the geometry only looks at the real part of the solution, unless you know how to tease out the complex solutions. However the roots of every polynomial can be in $\mathbb{C}$, so we may not ignore the imaginary roots, as Newton did. There is an important related fundamental theorem, known as Bézout’s theorem, that address this case, as described at the end of Section 1.3.5, 77.

**Composition of polynomials: Bézout’s theorem**

When we multiply two polynomials, the resulting degree is the sum of the degrees of the individual polynomials (Section 1.3.4, p. 70). However when we compose two polynomials, of degree $m$ and $n$, the resulting degree is the *product* of the degrees. For example given two polynomials of degree $n$ and $m$, $f_n(x)$ and $g_m(x)$, their composition is defined as $f_n \circ g_m = f(g(x))$. For example, the composition of a line with a line, is a line. The composition of a line with a circle, is a circle, and the composition of a circle with a circle is a quartic.

This is known as Bézout’s theorem, which states the number of roots of the composition of two functions is determined by the product of their degrees. The power of Bézout’s theorem is that it applies to functions of more than one variable. Composition of functions is also known as the *construct of equations* (Stillwell, 2010, p. 118). It was proved by Bézout (1779), who showed how one must count all the roots, including multiple roots, and roots at infinity (Stillwell, 2010, p. 295).

**Exercises:**

1. **Two Circles:** The composition of 2 independent circles give 4 points:

   \[ x^2 + y^2 = 1, \quad (x - 1)^2 + y^2 = 1 \]

   Two circles intersect at 2 point: How does this work on using algebra?

   Here we subtract the two equations, eliminating $y$

   \[ x^2 - (x - 1)^2 = x^2 - (x^2 - 2x + 1) = 2x - 1 = 0 \]

   Thus $x = 1/2$. Verifying gives

   \[ (1/2)^2 + y^2 = 1, \quad (1/2 - 1)^2 + y^2 = 1 \]

   or

   \[ y^2 = 1 - 1/4 = 3/4, \quad y^2 = 1 - 1/2 = 1/2 \]

   or

   \[ y = \pm \sqrt{3/4}, \quad y = \pm \sqrt{1/2}. \]

   In summary, there is only one value of $x = 1/2$ but there are two values of $y$, which is independent of the sign, since $y$ appears in the equations as $y^2$. Thus there are *four* solutions: $(1/2, \pm \sqrt{3/4})$ and $(1/2, \pm \sqrt{1/2})$. 

Totally rework
2. Composition of a line and a circle: clearly state the problem

Let the circle be \( x^2 + y^2 = 1 \) and the line be \( x + y = 1 \). Graphically it is easy to show that the solutions are \((0, 1)\) and \((1, 0)\). This problem has already been addressed with the search for Pythagorean triplets. In Fig. 2.3 we derive the real integer solutions for the line from \(-1, 0\) to \(x, y\) intersection the circle. The values of \(a, b, c\) are presented as Proof III in that figure.

But what can we say about the angles \(\Theta\)? The formula for the angles is

\[
e^{\imath \Theta} = \frac{1 + t}{1 - t}
\]

with \(t = p/q \in \mathbb{Q}\).

If we solve this equation for \(t\) we find

\[
(1 - t) e^{\imath \Theta} = 1 + t
\]

\[
(e^{\imath \Theta} - 1) = t(e^{\imath \Theta} + 1)
\]

\[
t = \frac{e^{\imath \Theta} - 1}{e^{\imath \Theta} + 1} = \frac{e^{\imath \Theta/2} - e^{-\imath \Theta/2}}{e^{\imath \Theta/2} + e^{-\imath \Theta/2}}
\]

3. Problem 3: Suppose we have two circles

\[
y^2 + x^2 = 1 \quad y^2 + x^2 = 1 \quad (3.6)
\]

There are an infinite number of solutions since the circles lie on top of each other. All values of \(x, y\) are a solution. If

\[
y^2 + x^2 = 1 \quad y^2 + x^2 = 2 \quad (3.7)
\]

there are no solutions, as the circles never touch. When

\[
y^2 + x^2 = 1 \quad y^2 + (x - 1)^2 = 1 \quad (3.8)
\]

there are two intersection points. How can we find them?

We may remove \(y\) from these equations by subtracting them, which results in

\[
(x - 1)^2 - x^2 = 0
\]

which is easily solved since \(x^2 - 2x + 1 - x^2 = 0, x = 1/2\). We can then put this back into the first equation to find \(y^2 + 1/4 = 1\) or \(y = \pm \sqrt{3/4}\). This is easily verified.

What if the two circles are

\[
y^2 + x^2 = 1 \quad y^2 + (x - 1)^2 = 1 \quad (3.9)
\]

Removing \(y\) as before gives \((x - 1)^2 - x^2 = 0\) or \(x^2 - 2x + 1 - x^2 = 0\). Thus \(x = 1/2\).

Putting this back in the first (or second) equation gives \(y^2 - 1/2 = 1\), or \(y = \pm \sqrt{5}/2\). Thus the final solution is \((x_0, y_0) = 1/2, \pm 5/2\).

What can we take away from this exercise? The geometry fails us, at least as far as we know how to view it, in real coordinates. However using algebra we can determine the complex solutions. It is no wonder that Newton, and many others, were fooled by such calculations, and called such results imaginary. But as we shall learn, only by understanding such equations, can we fully appreciate the physics that is built upon them.
4. **Problem 4:** This problem is similar to problem 3 of two circles, but is in 3 dimensions with the intersection of two spheres. A unit radius sphere is given by

$$x^2 + y^2 + z^2 = 1$$

If we take $[x, y, z] \in \mathbb{R}$ we can easily picture the intersection of two spheres. If they just touch, the intersection is a single point. If they overlap, the solution is a circle, and if one is inside the other, there are no (real) solutions.

Next is an example of a polynomial in three variables, with no linear terms. We can generalize it by a shift in space

$$(x - 1)^2 + (y + 2)^2 + (z - 3)^2 = 2^2.$$ 

Here we shift $x$ by 1, $y$ by -2 and $z$ by 3, and scaled the radius from 1 to 2. Clearly this equation has the same geometry (its still a sphere), in factored form. Expanding it out gives

$$x^2 - 2x + y^2 + 2y + z^2 - 3z + 10 = 0.$$ 

Clearly the factored format is much easier to interpret. It should be clear how find the factored form given the general quadratic polynomial in the three variables (use Eq. 1.26 three times). Another interesting question is “Why are there no $ZY, ZY, xyz$ terms?”

What if $[x, y, z] \in \mathbb{C}$? Does this make physical sense? The answer is yes, if the variables are not the spatial coordinates but something called the **complex wave number**. More on this later.

### 3.2.4 Lec 14a Inner products and the Schwarz inequality

Another important example of algebraic expressions in mathematics is Hilbert’s generalization of Eq. 1.1, known as the Schwarz inequality, shown in Fig. 3.1. What is special about this generalization is that it proves that when the vertex is $90^\circ$, the length of the leg is minimum.

Vectors may be generalize to have $\infty$ dimensions: $\vec{U}, \vec{V} = [v_1, v_2, \cdots, v_\infty)$. The inner product (i.e., dot product) between two such vectors generalizes the finite dimensional case

$$\vec{U} \cdot \vec{V} = \sum_{k=1}^\infty u_k v_k.$$

As with the finite case, the **norm** $||\vec{U}|| = \sqrt{\vec{U} \cdot \vec{U}} = \sqrt{\sum u_k^2}$ is the dot product of the vector with itself, defining the **length** of the infinite component vector. Obviously there is an issue of convergence, if the norm for the vector to have a finite length.

It is a somewhat arbitrary requirement that $a, b, c \in \mathbb{R}$ for the Pythagorean theorem (Eq. 1.1). This seems natural enough since the sides are lengths. But, what if they are taken from the complex numbers, as for the lossy vector wave equation, or the lengths of vectors in $\mathbb{C}^n$? Then the equation generalizes to

$$\mathbf{c} \cdot \mathbf{c} = ||\mathbf{c}||^2 = \sum_{k=1}^n |c_k|^2,$$

where $||\mathbf{c}||^2 = (\mathbf{c}, \mathbf{c})$ is the inner (dot) product of a vector $\mathbf{c}$ with itself, where $|c_k|$ is the magnitude the complex $c_k$. As before, $||\mathbf{c}|| = \sqrt{||\mathbf{c}||^2}$ is the **norm** of vector $\mathbf{c}$, akin to a length.
**Schwarz Inequality**  

The Schwarz inequality\(^5\) says that the magnitude of the inner product of two vectors is less than or equal to the product of their lengths

\[ U \cdot V \leq ||U|| ||V||. \]

This may be simplified by normalizing the vectors to have unit length \((\hat{U} = U / ||U||, \hat{V} = V / ||V||)\), in which case \(-1 < \hat{U} \cdot \hat{V} \leq 1\). Another simplification is to define the scalar product in terms of the direction cosine

\[ \cos(\theta) = |\hat{U} \cdot \hat{V}| \leq 1. \]

A proof of the Schwarz inequality is as follows: From these definitions we may define the minimum difference between the two vectors as the perpendicular from the end of one to the intersection of the second. As shown in Fig. 3.1, \(U \perp V\) may be found by minimizing the length of the vector difference:

\[
\min_{\alpha} ||V - \alpha U||^2 = ||V||^2 + 2\alpha V \cdot U + \alpha^2 ||U||^2 > 0
\]

\[
0 = \partial_{\alpha} (V - \alpha U) \cdot (V - \alpha U)
= V \cdot U - \alpha^* ||U||^2
\]

\[
\therefore \quad \alpha^* = V \cdot U / ||U||^2.
\]

The Schwarz inequality follows:

\[
I_{\text{min}} = ||V - \alpha^* U||^2 = ||V||^2 - \frac{|U \cdot V|^2}{||U||^2} > 0
\]

\[
0 \leq |U \cdot V| \leq ||U|| ||V||
\]

An important example of such a vector space includes the definition of the Fourier Transform, where we may set

\[
U(\omega) = e^{-\omega_0 t} \quad V(\omega) = e^{\omega_0 t} \quad U \cdot V = \int_{\omega} e^{2\omega t} e^{-j\omega_0 t} \frac{d\omega}{2\pi} = \delta(\omega - \omega_0).
\]

It seems that the Fourier transform is a result that follows from a minimization, unlike the Laplace transform that follows from causal system parameters. This explains the important differences between the two, in terms of their properties (Unlike the LT, the FT is not complex analytic). We also explored this topic in Lecture 1.3.11 (p. 87).

\(^5\)A simplified derivation is provided in Sect. 1.3.5 (p. 71).
3.2. WEEK 5

Power vs. power series, linear vs. nonlinear

Another place where equations of second degree appear in physical applications is in energy and power calculations. The electrical power is given by the product of the voltage \( v(t) \) and current \( i(t) \) (or in mechanics as the force times the velocity). For example if we define \( P = v(t)i(t) \) to be the power \( P \) [Watts], then the total energy [Joules] at time \( t \) is (Van Valkenburg, 1964a, Chapter 14)

\[
\mathcal{E}(t) = \int_0^t v(t)i(t)dt.
\]

From this observe that the power is the rate of change of the total energy

\[
P(t) = \frac{d}{dt}\mathcal{E}(t),
\]

reminiscent of the Fundamental Theorem of calculus [Eq. 1.75, (p. 100)].

**Ohm’s Law and impedance:** The ratio of voltage over the current is call the *impedance* which has units of [Ohms]. For example given a resistor of \( R = 10 \) [ohms],

\[
v(t) = R \; i(t).
\]

Namely 1 amp flowing through the resistor would give 10 volts across it. Merging the linear relation due to Ohm’s law with the definition of power, shows that the instantaneous power in a resistor is quadratic in voltage and current

\[
P(t) = \frac{v(t)^2}{R} = i(t)^2R.
\]  

(3.10)

Note that Ohm’s law is linear in its relation between voltage and current whereas the power and energy are nonlinear.

Ohm’s Law generalizes in a very important way, allowing the impedance (e.g., resistance) to be a complex analytic function of complex frequency \( s = \sigma + \omega j \) (Kennelly, 1893; Brune, 1931b). Impedance is a fundamental concept in many fields of engineering. For example:6 Newton’s second law \( F = ma \) obeys Ohm’s law, with mechanical impedance \( Z(s) = sm \). Hooke’s Law \( F = kx \) for a spring is described by a mechanical impedance \( Z(s) = k/s \). In mechanics a “resistor” is called a *dashpot* and its impedance is a positive and real constant.7

**Kirchhoff’s Laws KCL, KVL:** The laws of electricity and mechanics may be written down using Kirchhoff’s Laws current and voltage laws, (KCL, KVL), which lead to linear systems of equations in the currents and voltages (velocities and forces) of the system under study, with complex coefficients having positive real parts.

Points of major confusion are a number of terms that are misused, and overused, in the fields of mathematics, physics and engineering. Some of the most obviously abused terms are linear/nonlinear, energy, power, power series. These have multiple meanings, which can, and are, fundamentally in conflict.

6In acoustics the pressure is a potential, like voltage. The force per unit area is given by \( f = -\nabla p \) thus \( F = -\int \nabla p \; dS \). Velocity is analogous to a current. In terms of the velocity potential, the velocity per unit area is \( v = -\nabla \phi \).

7https://en.wikipedia.org/wiki/Impedance_analogy
Transfer functions (Transfer matrix): The only method that seems to work, to sort this out, is to cite the relevant physical application, in specific contexts. The most common touch point is a physical system that has an input \( x(t) \) and an output \( y(t) \). If the system is linear, then it may be represented by its impulse response \( b(t) \). In such cases the system equation is

\[
y(t) = h(t) \star x(t) \leftrightarrow Y(\omega) = H(s)|_{\sigma=0} X(\omega),
\]

namely the convolution of the input with the impulse response gives the output. From Fourier analysis this relation may be written in the real frequency domain as a product of the Laplace transform of the impulse response, evaluated on the \( \omega j \) axis and the Fourier transform of the input \( X(\omega) \leftrightarrow x(t) \) and output \( Y(\omega) \leftrightarrow y(t) \).

Mention ABCD Transfer matrix

If the system is nonlinear, then the output is not given by a convolution, and the Fourier and Laplace transforms have no obvious meaning.

The question that must be addressed is why is the power said to be nonlinear whereas a power series of \( H(s) \) said to be linear. Both have powers of the underlying variables. This is massively confusing, and must be addressed. The question will be further addressed in Section 3.5.1 in terms of the system postulates of physical systems.

What’s going on? The domain variables must be separated from the codomain variables. In our example, the voltage and current are multiplied together, resulting in a nonlinear output, the power. If the frequency is squared, this is describing the degree of a polynomial. This is not nonlinear because it does not impact the signal output, it characterizes the Laplace transform of the system response.

3.3 Week 6

3.3.1 Lec 15 Gaussian Elimination of linear equations

Example problems using Gaussian Elimination: Gaussian Elimination (Appendix A.4, p. 235) is valid for nonlinear systems of equations. Till now we have emphasized the reduction of linear systems of equations.

Problem 1: Two lines in a plane either intersect or are parallel, in which case they are said to meet at \( \infty \). Does this make sense? The two equations that describe this may be written in matrix form as \( Ax = b \), which written out as

\[
\begin{bmatrix}
a_{11} & a_{12} \\
a_{21} & a_{22}
\end{bmatrix}
\begin{bmatrix}
x_1 \\
x_2
\end{bmatrix}
= 
\begin{bmatrix}
b_1 \\
b_2
\end{bmatrix}
\]

(3.11)

The intersection point \( x_0, y_0 \) is given by the solution of two equations:

\[
\begin{bmatrix}
x_1 \\
x_2
\end{bmatrix}
= 
\frac{1}{\Delta}
\begin{bmatrix}
a_{22} & -a_{12} \\
-a_{21} & a_{11}
\end{bmatrix}
\begin{bmatrix}
b_1 \\
b_2
\end{bmatrix},
\]

(3.12)

where \( \Delta = a_{11}a_{22} - a_{12}a_{21} \) is the determinant of matrix \( A \) (Matlab’s \( \text{det}(A) \) function).

It is useful to give an interpretation of these two equations. Each row of the 2x2 matrix defines a line in the \((x, y)\) plane. The top row is

\[a_{11}x + a_{12}y = b_1.\]
Normally we would write this equation as \( y(x) = \alpha x + \beta \), where \( \alpha \) is the slope and \( \beta \) is the intercept (i.e., \( y(0) = \beta \)). In terms of the elements of matrix \( A \), the slope of the first equation is \( \alpha = -a_{11}/a_{12} \) while the slope of the second is \( \alpha = -a_{21}/a_{22} \). The two slopes are equal (the lines are parallel) when \( -a_{11}/a_{12} = -a_{21}/a_{22} \), or written out
\[
\Delta = a_{11}a_{22} - a_{12}a_{21} = 0.
\]
Thus when the determinate is zero, the two lines are parallel and there is no solution to the equations.

This 2x2 matrix equation is equivalent to a 2\text{d} degree polynomial. If we seek an eigenvector solution \( [e_1, e_2]^T \) such that
\[
\begin{bmatrix}
a_{11} & a_{12} \\
a_{21} & a_{22}
\end{bmatrix}
\begin{bmatrix}
e_1 \\
e_2
\end{bmatrix}
= \lambda
\begin{bmatrix}
e_1 \\
e_2
\end{bmatrix}
\] (3.13)
the 2x2 equation becomes singular, and \( \lambda \) is one of the roots of the polynomial. One may proceed by merging the two terms to give
\[
\begin{bmatrix}
a_{11} - \lambda & a_{12} \\
a_{21} & a_{22} - \lambda
\end{bmatrix}
\begin{bmatrix}
e_1 \\
e_2
\end{bmatrix}
= \begin{bmatrix} 0 \\ 0 \end{bmatrix}.
\] (3.14)
Clearly this new matrix has no solution, since if it did, \( [e_1, e_2]^T \) would be zero, which is nonsense. If it has no solution, then the determinant of the matrix must be zero. Forming this determinate gives
\[
(a_{11} - \lambda)(a_{22} - \lambda) - a_{12}a_{21} = 0
\]
thus we obtain the following quadratic equation for the roots \( \lambda_{\pm} \) (eigenvalues)
\[
\lambda_{\pm}^2 - (a_{11} + a_{22})\lambda_{\pm} + \Delta = 0.
\]
When \( \Delta = 0 \), one eigenvalue is zero while the other is \( a_{11} + a_{22} \), which is known as the trace of the matrix.

**In summary:** Given a “linear” equation for the point of intersection of two lines, we see that there must be two points of intersection, as there are always two roots of the quadratic characteristic polynomial. However the two lines only intersect at one point. What is going on? What is the meaning of this second root?

Some simple examples will help. The eigenvalues depend on the relative slopes of the lines, which in general can become complex. The intercepts are dependent on \( b \). Thus when the RHS is zero, the eigenvalues are irrelevant. This covers the very simple examples. When one eigenvalue is real and the other is imaginary, more interesting things are happening since the slope of one line is real and the slope of the other is pure imaginary. The lines can intersect in the real plane, and again in the complex plane.

Lets try an example of two lines, slopes of 1 and 2: \( y_1 = x + a \) and \( y_2 = 2x + b \). In matrix form Let
\[
\begin{bmatrix} 1 & -1 \\ 1 & -2 \end{bmatrix}
\begin{bmatrix} y \\ x \end{bmatrix}
= \begin{bmatrix} a \\ b \end{bmatrix}
\] (3.15)
The determinate is \( \Delta = -1 \), thus the solution is
\[
\begin{bmatrix} y_0 \\ x_0 \end{bmatrix}
= -1 \begin{bmatrix} -2 & 1 \\ -1 & 1 \end{bmatrix}
\begin{bmatrix} a \\ b \end{bmatrix}
= \begin{bmatrix} 2 & -1 \\ 1 & -1 \end{bmatrix}
\begin{bmatrix} a \\ b \end{bmatrix}
= \begin{bmatrix} 2a - b \\ a - b \end{bmatrix}.
\] (3.16)
Thus the two real lines having slopes of 1 and 2 having intercepts of \( a \) and \( b \), meet \((x_0, y_0) = (2a - b, a - b)\). We may verify by substituting \( x = a - b \) into the starting equations \( y_1 = (a - b) + a = 2a - b \) and \( y_2 = 2(a - b) + b = 2a - b \), which each \( 2a - b \).

While there is a unique solution, there are two eigenvalues, given by the roots of
\[
(1 - \lambda) (-2 - \lambda) + 1 = 0.
\]
If we transfer the sign from one monomial to the other
\[
(-1 + \lambda)(2 + \lambda) + 1 = 0
\]
and reorder for simplicity
\[
(\lambda - 1)(\lambda + 2) + 1 = 0
\]
we obtain the quadratic for the roots
\[
\lambda^2 + \lambda - 1 = 0.
\]
Completing the square gives
\[
(\lambda + 1/2)^2 = 3/4.
\]
or
\[
\lambda = -1/2 \pm \sqrt{3}/2.
\]
The question is, what is the relationship between the eigenvalues and the final solution, if any? Maybe none. The solution \((x_0, y_0)\) is reasonable, and it’s not clear that the eigenvalues play any useful role here, other than to predict there is a second solution. I’m confused.

Two lines in 3-space: In three dimensions
\[
\begin{bmatrix}
a_{11} & a_{12} & a_{13} \\
a_{21} & a_{22} & a_{23} \\
a_{31} & a_{32} & a_{33}
\end{bmatrix}
\begin{bmatrix}
x \\
y \\
z
\end{bmatrix} =
\begin{bmatrix}
b_1 \\
b_2 \\
b_3
\end{bmatrix}
\]
(3.17)
Each row of the matrix describes a plane, which is said to be linear in the unknowns \((x, y, z)\). Thus the system of linear equations represents three planes, which must intersect at one point. If two planes are parallel, there is no real solution. In this case the intersection by the third plane generates two parallel lines.

As in the 2x2 case, one may convert this linear equation into a cubic polynomial by setting the determinant of the matrix, with \(-\lambda\) subtracted from the diagonal, equal to zero. That is, \(\text{det}(A - \lambda I) = 0\). Here \(I\) is the matrix with 1 on the diagonal and zero off the diagonal.

Simple example: As a simple example, let the first plane be \(z = 0\) (independent of \(x, y\)), the second parallel plane be \(z = 1\) (independent of \((x, y)\)) and the third plane be \(x = 0\) (independent of \(y, z\)). This results in the system of equations
\[
\begin{bmatrix}
0 & 0 & a_{13} \\
0 & 0 & a_{23} \\
a_{31} & 0 & 0
\end{bmatrix}
\begin{bmatrix}
x \\
y \\
z
\end{bmatrix} =
\begin{bmatrix}
0 \\
1 \\
0
\end{bmatrix}
\]
(3.18)
Writing out the three equations we find \(a_{13}z = 0, a_{23}z = 1, \) and \(a_{31}x = 0\). Note that \(\text{det}(A) = 0\) (we need to learn how to compute the 3x3 determinant). This means the three planes never intersect at one point. Use Matlab to find the eigenvalues.
3.3.2 Lec 16 Matrix composition: Bilinear and ABCD transformations

The Transmission matrix

A transmission matrix is a 2x2 matrix that characterizes a 2-port circuit, one having an input and output voltage and current, as shown in Fig. 1.14. The input is the voltage and current \( V_1, I_1 \) and the output is the voltage and current \( V_2, -I_2 \), with the current always defined to flow into the port. For any such a linear network, the input-output relations may be written in a totally general way as

\[
\begin{bmatrix} V_1 \\ I_1 \end{bmatrix} = \begin{bmatrix} A(s) & B(s) \\ C(s) & D(s) \end{bmatrix} \begin{bmatrix} V_2 \\ -I_2 \end{bmatrix}.
\]

In Section 1.3.7 we showed that a cascade of such matrices is composition. We shall show below that the justification of this relationship is based on the composition of bilinear transformations.

Expanding Eq. 1.61 into its individual equations demonstrates the linear form of the relations

\[
V_1 = A(s) V_2 - B(s) I_2 \quad \quad I_1 = C(s) V_2 - D(s) I_2,
\]

quantifying the relationship between the input voltage and current to its output voltage and current.

Define \( H(s) = V_2/V_1 \) as the transfer function, as the ratio of the output voltage \( V_2 \) over the input voltage \( V_1 \), under the constraint that the output current \( I_2 = 0 \). From this definition \( H(s) = 1/A(s) \).

In a similar fashion we may define the meaning of all four functions as

\[
A(s) \equiv \frac{V_1}{V_2} \bigg|_{I_2=0} \quad \quad B(s) \equiv -\frac{V_1}{I_2} \bigg|_{V_2=0} \quad \quad C(s) \equiv \frac{I_1}{V_2} \bigg|_{I_2=0} \quad \quad D(s) \equiv -\frac{I_1}{I_2} \bigg|_{V_2=0}
\] (3.19)

From Eq. 1.61 one may compute any desired quantity, specifically those quantities defined in Eq. 3.20, the open circuit voltage transfer function \((1/A(s))\), the short-circuit transfer current \((1/D(s))\) and the two transfer impedances \(B(s)\) and \(1/C(s)\).

In the engineering fields this matrix composition is called the Transmission matrix, also known as the ABCD method. It is a powerful method that is easy to learn and use, that gives important insights into transmission lines, and thus even the 1 dimensional wave equation. This method is exquisitely presented in the physics literature, with solid state applications, in an amazing book by Brillouin (1953).

Derivation of ABCD matrix for example of Fig. 1.14 (p. 79).

The derivation is straightforward by the application of Ohm’s Law, as shown in Section 1.3.7.

The convenience of the ABCD matrix method is that the output of one is identically the input of the next. Cascading (composing) the results for the series inductor with the shunt compliance leads to the 2x2 matrix form that precisely corresponds to the transmission line CFA shown in Fig. 2.2,

\[
\begin{bmatrix} V_n(s) \\ I_n(s) \end{bmatrix} = \begin{bmatrix} 1 & sL_n \\ 0 & 1 \end{bmatrix} \begin{bmatrix} V_{n+1}(s) \\ -I_{n+1}(s) \end{bmatrix}.
\] (3.21)

This matrix relation characterizes the series mass term \( sL_n \). A second equation maybe be used for the shunt capacitance term \( sY_n \)

\[
\begin{bmatrix} V_n(s) \\ I_n(s) \end{bmatrix} = \begin{bmatrix} 1 & 0 \\ sC_n & 1 \end{bmatrix} \begin{bmatrix} V_{n+1}(s) \\ -I_{n+1}(s) \end{bmatrix}.
\] (3.22)
The positive constants $L_n, C_n \in \mathbb{R}$ represent the series mass (inductance) and the shunt compliance (capacitance) of the mechanical (electrical) network. The integer $n$ indexes the series and shunt sections, that are composed one following the next.

A more detailed analysis shows that for the case of no losses, the wave velocity is

$$c = 1/\sqrt{LC}. \quad (3.23)$$

Also the wave number is

$$\kappa = \sqrt{Z \bar{Y}} = \sqrt{sL \cdot sC} = s\sqrt{LC} = \frac{s}{c}. \quad (3.24)$$

Finally the characteristic resistance is

$$r_o = \sqrt{Z / \bar{Y}} = \sqrt{sL / sC} = \sqrt{L / C}. \quad (3.25)$$

All of these based on a unit length $\Delta x$ as shown in Fig. 2.2 (p. 169).

**Matrix composition and the bilinear transform:** Now that we have defined the composition of two functions, we will use it to define the Möbius or bilinear transformation. Once you understand how this works, hopefully you will understand why it is the unifying element in many important engineering problems.

The bilinear transformation is given by

$$w = \frac{a + bz}{c + dz}$$

This takes one complex number $z = x + iy$ and transforms it into another complex number $w = u + iv$. This transformation is bilinear in the sense that its linear in both the input and output side of the equation. This may be seen when written as

$$(c + dz)w = a + bz,$$

since this relation is linear in the coefficients $[a, b, c, d]$. An important example is the transformation between impedance $Z(s)$ and reflectance $\Gamma(s)$,

$$\Gamma(s) = \frac{Z(s) - r_0}{Z(s) + r_0},$$

which is widely used in transmission line problems. In this example $w = \Gamma, z = Z(s), a = -r_0, b = 1, c = r_0, d = 1$.

If we define a second bilinear transformation (this could be the transformation from reflectance back to impedance)

$$r = \frac{\alpha + \beta w}{\gamma + \delta w},$$

and then compose the two something astray wrt arguments

$$w \circ r = \frac{a + b r}{c + d r} = \frac{a(\gamma + \delta w) + b(\alpha + \beta w)}{c(\gamma + \delta w) + d(\alpha + \beta)w} = \frac{a\gamma + b\alpha + (a\delta + b\beta)w}{c\gamma + d\alpha + (c\delta + d\beta)w},$$
something surprising happens. The composition \( w \circ r \) may be written in matrix form, as the product of two matrices that represents each bilinear transform. This may be seen as true by inspecting the coefficients of the composition \( w \circ r \) (shown above) and the product of the two matrices

\[
\begin{bmatrix} a & b \\ c & d \end{bmatrix} \begin{bmatrix} \alpha & \beta \\ \gamma & \delta \end{bmatrix} = \begin{bmatrix} (a\gamma + b\alpha) & (a\delta + b\beta) \\ (c\gamma + d\alpha) & (c\delta + d\beta) \end{bmatrix}.
\]

The the power of this composition property of the bilinear transform may be put to work solving important engineering problems, using transmission matrices.

### 3.3.3 Lec 16a Transmission line example

In this problem, we will look at the transfer function of a two-port network, shown in Fig. 3.2. We wish to model the dynamics of a freight-train having \( N \) such cars. The model of the train consists of masses connected by springs.

**Transfer functions:** The velocity transfer function for this system is defined as the ratio of the output to the input velocity. Consider the engine on the left pulling the train at velocity \( V_1 \) and each car responding with a velocity of \( V_n \). Then

\[
H(s) = \frac{V_N(s)}{V_1(s)}
\]

is the frequency domain ratio of the last car having velocity \( V_N \) to \( V_1 \), the velocity of the engine, at the left most spring (i.e., coupler).

**The transmission matrix** Consistent with the figure, break the model into cells each consisting of three elements: a series inductor representing half the mass (\( L = M/2 \)), a shunt capacitor representing the spring (\( C = 1/K \)), and another series inductor representing half the mass (\( L = M/2 \)). Each cell is symmetric, making the model a cascade of identical cells.

**To do:** Use the ABCD method to find the matrix representation of Fig. 3.2.

At each node define the force \( f_n(t) \leftrightarrow F_n(\omega) \) and the velocity \( v_n(t) \leftrightarrow V_n(\omega) \) at junction \( n \).
1. Write the ABCD matrix $T$ for a single cell, composed of series mass $M/2$, shunt compliance $C$ and series mass $M/2$, that relates the input node 1 to node 2 where

$$\begin{bmatrix} F_1 \\ V_1 \end{bmatrix} = T \begin{bmatrix} F_2(\omega) \\ -V_2(\omega) \end{bmatrix}$$

Note that here the mechanical force $F$ is analogous to electrical voltage, and the mechanical velocity $V$ is analogous to electrical current.

Sol:

$$T = \begin{bmatrix} 1 & sM/2 \\ 0 & 1 \\ sC & 1 \end{bmatrix} \begin{bmatrix} 1 & 0 \\ 0 & 1 \end{bmatrix} \begin{bmatrix} 1 & sM/2 \\ sC & 1 \end{bmatrix} \begin{bmatrix} 1 & 0 \\ 0 & 1 \end{bmatrix} \begin{bmatrix} 1 & sM/2 \\ 0 & 1 \end{bmatrix}$$

(3.26)

$$= \begin{bmatrix} 1 + \frac{1}{2}s^2/c^2 & sM(1 + \frac{1}{2}s^2/c^2) \\ \frac{sC}{sC} & 1 + \frac{1}{2}s^2/c^2 \end{bmatrix}$$

(3.27)

$$\approx \begin{bmatrix} 1 & sM/2 \\ sC & 1 \end{bmatrix}$$

(3.28)

2. Assuming that $N = 2$ and that $F_2 = 0$ (two mass problem), find the transfer function $H(s) \equiv V_2/V_1$. From the results of the $T$ matrix you determined above, find

$$H_{21}(s) = \frac{V_2}{V_1} \bigg|_{F_2=0}$$

Sol: From the lower equation we see that $V_1 = sCF_2 - (s^2MC/2 + 1)V_2$. Recall that $F_2 = 0$, thus

$$\frac{V_2}{V_1} = \frac{-1}{s^2MC/2 + 1} = \left( \frac{c_+}{s - s_+} + \frac{c_-}{s - s_-} \right).$$

with $s_\pm = \pm j\sqrt{\frac{2}{MC}}$ and $c_\pm = \pm j/\sqrt{2MC}$.

3. Find $h_{21}(t)$, the inverse Laplace transform of $H_{21}(s)$. Sol:

$$h(t) = \int_{\sigma_0-j\infty}^{\sigma_0+j\infty} e^{st} \frac{d\sigma}{\pi j} = c_+e^{-s_+t}u(t) + c_-e^{-s_-t}u(t).$$

The integral follows from the CRT. The poles are at $s_\pm = \pm j\sqrt{\frac{2}{MC}}$ and the residues are $c_\pm = \pm j/\sqrt{2MC}$.

4. What is the input impedance $Z_2 = F_2/V_2$ if $F_3 = -r_0V_3$? Sol:

Starting from $T$ calculated above, find $Z_2$

$$Z_2(s) = \frac{F_2}{V_2} = T \begin{bmatrix} F_3 \\ -V_3 \end{bmatrix} = \begin{bmatrix} F_2(\omega) \\ -V_2(\omega) \end{bmatrix}$$

$$= \frac{-(1 + s^2CM/2)r_0V_3 - (sM/2)(2 + s^2CM/2)V_3}{-sCr_0V_3 - (1 + s^2CM/2)V_3}$$
5. Simplify the expression for \( Z_2 \) with \( N \to \infty \) by assuming that:

1) \( F_3 = -r_0 V_3 \) (i.e., \( V_3 \) cancels),
2) \( s^2 MC << 1 \):
3) \( r_0 = \sqrt{M/C} \)

\[ Z_2(s) = \frac{(1 + s^2 CM/2)r_0 + (sM/2)(2 + s^2 CM/2)}{sC r_0 + (1 + s^2 CM/2)} \approx \frac{r_0 + sM}{sC r_0 + 1} \]

Assumption 3 gives

\[ Z_2 = \frac{sM + \sqrt{M}}{sC \sqrt{M/C} + 1} = r_0 \left( 1 + \frac{s \sqrt{MC}}{1 + s \sqrt{MC}} \right) \]

The constant \( r_0 \) is called the \textit{characteristic impedance}. This shows that below the cutoff frequency (approximation 2), the system approximates a transmission line.

6. State the ABCD matrix relationship between the first and \( N \)th node, in terms of the cell matrix.

\[ \begin{bmatrix} F_1 \\ V_1 \end{bmatrix} = T^N \begin{bmatrix} F_N(\omega) \\ -V_N(\omega) \end{bmatrix} \]

7. Given a \( T \) (ABCD) transmission matrix, the eigenvalues are and vectors are given in Appendix C of the Notes (p. 143), repeated here.

**Eigenvalues:**

\[ \begin{bmatrix} \lambda_+ \\ \lambda_- \end{bmatrix} = \frac{1}{2} \begin{bmatrix} (A + D) - \sqrt{(A-D)^2 + 4BC} \\ (A + D) + \sqrt{(A-D)^2 + 4BC} \end{bmatrix} \]

Due to symmetry, \( A = D \), this simplifies to \( \lambda_\pm = A \mp \sqrt{BC} \) so that the eigen matrix is

\[ \Lambda = \begin{bmatrix} A - \sqrt{BC} & 0 \\ 0 & A + \sqrt{BC} \end{bmatrix} \]

**Eigenvectors:** The eigenvectors are

\[ \begin{bmatrix} E_+ \\ E_- \end{bmatrix} = \frac{1}{2} \begin{bmatrix} (A - D) \mp \sqrt{(A-D)^2 + 4BC} \\ 1 \end{bmatrix} \]

The determinant of \( E \) is

\[ \Delta_E = -\frac{1}{2C} \sqrt{(A-D)^2 + 4BC} \]

In the symmetric case when \( A = D \)

\[ E = \begin{bmatrix} -\sqrt{\frac{B}{C}} & +\sqrt{\frac{B}{C}} \\ 1 & 1 \end{bmatrix}, \quad E^{-1} = \frac{1}{2} \begin{bmatrix} -\sqrt{\frac{C}{B}} & 1 \\ +\sqrt{\frac{C}{B}} & 1 \end{bmatrix} \]

and

\[ \Delta_E = -2 \sqrt{\frac{B}{C}} = -2r_o. \]

There are two important invariants for the transmission lines, the wave number \( k = \sqrt{BC} = s/c \), and the characteristic resistance \( r_o = \sqrt{B/C} \).
Transfer functions of the transmission line: The velocity transfer function $H_{N,1} = \frac{V_N}{V_1}$ is given by

$$\begin{bmatrix} F_1 \\ V_1 \end{bmatrix} = T^N \begin{bmatrix} F_N(\omega) \\ -V_N(\omega) \end{bmatrix}$$

along with the eigenvalue expansion

$$T^N = E \Lambda^N E^{-1} = E \begin{bmatrix} \lambda_+^N & 0 \\ 0 & \lambda_-^N \end{bmatrix} E^{-1}.$$

### 3.3.4 Lec 17 Introduction to the Branch cut and Riemann sheets

Branch cuts are required to preserve the single-valued nature of complex analytic functions. When an analytic function is multi-valued, some method needs to be devised to allow the multi-valued complex analytic function to be expanded as a Taylor series, which is necessarily single-valued. It follows that each single-valued sheet must have a different expansion, valid out to the nearest pole (or singularity). We shall explain these ideas with the simplest case, the double-valued square root function $w(z) = \pm \sqrt{z}$, as shown in Fig. 1.22 (p. 109).

### 3.4 Week 7

#### 3.4.1 Lec 18 Complex analytic mappings (domain coloring)

When one uses complex analytic functions it is helpful to understand their properties in the complex plane. In this sections we explore several well-known functions using domain coloring, first discussed in Section 1.3.10, p. 85. For the following figures the coordinate systems are defined by

$$s = \sigma + \omega j = \sqrt{x^2 + y^2} e^{i\theta} \text{ and } w = u + v j = \sqrt{u^2 + v^2} e^{i\psi}.$$  

For the first example (Section 1.3.10), $w(s) = s^2$ (Fig. 1.24, p. 109, left) and its inverse $s(w) = \sqrt{w}$ (right) are shown. On the left the red region, corresponding to $0^\circ$ [degrees], appears at both $0$ and $180$ ($u = \pm 1$) in the $w$ plane. This is because in polar coordinates $s^2 = |s|^2 e^{2i\theta}$ where $\theta$ is the angle of $s = |s| e^{i\theta}$. The square causes the phase to rotate twice around for once around the $s$ plane. Namely the angle is doubled, and the magnitude squared. Due to the faster changing phase in $w$, there are two red regions, one when $\theta = 0$ and the second at $\theta = \pi$ ($\angle w(s) = 2\theta$). The black spot is dilated due to the squaring of the radius (expanding it).

On the right the $\sqrt{w} = \sqrt{|w|} e^{i\theta/2}$. Because the angle of $w$ is divided by two, it takes twice as much phase (in $w$) to cover the same angle. Thus the red region ($0^\circ$) is expanded. We barely see the violet $90^\circ$ and yellow $-90^\circ$ angles. There is a branch cut running from $w = 0$ to $w = \infty$. As the branch cut is crossed, the function switches Riemann sheets, going from the top sheet (shown here) to the bottom sheet (not shown). Figure 1.22 in Section 3.3.4 depicts what is going on with these two sheets, and show the branch cut from the origin (point O) to $\infty$. In this depiction the first sheet ($+\sqrt{z}$) is on the bottom, while the second sheet ($\sqrt{z}$) is on top. For every value of $z$ there are two possible outcomes, $\pm \sqrt{z}$, represented by the two sheets.

Two more examples are given in Fig. 3.3 to interpret the two complex colorized mappings $w = \cos(\pi s)$ (left) and the Bessel function $J_0(\pi z)$. Note how the white and black contour lines are...
3.4. WEEK 7

$$u+jv = besselj(0,\pi(x+jy))$$

$$x$$

-2 -1  0  1  2

$$jy$$

-2 -1  0  1  2

$$u+jv = besselh(0,1,\pi(x+jy)/2)$$

$$x$$

-2 -1  0  1  2

$$jy$$

-2 -1  0  1  2

Figure 3.3: On the left is the Bessel function $J_0(\pi z)$, which is similar to $\cos(\pi z)$, except the zeros are distorted away from $s = 0$ by a small amount, due to the cylindrical geometry. On the right is the related Hankel function $H^{(1)}_0(u^2)$. The Bessel and Hankel functions are solution to the wave equation in cylindrical coordinates with different boundary conditions. The zeros in the function are the places where the pinned boundary condition is satisfied (where the string is restricted, by the boundary, from moving). The Hankel function $H^{(1)}_0(\pi z/2)$ has a branch cut, and a zero $z_{0,1}$ at $\pi * z_{0,1}/2 = -1.5 - 0.1j$.

always perpendicular where they cross, just as in the calibration plots for the x and y axes, shown in Fig. 1.17 in Section 1.3.10 (p. 85).

Along the $\sigma$ axis the $\cos(\pi x)$ is the periodic with a period of $\pi$. The dark spots are at the zeros at $\pm \pi/2, \pm 3\pi/2, \ldots$. When we stray off the $\omega j = 0$ axis, the function either goes to zero (black) or $\infty$ (white). This behavior carries the same $\pi$ periodicity as it has along the $\omega = 0$ line. On the right is the Bessel function $J_0(\pi z)$, which is similar to $\cos(\pi z)$, except the zeros are distorted away from the origin. These figure are worthy of careful study to develop an intuition for complex functions of complex variables. In Section 1.3.10 we shall explore more complex mappings, and in greater detail.

$$u+jv = \tan((x+jy))$$

$$x$$

-2 -1  0  1  2

$$jy$$

-2 -1  0  1  2

$$R+Xj$$

$$\sigma$$

-2 -1  0  1  2

Figure 3.4: On the left is the function $w(z) = \tan(z)$. On the right is the inverse $s = \tan^{-1}(w)$. Of special interest is $zviz \ atan(i*Z) = (i/2)*\log((1+Z)/(1-Z))$.

In the third example (Fig. 3.4) we show $w = \tan(z)$ and its inverse $z = \tan^{-1}(w)$. The tangent function has zeros where $\sin(z)$ has zeros (e.g., at $z = 0$) and poles where $\cos(z)$ is zero (e.g., at $\pm \pi/2$. The inverse function $s = \atan(w)$ has a zero at $w = 0$ and branch cuts eliminating from
$z = \pm \pi$.

The command \texttt{zviz besselj(0,pi*Z) besseli(0,j*pi*Z)} gives exactly the same plot, demonstrating that $I_0(z) = J_0(z)$.

It is fun, easy and interesting to study polynomials, say of degree 5 and 4 (i.e, with one zero removed), to demonstrate the Fundamental Theorem of Algebra.

### 3.4.2 Lec 20: Laplace transform and the Riemann zeta function (I)

### 3.4.3 Filter classification

Let $z \equiv e^{sT}$ where $T$ is the “sample period” at which data is taken (every $T$ seconds). For example if $T = 22.7 = 1/44100$ seconds then the data is sampled at 44100 k [Hz]. This is how a CD player works with high quality music. Thus the unit-time delay operator $z^{-1}$ as

$$
\delta(t - T) \leftrightarrow e^{-sT}
$$

#### Filter properties

Given the function

$$
F(s) = \frac{(s + 1)(s - 1)}{(s + 2)},
$$

1. find the minimum phase $M(s)$ and all-pass $A(s)$ parts. The minimum phase part has all of its poles and zeros in the left half plane (LHP), while the all-pass part has its poles in the LHP and mirrored zeros in the RHP. Thus we place a removable pole zero pair symmetrically across from the RHP zero, and then write the expression as the product, that is $F(s) = M(s) \cdot A(s)$:

$$
F(s) = \frac{(s + 1)(s - 1)}{(s + 2)} \cdot \frac{s + 1}{s + 2} \cdot \frac{s - 1}{s + 1}
$$

Thus $M(s) \equiv \frac{(s+1)^2}{s+2}$ and $A(s) \equiv \frac{s-1}{s+1}$

2. Find the magnitude of $M(s)$ Take the real part of the log of $M$ and then the anti-log. Thus $|M| = e^{\Re \ln M(s)}$

3. Find the phase of $M(s)$ In this case we use the imaginary part: $\angle M = \Im \ln M(s)$

4. Find the magnitude of $A(s)$ 1, by definition.

5. Find the phase of $A(s)$ $\angle A = \Im \ln(A)$

#### More questions

There are a number of question I would like to address in this report. Some of these are

1. Can we interpret the zeta function as a frequency domain quantity, and then inverse transform it into the time domain?

   The answer to this is yes, and the results are quite interesting.

2. Make a histogram of the entropy for the first million integers.

   This is a 5 minute job in Matlab. It goes something like this: $K=1e5; N=1:K; F=\text{zeros}(K,10);$ for $n=1:K; f=\text{factor}(n); F(n,1:length(f))=f; \text{end; hist}(F);$
Table 3.1: The following table provides an extended table of Laplace Transforms. $J_0, K_1$ are Bessel functions of the first and second kind.

<table>
<thead>
<tr>
<th>$f(t) \leftrightarrow F(s)$</th>
<th>Name</th>
</tr>
</thead>
<tbody>
<tr>
<td>$\delta(at) \leftrightarrow \frac{1}{a}$</td>
<td>$a \neq 0$; Time-scaled Dirac</td>
</tr>
<tr>
<td>$\delta(t + T_0) \leftrightarrow e^{sT_0}$</td>
<td>negative delay</td>
</tr>
<tr>
<td>$u(at) \leftrightarrow \frac{a}{s}$</td>
<td>$a \neq 0$; dilate</td>
</tr>
<tr>
<td>$u(-t) \leftrightarrow -\frac{1}{s}$</td>
<td>anti-causal step</td>
</tr>
<tr>
<td>$e^{at}u(-t) \leftrightarrow \frac{1}{-s + a}$</td>
<td>anticausal damped step</td>
</tr>
</tbody>
</table>

| $\frac{d^{1/2}}{dt^{1/2}} f(t)u(t) \leftrightarrow \sqrt{s}F(s)$ | “half” derivative |
| $\frac{d^{1/2}}{dt^{1/2}} u(t) \leftrightarrow \sqrt{s}$ | “half” derivative |
| $\frac{1}{\sqrt{\pi}t} u(t) \leftrightarrow \frac{1}{\sqrt{s}}$ | “half” integration |
| $\operatorname{erfc}(\alpha \sqrt{t}) \leftrightarrow \frac{1}{s}e^{-2\alpha \sqrt{s}}$ (Morse-Feshbach-II, p. 1582) | $\alpha > 0$; erfc |

| $J_0(at)u(t) \leftrightarrow \frac{1}{s^2 + a^2}$ | Bessel |
| $J_1(t)u(t)/t \leftrightarrow \sqrt{s^2 + 1} - s$ |
| $J_1(t)u(t)/t + 2u(t) \leftrightarrow \sqrt{s^2 + 1} + s = e^{\sinh^{-1}(s)}$ |
| $\delta(t) + J_1(t)u(t)/t \leftrightarrow \sqrt{s^2 + 1}$ |
| $I_0(t)u(t) \leftrightarrow 1/\sqrt{s^2 - 1}$ |
| $u(t)/\sqrt{t + 1} \leftrightarrow e^{s/\sqrt{\pi}} \operatorname{erfc}(\sqrt{s})$ |
| $\sqrt{tu(t)} * \sqrt{1 + tu(t)} \leftrightarrow e^{s/2}K_1(s/2)/2s$ |
Entropy analysis

According to the Fundamental theorem of arithmetic, every integer may be uniquely written as the product of primes. Thus \( n = \prod_k n^\alpha_k \) (e.g., \( 12 = 2^2 \cdot 3 \) with \( \alpha_2 = 2 \) and \( \alpha_3 = 1 \)). If one thinks of \( K_n \) as a random variable on \( n \), then the constant \( K_n \) may be characterized by the concept of entropy. Thus each integer is associated with an entropy, defined by \( H_n = \sum K_n \log_2(K_n) \).

The zeta function The zeta function depends explicitly on the primes, which makes it very important. In 1737 Euler proposed the real-valued function \( \zeta(x) \in \mathbb{R} \) and \( x \in \mathbb{R} \), to prove that the number of primes is infinite (Goldstein, 1973). Euler’s definition of \( \zeta(x) \in \mathbb{R} \) is given by the power series

\[
\zeta(x) = \sum_{n=1}^{\infty} \frac{1}{n^x} \quad \text{for } x > 0 \in \mathbb{R}.
\]

This series converges for \( x > 0 \), since \( R = n^{-x} < 1, n > 1 \in \mathbb{N} \).

In 1860 Riemann extended the zeta function into the complex plane, resulting in \( \zeta(s) \), defined by the complex analytic power series, identical to the Euler formula, except \( x \in \mathbb{R} \) has been replaced by \( s \in \mathbb{C} \)

\[
\zeta(s) \equiv \frac{1}{1^s} + \frac{1}{2^s} + \frac{1}{3^s} + \frac{1}{4^s} + \cdots = \sum_{n=1}^{\infty} \frac{1}{n^s} = \sum_{n=1}^{\infty} n^{-s} \quad \text{for } \Re\{s\} = \sigma > 0.
\]

This formula converges for \( \Re\{s\} > 1 \) (Goldstein, 1973). To determine the formula in other regions of the \( s \) plane one, must extend the series via analytic continuation. As it turns out, Euler’s formulation provided detailed information about the structure of primes, going far beyond his original goal.

Euler product formula

As was first published by Euler in 1737, one may recursively factor out the leading prime term, resulting in Euler’s product formula. Euler’s procedure is an algebraic implementation of the sieve of Eratosthenes (Section 1.2.3, p. 37 and Section 1.5.1, page 119).

Multiplying \( \zeta(s) \) by the factor \( 1/2^s \), and subtracting from \( \zeta(s) \), removes all the even terms \( \propto 1/(2n)^s \) (e.g., \( 1/2^s + 1/4^s + 1/6^s + 1/8^s + \cdots \))

\[
\left(1 - \frac{1}{2^s}\right) \zeta(s) = 1 + \frac{1}{3^s} + \frac{1}{4^s} + \frac{1}{5^s} + \cdots - \left(\frac{1}{2^s} + \frac{1}{4^s} + \frac{1}{6^s} + \frac{1}{8^s} + \cdots\right),
\]

which results in

\[
\left(1 - \frac{1}{2^s}\right) \zeta(s) = 1 + \frac{1}{3^s} + \frac{1}{5^s} + \frac{1}{7^s} + \frac{1}{9^s} + \frac{1}{11^s} + \frac{1}{13^s} + \cdots.
\]

Repeating this with a lead factor \( 1/3^s \) applied to Eq. 3.32 gives

\[
\frac{1}{3^s} \left(1 - \frac{1}{2^s}\right) \zeta(s) = \frac{1}{3^s} + \frac{1}{9^s} + \frac{1}{15^s} + \frac{1}{21^s} + \frac{1}{27^s} + \frac{1}{33^s} + \cdots.
\]

\(^8\)Sanity check: For example let \( n = 2 \) and \( x > 0 \). Then \( R = 2^{-x} < 1 \), where \( \epsilon \equiv \lim x \to 0^+ \). Taking the log gives \( \log_2 R = -\epsilon \log_2 2 = -\epsilon < 0 \). Since \( \log R < 0 \), \( R < 1 \).

\(^9\)This is known as Euler’s sieve, as distinguish from the Eratosthenes sieve.
Subtracting Eq. 3.33 from Eq. 3.32 cancels the RHS terms of Eq. 3.32

\[
\left(1 - \frac{1}{3^s}\right) \left(1 - \frac{1}{2^s}\right) \zeta(s) = 1 + \frac{1}{5^s} + \frac{1}{7^s} + \frac{1}{11^s} + \frac{1}{13^s} + \frac{1}{17^s} + \frac{1}{19^s} \cdots.
\]

Further repeating this process, with prime scale factors, (i.e., \(1/5^s, 1/7^s, \ldots, 1/\pi_k^s, \ldots\)), removes all the terms on the RHS but 1, results in Euler’s analytic product formula \((s = x \in \mathbb{R})\) and Riemann’s complex analytic product formula \((s \in \mathbb{C})\)

\[
\zeta(s) = \prod_k \frac{1}{1 - \pi_k^{-s}} = \frac{1}{1 - 2^{-s}} \cdot \frac{1}{1 - 3^{-s}} \cdot \frac{1}{1 - 5^{-s}} \cdot \frac{1}{1 - 7^{-s}} \cdots \cdot \frac{1}{1 - \pi_n^{-s}} \cdots
\]

(3.34)

where \(\pi_k\) represents the \(k^{th}\) prime. The above defines each prime factor

\[
\zeta_k(s) = \frac{1}{1 - \pi_k^{-s}} \quad (3.35)
\]

as the \(k^{th}\) term of the complex analytic Riemann product formula, having an RoC of \(|\pi_k^{-s}| < 1\), but valid for \(|\pi_k^{-s}| \neq 1\). Each recursive step in this construction assures that the lead term, along with all of its multiplicative factors, are subtracted out, just as with the cancellations with the sieve of Eratosthenes. It is instructive to compare each iteration with that of the sieve (Fig. 1.5, p. 39).

**Exercise:** Work out the RoC for \(k = 2\).

**Poles of \(\zeta_k(s)\)**

Discuss RoC for each term in the product representation, and the anti-causal nature of \(\zeta_k(1 - s)\). Riemann proposed that Euler’s zeta function \(\zeta(s) \in \mathbb{C}\) have a complex argument (first actually explored by Chebyshev in 1850, (Bombieri, 2000)), extending \(\zeta(s)\) into the complex plane, \((s \in \mathbb{C})\), making it a complex analytic function.

Given that \(\zeta(s)\) is a complex analytic function, one might naturally wonder if \(\zeta(s)\) has an inverse Laplace transform. There seems to be very little written on this topic (Hill, 2007). We shall explore this question further here.

One may now identify the poles of \(\zeta_k(s) (p \in \mathbb{N})\), which are required to determining the RoC. For example, the \(k^{th}\) factor of Eq. 3.34, expressed as an exponential, is

\[
\zeta_k(s) \equiv \frac{1}{1 - \pi_k^{-s}} = \frac{1}{1 - e^{-sT_k}} = \sum_{k=0}^{\infty} e^{-skT_k},
\]

(3.36)

where \(T_k \equiv \ln \pi_k\). Thus \(\zeta_p(s)\) has poles at \(-s_n T_p = 2\pi n j\) (when \(e^{-sT_p} = 1\), thus

\[\omega_n = \frac{2\pi n}{T_k},\]

with \(-\infty < n \in \mathbb{Z} < \infty\). These poles are the eigen-modes of the zeta function. A domain-colorized plot of this function is provided in Fig. 3.5. It is clear that the RoC of \(\zeta_k\) is \(> 0\). It would be helpful to determine why \(\zeta(s)\) as such a more restrictive RoC than each of its factors.
CHAPTER 3. ALGEBRAIC EQUATIONS: STREAM 2

\[ u + jv = \frac{1}{1 - e^{-s \pi}} \]

Figure 3.5: Plot of \( w(s) = \frac{1}{1 - e^{-s \pi}} \) (note \( \pi_1 = 2 \)), which is factor \( \zeta_1(s) \) of Eq. 3.34. Here \( w(s) \) has poles where \( e^{s \ln 2} = 1 \), namely where \( \omega_n \ln 2 = n2\pi \), as seen in the colorized–map (\( s = \sigma + \omega j \) is the Laplace frequency [rad]).

\[ Z_{Z_p}^{\text{cla}}(t) = \alpha Z_{Z_p}^{\text{cla}}(t - T_p) + \delta(t) \]

Figure 3.6: This feedback network is described by a time-domain difference equation with delay \( T_p = \ln \pi_k \), has an all-pole transfer function, given by Eq. 3.39. Physically this delay corresponds to a delay of \( T_p \) [s].

Inverse Laplace transform

The inverse Laplace transform of Eq. 3.36 is an infinite series of delays of delay \( T_p \)

\[ Z_{Z_p}^{\text{cla}}(t) = \delta(t) T_p \equiv \sum_{k=0}^{\infty} \delta(t - kT_p) \leftrightarrow \frac{1}{1 - e^{-sT_p}}. \quad (3.37) \]

Inverse transform of Product of factors

The time domain version of Eq. 3.34 may be written as the convolution of all the \( Z_{Z_k}^{\text{cla}}(t) \) factors

\[ Z^{\text{cla}}(t) \equiv Z_2^{\text{cla}} * Z_3^{\text{cla}}(t) * Z_5^{\text{cla}}(t) * Z_7^{\text{cla}}(t) \cdots * Z_{Z_p}^{\text{cla}}(t) \cdots, \quad (3.38) \]

where \( * \) represents time convolution (Table 1.5, p. 93).

\(^{10}\)Here we use a shorthand double-parentheses notation to define the one-sided infinite sum \( f(t) T \equiv \sum_{k=0}^{\infty} f(t - kT) \).
Physical interpretation

Such functions may be generated in the time domain as shown in Fig. 3.6, using a feedback delay of \( T_p [s] \) as described by the equation in the figure, with a unity feedback gain \( \alpha = 1 \).

\[
Z_{\text{eta}}(t) = Z_{\text{eta}}(t - T_n) + \delta(t).
\]

Taking the Laplace transform of the system equation we see that the transfer function between the state variable \( q(t) \) and the input \( x(t) \) is given by \( Z_p \). Taking the \( LT \), we see that \( \zeta(s) \) is an all-pole function

\[
\zeta_p(s) = e^{-sT_p} \zeta_p(1) + 1(t), \quad \text{or} \quad \zeta_p(s) = \frac{1}{1 - e^{-sT_p}}.
\]  

(3.39)

Discussion: In terms of the physics, these transmission line equations are telling us that \( \zeta(s) \) may be decomposed into an infinite cascade of transmission lines (Eq. 3.38), each having a unique delay given by \( T = \ln \pi_k, \pi_k \in \mathcal{P}, \) the log of the primes. The input admittance of this cascade may be interpreted as an analytic continuation of \( \zeta(s) \) which defines the eigen-modes of that cascaded impedance function.

Working in the time domain provides a key insight, as it allows us to determine the analytic continuation of the infinity of possible continuations, which are not obvious in the frequency domain. Transforming to the time domain is a form of analytic continuation of \( \zeta(s) \), that depends on the assumption that \( Z_{\text{eta}}(t) \) is one-sided in time (causal).

Additional relations: Some important relations provided by both Euler and Riemann (1859) are needed when studying \( \zeta(s) \).

With the goal of generalizing his result, Euler extended the definition with the functional equation

\[
\zeta(s) = 2^s \pi^{s-1} \sin \left( \frac{\pi s}{2} \right) \Gamma(1-s) \zeta(1-s).
\]  

(3.40)

This seems closely related to Riemann’s time reversal symmetry properties (Bombieri, 2000)

\[
\pi^{-s/2} \Gamma \left( \frac{s}{2} \right) \zeta(s) = \pi^{-(1-s)/2} \Gamma \left( \frac{1-s}{2} \right) \zeta(1-s).
\]

This equation is of the form \( F \left( \frac{z}{2} \right) \zeta(s) = F \left( \frac{1-z}{2} \right) \zeta(1-s) \) where \( F(s) = \Gamma(s)/\pi^s \).

As shown in Table 1.5 the \( LT^{-1} \) of \( f(-t) \leftrightarrow F(-s) \) is simply a time-reversal. This leads to a causal and anti-causal function that are symmetric about \( \Re\{s\} = 1/2 \) (Riemann, 1859). It seems likely this is an important insight into the Euler’s functional equation.

Riemann (1859, page 2) provides an alternate integral definition of \( \zeta(s) \), based on the complex contour integration

\[
2 \sin(\pi s) \Gamma(s - 1) \zeta(s) = \int_{c-x \to \infty} \frac{(-x)^{s-1}}{e^x - 1} dx. \quad \text{or} \quad \int_{y \to \infty} \frac{(y)^{s-1}}{e^{-y} - 1} dy.
\]

Given the \( \zeta_k(s) \) it seems important to look at the inverse \( LT \) of \( \zeta_k(1-s) \), to gain insight into the analytically extended \( \zeta(s) \)

---

11Verify Riemann’s use of \( x \), which is taken to be real rather than complex. This could be more natural (i.e., modern Laplace transformation notation) if \( -x \to y \to z \).
**Integral definition of the complex Gamma function** $\Gamma(s)$  
The definition of the complex analytic Gamma function (p. 93) 

$$\Gamma(s + 1) = s\Gamma(s) \equiv \int_0^\infty \xi^s e^{-\xi} d\xi,$$

which is a generalization of the real integer factorial function $n!$.

**Zeros of $\zeta(s)$**  
We are still left with the most important question of all “Where are the zeros of $\zeta(s)$?” Equation 3.39 has no zeros, it is an all-pole system. The cascade of many such systems is also all-pole. As I see it, the issue is what is the actual formula for $\zeta(s)$?

### 3.4.4 Lec 20b: $\mathcal{L}$ of the Riemann zeta function

This analytic function is a strong argument for the need for a deeper understanding of the analytic function. Just like the Pythagorean theorem is important to all mathematics, the zeta function is important to analysis, with many streams of analysis emanating from this form. For example the analytic Gamma function $\Gamma(s)$ is a generalization of the factorial by the relationship

$$n! = n(n - 1)! \Rightarrow \Gamma(s) = s\Gamma(s - 1).$$

Another useful relationship is 

$$\sum_{k=n}^{\infty} k = nu_n = u_n \ast u_n$$

where the $\ast$ represents convolution. If this is treated in the frequency domain we obtain $z$-transforms of a very simple second-order pole

$$nu_n \leftrightarrow \frac{2}{(z - 1)^2}.$$

This follows from the geometric series

$$\frac{1}{1 - z} = \sum_{n} z^n$$

with $z = e^s$, and the definition of convolution.

The Laplace transform does not require that the series converge, rather that the series have a region of convergence that is properly specified. Thus the non-convergent series $nu_n$ is perfectly well defined, just like

$$tu(t) = u(t) \ast u(t) \leftrightarrow \frac{1!}{s^2}$$

is well defined, in the Laplace transform sense. More generally

$$t^n u(t) \leftrightarrow \frac{n!}{s^{n+1}}.$$

From this easily understood relationship we can begin to understand $\Gamma(s)$, as the analytic extension of the factorial. Its definition is simply related to the inverse Laplace transform, which is an

---

12 Need to verify the exact form of these relationships, not work from my memory.
integral. But to go there we must be able to think in the complex frequency domain. In fact we have the very simple definition for $\Gamma(p)$ with $p \in \mathbb{C}$

$$t^{p-1}u(t) \leftrightarrow \frac{\Gamma(p)}{s^p}$$

which totally explains $\Gamma(p)$. Thinking in the time domain is crucial for my understanding.

Since

$$\sum_{k=0}^{\infty} \delta(t - kT_p) \leftrightarrow \zeta_p(s) = \frac{1}{1 - e^{-st_p}} = \sum_{0}^{\infty} e^{-snT_p}, \quad (3.41)$$

by changing the sign of $s$ we have

$$\sum_{k=0}^{\infty} \delta(t + kT_p) \leftrightarrow \zeta_p(-s) = \frac{1}{1 - e^{sT_p}} = \sum_{0}^{\infty} e^{snT_p}, \quad (3.42)$$

Alternatively we may “shift” $s \rightarrow s + 1$, resulting in

$$\sum_{k=0}^{\infty} \delta(t - kT_p) \leftrightarrow \zeta_p(s - 1) = \frac{1}{1 - e^{-(s-1)t_p}} = \sum_{0}^{\infty} e^{snT_p}e^{-snT_p}, \quad (3.43)$$

3.5 Week 8

3.5.1 Lec 21 The ten postulates of System of algebraic Networks

Physical system obey very important rules, that follow from the physics. It is helpful to summarize these physical restrictions in terms of postulates, presented in terms of a taxonomy, or categorization method, of the fundamental properties of physical systems. Nine of these are listed below. These nine come from a recently published paper (Kim and Allen, 2013). It is possible that given time, others could be added.

A taxonomy of physical systems comes from a systematic summary of the laws of physics, which includes at least the nine basic network postulates, described in Section 1.3.13. To describe
CHAPTER 3. ALGEBRAIC EQUATIONS: STREAM 2

Figure 3.8: A schematic representation of a 2-port ABCD electro-mechanic system using Hunt parameters $Z_e(s)$, $z_m(s)$, and $T(s)$: electrical impedance, mechanical impedances, and transduction coefficient (Hunt, 1952; Kim and Allen, 2013). Also $V(f)$, $I(f)$, $F(f)$, and $U(f)$ are the frequency domain voltage, current, force, and velocity respectively. Notice how the matrix method ‘factors’ the 2-port model into three $2 \times 2$ matrices. This allows one to separate the physical modeling from the algebra. It is a standard impedance convention that the flows $I(f)$, $U(f)$ are always defined into the port. Thus it is necessary to apply a negative sign on the velocity $-U(f)$ so that it has an outward flow, to feed the next cell with an inward flow. Replace $\Phi$ with $V$.

each of the network postulates it is helpful to begin with the 2-port transmission (aka ABCD, chain) matrix representation, discussed in Section 3.3.2 (p. 191).

As a specific example we show the 2-port transmission matrix for an acoustic transducer (loudspeaker), shown in Fig. 3.8, characterized by the equation

$$
\begin{bmatrix}
\Phi_i \\
I_i
\end{bmatrix} =
\begin{bmatrix}
A(s) & B(s) \\
C(s) & D(s)
\end{bmatrix}
\begin{bmatrix}
F_i \\
-U_i
\end{bmatrix} =
\frac{1}{T}
\begin{bmatrix}
z_m(s) & z_e(s)z_m(s) + T^2 \\
1 & z_e(s)
\end{bmatrix}
\begin{bmatrix}
F_i \\
-U_i
\end{bmatrix}.
$$

(3.44)

This equation comes from the product of the three $2 \times 2$ matrices representing each of the three elements of the figure.

This figure represents the motor of the loudspeaker (not the box that it comes in). The system consists of three elements, the electrical input impedance $Z_e(s)$, a gyrator, which is similar to a transformer, but relates current to force, and an output mechanical impedance $Z_m(s)$. This circuit describes what is needed to fully characterize it operation, from electrical input to mechanical (acoustical) output.

The input is electrical (voltage and current) $[\Phi_i, I_i]$ and the output (load) are the mechanical (force and velocity) $[F_i, U_i]$. The first matrix is the general case, expressed in terms of four unspecified functions $A(s), B(s), C(s), D(s)$, while the second matrix is for the specific example of Fig. 3.8. The four entries are the electrical driving point impedance $Z_e(s)$, the mechanical impedance $z_m(s)$ and the transduction $T = B_0l$ where $B_0$ is the magnetic flux strength and $l$ is the length of the wire crossing the flux. Since the transmission matrix is anti-reciprocal, its determinate $\Delta_T = -1$, as is easily verified.

Other common transduction examples of cross-modality transduction include current–thermal (thermoelectric effect) and force–voltage (piezoelectric effect). These systems are all reciprocal, thus the transduction has the same sign.

Impedance matrix

These nine postulates describe the properties of a system having an input and an output. For the case of an electromagnetic transducer (Loudspeaker) the system is described by the 2-port, as show in Fig. 3.8. The electrical input impedance of a loudspeaker is $Z_e(s)$, defined by

$$
Z_e(s) = \frac{V(\omega)}{I(\omega)} \bigg|_{U=0}.
$$

Note that this driving-point impedance must be causal, thus it has a Laplace transform and therefore is a function of the complex frequency $s = \sigma + j\omega$, whereas the Fourier transforms of the voltage
3.5. WEEK 8

$V(\omega)$ and current $I(\omega)$ are functions of the real radian frequency $\omega$, since the time-domain voltage $v(t) \leftrightarrow V(\omega)$ and the current $i(t) \leftrightarrow I(\omega)$ are signals that may start and stop at any time (they are not typically causal).

The corresponding 2-port impedance matrix for Fig. 3.8 is

$$\begin{bmatrix}
\Phi_i \\
F_i
\end{bmatrix} = \begin{bmatrix}
z_{11}(s) & z_{12}(s) \\
z_{21}(s) & z_{22}(s)
\end{bmatrix} \begin{bmatrix}
I_i \\
U_i
\end{bmatrix} = \begin{bmatrix}
Z_c(s) & -T(s) \\
T(s) & z_m(s)
\end{bmatrix} \begin{bmatrix}
I_i \\
U_i
\end{bmatrix}. \tag{3.45}
$$

Such a description allows one to define Thévenin parameters, a very useful concept used widely in circuit analysis and other network models from other modalities.

The impedance matrix is an alternative description of the system, but with generalized forces $[\Phi_i, F_i]$ on the left and generalized flows $[I_i, U_i]$ on the right. A rearrangement of the equations allows one to go from the ABCD to impedance set of parameters (Van Valkenburg, 1964b). The electromagnetic transducer is anti-reciprocal (P6), $z_{12} = -z_{21} = T = B_0 l$.

Additional or modified postulates

The postulates must go beyond postulates P1-P6 defined by Carlin and Giordano (Section 1.3.13, p. 95), when there are interaction of waves and a structured medium, along with other properties not covered by classic network theory. Assuming QS, the wavelength must be large relative to the medium’s lattice constants. Thus the QS property must be extended to three dimensions, and possibly to the cases of an-isotropic and random media.

Causality: P1  As stated above, due to causality the negative properties (e.g., negative refractive index) must be limited in bandwidth, as a result of the Cauchy-Riemann conditions. However even causality needs to be extended to include the delay, as quantified by the d’Alembert solution to the wave equation, which means that the delay is proportional to the distance. Thus we generalize P1 to include the space dependent delay. When we wish to discuss this property we denote it Einstein causality, which says that the delay must be proportional to the distance $x$, with impulse response $\delta(t - x/c)$.

Linearity: P2  The wave properties of may be non-linear. This is not restrictive as most physical systems are naturally nonlinear. For example, a capacitor is inherently nonlinear: as the charge builds up on the plates of the capacitor, a stress is applied to the intermediate dielectric due to the electrostatic force $F = qE$. In a similar manner, an inductor is nonlinear. Two wires carrying a current are attracted or repelled, due to the force created by the flux. The net force is the product of the two fluxes due to each current.

In summary, most physical systems are naturally nonlinear, it’s simply a matter of degree. An important counter example is a amplifier with negative feedback, with very large open-loop gain. There are, therefore, many types of non-linear, instantaneous and those with memory (e.g., hysteresis). Given the nature of P1, even an instantaneous non-linearity may be ruled out. The linear model is so critical for our analysis, providing fundamental understanding, that we frequently take P1 and P2 for granted.

Passive/Active: P3  This postulate is about conservation of energy and Otto Brune’s positive Real (PR aka physically realizable) condition, that every passive impedance must obey. Following up on the earlier work of his primary PhD thesis advisor Wilhelm Cauer (1900-1945) and Ernst
Guillemin, along with Norbert Weiner and Vannevar Bush at MIT, Otto Brune mathematically characterized the properties of every PR 1-port driving point impedance (Brune, 1931a).

When the input resistance of the impedance is real, the system is said to be passive, which means the system obeys conservation of energy. The real part of $Z(s)$ is positive if and only if the corresponding reflectance is less than 1 in magnitude. The definition of the reflectance of $Z(s)$ is defined as a bilinear transformation of the impedance, normalized by its surge resistance $r_0$ (Campbell, 1903)

$$
\Gamma(s) = \frac{Z(s) - r_0}{Z(s) + r_0}.
$$

The surge resistance is defined in terms of the inverse Laplace transform of $Z(s) \leftrightarrow z(t)$, which must have the form

$$
z(t) = r_0\delta(t) + \zeta(t),
$$

where $\zeta(t) = 0$ for $t < 0$. It naturally follows that $\gamma(t) \leftrightarrow \Gamma(s)$ is zero for negative and zero time, namely $\gamma(0) = 0$, $t \leq 0$. at

Given any linear PR impedance $Z(s) = R(\sigma, \omega) + jX(\sigma, \omega)$, having real part $R(\sigma, \omega)$ and imaginary part $X(\sigma, \omega)$, the impedance is defined as being PR (Brune, 1931a) if and only if

$$
R(\sigma \geq 0, \omega) \geq 0.
$$

That is, the real part of any PR impedance is non-negative everywhere in the right half $s$ plane ($\sigma \geq 0$). This is a very strong condition on the complex analytic function $Z(s)$ of a complex variable $s$. This condition is equivalent to any of the following statements (Van Valkenburg, 1964a):

1. There are no poles or zeros in the right half plane ($Z(s)$ may have poles and zeros on the $\sigma = 0$ axis).

2. If $Z(s)$ is PR then its reciprocal $Y(s) = 1/Z(s)$ is PR (the poles and zeros swap).

3. If the impedance may be written as the ratio of two polynomials (a limited case, related to the quasi-statics approximation, P9) having degrees $N$ and $L$, then $|N - L| \leq 1$.

4. The angle of the impedance $\theta \equiv \angle Z$ lies between $[-\pi \leq \theta \leq \pi]$.

5. The impedance and its reciprocal are complex analytic in the right half plane, thus they each obey the Cauchy Riemann conditions there.

**Energy and Power:** The PR (positive real or Physically realizable) condition assures that every impedance is positive-definite (PD), thus guaranteeing conservation of energy is obeyed (Schwinger and Saxon, 1968, p.17). This means that the total energy absorbed by any PR impedance must remain positive for all time, namely

$$
E(t) = \int_{-\infty}^{t} v(t)i(t) \ dt = \int_{-\infty}^{t} i(t) \ast z(t) \ i(t) \ dt > 0,
$$

where $i(t)$ is any current, $v(t) = z(t) \ast i(t)$ is the corresponding voltage and $z(t)$ is the real causal impulse response of the impedance, e.g., $z(t) \leftrightarrow Z(s)$ are a Laplace Transform pair. In summary, if $Z(s)$ is PR, $E(t)$ is PD.
As discussed in detail by Van Valkenburg, any rational PR impedance can be represented as a partial fraction expansion, which can be expanded into first-order poles as

$$Z(s) = K \prod_{i=1}^{L} \frac{1}{(s - n_i)} \prod_{k=1}^{N} \frac{1}{(s - d_k)} = \sum_{n} \frac{\rho_n}{s - s_n} e^{j(\theta_n - \theta_d)}, \quad (3.47)$$

where \(\rho_n\) is a complex scale factor (residue). Every pole in a PR function has only simple poles and zeros, requiring that \(|L - N| \leq 1\) (Van Valkenburg, 1964b).

Whereas the PD property clearly follows P3 (conservation of energy), the physics is not so clear. Specifically what is the physical meaning of the specific constraints on \(Z(s)\)? In many ways, the impedance concept is highly artificial, as expressed by P1-P7.

When the impedance is not rational, special care must be taken. An example of this is the semi-inductor \(M/\sqrt{s}\) and semi-capacitor \(K/\sqrt{s}\) due, for example, to the skin effect in EM theory and viscous and thermal losses in acoustics, both of which are frequency dependent boundary-layer diffusion losses (Vanderkooy, 1989). They remain positive-real but have a branch cut, thus are double valued in frequency.

**Real time response: P4** The impulse response of every physical system is real, vs. complex. This requires that the Laplace Transform have conjugate-symmetric symmetry \(H(s) = H^*(s^*)\), where the * indicates conjugation (e.g., \(R(\sigma, \omega) + X(\sigma, \omega) = R(\sigma, \omega) - X(\sigma, -\omega)\)).

**Time invariant: P5** The meaning of time-invariant requires that the impulse response of a system does not change over time. This requires that the system coefficients of the differential equation describing the system are constant (independent of time).

**Rayleigh Reciprocity: P6** Reciprocity is defined in terms of the unloaded output voltage that results from an input current. Specifically

$$\begin{bmatrix} z_{11}(s) & z_{12}(s) \\ z_{21}(s) & z_{22}(s) \end{bmatrix} = \frac{1}{C(s)} \begin{bmatrix} A(s) & \Delta_T \\ 1 & D(s) \end{bmatrix}, \quad (3.48)$$

where \(\Delta_T = A(s)D(s) - B(s)C(s) = \pm 1\) for the reciprocal and anti-reciprocal systems respectively. This is best understood in term of Eq. 3.45. The off-diagonal coefficients \(z_{12}(s)\) and \(z_{21}(s)\) are defined as

$$z_{12}(s) = \frac{\Phi_i}{U_i} \bigg|_{I_i=0} \quad z_{21}(s) = \frac{F_i}{I_i} \bigg|_{U_i=0}. \quad \text{When these off-diagonal elements are equal \([z_{12}(s) = z_{21}(s)]\) the system is said to obey Rayleigh reciprocity. If they are opposite in sign \([z_{12}(s) = -z_{21}(s)]\), the system is said to be anti-reciprocal. If a network has neither of the reciprocal or anti-reciprocal characteristics, then we denote it as non-reciprocal (McMillan, 1946). The most comprehensive discussion of reciprocity, even to this day, is that of Rayleigh (1896, Vol. I). The reciprocal case may be modeled as an ideal transformer (Van Valkenburg, 1964a) while for the anti-reciprocal case the generalized force and flow are swapped across the 2-port. An electromagnetic transducer (e.g., a moving coil loudspeaker or electrical motor) is anti-reciprocal (Kim and Allen, 2013; Beranek and Mellow, 2012), requiring a gyrator rather than a transformer, as shown in Fig. 3.8.
Reversibility: P7  A second 2-port property is the reversible/non-reversible postulate. A reversible system is invariant to the input and output impedances being swapped. This property is defined by the input and output impedances being equal.

Referring to Eq. 3.48, when the system is reversible $z_{11}(s) = z_{22}(s)$ or in terms of the transmission matrix variables $A(s) = \frac{D(s)}{C(s)}$ or simply $A(s) = D(s)$ assuming $C(s) \neq 0$.

An example of a non-reversible system is a transformer where the turns ratio is not one. Also an ideal operational amplifier (when the power is turned on) is non-reversible due to the large impedance difference between the input and output. Furthermore it is active (it has a power gain, due to the current gain at constant voltage) (Van Valkenburg, 1964b).

Generalizations of this lead to group theory, and Noether’s theorem. These generalizations apply to systems with many modes whereas quasi-statics holds when operate below a cutoff frequency (Table 3.2), meaning that like the case of the transmission line, there are no propagating transverse modes. While this assumption is never exact, it leads to highly accurate results because the non-propagating evanescent transverse modes are attenuated over a short distance, and thus, in practice, may be ignored (Montgomery et al., 1948; Schwinger and Saxon, 1968, Chap. 9-11).

We extend the Carlin and Giordano postulate set to include (P7) Reversibility, which was refined by Van Valkenburg (1964a). To satisfy the reversibility condition, the diagonal components in a system’s impedance matrix must be equal. In other words, the input force and the flow are proportional to the output force and flow, respectively (i.e., $Z_e = z_m$).

Spatial invariant: P8  The characteristic impedance and wave number $\kappa(s, x)$ may be strongly frequency and/or spatially dependent, or even be negative over some limited frequency ranges. Due to causality, the concept of a negative group velocity must be restricted to a limited bandwidth (Brillouin, 1960). As is made clear by Einstein’s theory of relativity, all materials must be strictly causal (P1), a view that must therefore apply to acoustics, but at a very different time scale. We first discuss generalized postulates, expanding on those of Carlin and Giordano.

The QS constraint: P9  When a system is described by the wave equation, delay is introduced between two points in space, which depends on the wave speed. When the wavelength is large compared to the delay, one may successfully apply the quasi-static approximation. This method has wide-spread application, and is frequency used without mention of the assumption. This can lead to confusion, since the limitations of the approximation may not be appreciated. An example is the use of QS in Quantum Mechanics. The QS approximation has wide spread use when the signals may be accurately approximated by a band-limited signal. Examples include KCL, KVL, wave guides, transmission lines, and most importantly, impedance. The QS property is not mentioned in the six postulates of Carlin and Giordano (1964), thus they need to be extended in some fundamental ways.

When the dimensions of a cellular structure in the material are much less than the wavelength, can the QS approximation be valid. This effect can be viewed as a mode filter that suppresses unwanted (or conversely enhances the desired) modes (Ramo et al., 1965). QSs may be applied to a 3 dimensional specification, as in a semiconductor lattice. But such applications fall outside the scope of this text (Schwinger and Saxon, 1968).

Although I have never seen the point discussed in the literature, the QS approximation is applied when defining Green’s theorem. For example, Gauss’s Law is not true when the volume of the container violates QS, since changes in the distribution of the charge have not reached the boundary, when doing the integral. Thus such integral relationships assume that the system is in
quasi steady-state (i.e., that QS holds).

Table 3.2: There are several ways of indicating the quasi-static (QS) approximation. For network theory there is only one lattice constant $a$, which must be much less than the wavelength (wavelength constraint). These three constraints are not equivalent when the object may be a larger structured medium, spanning many wavelengths, but with a cell structure size much less than the wavelength. For example, each cell could be a Helmholtz resonator, or an electromagnetic transducer (i.e., an earphone).

<table>
<thead>
<tr>
<th>Measure $ka &lt; 1$</th>
<th>Domain Wavenumber constraint</th>
</tr>
</thead>
<tbody>
<tr>
<td>$\lambda &gt; 2\pi a$</td>
<td>Domain Wavelength constraint</td>
</tr>
<tr>
<td>$f_c &lt; c/2\pi a$</td>
<td>Domain Bandwidth constraint</td>
</tr>
</tbody>
</table>

Formally, QS is defined as $ka < 1$ where $k = 2\pi/\lambda = \omega/c$ and $a$ is the cellular dimension or the size of the object. Other ways of expressing this include $\lambda/4 > a$, $\lambda/2\pi > a$, $\lambda > 4a$ or $\lambda > 2\pi a$. It is not clear if it is better to normalize $\lambda$ by 4 (quarter-wavelength constraint) or $2\pi \approx 6.28 > 4$, which is more conservative by a factor of $\pi/2 \approx 1.6$. Also $k$ and $a$ can be vectors, e.g., Eq. 1.21, p. 53, Section 1.4.1.

Schelkunoff may have been the first to formalize this concept (Schelkunoff, 1943) (but not the first to use it, as exemplified by the Helmholtz resonator). George Ashley Campbell was the first to use the concept in the important application of a wave-filter, some 30 years before Schelkunoff (Campbell, 1903). These two men were 40 years apart, and both worked for the telephone company (after 1929, called AT&T Bell Labs) (Fagen, 1975).

There are alternative definitions of the QS approximation, depending on the geometrical cell structure. The alternatives are outlined in Table 3.2.

Summary

A transducer converts between modalities. We propose the general definition of the nine system postulates, that include all transduction modalities, such as electrical, mechanical, and acoustical. It is necessary to generalize the concept of the QS approximation (P9) to allow for guided waves.

Given the combination of the important QS approximation, along with these space-time, linearity, and reciprocity properties, a rigorous definition and characterization a system can thus be established. It is based on a taxonomy of such materials, formulated in terms of material and physical properties and in terms of extended network postulates.

3.5.2 Lec 22 Exam II (Evening)
Chapter 4

Scalar Differential Equations: Stream 3a

4.1 Week 8-continued

4.1.1 Lec 23 Integration in the complex plane

The period of analytic discovery: Coming out of the dark ages, from algebra, to analytic geometry, to calculus.

Complex analytic functions: Cauchy I, II, III

Starting from real analytic functions by Euler, Cauchy moved to integration of complex analytic functions.

Exercise: Algebraically justify Eq. 1.73 (p. 99). Sol: Following the hint gives

\[ z(w) = -j \frac{e^{wj} - e^{-wj}}{e^{wj} + e^{-wj}} = -j \frac{e^{2wj} - 1}{e^{2wj} + 1}. \]

Solving this bilinear equation for \( e^{2wj} \) gives

\[ e^{2wj} = \frac{1 + zj}{1 - zj} = \frac{j - z}{j + z}. \]

Taking the log and using our definition of \( w(z) \) we find

\[ w(z) = \tan^{-1}(z) = -\frac{j}{2} \log \frac{j - z}{j + z}. \]
4.2 Week 9

4.2.1 Lec 24 Cauchy Riemann conditions
4.2.2 Lec 25a Complex analytic functions and Brune impedance (I)
4.2.3 Lec 25b Complex analytic functions and Brune impedance (II)
4.2.4 Lec 26: Summary: Branch points, cuts, Sheets & Transmission Lines

4.3 Week 10

4.3.1 Lec 27 Cauchy Theorem I
4.3.2 Lec 28 Cauchy Theorems: II, III
4.3.3 Lec 29 Inverse Laplace Transform: Causality $t < 0$

4.4 Week 11

4.4.1 Lec 30 Inverse Laplace Transforms II: $t > 0$
4.4.2 Lec 31 Laplace Transform properties
4.4.3 Lec 32: Review for Exam III
Chapter 5

Vector Calculus: Stream 3b

5.1 Stream 3b

5.1.1 Lec 33: Grad, Div, Curl and Laplacian

5.1.2 Lec 34: Scalar wave equation (I)

5.1.3 Lec 35: Scalar wave equation (II)

Equation 1.113 (p. 130) is the traditional Webster horn equation in terms of the average pressure. There is an alternative form of the Webster equation (Webster, 1919; Morse, 1948; Morse and Feshbach, 1953; Pierce, 1981), derived from Eq. 1.126 (p. 138, by taking the partial derivative (abbreviated \( \partial_x \)) of the average pressure, giving

\[
P_{xx} + \mathcal{Z}_x \nabla + \mathcal{Z}(s) \nabla_x = 0.
\]

Next, using Newton’s and Hooke’s equations a second time, one may remove the velocity, obtaining the horn pressure wave equation

\[
\mathcal{P}_{rr} - \left( \mathcal{Z}_r / \mathcal{Z} \right) \mathcal{P}_r = \frac{s^2}{c_o^2} \mathcal{P} \leftrightarrow \frac{1}{c_o^2} \mathcal{P}_{tt}.
\]

in terms of the Laplace Transform pair \( g(r, t) \leftrightarrow \mathcal{P}(r, \omega) \) for pressure, where

\[
\kappa(s) = \pm \sqrt{\mathcal{Z}(r, s) \mathcal{Z}(r, s)} = \pm \frac{s}{c_o}.
\]

The coefficient on \( \mathcal{P}_r \) may be further analyzed, as it represents the damping term in the equation

\[
-\mathcal{Z}_r / \mathcal{Z} = -\partial_r \ln \mathcal{Z}_s(r, s)
\]

\[
= -\partial_r \left( \ln s \frac{\rho_0}{A(r)} \right)
\]

\[
= \partial_r \ln A(r)
\]

\[
= \frac{1}{A(r)} \frac{\partial}{\partial r} A(r)
\]
Equation 1.138 in fact shows that by integration by parts, this “loss” term in Eq. 5.1 is related to the area function in an interesting way. More thought is required to figure this out. How does the formula for the characteristic impedance (Eqs. 1.132 & 5.29, p. 139) come about? This seems to be a very important question, that needs to be fully addressed.

Examples:
The primitiv solutions for the four horns are summarized in Table 5.1. In each case we may start from Eq. 5.3 with the area $A(r)$ given for a particular case to find the differential equation for that geometry, and then find the primitive solutions.

Exercise: By direct differentiation show that $\delta(t \mp x/c_o) \leftrightarrow e^{\mp \kappa x} e^{st}$ are solutions of the 1-dimensional wave equation. First show this for the time-domain solution and then again for the frequency domain solution. Third shown that this pair of functions are Laplace transforms of each other.

Conical horn: For the spherical geometry $A(r) = A_o r^2$ (See Table 5.1), resulting in the spherical horn wave equation

$$\frac{1}{A_o r^2} \frac{\partial}{\partial r} \left( A_o r^2 \frac{\partial \varrho}{\partial r} \right) = \frac{\partial^2 \varrho}{\partial r^2} = \frac{1}{c^2} \frac{\partial^2 \varrho}{\partial t^2}. \quad (5.3)$$

having primitive solutions

$$\mathcal{P}^\pm = e^{\mp \kappa r}/r.$$

Here the + is the “outbound” wave and the – is the inbound wave. As we shall see, when the area is variable, each of the primitive solutions has local reflections due to this variation. This gives rise to a reactive mass-component in the radiation impedance.

Surge admittance: Every radiation admittance may be written as $y_{rad}(t) \leftrightarrow Y_{rad}(s)$ which may be further split into a real surge admittance $y_o \delta(t)$ (Campbell, 1922), and a causal remainder admittance $y_r(t)$, as

$$y_{rad}(t) = y_o \delta(t) + y_r(t).$$

Alternatively this may also be written as the sum of an impedance surge and remainder components $z_{rad}(t) = z_o \delta(t) + z_r(t)$. These functions are inverses of each other in the convolution sense, namely $y_{rad}(t) \ast z_{rad}(t) = \delta(t)$, which follows from the definition $Z_{rad}(s)Y_{rad}(s) = 1$. Any function having a causal inverse is said to be minimum phase thus every impedance and admittance must be minimum phase. The remainder characterizes the dispersive component of the impedance, thus when it is zero, the impedance is purely real (e.g., the reactance is zero).

Wave admittance $Y(x,s)$: The driving-point wave admittance, defined as the ratio of the volume velocity $V(r,s)$ to the average pressure $P(r,s)$ at any point $r$ along the range axis, may be interpreted as follows: If the horn were split at any point $x$ (i.e., a node is defined), the pressure at the two throats are the same ($P^+ = P^-$). The velocities are defined as into the port, thus

---

1For spherical case we typically take the radius $r$ as the range variable.
2Since it is real it would best be called a surge conductance.
3It seem obvious that $y_o \equiv Y$?
$V^+ = -V^-$. Due to this definition of the flow into the port, the total velocity is the difference of these two driving point node velocities ($V = V^+ - V^-$). Consistent with Kirchhoff’s laws, the wave admittance as the sum of the two radiation admittances

$$Y(x, s) \equiv Y^+_{rad}(x, s) + Y^-_{rad}(x, s). \quad (5.4)$$

**Impedance Matrix:** For a finite section of horn, the $2 \times 2$ impedance matrix (a generalization of Ohm’s Law) may expressed in terms of the ABCD matrix elements (Van Valkenburg, 1964b) as

$$\begin{bmatrix} P_o \\ \mathcal{P}_L \end{bmatrix} = \frac{1}{C(s)} \begin{bmatrix} \mathcal{A}(s) & \Delta_T \\ 1 & \mathcal{D}(s) \end{bmatrix} \begin{bmatrix} \mathcal{V}_o \\ \mathcal{V}_L \end{bmatrix}. \quad (5.5)$$

Note that $\Delta_T = 1$ since the horn must be reciprocal (Morse and Feshbach, 1953; Hunt, 1982; Pierce, 1981).

While the Transmission (ABCD) matrix is convenient when modeling, the impedance matrix (and its inverse, the admittance matrix) are useful when one makes experimental measurements. For example

$$Y_o|_{\mathcal{V}_L=0} \equiv \frac{C(s)}{\mathcal{A}(s)} \quad \text{and} \quad Y_L|_{\mathcal{V}_o=0} \equiv \frac{C(s)}{\mathcal{D}(s)}$$

are the unloaded input admittances of the horn looking into the two ports (Eq. 5.30). These admittances are typically easily measured experimentally, given access to the endpoints.

In section 1.5.9 we work out these relationships for the trivial case of the 1D horn (Goldsmith and Minton, 1924; Olson, 1947). Then the cases for 2D, 3D and exponential horns are derived. The parameters for each of the cases is provided in Table 5.1.

**Transmission (ABCD) matrix:** The transmission matrix (Eq. 1.61, p. 79) is useful for computing the cascade of several system, such as a horn driven by a Thévenin system and loaded by the radiation impedance or a cascade of several horns. The solution of a horn having finite length may be expressed in terms of a 2-port ABCD matrix, that relates the pressure and volume velocity at the input and output ports (the two ends) of the horn ($x = 0$ and $x = L$)

$$\begin{bmatrix} P_o \\ \mathcal{P}_L \end{bmatrix} = \begin{bmatrix} \mathcal{A}(s) & \mathcal{B}(s) \\ \mathcal{C}(s) & \mathcal{D}(s) \end{bmatrix} \begin{bmatrix} P_o \\ \mathcal{P}_L \end{bmatrix}.$$ \quad (5.6)

Note that $\mathcal{A}(s) \equiv \frac{P_o}{\mathcal{P}_L}|_{\mathcal{V}_L=0}$ must not be confused with the horn area $A(x)$ (note the different font).

By definition, the output velocity $\mathcal{V}_L$, of an ABCD matrix is out of the port, hence the negative sign, since $\mathcal{V}_o, \mathcal{V}_L$ are defined into their respective ports (Orfanidis, 2009). When the system is reversible, $\mathcal{A} = \mathcal{D}$, reciprocal when $\Delta_T \equiv AD - BC = 1$, and anti-reciprocal when $\Delta_T = -1$. With the trivial exception of the uniform horn, all horns are non-reversible and reciprocal.

**Thanksgiving Holiday 11/18–11/26 2017**

**2D parabolic Horn**

For 2D cylindrical waves the area function is $A(r) = A_o r$, (horn radius $\propto \sqrt{r}$) thus the Webster horn equation reduces to the cylindrical wave equation (Appendix 1.5.7.)

$$P_{rr}(r, \omega) + \frac{1}{r} P_r(r, \omega) = \kappa^2(s) P(r, \omega)$$
having Bessel function primitive solutions

\[ \mathcal{P}^+(r, \omega) = J_0(\omega r/c) - iY_0(\omega r/c) = H^+_0(-jkr) \]

and

\[ \mathcal{P}^-(r, \omega) = J_0(\omega r/c) + iY_0(\omega r/c) = H^-_0(-jkr), \]

where \( J_0 \) and \( Y_0 \) are the standard two types Bessel functions (like \( \cos() \) and \( \sin() \)), and \( H^\pm_0(r) \) are the two kinds of Hankel function, of the first (-) and second (+) kind (similar to \( e^{\pm jkr} \)), all of order zero (indicated by the subscript) (Salmon, 1946; Olson, 1947; Morse and Feshbach, 1953).

2D input admittance: Given a half-infinite section of parabolic horn, from Eq. 1.134

\[ Y^\pm_{in}(r, s) = \mp Y_0' \frac{\partial}{\partial r} \ln H^\pm_0(-jkr) = \mp jY_0' \frac{H^\pm_0}{H^\pm_0}, \quad (5.7) \]

2D ABCD Transmission matrix: Based on Eq. 5.27 (p. 223)

\[
\begin{bmatrix}
\mathcal{P}_o \\
\mathcal{V}_o
\end{bmatrix}
= \frac{1}{\Delta_L}
\begin{bmatrix}
1 & \mathcal{P}_{-L}^- \\
Y_{in}^+ & -Y_{in}^- \mathcal{P}_{-L}^-
\end{bmatrix}
\begin{bmatrix}
-1 & Y_{in}^- \\
\mathcal{P}_L^+ & \mathcal{P}_L^-
\end{bmatrix}
\begin{bmatrix}
\mathcal{P}_L \\
-\mathcal{V}_L
\end{bmatrix}, \quad (5.8)
\]

Verify the following

\[
\begin{bmatrix}
\mathcal{P}_o \\
\mathcal{V}_o
\end{bmatrix}
= \frac{1}{\Delta_L}
\begin{bmatrix}
1 & H^0_0(r - L) \\
Y_{in}^+ & -Y_{in}^- H^0_0(r - L)
\end{bmatrix}
\begin{bmatrix}
-1 & Y_{in}^- \\
\mathcal{P}_L^+ & \mathcal{P}_L^-
\end{bmatrix}
\begin{bmatrix}
\mathcal{P}_L \\
-\mathcal{V}_L
\end{bmatrix}, \quad (5.9)
\]

where the subscript on each matrix indicates the value of \( r \) at which it is evaluated.

Impedance matrix:

\[
\begin{bmatrix}
\mathcal{P}_o \\
\mathcal{P}_L
\end{bmatrix}
= \frac{1}{C(s)}
\begin{bmatrix}
A(s) & 1s \\
1 & D(s)
\end{bmatrix}
\begin{bmatrix}
\mathcal{V}_o \\
\mathcal{V}_L
\end{bmatrix}, \quad (5.10)
\]

5.1.4 Generalized impedance

5.1.5 Lec 37: Webster Horn Equation: Derivation

Here we transform the acoustic equations Eq. 1.114 and 1.115 (p. 131) into their equivalent integral form Eq. 1.113 (p. 130). This derivation is similar (but not identical) to that of Hanna and Slepian (1924); Pierce (1981, p. 360).

\[ \frac{\partial \ln H^\pm_0(kr)}{\partial r} = -k \frac{H^\pm_0(kr)}{H^\pm_0(kr)} \]
Conservation of momentum: The first step is an integration the normal component of Eq. 1.114 over the pressure iso-surface $S$, defined by $\nabla p = 0$,

$$- \int_S \nabla p(x, t) \cdot dA = \rho_o \frac{\partial}{\partial t} \int_S u(x, t) \cdot dA. \quad (5.11)$$

The average pressure $\rho(x, t)$ is defined by dividing by the total area

$$\rho(x, t) \equiv \frac{1}{A(x)} \int_S p(x, t) \hat{n} \cdot dA. \quad (5.12)$$

From the definition of the gradient operator

$$\nabla p = \frac{\partial p}{\partial x} \hat{n}, \quad (5.13)$$

where $\hat{n}$ is a unit vector perpendicular to the iso-pressure surface $S$. Thus the left side of Eq. 1.114 reduces to $\partial \rho(x, t)/\partial x$.

The integral on the right side defines the volume velocity

$$\nu(x, t) \equiv \int_S u(x, t) \cdot dA. \quad (5.14)$$

Thus the integral form of Eq. 1.114 becomes

$$\frac{\partial}{\partial x} \rho(x, t) = - \frac{\rho_o}{A(x)} \frac{\partial}{\partial t} \nu(x, t). \quad (5.15)$$

Conservation of mass: Integrating Eq. 1.115 over the volume $V$ gives

$$- \int_V \nabla \cdot u \, dV = \frac{1}{\eta_o \rho_o} \frac{\partial}{\partial t} \int_V p(x, t) \, dV.$$
Volume $V$ is defined by two iso-pressure surfaces between $x$ and $x + \delta x$ (Fig. 5.2, shaded blue). On the right hand side we use our definition for the *average pressure* (i.e., Eq. 5.12), integrated over the thickness $\delta x$.

Applying Gauss’ law to the left hand side, and using the definition of $\varrho$ (on the right), in the limit $\delta x \rightarrow 0$, gives

$$\frac{\partial \nu}{\partial x} = - \frac{A(x)}{\eta_0 P_0} \frac{\partial \varrho}{\partial t}. \quad (5.16)$$

Finally, writing Eq. 5.15 and Eq. 5.16 in matrix form, in the frequency domain ($\partial_t \leftrightarrow \mathbf{s}$), results in Eq. 1.126 (p. 138).

### 5.2 Laplacian operator in $N$ dimensions

To show that the Webster equation is in agreement with the wave equation in 2 and three dimensions, we need to express the Laplacian, and then determine $F(n)$ of Table 5.1. In general it may be shown that, in $N$ dimensions (Sommerfeld, 1949, p. 227)

$$\nabla^2 r P \equiv \frac{1}{r^{N-1}} \frac{\partial}{\partial r} \left( r^{N-1} \frac{\partial P}{\partial r} \right) \quad (5.17)$$

This may be expanded as

$$\frac{\partial^2 P}{\partial r^2} + F(r) \frac{\partial P}{\partial r}, \quad (5.18)$$

where $F(r) = (N - 1)/r$ is the same as in Table 5.1. In terms of the Webster Horn equation, $F(x) \equiv -\partial \ln \mathscr{Z}(x,s)/\partial x$. Thus we see there is a fundamental relation between $\mathscr{Z}$ and the dimensionality of the horn.

For each $N$, $r^{N-1}$ is proportional to the area function $A(x)$. This generalizes to the Webster Horn equation Eq. 1.126. For the case of $N = 3$ (the conical Horn) there is a special relationship

$$\nabla^2 r P \equiv \frac{1}{r^2} \partial_r r^2 \partial_r P \quad (5.19)$$

$$= \frac{1}{r} \partial_{rr} r P \quad (5.20)$$

\(^5\)As shown in Fig. 5.2, we convert the divergence into the difference between two volume velocities, namely $\nu(x + \delta x) - \nu(x)$, and $\partial \nu/\partial x$ as the limit of this difference over $\delta x$, as $\delta x \rightarrow 0$.

\(^6\)http://en.wikipedia.org/wiki/Laplacian#Three_dimensions
resulting in the general d’Alembert solutions
\[ P^\pm(r, s) = \frac{1}{r} e^{\mp \kappa(s) r}. \]

**Exercise:** Prove this last result by expanding Eq. 5.19, 5.20 using the chain rule.

**Sol:** To show these are equal, expanded them using the chain rule.

- Expanding Eq. 5.19:
  \[
  \frac{1}{r^2} \partial_r r^2 \partial_r P = \frac{1}{r^2} \left( 2r + r^2 \partial_r \right) \partial_r P = \frac{2}{r} P_r + P_{rr}.
  \]

- Expanding Eq. 5.20:
  \[
  \frac{1}{r} \partial_r r P = \frac{1}{r} \partial_r \left( P + r P_r \right) = \frac{1}{r} (P_r + P + r P_{rr}) = \frac{2}{r} P_r + P_{rr}.
  \]

**Boundary Conditions:** The pressure and velocity at any point \( x \) be written in terms of a superposition of the two homogeneous solutions \( P^+(x, s) \) and \( P^-(x, s) \) (aka, primitive solutions) of Eq. 1.126 (p. 138), since by d’Alembert principle \( V = V^+ - V^- \) and \( P = P^+ + P^- \). The formula for the input admittance at any point \( x \) on the line is given in terms of the forward and reverse primitive solutions

\[
Y_{in}(r) \equiv \frac{V(r, \omega)}{P(r, \omega)} = \frac{V^+ - V^-}{P^+ + P^-} = \frac{V_+}{P_+} \left( \frac{1 - V^-/V^+}{1 + P^-/P^+} \right) = \mathcal{Y} \frac{1 - \Gamma(r, s)}{1 + \Gamma(r, s)},
\]

(5.21)

Here the reflectance \( \Gamma(r, s) \) is defined as the ratio of the forward and retrograde traveling waves

\[
\Gamma(r, s) \equiv \frac{V^-(r, \omega)}{V^+(r, \omega)} = \frac{P^-(r, \omega)}{P^+(r, \omega)}.
\]

By a rearrangement of the above terms we may recover the definition of the characteristic admittance

\[
\mathcal{Y}(r) = \frac{V^+(r, \omega)}{P^+(r, \omega)} = \frac{V^-(r, \omega)}{P^-(r, \omega)},
\]

where \( \mathcal{Y}(r) = \rho_o c_o / A(r) \) for lossless horns.

It follows that in matrix notation d’Alembert’s principle may be written as

\[
[ P(x) \quad V(x) ] = \begin{bmatrix} 1 & 1 \\ \mathcal{Y}(x) & -\mathcal{Y}(x) \end{bmatrix} [ P^+(x) \quad P^-(x) ].
\]

(5.22)
Complex-analytic nature of $\Gamma(s)$ and $Z_{in}(s)$: A very important assumption has been made here when expressing the complex reflectance $\Gamma(s)$ as a function of the complex frequency $s = \sigma + j\omega$, even though it is defined by the ratio of two functions of real (radian) frequency $\omega$. This assumption is based on the fact that, like the impedance, the reflectance must be causal. Namely $\gamma(t) \leftrightarrow \Gamma(s)$ is zero for $t < 0$. The same may be assumed of the time-domain impedance $\zeta(t) \leftrightarrow Z_{in}(s)$. That $\gamma(t)$ and $\zeta(t)$ are causal is required by the physics. It follows that both $\Gamma(s)$ and $Z_{in}(x)$ are complex analytic functions of $s$, which means they must have a Taylor series expansion in $s$ everywhere in the right-half $s$ plane ($\sigma > 0$). This follows from the Cauchy’s integral theorem (aka The fundamental theorem of complex calculus.), which is a special case of the Cauchy’s integral formula (aka, The Residue theorem).

The forward and retrograde waves may be any function of frequency $\omega$, since they depend on the source pressure (or velocity) at the input to the horn. The reflectance is a transfer function (thus the source term cancels), that only depends on the impedance (or reflectance) looking into the system (at any position $x$).

As an alternative way to derive $\Gamma(r, s)$, invert Eq. 5.22

$$
\begin{bmatrix}
P^+(x) \\
P^-(x)
\end{bmatrix} = \frac{1}{2Y(x)} \begin{bmatrix}
Y(x) & 1 \\
Y(x) & -1
\end{bmatrix} \begin{bmatrix}
P \\
V
\end{bmatrix} = \frac{1}{2} \begin{bmatrix}
1 & Z(x) \\
1 & -Z(x)
\end{bmatrix} \begin{bmatrix}
P \\
V
\end{bmatrix}
$$

and form the ratio of reflected to incident waves

$$
\Gamma(s) = \frac{P^-}{P^+} = \frac{P - ZV}{P + ZV} = \frac{Z_{in} - Z}{Z_{in} + Z}
$$

It is convenient to define the normalized input impedance $Z_{in}/Z$ when working with this form of $\Gamma(s)$.

Given some experience with these two forms, $Z_{in} \equiv 1/Y_{in}$ and $\Gamma(s)$, one may quickly begin to appreciate the advantage of working with the reflectance over the radiation impedance/admittance (aka immittance). The impedance has so many different forms, each of which are complicated, whereas the reflectance is easily understood, as it is closer to the physics.

Finite length Horns: For a horn of fixed length $L$ these expressions may be rewritten in terms of re-normalized primitive waves. If we define the forward wave $P^+(x)$ as launched from $x = 0$ and the retrograde wave $P^-(x)$ as launched from $x = L$, we may also write the pressure and velocity in terms of the primitives.

$$
\begin{bmatrix}
P(x) \\
V(x)
\end{bmatrix} = \begin{bmatrix}
P^+(x) & P^-(x - L) \\
Y^+(s)P^+(x) & -Y^-(s)P^-(x - L)
\end{bmatrix} \begin{bmatrix}
\alpha \\
\beta
\end{bmatrix}
$$

The characteristic admittance $Y^\pm(x, s)$ depends on the range $x$ and Laplace frequency $s$. Coefficients $\alpha(\omega)$ and $\beta(\omega)$, which depend on frequency $\omega$ (but not $x$), are determined by the boundary condition at $x = L$. To find the four parameters $[A(s), B(s), C(s), D(s)]$ we evaluate the inverse of Eq. 5.24 at $x = L$, substitute this result into 5.24, and then evaluated the matrix product at $x = 0$.

For compactness we adopt the following simplified subscript notation: $P^+_o \equiv P^+(x = 0, s)$, $P^+_L \equiv P^+(x = L, s)$, i.e., $P_L \equiv P(x = L)$ and $V_L \equiv V(x = L)$. The normalization of the primitive solutions are taken as $P^+_o = 1$ and $P^-_o = 1$.

To compute $\Delta_L$, we start with

$$
-\Delta_L(x, s) = P^+(x) Y^-_{rad}(x) P^-(x - L) + P^-(x - L) Y^+_{rad}(x) P^+(x)
$$
5.2. LAPLACIAN OPERATOR IN N DIMENSIONS

evaluated at \( x = L \) (recall \( Y_{\text{rad}} \equiv Y_{\text{rad}}^+(x = L), P_o^+ = P_L^- = 1 \),

\[
-\Delta_L = Y_{\text{rad}} \left[ P_L^- P_o^- + P_o^+ P_L^+ \right],
\]

thus

\[
-\Delta_L = Y_{\text{rad}} \left[ P_L^- P_o^- + P_o^+ P_L^+ \right].
\]

\[
\begin{bmatrix} \alpha \\ \beta \end{bmatrix} = \frac{-1}{\Delta_L} \begin{bmatrix} Y_{\text{rad}}^-(x - L) & P^-(x - L) \\ Y_{\text{rad}}^+(x) & -P^+(x) \end{bmatrix}_L \begin{bmatrix} P_L \\ V_L \end{bmatrix}, \tag{5.25}
\]

which when simplified is

\[
\begin{bmatrix} \alpha \\ \beta \end{bmatrix} = \frac{-1}{\Delta_L} \begin{bmatrix} Y_{\text{rad}}(L) & -1 \\ Y_{\text{rad}}(r_o & P_L^+ \end{bmatrix}_o \begin{bmatrix} P_L \\ V_L \end{bmatrix}. \tag{5.26}
\]

This detailed set of computations results in the following:

\[
\begin{bmatrix} P_o \\ V_o \end{bmatrix} = \frac{-1}{\Delta_L} \begin{bmatrix} 1 & P_o^- \\ Y_{\text{rad}}(r_o) & -Y_{\text{rad}}(r_o) P_o^- \end{bmatrix}_o \begin{bmatrix} P_L \\ P_L^+ \end{bmatrix}_L \begin{bmatrix} P_L \\ -V_L \end{bmatrix}. \tag{5.27}
\]

The subscript to the right of each matrix indicates it is evaluated at \( x = 0 \) or \( x = L \). Here \( P_o^- \) is \( P^-(x - L) \) at \( x = 0 \). The sign of \( \gamma_L \) must be negative to satisfy the definition of every ABCD matrix, that the output velocity (i.e., \( -V_L \)) is out of the port.

The relationship of \( \beta/\alpha \) has special significance because it specifies the ratio of the reflected wave amplitude \( \beta(\omega) \) in terms of the incident wave amplitude \( \alpha(\omega) \). This ratio is know as the reflectance

\[
\gamma_L(t) \leftrightarrow \Gamma_L(s) \equiv \frac{\beta}{\alpha}. \tag{5.28}
\]

It has a critical role in the theory of horns, as we shall see as it is determined by the relative rate of change of the impedance (i.e., area) with range (i.e., \( d \ln(Z)/dx \)).

**Examples:** Given the equation for the two d’Alembert pressure waves \( P^\pm(x, s) \) for the case of the uniform horn (Eq. 1.119, p. 133), with \( \kappa(s) = s/c_o \), the volume velocity is

\[
V^\pm(x, \omega) = \mp \frac{1}{\mathcal{Z}(x, s)} \frac{d}{dx} e^{\mp \kappa(s)x}
\]

\[
= \pm \frac{\kappa(s)}{c_o} P^\pm(x, s)
\]

\[
= V_o \ P^\pm(x, s)
\]

\[
= A_o \rho_o c \ P^\pm(x, s). \]

Thus

\[
\mathcal{Y}^\pm = \frac{A_o}{\rho_o c_o} > 0 \tag{5.29}
\]

is the characteristic admittance of the uniform horn, and independent of the wave direction. This is the simplest case.

For the spherical geometry Eq. 1.5.4 (p. 133):
Thus the input admittance is

$$Y^\pm(r, \omega) = -\frac{1}{\mathcal{Z}(r, s)} \frac{d}{dr} \left[ e^{\mp \kappa(s)r}/r \right]$$

$$= -\frac{1}{\mathcal{Z}(r, s)} \left[ \mp \kappa(s) \frac{e^{\mp \kappa(s)r}/r}{r^2} - \frac{e^{\mp \kappa(s)r}/r}{r^2} \right]$$

$$= \frac{1}{\mathcal{Z}(r, s)} \left[ \pm \kappa(s) \frac{e^{\mp \kappa(s)r}/r}{r^2} + \frac{e^{\mp \kappa(s)r}/r}{r^2} \right]$$

$$= \pm \frac{1}{\mathcal{Z}(r, s)} \left[ \kappa(s) \mathcal{P}^\pm \mp \frac{1}{r} \mathcal{P}^\pm \right]$$

$$= \pm \frac{1}{\mathcal{Z}(r, s)} \left[ \kappa(s) \pm \frac{1}{r} \right] \mathcal{P}^\pm(r, s)$$

$$= \frac{\kappa(s)}{\mathcal{Z}(r, s)} \left[ 1 \pm \frac{1}{\kappa(s) r} \right] \mathcal{P}^\pm(r, s)$$

$$= \mathcal{Y}(r) \left[ 1 \pm \frac{1}{\kappa(s) r} \right] \mathcal{P}^\pm(r, s),$$

where \( A(r) = A_o(r/a)^2 \), \( a \) is the radius at the mouth, \( \mathcal{Y}(r) = A_o r^2/4^2 \rho_o c_o \) and \( \kappa(s) = s/c_o \). Thus the input admittance is

$$Y^\pm_{in}(r, s) = \frac{A_o r^2}{a^2 \rho_o c_o} \left[ 1 \pm \frac{c_o}{sr} \right] \leftrightarrow \frac{A_o r^2}{a^2 \rho_o c_o} \left( \delta(t) \pm \frac{c_o}{r} u(t) \right). \quad (5.30)$$

Since the input impedance is the reciprocal of the admittance,

$$Z^\pm_{in}(r, s) = \frac{a^2 \rho_o c_o}{A_o r^2} \left[ 1 \pm \frac{c_o}{sr} \right] = \frac{\rho_o c_o}{A(r)} \left[ sr/c_o \pm 1 \right] \leftrightarrow \frac{\rho_o c_o}{A(r)} e^{\mp r/c_o} u(t). \quad (5.31)$$

Note that both \( \Re\{Y_{in}\} = \mathcal{Y} > 0 \) and \( \Re\{Z_{in}\} \geq 0 \), as is physically required.

Writing \( Y_{in}(r, s) \) in terms of \( \ln \mathcal{P}^\pm \) greatly simplifies the algebra, and thus provides improved insight into the properties of the radiation admittance. For example, for the spherical wave,

$$\frac{\partial \ln \mathcal{P}^\pm}{\partial r} = \frac{\partial}{\partial r} (\mp \kappa(s)r - \ln r) = \mp \kappa(s) - \frac{1}{r} \leftrightarrow \mp \quad (5.32)$$

Frequently the velocity and pressure are specified as functions of frequency \( \omega \), not complex frequency \( s = \sigma + i\omega \). However since \( Y_{in}(x, s) \) must be causal and be non-negative, we indicate it as a function of \( s \). Since the primitive solutions must be causal and stable functions, they also must be analytic functions of \( s \) for \( \sigma > 0 \).

**Summary of four classic horns:** Table 5.1 is a summary of the properties for four different cases of \( A(r) \) for the uniform, parabolic, conical and exponential horns.

**The quasi-static approximation:** Since the velocity perpendicular to the walls of the horn must be zero, any radial wave propagation is exponentially attenuated (\( \kappa(s) \) is real and negative, i.e., Eq. 1.133 will not describe radial wave propagation), with a space constant of about 1 diameter. The assumption that these radial waves can be ignored (i.e., more than 1 diameter from their source) is called the quasi-static approximation. As the frequency is increased, once \( f \geq f_c = 2c_o/\lambda, \) the
When a finite section of horn is terminated in a load impedance, the forward wave can satisfy the zero normal velocity wall boundary condition, and therefore will not be attenuated. Thus above this critical frequency, radial waves (aka, higher order modes) are supported (κ becomes imaginary). For Eq. 1.113 to describe guided wave propagation, \( f < f_c \). But even under this condition, the solution will not be precise within a diameter (or so) of any discontinuities (i.e., rapid variations) in the area.

Each horn, as determined by the area function \( A(r) \), has a distinct wave equation, and thus a distinct solution. Note that the area function determines the upper cutoff frequency via the quasi-static approximation since \( f_c = c_0/\lambda_c, \lambda_c/2 > d, A(r) = \pi(d/2)^2 \). Thus to satisfy the quasi-static approximation, the frequency \( f \) must be less than the cutoff frequency

\[
 f < f_c(r) = \frac{c_0}{4} \sqrt{\frac{\pi}{A(r)}}. 
\]  
(5.33)

We shall discuss two alternative matrix formulations of these equations, the ABCD transmission matrix, use for computation, and the impedance matrix, used when working with experimental measurements (Pierce, 1981, Chapter 7). For each formulation reciprocity and reversibility show up as different matrix symmetries, as addressed in Section 1.3.13 (p. 95) (Pierce, 1981, p. 195-203).

**Finite horns:** When a finite section of horn is terminated in a load impedance, the forward wave \( \mathcal{P}^+(r,s) \) is reflected at the load, producing a backward traveling wave \( \mathcal{P}^-(r,s) \). The way this is treated is by introducing boundary conditions at the termination, a key concept that needs an explanation.

**Exercise:** Prove that \( \mathcal{Y}^+(r) \) is independent of frequency even when Eq. 1.126 has losses. Sol: This proof rests on the fact that \( \gamma(t) \leftrightarrow \Gamma(s) \) is strictly causal, that is \( \gamma(t) = 0 \) for \( t \leq 0 \). Since \( \Gamma(s) \) is complex analytic, it is causal (i.e., \( \gamma(t < 0) = 0 \)). Thus we need to look at \( t = 0 \). The magnitude of any delta function term in \( \gamma(t) \leftrightarrow \Gamma(s) \), that is

\[
 \gamma(t) = \gamma_o \delta(t) + \tilde{\gamma}(t)
\]

is zero. Lundberg et al. (2007) have shown that

\[
 \gamma_o = \gamma(0^+) - \gamma(0^-). 
\]  
(5.34)
For any causal (complex analytic) function, \( \gamma(0^-) = 0 \). Furthermore from the initial-value theorem,

\[
\gamma(0^+) = \lim_{\sigma \to \infty} s \hat{F}(s)
\]

where \( \hat{F}(s) \) is that part of \( F(s) \) which converges to \( \gamma(0^+) \) as \( \sigma \to \infty \).

But since \( \Gamma(s) \) is complex analytic and bounded in magnitude (\(|\Gamma(\omega j)| \leq 1\)), this limit must be zero. This proof is similar to that of Brune (1931b) for positive real impedances.

Imagine a function that is complex analytic for \( \sigma \geq 1 \), that is bounded by 1 on the \( \sigma = 0 \) axis. Since it is causal, the condition of Eq. 5.34 must be true. The corresponding physical argument is interesting. When a plane wave hits a junction, the only reflection is due to the change in impedance. If there is no change, then there is no reflection. If the change is only due to the losses, then there must be a delay in that reflection, which physically means \( \gamma(0) = 0 \).

While such cases are interesting, and realistic, they lie outside the traditional loss-less Webster horn equation formulation.

5.2.1 Week 14

5.2.2 Lec 39 Geometry of gradient, divergence and curl vector operators

Geometry of gradient, divergence and curl vector operators

5.2.3 Lec 40

5.2.4 Lec 41: The Quasi-static approximation and more applications

Derivation of the vector wave equation

Here we do a full derivation of the vector wave equation for Maxwell’s equations, similar to the derivation of the Webster Horn equation.

Starting from Maxwell’s equations in free space, and using a matrix format

\[
\nabla \times \begin{bmatrix} E(x, t) \\ H(x, t) \end{bmatrix} = \frac{\partial}{\partial t} \begin{bmatrix} -B(x, t) \\ + D(x, t) \end{bmatrix} = \begin{bmatrix} 0 & -\mu_0 \\ \epsilon_0 & 0 \end{bmatrix} \frac{\partial}{\partial t} \begin{bmatrix} E(x, t) \\ H(x, t) \end{bmatrix} \leftrightarrow \begin{bmatrix} 0 & -s \mu_0 \\ s \epsilon_0 & 0 \end{bmatrix} \begin{bmatrix} E(x, \omega) \\ H(x, \omega) \end{bmatrix}
\]

Working in the frequency domain, and taking the curl of both sides gives

\[
\nabla \times \nabla \times \begin{bmatrix} E \\ H \end{bmatrix} = \begin{bmatrix} 0 & -s \mu_0 \\ s \epsilon_0 & 0 \end{bmatrix} \nabla \times \begin{bmatrix} E \\ H \end{bmatrix} = \begin{bmatrix} 0 & -s \mu_0 \\ s \epsilon_0 & 0 \end{bmatrix} \begin{bmatrix} d \gamma(x, \omega) \\ 0 \end{bmatrix}
\]

Using the identity \( \nabla \times \nabla \times () = \nabla \nabla \cdot () - \nabla^2 () \) (Eq. 1.152, p. 150) we obtain

\[
\nabla^2 \begin{bmatrix} E \\ H \end{bmatrix} = s^2 \begin{bmatrix} E \\ H \end{bmatrix}
\]

or

\[
\nabla^2 \begin{bmatrix} E \\ H \end{bmatrix} - s^2 \begin{bmatrix} E \\ H \end{bmatrix} = \nabla \left[ \frac{1}{\epsilon_0} \nabla \cdot D \right] = \frac{1}{\epsilon_0} \begin{bmatrix} \nabla \rho(x, s) \\ 0 \end{bmatrix}
\]

This is the vector wave equation, with an electric excitation term \( \nabla \rho(x, s) \).
5.3 Week 15

5.3.1 Lec 42: The Quasi-static approximation and more applications

The quasi-static approximation failure at high frequencies: At high frequencies the quasi-static approximation must break down. Thus at higher frequencies we need to consider other significant physics of the system, known as higher order modes. A further complication is that at higher frequencies, damping becomes an issue.

In acoustics viscosity and thermal effects are typically ignored, by assuming that wave propagation is irrotational, thus is described by the scalar wave equation. It turns out that these two loss mechanisms are related (Rayleigh, 1896). But to understand why is complex. Helmholtz, with help from Kirchhoff, explained this interaction, and independently published them between 1863 (Helmholtz, 1863b) and 1868 (Kirchhoff, 1868). Their collective theory is summarized by Lord Rayleigh (Rayleigh, 1896), and experimentally verified by Warren P. Mason (Mason, 1928). The nature of the correction is that the wave number is extended to be of form

\[ \kappa(s) = \frac{s + \beta_0 \sqrt{s}}{c_0}, \]

(5.35)

where the forwards \( P^- \) and backwards \( P^+ \) pressure waves propagate as

\[ P_{\pm}(s, x) = e^{-\kappa(s)x}, e^{-\kappa(s)x} \]

(5.36)

with \( \kappa(s) \) the complex conjugate of \( \kappa(s) \), and \( \Re\kappa(s) > 0 \). The term \( \beta_0 \sqrt{s} \) affects both the real and imaginary parts of \( \kappa(s) \). The real part is a frequency dependent loss and the imaginary part introduces a frequency dependent velocity (Mason, 1928).

The frequency where the loss-less part equals the lossy part is an important parameter of the system. This frequency is \( s_0 + \beta_0 \sqrt{s_0} = 0 \), or \( \sqrt{s_0} = \beta_0 \) or \( f_0 = \beta^2 / 2\pi \).

Assuming air at 23.5°C [C], \( c_0 = \sqrt{\eta_0 P_0 / \rho_0} \approx 344 \) [m/s] is the speed of sound, \( \eta_0 = c_p / c_v = 1.4 \) is the ratio of specific heats, \( \mu_0 = 18.5 \times 10^{-6} \) [Pa-s] is the viscosity, \( \rho_0 \approx 1.2 \) [kg/m²] is the density, \( P_0 = 10^5 \) [Pa] (1 atm).

The constant \( \beta_0 = P \eta / 2S \sqrt{\rho_0} \)

\[ \eta' = \sqrt{\mu_0} \left[ 1 + \sqrt{5/2} \left( \eta_0^{1/2} - \eta_0^{-1/2} \right) \right] \]

is a thermodynamic constant, \( P \) is the perimeter of the tube and \( S \) the area (Mason, 1928).

For a cylindrical tube having radius \( R = 2S / P, \beta_0 = \eta'_0 / R \sqrt{\rho} \). To get a feeling for the magnitude of \( \beta_0 \) consider a 7.5 [mm] tube (i.e., the average diameter of the adult ear canal). Then \( \eta'_0 = 6.6180 \times 10^{-3} \) and \( \beta_0 = 1.6110 \). Using these conditions the wave-number cutoff frequency is \( 1.611^2 / 2\pi = 0.4131 \) [Hz]. At 1 kHz the ratio of the loss over the propagation is \( \beta_0 / \sqrt{|s|} = 1.6011 / \sqrt{2\pi 10^3} \approx 2\% \). At 100 [Hz] this is a 6.4% effect.\(^7\)

Mason shows that the wave speed drops from 344 [m/s] at 2.6 [kHz] to 339 [m/s] at 0.4 [kHz], which is a 1.5% reduction in the wave speed. In terms of the losses, this is much larger effect. At 1 [kHz] the loss is 1 [dB/m] for a 7.5 [mm] tube. Note that the loss and the speed of sound vary inversely with the radius. As the radius approaches the boundary layer thickness (the radial distance such that the loss is \( e^{-1} \)), the effect of loss dominates.

In Section 5.3.1 we shall look at some simple problems where we use the quasi-static effect and derive the Kirchhoff voltage and current equations, starting from Maxwell’s equations.

\(^7\)home/jba/Mimosa/2C-FindLengths.16/doc.2-c_calib.14/m/MasonKappa.m
5.3.2 Lec 43: Summary review of Fund Thms of Mathematics
Appendix A

Notation

A.1 Number systems

The notation used in this book is defined in this appendix so that it may be quickly accessed.¹ Where the definition is sketchy, page numbers are provided where these concepts are fully explained, along with many other important and useful definitions. For example a discussion of \( \mathbb{N} \) may be found on page 25. Math symbols such as \( \mathbb{N} \) may be found at the top of the index, since they are difficult to alphabetize.

A.1.1 Units

Strangely, or not, classical mathematics (as taught today in schools) does not contain the concept of units. It seems units have been abstracted away. This makes mathematics distinct from physics, where almost everything has units. Presumably this makes mathematics more general (i.e., abstract). But for the engineering mind, this is not ideal, as it necessarily means that important physical meaning has been surgically removed, by design. We shall stick to SI units when ever possible. Spatial coordinates are quoted in meters \([m]\), and time in seconds \([s]\). Angles in degrees have no units, whereas radians have units of inverse-seconds \([s^{-1}]\).

A.1.2 Symbols and functions

We use \( \ln \) as the log function base \( e \), \( \log \) as base 2, and \( \pi_k \) to indicate the \( k \)th prime (e.g., \( \pi_1 = 2, \pi_2 = 3 \)).

When working with Fourier \( \mathcal{F}T \) and Laplace \( \mathcal{L}T \) transforms, lower case symbols are in the time domain while upper case indicates the frequency domain, as \( f(t) \leftrightarrow F(\omega) \). An important exception are Maxwell’s equations, because they are so widely used as upper case bold letters (e.g., \( \mathbf{E}(x, \omega) \)). It seems logical to change this to conform to lower case, with \( e(x,t) \leftrightarrow \mathbf{E}(x, \omega) \) as the preferred notation.

A.1.3 Special symbols common to mathematical:

There are many pre-defined symbols in mathematics, too many to summarize here. We shall only use a small subset, defined here.

¹https://en.wikipedia.org/wiki/List_of_mathematical_symbols_by_subject#Definition_symbols
A set is a collection of objects that have a common property. A set is defined using braces. For example, if set $P = \{a, b, c\}$ such that $a^2 + b^2 = c^2$, then members of $P$ obey the Pythagorean theorem. Thus we could say that $1, 1, \sqrt{2} \in P$. In this last case the three numbers $1, 1, \sqrt{2}$ are not a set, rather than are in (members of) the Pythagorean set $P$.

- Number sets: $\mathbb{N}$, $\mathbb{P}$, $\mathbb{Z}$, $\mathbb{Q}$, $\mathbb{F}$, $\mathbb{I}$, $\mathbb{R}$, $\mathbb{C}$ are briefly discussed below, and in more detail in Section 1.2.1 on p. 25.

- One can define sets of sets and subsets of sets, and this is prone (in my experience) to error. For example, what is the difference between the number 0 and the null set $\emptyset = \{0\}$? Is $0 \in \emptyset$? Ask a mathematician. It is not a useful construction in the world of engineering.

- They symbol $\perp$ is used in many different ways to indicate two things are perpendicular, orthogonal, or in disjoint sets. In set theory $A \perp B$ is equivalent to $A \cap B = \emptyset$. If two vectors $E, H$ are perpendicular $E \perp H$, then their inner product $E \cdot H = 0$ is zero. One must infer the meaning of $\perp$ from its usage (the context).

A.1.4 Greek letters

The Greek letters used in this text include (at least) $\alpha, \beta, \gamma, \delta, \epsilon, \kappa, \rho, \xi, \omega, \sigma, \phi, \psi, \zeta$, and upper-case $\Gamma, \Xi, \Phi, \Psi, \Delta, \Omega$. Unfortunately some common Greek letters have no upper-case symbol, such as $\alpha, \zeta$.

Many of these are pre-associated in engineering and physics with a specific physical meaning. For example, $\omega [\text{rad}]$ is the radian frequency $2\pi f$, $\rho [\text{kg/m}^3]$ is commonly the density. $\phi, \psi$ are commonly used to indicate angles of a triangle, and $\zeta(s)$ is the Riemann zeta function. Many of these are so well established it makes no sense to define new terms, so we will adopt these common terms (and define them).

Likely you do not know all of these Greek letters, commonly used in mathematics. Some of them are pronounced in strange ways. The symbol $\xi$ is pronounced “see,” $\zeta$ is “zeta,” $\beta$ is “beta,” and $\chi$ is “kie” (rhymes with pie and sky). I will assume you know how to pronounce the others, which are more phonetic in English. One advantage of learning \LaTeX is that all of these math symbols are built in, and thus more easily learned, once you have adopted this powerful open-source math-oriented word-processing system (e.g., used to write this book).

<table>
<thead>
<tr>
<th>Symbol (p. #)</th>
<th>Genus</th>
<th>Examples</th>
<th>Counter Examples</th>
</tr>
</thead>
<tbody>
<tr>
<td>$\mathbb{N}$ (25)</td>
<td>Counting</td>
<td>1, 2, 17, 3, $10^{20}$</td>
<td>0, -10, 5j</td>
</tr>
<tr>
<td>$\mathbb{P}$ (26)</td>
<td>Prime</td>
<td>2, 17, 3, $10^{20}$</td>
<td>0, 1, 4, $3^2$, 12, $-5$</td>
</tr>
<tr>
<td>$\mathbb{Z}$ (26)</td>
<td>Integer</td>
<td>-1, 0, 17, 5j, $-10^{20}$</td>
<td>$1/2, \pi$, $\sqrt{5}$</td>
</tr>
<tr>
<td>$\mathbb{Q}$ (27)</td>
<td>Rational</td>
<td>2/1, 3/2, 1.5, 1.14</td>
<td>$\sqrt{2}$, $3^{-1/3}$, $\pi$</td>
</tr>
<tr>
<td>$\mathbb{F}$ (27)</td>
<td>Fractional</td>
<td>1/2, 7/22</td>
<td>$2/1$, 1/$\sqrt{2}$</td>
</tr>
<tr>
<td>$\mathbb{I}$ (27)</td>
<td>Irrational</td>
<td>$\sqrt{2}$, $3^{-1/3}$, $\pi, e$</td>
<td>Vectors</td>
</tr>
<tr>
<td>$\mathbb{R}$ (27)</td>
<td>Reals</td>
<td>$\sqrt{2}$, $3^{-1/3}$, $\pi$</td>
<td>$2\pi j$</td>
</tr>
<tr>
<td>$\mathbb{C}$ (28)</td>
<td>Complex</td>
<td>1, $\sqrt{2}j$, $3^{-j/3}$, $\pi j$</td>
<td>Vectors</td>
</tr>
</tbody>
</table>
A.1. **NUMBER SYSTEMS**

### A.1.5 Double-Bold notation

Table A.1 indicates the symbol followed by a page number indication where it is discussed, and the Genus (class) of the number type. For example, \( \mathbb{N} > 0 \) indicates the infinite set of *counting numbers* \( \{1, 2, 3, \ldots \} \), not including zero. Starting from any counting number, you get the next one by adding 1. Counting numbers are also know as the *Cardinal numbers*.

We say that a number is in the set with the notation \( 3 \in \mathbb{N} \in \mathbb{R} \), which is read as “3 is in the set of counting numbers, which in turn in the set of real numbers,” or in vernacular language “3 is a real counting number.”

Prime numbers \( (\mathbb{P} \in \mathbb{N} > 1) \) are taken from the counting numbers, do not include 1, and cannot be factored.

The signed integers \( \mathbb{Z} \) include 0 and negative integers. Rational numbers \( \mathbb{Q} \) are historically defined to include \( \mathbb{Z} \), a somewhat inconvenient definition, since the more interesting class are the *fractionals* \( \mathbb{F} \), a subset of rationals \( \mathbb{F} \in \mathbb{Q} \) that exclude the integers (i.e., \( \mathbb{F} \perp \mathbb{Z} \)). This is a useful definition because the rationals \( \mathbb{Q} = \mathbb{Z} \cup \mathbb{F} \) are formed from the union of integers and fractionals.

The rationals may be define, using set notation (a very sloppy language, with incomprehensible syntax) as

\[
\mathbb{Q} = \{p/q : q \neq 0 \& p, q \in \mathbb{Z}\}
\]

which may be read as “the set ‘\( \{ \ldots \} \)’ of all \( p/q \) such that ‘:\( q \neq 0 \), ‘and’ \( p, q \in \mathbb{Z} \). The translation of the symbols is in single (‘\( \ldots \)’) quotes.

Irrational numbers \( \mathbb{I} \) are very special: They are formed by taking a limit of fractionals, as the numerator and denominator \( \to \infty \), and approach a limit point. It follows that irrational numbers must be approximated by fractionals.

The reals \( (\mathbb{R}) \) include complex numbers \( (\mathbb{C}) \) having a zero imaginary part (i.e., \( \mathbb{R} \subset \mathbb{C} \)).

The *size* of a set is denoted by taking the absolute value (e.g., \( |\mathbb{N}| \)). Normally in mathematics this symbol indicates the cardinality, so we are defining it differently from the standard notation.

**Classification of numbers:** From the above definitions there exists a natural heretical structure of numbers:

\[
\mathbb{P} \in \mathbb{N}, \quad \mathbb{Z} : \{\mathbb{N}, 0, -\mathbb{N}\}, \quad \mathbb{F} \perp \mathbb{Z}, \quad \mathbb{Q} : \mathbb{Z} \cup \mathbb{F}, \quad \mathbb{R} : \mathbb{Q} \cup \mathbb{I} \subset \mathbb{C}
\]

1. The primes are a subset of the counting numbers: \( \mathbb{P} \subset \mathbb{N} \).
2. The signed integers \( \mathbb{Z} \) are composed of \( \pm \mathbb{N} \) and 0, thus \( \mathbb{N} \subset \mathbb{Z} \).
3. The fractionals \( \mathbb{F} \) do not include of the signed integers \( \mathbb{Z} \).
4. The rationals \( \mathbb{Q} = \mathbb{Z} \cup \mathbb{F} \) are the union of the signed integers and fractionals
5. Irrational numbers \( \mathbb{I} \) have the special properties \( \mathbb{I} \perp \mathbb{Q} \).
6. The reals \( \mathbb{R} : \mathbb{Q}, \mathbb{I} \) are the union of rationals and *irrationals* \( \mathbb{I} \)
7. Reals \( \mathbb{R} \) may be defined as a subset of those complex numbers \( \mathbb{C} \) having zero imaginary part.
A.2 Vectors

Vectors as columns of ordered sets of scalars $\in \mathbb{C}$. When we write them out in text, we typically use row notation, with the transpose symbol:

$$[a, b, c]^T = \begin{bmatrix} a \\ b \\ c \end{bmatrix}.$$  

This is strictly to save space on the page. The notation for conjugate transpose is $^\dagger$, for example

$$\begin{bmatrix} a \\ b \\ c \end{bmatrix}^\dagger = \begin{bmatrix} a^* \\ b^* \\ c^* \end{bmatrix}.$$  

The above example is said to be a 3-dimensional vector, because it has three components.

**Row vs. column vectors:** With rare exceptions, vectors are columns, denoted column-major.\(^2\) To avoid confusion, it is a good rule to make your mental default column-major, in keeping with most signal processing (vectorized) software.\(^3\) Column vectors are the unstated default of Matlab/Octave, only revealed when matrix operations are performed. The need for the column (or row) major is revealed as a consequence of efficiency when accessing long sequences of numbers from computer memory. For example, when forming the sum of many numbers using the Matlab/Octave command $\text{sum}(A)$, where $A$ is a matrix, by default Matlab/Octave operates on the columns, returning a row vector, of column sums. Specifically

$$\text{sum}\left[ \begin{bmatrix} 1 \\ 2 \\ 3 \\ 4 \end{bmatrix} \right] = [4, 6].$$

If the data were stored in “row-major” order, the answer would have been the column vector $\begin{bmatrix} 3 \\ 7 \end{bmatrix}$.

**Dot products:** A vector dot product is defined to weight vector elements before summing them, resulting in a scalar. The transpose of a vector (a row-vector), typically used as weights on the elements of a vector. Dot products play an important role in vector algebra and calculus. For example

$$\begin{bmatrix} 1 \\ 1 \\ 1 \\ 3 \end{bmatrix} \cdot \begin{bmatrix} 1 \\ 2 \\ 3 \end{bmatrix} = \begin{bmatrix} 1 \\ 1 \\ 2 \\ 3 \end{bmatrix}^T = \begin{bmatrix} 1 \\ 1 \\ 2 \\ 3 \end{bmatrix} = 1 + 2 + 3 = 6.$$  

**Dialects of vector notation:** Physical fields are, by definition, functions of space $\mathbf{x}$ [m], and in the most general case, time $t$[s]. When Laplace transformed, the fields become functions of space and complex frequency (e.g., $\mathbf{E}(\mathbf{x}, t) \leftrightarrow \mathbf{E}(\mathbf{x}, s)$). As before, there are several equivalent vector notations. For example, vector $\mathbf{E}(\mathbf{x}, t)$ may also written as

$$\mathbf{E}(\mathbf{x}, t) = \begin{bmatrix} E_x(\mathbf{x}, t) \\ E_y(\mathbf{x}, t) \\ E_z(\mathbf{x}, t) \end{bmatrix} = \begin{bmatrix} E_x \\ E_y \\ E_z \end{bmatrix}(\mathbf{x}, t) = \begin{bmatrix} E_x, \\ E_y, \\ E_z \end{bmatrix}^T = E_x\hat{x} + E_y\hat{y} + E_z\hat{z}.$$  

\(^2\)https://en.wikipedia.org/wiki/Row-_and_column-major_order  

\(^3\)In contrast, reading words in English is ‘row-major.’
A.2. VECTORS

The above equation, with an equation number, is called a “displayed” equation. The in-line dialect is $E(x, t) = [E_x, E_y, E_z]^T E_x(x, t) \hat{x} + E_y(x, t) \hat{y} + E_z(x, t) \hat{z}$. These are all shorthand notations for expressing the vector. Such usage is similar to a dialect in a language. Another dialect is column-transpose or “in-line” notation, so to place the vector on one line, saving space: $E(x, t) = [E_x, E_y, E_z](x, t)^T$. These several variants on vector notation are frequently used, even interchangeably.

**Complex elements:** When the elements are complex ($\in \mathbb{C}$), the transpose is defined as the complex conjugate of the elements. In such complex cases the transpose conjugate is denoted with $a^\dagger$ rather than $T$. Vectors are also frequency written using a bold font $a^T = \begin{bmatrix} 2j \\ -3j \\ 1 \end{bmatrix} \in \mathbb{C}$.

For this case when the elements are complex, the dot product is a real number

$$a \cdot b = a^\dagger b = \begin{bmatrix} a_1^* \\ a_2^* \\ a_3^* \end{bmatrix} \begin{bmatrix} b_1 \\ b_2 \\ b_3 \end{bmatrix} = a_1^* b_1 + a_2^* b_2 + a_3^* b_3 \in \mathbb{R}.$$  

**Norm of a vector:** The dot product of a vector with itself is called the norm of $a$, designated as

$$||a|| = \sqrt{a^\dagger a} \geq 0$$

which is always non-negative.

Such a construction is useful when $a$ and $b$ are related by an impedance matrix $V(s) = Z(s)I(s)$ and we wish to compute the power. For example, the impedance of a mass is $ms$ and a capacitor is $1/sC$. When given a system of equations (a mechanical or electrical circuit) one may define an impedance matrix.

**Complex power:** In this special case, the complex power $P(s) \in \mathbb{R}(s)$ is defined, in the complex frequency domain ($s$) as

$$P(s) = I^\dagger(s)V(s) = I^\dagger(s)Z(s)I(s) \leftrightarrow p(t). \quad [W]$$

The real part of the complex power must be positive. The imaginary part corresponds to available stored energy.

**GIVE MORE EXAMPLES**

The case of three-dimensions is special, allowing definitions that are not defined in more dimensions. A vector in $\mathbb{R}^3$ labels the point having the coordinates of that vector.

A.2.1 Vectors in $\mathbb{R}^3$

**Dot product in $\mathbb{R}^3$:** The dot product $B \cdot C = ||B|| ||C|| \cos(\theta)$, and $\cos(\theta)$ is called the direction cosine between $B$ and $C$. 

Norm of a vector in $\mathbb{R}^3$: The norm of a vector is the dot product of it with itself

$$||A|| = \sqrt{A \cdot A}$$

Euclidean distance between two points in $\mathbb{R}^3$: The dot product of the difference between two vectors $(A - B) \cdot (A - B)$ is the Euclidean distance between the points they define

$$||A - B|| = \sqrt{(a_1 - b_1)^2 + (a_2 - b_2)^2 + (a_3 - b_3)^2}.$$ 

Cross product: and cross product $A \times B = ||A|| ||B|| \sin(\theta)$ are defined between the two vectors $A$ and $B$.

The triple product: This is defined between three vectors as

$$A \cdot (B \times C) = \det \begin{vmatrix} a_1 & a_2 & a_3 \\ b_1 & b_2 & b_3 \\ c_1 & c_2 & c_3 \end{vmatrix}$$

also defined in Fig. 1.13. This may be indicated without the use of parentheses, since there can be no other meaningful interpretation. However for rigor, parentheses should be used. The triple product is the volume of the parallelepiped (3D-crystal shape) outlined by the three vectors, shown in Fig. 1.13, p. 74.

A.3 Matrices

When working with matrices, the role of the weights and vectors can change, depending on the context. A useful way to view a matrix is as a set of column vectors, weighted by the elements of the column-vector of weights multiplied from the right. For example

$$\begin{bmatrix} a_{11} & a_{12} & a_{13} & \cdots & a_{1M} \\ a_{21} & a_{22} & a_{32} & \cdots & a_{3M} \\ \vdots & \vdots & \vdots & \ddots & \vdots \\ a_{N1} & a_{N2} & a_{N3} & \cdots & a_{NM} \end{bmatrix} \begin{bmatrix} w_1 \\ w_2 \\ \vdots \\ w_M \end{bmatrix} = w_1 \begin{bmatrix} a_{11} \\ a_{21} \\ \vdots \\ a_{N1} \end{bmatrix} + w_2 \begin{bmatrix} a_{12} \\ a_{22} \\ \vdots \\ a_{N2} \end{bmatrix} + \cdots + w_M \begin{bmatrix} a_{1M} \\ a_{2M} \\ \vdots \\ a_{NM} \end{bmatrix},$$

where the weights are $[w_1, w_2, \ldots, w_M]^T$. Alternatively the matrix is a set of row vectors of weights, each of which are applied to the column vector on the right ($[w_1, w_2, \ldots, W_M]^T$).

The determinant of a matrix is denoted either as $\det A$ or simply $|A|$, (as in the absolute value.) The inverse of a square matrix is $A^{-1}$ or $\text{inv} A$. If $|A| = 0$, the inverse does not exist. $AA^{-1} = A^{-1}A$.

Matlab’s notional convention for a row-vector is $[a, b, c]$ and a column-vector is $[a; b; c]$. A prime on a vector takes the complex conjugate transpose. To suppress the conjugation, place a period before the prime. The : argument converts the array into a column vector, without conjugation. A tacit notation in Matlab is that vectors are columns and the index to a vector is a row vector. Matlab defines the notation $1:4$ as the “row-vector” $[1, 2, 3, 4]$, which is unfortunate as it leads users to assume that the default vector is a row. This can lead to serious confusion later, as
Matlab’s default vector is a column. I have not found the above convention explicitly stated, and it took me years to figure this out for myself.

When writing a complex number we shall adopt $\sqrt{-1}$. Matlab/Octave prefer this as well, as its explicit.

Units are SI; Angles in degrees [deg] unless otherwise noted. The units for $\pi$ are always in radians [rad]. Ex: $\sin(\pi)$, $e^{j90^\circ} e^{j\pi/2}$.

A.4 Gaussian elimination

Definitions (Appendix A, Class Notes):

1. **Scalar**: A number, e.g. $\{a, b, c, \alpha, \beta, \ldots\} \in \{\mathbb{Z}, \mathbb{Q}, \mathbb{I}, \mathbb{R}, \mathbb{C}\}$

2. **Vector**: A quantity having direction as well as magnitude, often denoted by a bold-face letter with an arrow, $\vec{x}$. In matrix notation, this is typically represented as a single row $[x_1, x_2, x_3, \ldots]$ or single column $[x_1, x_2, x_3, \ldots]^T$ (where $T$ indicates the transpose). In this class we will typically use column vectors. The vector may also be written out using unit vector notation to indicate direction. For example: $x_3 \times 1 = x_1 \hat{x} + x_2 \hat{y} + x_3 \hat{z} = [x_1, x_2, x_3]^T$, where $\hat{x}, \hat{y}, \hat{z}$ are unit vectors in the $x, y, z$ cartesian directions (here the vector’s subscript $3 \times 1$ indicates its dimensions). The type of notation used may depend on the engineering problem you are solving.

3. **Matrix**: $A = [a_{11}, a_{12}, a_{13}, \ldots, a_{M1}, a_{M2}, a_{M3}, \ldots, a_{MN}]_{N \times M} = \{a_{n,m}\}_{N \times M}$, can be a non-square matrix if the number of elements in each of the vectors ($N$) is not equal to the number of vectors ($M$). When $M = N$, the matrix is square. It may be inverted if its determinant $|A| = \prod \lambda_k \neq 0$ (where $\lambda_k$ are the eigenvalues).

We shall only work with $2 \times 2$ and $3 \times 3$ square matrices throughout this course.

4. **Linear system of equations**: $Ax = b$ where $x$ and $b$ are vectors and matrix $A$ is a square.

   (a) **Inverse**: The solution of this system of equations may be found by finding the inverse $x = A^{-1}b$

   (b) **Equivalence**: If two systems of equations $A_0x = b_0$ and $A_1x = b_1$ have the same solution (i.e., $x = A_0^{-1}b_0 = A_1^{-1}b_1$), they are said to be equivalent.

   (c) **Augmented matrix**: The first type of augmented matrix is defined by combining the matrix with the right-hand-side. For example, given the linear system of equations $Ax = y$

   $$\begin{bmatrix} a & b \\ c & d \end{bmatrix} \begin{bmatrix} x_1 \\ x_2 \end{bmatrix} = \begin{bmatrix} y_1 \\ y_2 \end{bmatrix},$$

   then the augmented matrix is

   $$A|y = \begin{bmatrix} a & b \\ c & d \end{bmatrix} \begin{bmatrix} y_1 \\ y_2 \end{bmatrix}$$

   A second type of augmented matrix may be used for finding the inverse of a matrix (rather than solving a specific instance of linear equations $Ax = b$). In this case the
augmented matrix is

\[
A|I = \begin{bmatrix}
 a & b & 1 & 0 \\
 c & d & 0 & 1 \\
\end{bmatrix}
\]

Performing Gaussian elimination on this matrix, until the left side becomes the identity matrix, yields \(A^{-1}\). This is because multiplying both sides by \(A^{-1}\) gives \(A^{-1}A|A^{-1}I = I\).

5. **Permutation matrix** \((P)\): A matrix that is equivalent to the identity matrix, but with scrambled rows (or columns). Such a matrix has the properties \(\det(P) = \pm 1\) and \(P^2 = I\). For the 2x2 case, there is only one permutation matrix:

\[
P = \begin{bmatrix}
 0 & 1 \\
 1 & 0 \\
\end{bmatrix}
\]

\[
P^2 = \begin{bmatrix}
 0 & 1 \\
 1 & 0 \\
\end{bmatrix} \begin{bmatrix}
 0 & 1 \\
 1 & 0 \\
\end{bmatrix} = \begin{bmatrix}
 1 & 0 \\
 0 & 1 \\
\end{bmatrix}
\]

A permutation matrix \(P\) swaps rows or columns of the matrix it operates on. For example, in the 2x2 case, pre-multiplication swaps the rows

\[
PA = \begin{bmatrix}
 0 & 1 \\
 1 & 0 \\
\end{bmatrix} \begin{bmatrix}
 a & b \\
 \alpha & \beta \\
\end{bmatrix} = \begin{bmatrix}
 \alpha & \beta \\
 a & b \\
\end{bmatrix}
\]

whereas post-multiplication swaps the columns

\[
AP = \begin{bmatrix}
 a & b \\
 \alpha & \beta \\
\end{bmatrix} \begin{bmatrix}
 0 & 1 \\
 1 & 0 \\
\end{bmatrix} = \begin{bmatrix}
 b & a \\
 \beta & \alpha \\
\end{bmatrix}
\]

For the 3x3 case there are \(3 \cdot 2 = 6\) such matrices, including the original 3x3 identity matrix (swap a row with the other 2, then swap the remaining two rows).

6. **Gaussian elimination (GE) matrices** \(G_k\): There are 3 types of elementary row operations, which may be performed without fundamentally altering a system of equations (e.g. the resulting system of equations is equivalent). These operations are (1) swap rows (e.g. using a permutation matrix), (2) scale rows, or (3) perform addition/subtraction of two scaled rows. All such operations can be performed using matrices.

For lack of a better term, we’ll describe these as ‘gaussian elimination’ or ‘GE’ matrices.\(^4\) We will categorize any matrix that performs only elementary row operations (but any number of them) as a ‘GE’ matrix. Therefore, cascade of GE matrices is also a GE matrix.

Consider the GE matrix

\[
G = \begin{bmatrix}
 1 & 0 \\
 1 & -1 \\
\end{bmatrix}
\]

(a) Pre-multiplication scales and adds the rows

\[
GA = \begin{bmatrix}
 1 & 0 \\
 1 & -1 \\
\end{bmatrix} \begin{bmatrix}
 a & b \\
 \alpha & \beta \\
\end{bmatrix} = \begin{bmatrix}
 a & b \\
 a - \alpha & b - \beta \\
\end{bmatrix}
\]

The result is a Gaussian elimination operation.

\(^4\)The term ‘elementary matrix’ may also be used to refer to a matrix that performs an elementary row operation. Typically, each elementary matrix differs from the identity matrix by one single row operation. A cascade of elementary matrices could be used to perform Gaussian elimination.
A.5 Periodic functions

(b) Post-multiplication adds and scales columns.

\[ AG = \begin{bmatrix} a & b \\ \alpha & \beta \end{bmatrix} \begin{bmatrix} 1 & 0 \\ -1 & 1 \end{bmatrix} = \begin{bmatrix} a-b & b \\ \alpha-\beta & \beta \end{bmatrix} \]

Here the second column is subtracted from the first, and placed in the first. The second column is untouched. This operation is not a Gaussian elimination. Therefore, to put Gaussian elimination operations in matrix form, we form a cascade of pre-multiply matrices.

Here \( \det(G) = 1 \), \( G^2 = I \), which won’t always be true if we scale by a number greater than 1. For instance, if \( G = \begin{bmatrix} 1 & 0 \\ m & 1 \end{bmatrix} \) (scale and add), then we have \( \det(G) = 1 \), \( G^n = \begin{bmatrix} 1 & 0 \\ n \cdot m & 1 \end{bmatrix} \).

A.5 Periodic functions

Fourier series tells us that periodic functions are discrete in frequency, with frequencies given by \( nT_x \), where \( T_x \) is the sample period (\( T_x = 1/2F_{\text{max}} \) and \( F_{\text{min}} = F_{\text{max}}/N\text{FT} \)).

This concept is captured by the Fourier series, which is a frequency expansion of a periodic function. This concept is quite general. Periodic in frequency implies discrete in time. Periodic and discrete in time requires periodic and discrete in frequency (the case of the DFT). The modulo function \( \text{mod}n \) is periodic with period \( n \).

A periodic function may be conveniently indicated using double-parentheses notation. This is sometimes known as modular arithmetic. For example

\[ f((t))_T = f(t) = f(t \pm kT), \]

is periodic on \( t, T \in \mathbb{R} \) with a period of \( T \) and \( k \in \mathbb{Z} \). This notation is useful when dealing with Fourier series of periodic functions.

When a discrete valued (e.g., time) sequence is periodic we use square brackets

\[ f[[n]]_N = f[n] = f[n \pm kN], \]

with \( n, k, N \in \mathbb{Z} \) and period \( N \). This notation will be used with discrete-time signals that are periodic, such as the case of the DFT.

A.6 Differential equations vs. Polynomials

A polynomial has degree \( N \) defined by the largest power. A quadratic equation is degree 2, and a cubic has degree 3. We shall indicate a polynomial by the notation

\[ P_N(z) = z^N + a_{N-1}z^{N-1} \cdots a_0. \]

It is a good practice to normalize the polynomial so that \( a_N = 1 \). This will not change the roots, defined by Eq. 1.24 (p. 55). The coefficient on \( z^{N-1} \) is always the sum of the roots \( z_n \) \( (a_{N-1} = \Sigma_n^N z_n) \), and \( a_0 \) is always their product \( (a_0 = \prod_n^N z_n) \).
Differential equations have order (polynomials have degree). If a second order differential equation is Laplace transformed (Lec. 1.3.12, p. 91), one is left with a degree 2 polynomial. For example:

\[
\frac{d^2}{dt^2} y(t) + b \frac{d}{dt} y(t) + c y(t) = \alpha \left( \frac{d}{dt} x(t) + \beta x(t) \right) \leftrightarrow (A.1) \\
(s^2 + bs + c)Y(s) = \alpha (s + \beta)X(s). \quad (A.2) \\
\frac{Y(s)}{X(s)} = \alpha \frac{s + \beta}{s^2 + bs + c} \equiv H(s) \leftrightarrow h(t). \quad (A.3)
\]

Using the same argument as for polynomials, the lead coefficient must always be 1. The coefficient \( \alpha \in \mathbb{R} \) is called the gain. The complex variable \( s \) is the Laplace frequency. Discuss half-derivatives, etc.

The ratio of the output \( Y(s) \) over the input \( X(s) \) is called the system transfer function \( H(s) \). When \( H(s) \) is the ratio of two polynomials in \( s \), the transfer function is said to be bilinear, since it is linear in both the input and output. The roots of the numerator are called the zeros and those of the denominator, the poles. The inverse Laplace transform of the transfer function is called the system impulse response, which describes the system’s output signal \( y(t) \) for any given input signal \( x(t) \), via convolution (i.e., \( y(t) = h(t) \ast x(t) \)).
Appendix B

Linear algebra of 2x2 matrices

B.1 Notation

Definitions:

1. **Scalar**: A number, e.g. \{a, b, c, α, β, ⋯\} ∈ \{Z, Q, I, R, C\}

2. **Vector**: A quantity having direction as well as magnitude, often denoted by a bold-face letter with an arrow, \textbf{x}. In matrix notation, this is typically represented as a single row \([x_1, x_2, x_3, \ldots]\) or single column \([x_1, x_2, x_3, \ldots]^T\) (where \(T\) indicates the transpose). In this class we will typically use column vectors. The vector may also be written out using unit vector notation to indicate direction. For example: \(x_{3×1} = x_1\hat{x} + x_2\hat{y} + x_3\hat{z} = [x_1, x_2, x_3]^T\), where \(\hat{x}, \hat{y}, \hat{z}\) are unit vectors in the \(x, y, z\) Cartesian directions (here the vector’s subscript \(3×1\) indicates its dimensions). The type of notation used may depend on the engineering problem you are solving.

3. **Matrix**: \(A = \begin{bmatrix} a_{11}, a_{21}, a_{31}, \cdots, a_{M1} \\ a_{12}, a_{22}, a_{32}, \cdots, a_{M2} \\ \vdots \end{bmatrix}_{N×M} = \{a_{n,m}\}_{N×M}\), can be a non-square matrix if the number of elements in each of the vectors (\(N\)) is not equal to the number of vectors (\(M\)). When \(M = N\), the matrix is square. It may be inverted if its determinant \(|A| = \prod \lambda_k \neq 0\) (where \(\lambda_k\) are the eigenvalues).

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   (c) **Augmented matrix**: The first type of augmented matrix is defined by combining the matrix with the right-hand-side. For example, given the linear system of equations \(Ax = y\)

\[
\begin{bmatrix}
a & b \\
c & d
\end{bmatrix}
\begin{bmatrix}
x_1 \\
x_2
\end{bmatrix} =
\begin{bmatrix}
y_1 \\
y_2
\end{bmatrix},
\]

239
then the augmented matrix is
\[
A|y = \begin{bmatrix}
a & b \\
c & d
\end{bmatrix}
\begin{bmatrix}
y_1 \\
y_2
\end{bmatrix}
\]

A second type of augmented matrix may be used for finding the inverse of a matrix (rather than solving a specific instance of linear equations \(Ax = b\)). In this case the augmented matrix is
\[
A|I = \begin{bmatrix}
a & b & 1 & 0 \\
c & d & 0 & 1
\end{bmatrix}
\]

Performing Gaussian elimination on this matrix, until the left side becomes the identity matrix, yields \(A^{-1}\). This is because multiplying both sides by \(A^{-1}\) gives \(A^{-1}A|A^{-1}I = I|A^{-1}\).

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1 & 0
\end{bmatrix}
\begin{bmatrix}
a & b \\
\alpha & \beta
\end{bmatrix} = \begin{bmatrix}
\alpha & \beta \\
a & b
\end{bmatrix},
\]
whereas post-multiplication swaps the columns
\[
AP = \begin{bmatrix}
a & b \\
\alpha & \beta
\end{bmatrix}
\begin{bmatrix}
0 & 1 \\
1 & 0
\end{bmatrix} = \begin{bmatrix}
b & a \\
\beta & \alpha
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Consider the GE matrix
\[
G = \begin{bmatrix}
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1 & -1
\end{bmatrix}
\]

\(^1\)The term ‘elementary matrix’ may also be used to refer to a matrix that performs an elementary row operation. Typically, each elementary matrix differs from the identity matrix by one single row operation. A cascade of elementary matrices could be used to perform Gaussian elimination.
(a) Pre-multiplication scales and adds the rows

\[
GA = \begin{bmatrix} 1 & 0 \\ 1 & -1 \end{bmatrix} \begin{bmatrix} a \\ \alpha \end{bmatrix} = \begin{bmatrix} a \\ a - \alpha \end{bmatrix}
\]

The result is a Gaussian elimination operation.

(b) Post-multiplication adds and scales columns.

\[
AG = \begin{bmatrix} a & b \\ \alpha & \beta \end{bmatrix} \begin{bmatrix} 1 & 0 \\ -1 & 1 \end{bmatrix} = \begin{bmatrix} a - b & b \\ \alpha - \beta & \beta \end{bmatrix}
\]

Here the second column is subtracted from the first, and placed in the first. The second column is untouched. This operation is not a Gaussian elimination. Therefore, to put Gaussian elimination operations in matrix form, we form a cascade of pre-multiply matrices.

Here \(\det(G) = 1\), \(G^2 = I\), which won’t always be true if we scale by a number greater than 1. For instance, if \(G = \begin{bmatrix} 1 & 0 \\ m & 1 \end{bmatrix}\) (scale and add), then we have \(\det(G) = 1\), \(G^n = \begin{bmatrix} 1 & 0 \\ n \cdot m & 1 \end{bmatrix}\).

**B.1.1 Gaussian elimination exercises**

Find the solution to the following 3x3 matrix equation \(Ax = b\) by Gaussian elimination. Show your intermediate steps. You can check your work at each step using Matlab.

\[
\begin{bmatrix} 1 & 1 & -1 \\ 3 & 1 & 1 \\ 1 & -1 & 4 \end{bmatrix} \begin{bmatrix} x_1 \\ x_2 \\ x_3 \end{bmatrix} = \begin{bmatrix} 1 \\ 9 \\ 8 \end{bmatrix}
\]

1. Show (i.e., verify) that the first GE matrix \(G_1\), which zeros out all entries in the first column, is given by

\[
G_1 = \begin{bmatrix} 1 & 0 & 0 \\ -3 & 1 & 0 \\ -1 & 1 & 1 \end{bmatrix}
\]

Identify the elementary row operations that this matrix performs. Sol: Operate with GE matrix on \(A\)

\[
G_1[A|b] = \begin{bmatrix} 1 & 0 & 0 \\ -3 & 1 & 0 \\ -1 & 0 & 1 \end{bmatrix} \begin{bmatrix} 1 & 1 & -1 \\ 3 & 1 & 1 \\ 1 & -1 & 4 \end{bmatrix} = \begin{bmatrix} 1 & 1 & -1 \\ 0 & -2 & 4 \\ 0 & -2 & 5 \end{bmatrix}
\]

It scales the first row by -3 and adds it to the second row, and scales the first row by -1 and adds it to the third row.

2. Find a second GE matrix, \(G_2\), to put \(G_1A\) in upper triangular form. Identify the elementary row operations that this matrix performs. Sol:

\[
G_2 = \begin{bmatrix} 1 & 0 & 0 \\ 0 & 1 & 0 \\ 0 & -1 & 1 \end{bmatrix}
\]
which scales the second row by -1 and adds it to the third row. Thus we have

$$G_2G_1[A|b] = \begin{bmatrix} 1 & 0 & 0 \\ 0 & 1 & 0 \\ 0 & -1 & 1 \end{bmatrix} \begin{bmatrix} 1 & 0 & 0 \\ -3 & 1 & 0 \\ -1 & 0 & 1 \end{bmatrix} \begin{bmatrix} 1 & 1 & -1 \\ 3 & 1 & 1 \\ 1 & -1 & 4 \end{bmatrix} = \begin{bmatrix} 1 & 1 & -1 \\ 0 & -2 & 4 \\ 0 & 0 & 1 \end{bmatrix}$$

3. Find a third GE matrix, $G_3$, which scales each row so that its leading term is 1. Identify the elementary row operations that this matrix performs. Sol:

$$G_3 = \begin{bmatrix} 1 & 0/2 \\ 0 & -1/2 \\ 0 & 0 \end{bmatrix}$$

which scales the second row by -1/2. Thus we have

$$G_3G_2G_1[A|b] = \begin{bmatrix} 1 & 0 & 0 \\ 0 & -1/2 & 0 \\ 0 & 0 & 1 \end{bmatrix} \begin{bmatrix} 1 & 0 & 0 \\ 1 & 0 & 0 \\ 0 & -1 & 1 \end{bmatrix} \begin{bmatrix} 1 & 1 & -1 \\ 3 & 1 & 1 \\ -1 & 0 & 1 \end{bmatrix} = \begin{bmatrix} 1 & 1 & -1 \\ 0 & 1 & -2 \\ 0 & 0 & 1 \end{bmatrix}$$

4. Finally, find the last GE matrix, $G_4$, that subtracts a scaled version of row 3 from row 2, and scaled versions of rows 2 and 3 from row 1, such that you are left with the identity matrix ($G_4G_3G_2G_1A = I$). Sol:

$$G_4 = \begin{bmatrix} 1 & -1 & -1 \\ 0 & 1 & 2 \\ 0 & 0 & 1 \end{bmatrix}$$

Thus we have

$$G_4G_3G_2G_1[A|b] = \begin{bmatrix} 1 & -1 & -1 \\ 0 & 1 & 2 \\ 0 & 0 & 1 \end{bmatrix} \begin{bmatrix} 1 & 0 & 0 \\ 0 & -1/2 & 0 \\ 0 & 0 & 1 \end{bmatrix} \begin{bmatrix} 1 & 0 & 0 \\ 1 & 0 & 0 \\ 0 & -1 & 1 \end{bmatrix} \begin{bmatrix} 1 & 1 & -1 \\ -3 & 1 & 1 \\ 1 & -1 & 4 \end{bmatrix} = \begin{bmatrix} 1 & 0 & 0 \\ 0 & 3 & 0 \\ 0 & 0 & 6 \end{bmatrix}$$

5. Solve for $[x_1, x_2, x_3]^T$ using the augmented matrix format $G_4G_3G_2G_1[A|b]$ (where $[A|b]$ is the augmented matrix). Note that if you’ve performed the preceding steps correctly, $x = G_4G_3G_2G_1b$. Sol: From the preceding problems, we see that $[x_1, x_2, x_3]^T = [3, -1, 1]^T$

### B.2 Inverse of the 2x2 matrix

We shall now apply Gaussian elimination to find the solution $[x_1, x_2]$ for the 2x2 matrix equation $Ax = y$ (Eq. 1.53, left). We assume to know $[a, b, c, d]$ and $[y_1, y_2]$. We wish to show that the intersection (solution) is given by the equation on the right.

Here we wish to prove that the left equation has an inverse given by the right equation:

$$\begin{bmatrix} a & b \\ c & d \end{bmatrix} \begin{bmatrix} x_1 \\ x_2 \end{bmatrix} = \begin{bmatrix} y_1 \\ y_2 \end{bmatrix} \quad \Rightarrow \quad \begin{bmatrix} x_1 \\ x_2 \end{bmatrix} = \frac{1}{\Delta} \begin{bmatrix} d & -b \\ -c & a \end{bmatrix} \begin{bmatrix} y_1 \\ y_2 \end{bmatrix}.$$

How to take inverse:

1) Swap the diagonal, 2) change the signs of the off-diagonal, and 3) divide by $\Delta$. 

Now let's consider the inverse of the matrix $A = \begin{bmatrix} a & b \\ c & d \end{bmatrix}$.
B.2. INVERSE OF THE 2X2 MATRIX

B.2.1 Derivation of the inverse of a 2x2 matrix

1. Step 1: normalize the first column to 1.

2. Step 2: subtract the top equation from the lower.

3. Step 3: express result in terms of the determinate $\Delta = ad - bc$.

$$\begin{pmatrix} 1 & b \\ 0 & c \end{pmatrix} \begin{pmatrix} x_1 \\ y_1 \end{pmatrix} = \begin{pmatrix} \frac{1}{a} \\ \frac{1}{c} \end{pmatrix} \begin{pmatrix} a \\ b \end{pmatrix} \begin{pmatrix} x_2 \\ y_2 \end{pmatrix} = \begin{pmatrix} \frac{1}{a} & 0 \\ -\frac{1}{c} & 1 \end{pmatrix} \begin{pmatrix} y_1 \\ y_2 \end{pmatrix} \begin{pmatrix} 1 & b \Delta \\ 0 \end{pmatrix} \begin{pmatrix} x_1 \\ y_2 \end{pmatrix} = \begin{pmatrix} \frac{1}{a} & 0 \\ -\frac{1}{c} & a \end{pmatrix} \begin{pmatrix} y_1 \\ y_2 \end{pmatrix}$$

4. Step 4: These steps give the solution for $x_2$: $x_2 = -\frac{c}{\Delta} y_1 + \frac{a}{\Delta} y_2$.

5. Step 5: Finally the top equation is solved for $x_1$: $x_1 = \frac{1}{a} y_1 - \frac{b}{a} x_2 \Rightarrow x_1 = \left[ \frac{1}{a} + \frac{b}{a} \frac{c}{\Delta} \right] y_1 - \frac{b}{\Delta} \frac{a}{\Delta} y_2$.

In matrix format, in terms of the determinate $\Delta = ab - cd$ becomes:

$$\begin{pmatrix} x_1 \\ x_2 \end{pmatrix} = \begin{pmatrix} \frac{1}{a} + \frac{bc}{a\Delta} & -\frac{b}{a} \\ -\frac{c}{\Delta} & \frac{a}{\Delta} \end{pmatrix} \begin{pmatrix} y_1 \\ y_2 \end{pmatrix} \begin{pmatrix} x_1 \\ x_2 \end{pmatrix} = \frac{1}{\Delta} \begin{pmatrix} \Delta + bc/a & -b/a \\ -c & a \end{pmatrix} \begin{pmatrix} y_1 \\ y_2 \end{pmatrix} \begin{pmatrix} x_1 \\ x_2 \end{pmatrix} = \frac{1}{\Delta} \begin{pmatrix} d & -b/a \Delta \\ -c & a \end{pmatrix} \begin{pmatrix} y_1 \\ y_2 \end{pmatrix}.$$

Summary: This is a lot of algebra, that is why it is essential you memorize the formula for the inverse.
Appendix C

Symbolic analysis of $AE = E\Lambda$

Here we find $E, \Lambda$ given a 2x2 matrix $A = \begin{bmatrix} A & B; C & D \end{bmatrix}$, such that $AE = E\lambda$ using Matlab/Octave’s `syms A B C D M L M=[A B;C D] [E,L]=eig(M)`.

C.1 General case

The eigenvectors $E_{\pm}$ are

$$e_{\pm} = \left( \frac{1}{2i\sqrt{(A-D)^2 + 4BC}} \begin{bmatrix} (A-D) \mp \sqrt{(A-D)^2 + 4BC} \\ 1 \end{bmatrix} \right) \quad (C.1)$$

and eigenvalues are

$$\lambda_{\pm} = \frac{1}{2} \left( \frac{(A+D) - \sqrt{(A-D)^2 + 4BC}}{(A+D) + \sqrt{(A-D)^2 + 4BC}} \right) \quad (C.2)$$

C.2 Symmetry

C.2.1 Reciprocal systems

When the matrix is symmetrical ($B = C$), the corresponding system is said to be reciprocal. Electrical networks composed of inductors, capacitors and resistors are always reciprocal (the impedance matrix is symmetric).

C.2.2 Hermitian

When a system is Hermitian it is conjugate symmetric

$$Z(s) = Z(s)^\dagger,$$

a stronger condition that reciprocal, but not the symmetry of an impedance matrix, which is symmetric..

In this case the eigenvalues are always real. To show this start from the definition of an impedance eigen-equation ($V$ is a vector of voltages, $I$ a current vector $Z$ and an impedance matrix)

$$V = ZI = IA,$$
where $Z, I, V, \Lambda \in \mathbb{C}$, $A = A^\dagger$ is a square conjugate-symmetric matrix, and $I, V$ are vectors of the size of $Z$. Here $Z^\dagger$ is the complex transpose (see Appendix A, p. 233). Pre multiplying by $I^\dagger$ defines the power

$$P = V^\dagger I = I^\dagger ZI = I^\dagger Z^\dagger = I^\dagger \lambda I.$$ 

Subtracting the two equations gives

$$I^\dagger Z^\dagger - ZI = I^\dagger \lambda I - \lambda I$$

and using the fact that $Z^\dagger = Z$ gives

### C.2.3 Reversible systems

When the values of $T$ on its diagonal are equal ($A = D$), the matrix is called reversible, and the eigenvectors and eigenvalues greatly simplify to

$$E = \begin{bmatrix} -\sqrt{\frac{B}{C}} & +\sqrt{\frac{B}{C}} \\ 1 & 1 \end{bmatrix} \quad \Lambda = \begin{bmatrix} A - \sqrt{BC} & 0 \\ 0 & A + \sqrt{BC} \end{bmatrix}$$

(C.3)

### C.2.4 Transmission matrices and symmetry

The transmision matrix fully characterizes a two-port network (Sect. 1.3.7, p. 79).
Appendix D

Analysis of Pell equation (N=2)

Section 2.2.3 (p. 173) showed that the solution $[x_n, y_n]^T$ to Pell’s equation, for $N = 2$, is given by powers of Eq. 1.10. To find an explicit formula for $[x_n, y_n]^T$, one must compute powers of

$$A = 1j \begin{bmatrix} 1 & 2 \\ 1 & 1 \end{bmatrix}. \tag{D.1}$$

We wish to find the solution to Pell’s equation (Eq. 1.10), based on the recursive solution, Eq. 1.11 (p. 47). Thus we need is powers of $A$, that is $A^n$, which gives the a closed form expression for $[x_n, y_n]^T$. By the diagonalization of $A$, its powers are simply the powers of its eigenvalues. This diagonalization is called an eigenvalue analysis, a very general method rooted in linear algebra. This type of analysis allows us to find the solution to most of the linear equations we encounter.

From Matlab with $N = 2$ the eigenvalues of Eq. D.1 are $\lambda_\pm \approx [2.4142, -0.4142]$ (i.e., $\lambda_\pm = 1j(1 \pm \sqrt{2})$). The final solution to Eq. D.1 is given in Eq. 2.11 (p. 173). The solution for $N = 3$ is provided in Appendix D.1.1 (p. 250).

Once the matrix has been diagonalized, one may compute powers of that matrix as powers of the eigenvalues. This results in the general solution given by

$$\begin{bmatrix} x_n \\ y_n \end{bmatrix} = 1j^n A^n \begin{bmatrix} 1 \\ 0 \end{bmatrix} = 1j^n E \Lambda^n E^{-1} \begin{bmatrix} 1 \\ 0 \end{bmatrix}. \tag{D.1}$$

The eigenvalue matrix $D$ is diagonal with the eigenvalues sorted, largest first. The Matlab command $[E, D] = \text{eig}(A)$ is helpful to find $D$ and $E$ given any $A$. As we saw above,

$$\Lambda = 1j \begin{bmatrix} 1 + \sqrt{2} & 0 \\ 0 & 1 - \sqrt{2} \end{bmatrix} \approx \begin{bmatrix} 2.414j & 0 \\ 0 & -0.414j \end{bmatrix}. \tag{D.1}$$

D.1 Pell equation eigenvalue-eigenvector analysis

Here we show how to compute the eigenvalues and eigenvectors for the 2x2 Pell matrix for $N = 2$

$$A = \begin{bmatrix} 1 & 2 \\ 1 & 1 \end{bmatrix}. \tag{D.1}$$

The Matlab command $[E, D] = \text{eig}(A)$ returns the eigenvector matrix $E$

$$E = [e_+, e_-] = \frac{1}{\sqrt{3}} \begin{bmatrix} \sqrt{2} & \sqrt{2} \\ 1 & 1 \end{bmatrix} = \begin{bmatrix} 0.8165 & 0.8165 \\ 0.5774 & 0.5774 \end{bmatrix}. \tag{D.1}$$
and the eigenvalue matrix $\Lambda$ (Matlab’s $D$)

$$\Lambda \equiv \begin{bmatrix} \lambda_+ & 0 \\ 0 & \lambda_- \end{bmatrix} = \begin{bmatrix} 1 + \sqrt{2} & 0 \\ 0 & 1 - \sqrt{2} \end{bmatrix} = \begin{bmatrix} 2.4142 & 0 \\ 0 & -0.4142 \end{bmatrix}.$$ 

The factor $\sqrt{3}$ on $E$ normalizes each eigenvector to 1 (i.e., The Matlab command `norm([sqrt(2), 1])` gives $\sqrt{3}$).

In the following discussion we show how to determine $E$ and $D$ (i.e., $\Lambda$), given $A$.

**Calculating the eigenvalue matrix ($\Lambda$):** The matrix equation for $E$ is

$$AE = E\Lambda.$$  \hspace{1cm} (D.2)

Pre-multiplying by $E^{-1}$ diagonalizes $A$, given the eigenvalue matrix ($D$ in Matlab)

$$\Lambda = E^{-1}AE.$$ \hspace{1cm} (D.3)

Post-multiplying by $E^{-1}$ recovers $A$

$$A = E\Lambda E^{-1}.$$ \hspace{1cm} (D.4)

**Matrix power formula:** This last relation is the entire point of the eigenvector analysis, since it shows that any power of $A$ may be computed from powers of the eigenvalues. Specifically

$$A^n = E\Lambda^n E^{-1}.$$ \hspace{1cm} (D.5)

For example, $A^2 = AA = E\Lambda (E^{-1}E) \Lambda E^{-1} = E\Lambda^2 E^{-1}$.

Equations D.2, D.3 and D.4 are the key to eigenvector analysis, and you need to memorize them. You will use them repeatedly throughout this course, and possibly for a long time after it is over.

**Showing that $A - \lambda \pm I_2$ is singular:** If we restrict Eq. D.2 to a single eigenvector (one of $e_\pm$), along with the corresponding eigenvalue $\lambda_\pm$, we obtain a matrix equations

$$Ae_\pm = E_\pm \lambda_\pm = \lambda_\pm E_\pm$$

Note the important swap in the order of $E_\pm$ and $\lambda_\pm$. Since $\lambda_\pm$ is a scalar, this is legal (and critically important), since this allows us to remove (factored out) $E_\pm$

$$(A - \lambda \pm I_2)E_\pm = 0.$$ \hspace{1cm} (D.6)

This means that the matrix $A - \lambda \pm I_2$ must be singular, since when it operates on $E_\pm$, which is not zero, it gives zero. It immediately follows that its determinant is zero (i.e., $|(A - \lambda \pm I_2)| = 0$). This equation is used to uniquely determine the eigenvalues $\lambda_\pm$. Note the important difference between $\lambda \pm I_2$ and $\Lambda$ (i.e., $|(A - \Lambda)| \neq 0$).

**Calculating the eigenvalues $\lambda_\pm$:** The eigenvalues $\lambda_\pm$ of $A$ may be determined from $|(A - \lambda \pm I_2)| = 0$

$$\begin{vmatrix} 1 - \lambda_\pm & N \\ 1 & 1 - \lambda_\pm \end{vmatrix} = (1 - \lambda_\pm)^2 - N^2 = 0.$$ 

For our case of $N = 2$, $\lambda_\pm = (1 \pm \sqrt{2})^1$.

\footnote{It is a convention to order the eigenvalues from largest to smallest.}
Calculating the eigenvectors $e_{\pm}$: Once the eigenvalues have been determined, they are substituted into Eq. D.6, which determines the eigenvectors $E = [e_+, e_-]$, by solving

$$(A - \lambda_{\pm})e_{\pm} = \begin{bmatrix} 1 - \lambda_{\pm} & 2 \\ 1 & 1 - \lambda_{\pm} \end{bmatrix} E_{\pm} = 0,$$

where $1 - \lambda_{\pm} = 1 - (1 \pm \sqrt{2}) = \mp \sqrt{2}$.

Recall that Eq. D.6 is singular, because we are using an eigenvalue, and each eigenvector is pointing in a unique direction (This is why it is singular). You might respectively suggest that this equation has no solution. In some sense you would be correct. When we solve for $E_{\pm}$, the two equations defined by Eq. D.6 are co-linear (the two equations describe parallel lines). This follows from the fact that there is only one eigenvector for each eigenvalue.

Expecting trouble, yet proceeding to solve for $E_+ = [e_1^+, e_2^+]^T$,

$$\begin{bmatrix} -\sqrt{2} & 2 \\ 1 & -\sqrt{2} \end{bmatrix} \begin{bmatrix} e_1^+ \\ e_2^+ \end{bmatrix} = 0.$$

This gives two identical equations $-\sqrt{2}e_1^+ + 2e_2^+ = 0$ and $e_1^+ - \sqrt{2}e_2^+ = 0$. This is the price of an over-specified equation (the singular matrix is degenerate). The most we can determine is $e_+ = c \begin{bmatrix} -\sqrt{2} \\ 1 \end{bmatrix}^T$, where $c$ is a constant. We can determine eigenvector direction, but not its magnitude.

Following exactly the same procedure for $\lambda_-$, the equation for $e_-$ is

$$\begin{bmatrix} \sqrt{2} & 2 \\ 1 & \sqrt{2} \end{bmatrix} \begin{bmatrix} e_1^- \\ e_2^- \end{bmatrix} = 0.$$

In this case the relation becomes $e_1^- + \sqrt{2}e_2^- = 0$, thus $E_- = c \begin{bmatrix} \sqrt{2} \\ 1 \end{bmatrix}^T$ where $c$ is a constant.

Normalization of the eigenvectors: The constant $c$ may be determined by normalizing the eigenvectors to have unit length. Since we cannot determine the length, we set it to 1. In some sense the degeneracy is resolved by this normalization. Thus $c = 1/\sqrt{3}$, since

$$(\pm \sqrt{2})^2 + 1^2 = 3 = 1/c^2.$$

Summary: Thus far we have shown

$$E = [e_+, e_-] = \frac{1}{\sqrt{3}} \begin{bmatrix} \sqrt{2} & -\sqrt{2} \\ 1 & 1 \end{bmatrix}$$

and

$$\Lambda = \begin{bmatrix} \lambda_+ & 0 \\ 0 & \lambda_- \end{bmatrix} = \begin{bmatrix} 1 + \sqrt{2} & 0 \\ 0 & 1 - \sqrt{2} \end{bmatrix}.$$
By definition for any matrix $E^{-1}E = EE^{-1} = I_2$. Taking the product gives

$$E^{-1}E = \frac{\sqrt{3}}{2\sqrt{2}} \begin{bmatrix} 1 & \sqrt{2} \\ -1 & \sqrt{2} \end{bmatrix} \cdot \frac{1}{\sqrt{3}} \begin{bmatrix} \sqrt{2} & -\sqrt{2} \\ 1 & 1 \end{bmatrix} = \begin{bmatrix} 1 & 0 \\ 0 & 1 \end{bmatrix} = I_2.$$ 

We wish to show that $\Lambda = E^{-1}AE$

$$\begin{bmatrix} 1 + \sqrt{2} & 0 \\ 0 & 1 - \sqrt{2} \end{bmatrix} = \frac{\sqrt{3}}{2\sqrt{2}} \begin{bmatrix} 1 & \sqrt{2} \\ -1 & \sqrt{2} \end{bmatrix} \cdot \frac{1}{\sqrt{3}} \begin{bmatrix} \sqrt{2} & -\sqrt{2} \\ 1 & 1 \end{bmatrix} ,$$

which is best verified with Matlab.

**Verify that $A = E\Lambda E^{-1}$:** We wish to show that

$$\begin{bmatrix} 1 & 2 \\ 1 & 1 \end{bmatrix} = \frac{1}{\sqrt{3}} \begin{bmatrix} \sqrt{2} & -\sqrt{2} \\ 1 & 1 \end{bmatrix} \cdot \begin{bmatrix} 1 + \sqrt{2} & 0 \\ 0 & 1 - \sqrt{2} \end{bmatrix} \cdot \frac{\sqrt{3}}{2\sqrt{2}} \begin{bmatrix} 1 & \sqrt{2} \\ -1 & \sqrt{2} \end{bmatrix} ,$$

which is best verified with Matlab (or Octave). All the above equations have been verified both with Matlab and Octave. 

**Exercise:** I suggest that you verify $E\Lambda \neq \Lambda E$ and $AE = EA$ with Matlab. Here is the Matlab program which does this:

```matlab
A = [1 2; 1 1]; %define the matrix
[E,D] = eig(A); %compute the eigenvector and eigenvalue matrices
A*E-E*D %This should be \approx 0$, within numerical error.
E*D-D*E %This is not zero
```

D.1.1 Pell equation for $N = 3$

In Fig. D.1 for $N = 3$ is given, with $\beta_0 = 1/\sqrt{2}$. Perhaps try other trivial solutions such as $[-1, 0]^T$ and $[\pm 1, 0]^T$.

Try other trivial solutions such as $[-1, 0]^T$ and $[\pm 1, 0]^T$. Perhaps this can provide a clue to the proper value of $\beta_0$ for cases where $N > 3$. 
Pell’s Equation for $N = 3$

Case of $N = 3$ & $[x_0, y_0]^T = [1, 0]^T$, $\beta_0 = \frac{j}{\sqrt{2}}$

Note: $x_n^2 - 3y_n^2 = 1$, $x_n/y_n \xrightarrow{\infty} \sqrt{3}$

$$
\begin{bmatrix} x_1 \\ y_1 \end{bmatrix} = \beta_0 \begin{bmatrix} 1 \\ 1 \end{bmatrix} = \beta_0 \begin{bmatrix} 1 & 3 \\ 1 & 1 \end{bmatrix} \begin{bmatrix} 1 \\ 0 \end{bmatrix}
$$

$(1\beta_0)^2 - 3(1\beta_0)^2 = 1$

$$
\begin{bmatrix} x_2 \\ y_2 \end{bmatrix} = \beta_0^2 \begin{bmatrix} 4 \\ 2 \end{bmatrix} = \beta_0^2 \begin{bmatrix} 1 & 3 \\ 1 & 1 \end{bmatrix} \begin{bmatrix} 1 \\ 1 \end{bmatrix}
$$

$(4\beta_0^2)^2 - 3(2\beta_0^2)^2 = 1$

$$
\begin{bmatrix} x_3 \\ y_3 \end{bmatrix} = \beta_0^3 \begin{bmatrix} 10 \\ 6 \end{bmatrix} = \beta_0^3 \begin{bmatrix} 1 & 3 \\ 1 & 1 \end{bmatrix} \begin{bmatrix} 4 \\ 2 \end{bmatrix}
$$

$(10\beta_0^3)^2 - 3(6\beta_0^3)^2 = 1$

$$
\begin{bmatrix} x_4 \\ y_4 \end{bmatrix} = \beta_0^4 \begin{bmatrix} 28 \\ 16 \end{bmatrix} = \beta_0^4 \begin{bmatrix} 1 & 3 \\ 1 & 1 \end{bmatrix} \begin{bmatrix} 10 \\ 6 \end{bmatrix}
$$

$(28\beta_0^4)^2 - 3(16\beta_0^4)^2 = 1$

$$
\begin{bmatrix} x_5 \\ y_5 \end{bmatrix} = \beta_0^5 \begin{bmatrix} 76 \\ 44 \end{bmatrix} = \beta_0^5 \begin{bmatrix} 1 & 3 \\ 1 & 1 \end{bmatrix} \begin{bmatrix} 28 \\ 16 \end{bmatrix}
$$

$(76\beta_0^5)^2 - 3(44\beta_0^5)^2 = 1$

Figure D.1: This summarizes the solution of Pell’s equation due to the Pythagoreans using matrix recursion, for the case of $N=3$. The integer solutions are shown on the right. Not that $x_n/y_n \rightarrow \sqrt{3}$, in agreement with the Euclidean algorithm. The Matlab program for generating this data is PellS013.m. It seem likely that $\beta_0$ could be absorbed in the starting solution, and then be removed from the generating function, other than as the known factor $\beta_0^0$. 

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**D.1. PELL EQUATION EIGENVALUE-EIGENVECTOR ANALYSIS**

251
Appendix E

Stillwell’s Intersection vs. composition
There is a natural symbiotic relationship between Physics, Mathematics and Engineering, as depicted by this Venn diagram. Physics explores the boundaries. Mathematics provides the method and rigor. Engineering transforms the method into technology. While these three disciplines work well together, there is poor communication due to a different vocabulary. For example, Mathematics rarely, if ever, uses a system of units, whereas Physics and Engineering depend critically on them. Mathematics strives to abstract the ideas into proofs. Physics rarely, if ever, uses a proof. When they attempt abstract, they usually get into some difficulty. Engineers blunder ahead, ignoring most of these defects.

1.1 Depiction of the argument of Galileo (unpublished book of 1638) as to why weights of different masses (i.e., weight) must fall with identical velocity, contrary to what Archimedes had proposed c250BC. By joining them with an elastic cord, they become one. If the velocity were proportional to the mass, the joined masses would fall even faster. This results in a logical fallacy: How can two masses fall faster than either? This may have been the first time that the principle of conservation of energy was clearly stated. It seems likely that Galileo was attracted to this model of two masses connected by a spring because he was interested in planetary motion, which consist of masses (sun, earth, moon) dominated by gravity, which represents the the spring, providing the mutual attraction between the masses. There are actually three masses involved in his experiment, the earth pulling the two masses down, which in Galileo’s mind would represent the sun, the larger of the two masses (the earth) and the smaller mass (the moon).

1.2 Mathematical time-line between 1500 BCE and 1650 CE. The western renaissance is considered to have occurred between the 14-17 centuries. However the Asian (Chinese) “renaissance” was likely well before the 1 century (i.e., 1500 BCE?). There is significant evidence that a Chinese ‘treasure ship’ visited Italy in 1434, initiating the Italian renaissance (Menzies, 2008). This was not the first encounter between the Italians and the Chinese, as documented in ‘The travels of Marco Polo’ (c1300 CE).

1.3 Above: Jakob (1655-1705) and Johann (1667-1748) Bernoulli; Below: Leonhard Euler (1707) and Jean le Rond d’Alembert (1717-1783). Euler was blind in his right eye, hence the left portrait view. The figure numbers are from Stillwell (2010).

1.4 Exploded time-line of the two centuries between the 16th and 18th CE. Given the time line, it seems likely that Bombelli’s discovery of Diophantus’ book on “Arithmetic” in the Vatican library, triggered many of the ideas presented by Galileo, Descartes and Fermat, followed by others (i.e., Newton). Bombelli’s discovery might be considered as a magic moment in mathematics. The vertical red lines connect mentor-student relationships. Figure 1.2 (p. 19) gives the time-line from 1500BCE to 1650CE. Figure 1.12 (p. 54) presents the 17-20 CE (Newton–Einstein) view from 1640–1950, and Fig. 1.21 (p. 98) outlines the full (Bombelli–Einstein, 1525-1925) period.
1.5 Sieve of Eratosthenes for the case of \( N = 50 \).  

1.6 The Euclidean algorithm for finding the GCD of two numbers is one of the oldest algorithms in mathematics, and is highly relevant today. It is both powerful and simple. It was used by the Chinese during the Han dynasty (Stillwell, 2010, p. 70) for reducing fractions. It may be used to find pairs of integers that are coprime (their gcd must be 1), and it may be used to identify factors of polynomials by long division. It has an important sister algorithm called the continued fraction algorithm (CFA), that is so similar in concept that Gauss referred to the Euclidean algorithm as the “continued fraction algorithm” (Stillwell, 2010, p. 48).

1.7 The expansion of \( \pi \) to various orders, using the CFA, along with the order of the error of each rational approximation, with rounding. For example \( \hat{\pi}_2 = 22/7 \) has an absolute error \(|22/7 - \pi|\) of about 0.13%.

1.8 Beads on a string form perfect right triangles when the numbers of unit lengths between beads for each side satisfies Eq. 1.1. For example, when \( p = 2, q = 1 \), the sides are [3, 4, 5].

1.9 “Plimpton-322” is a stone tablet from 1800 [BCE], displaying \( a \) and \( c \) values of the Pythagorean triplets \([a, b, c]\), with the property \( b = \sqrt{c^2 - a^2} \in \mathbb{N} \). Several of the \( c \) values are primes, but not the \( a \) values. The stone is item 322 (item 3 from 1922), from the collection of George A. Plimpton. –Stillwell (2010)

1.10 This summarizes the solution of Pell’s equation for \( N = 2 \) using a slightly modified matrix recursion. Note that \( x_n/y_n \to \sqrt{2} \) as \( n \to \infty \), which was what the Pythagoreans were pursuing, and that the odd solutions are pure-imaginary.

1.11 This construction is called the Fibonacci spiral. Note how it is constructed out of squares having areas given by the square of the Fibonacci numbers. In this way, the spiral is smooth and the radius increases as the Fibonacci numbers (e.g., 8=3+5, 13 = 5+8, etc.). (Adapted from https://en.wikipedia.org/wiki/Golden_spiral.)

1.12 Time-line of the three centuries from the 18th to 20th CE. This was one of the most productive of all times, perhaps starting with the deep work of von Helmholtz, educated an experienced as a military surgeon, who mastered classical music, acoustics, physiology, vision, hearing (Helmholtz, 1863b), and, most important of all, mathematics. Kirchhoff frequently expanded on Helmholtz’s contributions. It is reported that Lord Rayleigh learned German so he could read Helmholtz’s great works. The history during this time is complex. For example, Lord Kelvin wrote a letter to Stokes, suggesting that Stokes try to prove what is today known as “Stokes theorem.” As a result, Stokes posted a reward (the Smith Prize), searching for a prove of “Lord Kelvin’s theorem,” which was finally proved by Hankel (1839-73) (https://en.wikipedia.org/wiki/Hermann_Hankel). Many new concepts were being proved and appreciated over this productive period. In 1863-65, Maxwell published his famous equations, followed by a reformating in modern vector notation by Heaviside, Gibbs and Hertz. The vertical red lines connect mentor-student relationships. This figure should put to rest the idea that ones best work is done in the early years. Many of these scientists were fully productive to the end of old age. Those that were not, died early, due to poor health or accidents. Figure 1.4 (p. 22) (Newton–Gauss) gives a closer look at the 15-18 CE, and Fig. 1.21 (p.98) (Bombelli–Einstein) for the full view from 16-20 CE. [fig:TimeLine19CE]
1.13 Vectors $\mathbf{A}$, $\mathbf{B}$, $\mathbf{C}$ are used to define the scalar product $\mathbf{A} \cdot \mathbf{B}$, vector (cross) product $\mathbf{A} \times \mathbf{B}$ and triple product $\mathbf{C} \cdot (\mathbf{A} \times \mathbf{B})$. The two vector products complement each other, one proportional to the sin $\theta$ of the angle $\theta$ between them and the other to the cos $\theta$. It follows that $||\mathbf{A} \cdot \mathbf{B}||^2 + ||\mathbf{A} \times \mathbf{B}||^2 = ||\mathbf{A}||^2 ||\mathbf{B}||^2$. The scalar product computes the projection of one vector on the other. The vector product $\mathbf{A} \times \mathbf{B}$ computes the area of the trapezoid formed by the two vectors, while the triple product $\mathbf{C} \cdot (\mathbf{A} \times \mathbf{B})$ defines the volume of the formed parallelepiped (i.e., prism). When all the angles are 90°, the volume becomes a cuboid.  

1.14 A single LC cell of the LC transmission line (see Fig. 2.2 (p. 169). Every cell of any transmission line may be modeled by the ABCD method, as the product of two matrices. For the example shown here, the inductance $L$ of the coil and the capacitance $C$ of capacitor are in units of [Henry/m] and [Farad/m], thus they depend on length $\Delta x$ [m] that the cell represents. Note the flows are always defined as into the + node. 

1.15 Impedance is defined as the ratio of a force over a flow, a concept also holds in mechanics and acoustics. In mechanics, the ‘force’ is equal to the mechanical force on an element (e.g. a mass, dash-pot, or spring), and the ‘flow’ is the velocity. In acoustics, the ‘force’ is pressure, and the ‘flow’ is the volume velocity or particle velocity of air molecules. 

1.16 The left panel shows how the real line may be composed with the circle. Each real $x$ value maps to a corresponding point $x'$ on the unit circle. The point $x \to \infty$ maps to the north pole $N$. This simple idea may be extended with the composition of the complex plane with the unit sphere, thus mapping the plane onto the sphere. As with the circle, the point on the complex plane $z \to \infty$ maps onto the north pole $N$. This construction is important because while the plane is open (does not include $z \to \infty$), the sphere is analytic at the north pole. Thus the sphere defines the closed extended plane. Figure from Stillwell (2010, pp. 299-300). 

1.17 LEFT: Domain–colorized map showing the complex mapping from the $s = \sigma + \omega j$ plane to the $\sigma, \omega$ plane. This mapping may be visualized by the use of intensity (light/dark) to indicate magnitude, and color (hue) to indicate angle (phase) of the mapping. RIGHT: This shows the $w(z) = z - \sqrt{2}$ plane, shift to the right and up by $\sqrt{2}/2 = 0.707$. The white and black lines are the iso-real and iso-imaginary contours of the mapping. 

1.18 Plot of sin$(0.5\pi(z-i))$.

1.19 This domain–color map allows one to visualize complex mappings by the use of intensity (light/dark) to indicate magnitude, and color (hue) to indicate angle (phase). The white and black lines are the iso-real and iso-imaginary contours of the mapping. LEFT: This figure shows the domain–color map for the complex mapping from the $z = x + y j$ plane to the $w(u,v) = u + v j = e^{x+jy}$ plane, which goes to zero as $x \to -\infty$, causing the domain–color map to become dark for $x < -2$. The white and black lines are always perpendicular because $e^z$ is complex analytic everywhere. RIGHT: This shows the principal value of the inverse function $u(x,y) + vi(x,y) = \log(x+yi)$, which has a zero (dark) at $z = 1$, since there $\log(1) = 0$ (the imaginary part is zero). Note the branch cut from $x = 0$ to $x = -\infty j$. On branches other than the one shown, there is no zeros, since the phase ($\angle z = 2\pi n$) is not zero. 

1.20 This three element mechanical resonant circuit consisting of a spring, mass and dash-pot (e.g., viscous fluid).
1.21 Time-line of the four centuries from the 16th and 20th CE. Given the time line, it is likely that Bombelli’s discovery of Diophantus’ book on “Arithmetic” in the Vatican library, triggered many of the ideas presented by Descartes and Fermat, followed by others (i.e., Newton). Bombelli’s discovery might be considered as a magic moment in mathematics. The vertical red lines connect mentor-student relationships. Figure 1.4 (p. 22) for a closer look at the 15-18c, and Fig. 1.12 (p.54) for a closer look at the 18-20c. . [fig:TimeLineCE].

1.22 Here we see the mapping for the square root function \( w(z) = \pm \sqrt{z} \) which has two single-valued sheets, corresponding to the two signs of the square root. The lower sheet is \(+\sqrt{z}\), and the upper sheet is \(-\sqrt{z}\). The location of the branch cut may be moved by rotating the \( z \) coordinate system. For example, \( w(z) = \pm j \sqrt{z} \) and \( w(z) = \pm \sqrt{-z} \) have different branch cuts, as may be easily verified using the Matlab/Octave commands \( j \text{viz}(z) \) and \( \text{viz}(-z) \). A function is analytic on the branch cut, since the cut may be moved. If a Taylor series is formed on the branch cut, it will describe the function on the two different sheets. Thus the complex analytic series (i.e., the Taylor branch cut lies in the domain \((x, \infty)\)), does not depend on the location of a branch cut, as it only describes the function uniquely (as a single valued function), valid in its local region of convergence. This figure has been taken from Stillwell (2010, p. 303). A more comprehensive view presented in the text. The branch cut lies in the domain \((x \in \mathbb{C})\), not in the codomain \(w(x) \in \mathbb{C} \) (Kusse and Westwig, 2010). This becomes clear by studying how \( \text{viz}.m \) works, and understanding its output. A example that shows this is Fig. 1.23, where the axes are \( s \) and \( \omega_j \), with the branch cut along the negative \( \sigma \) axis \((\theta = \pi)\).

1.23 Colorized plots of two sheets of \( w(s) = \pm \sqrt{s} \) with the branch cut at \( \theta = \pi \). Left: Color map reference plane \((z = s)\). Center: \( e^{\pi j \sqrt{s}} = -\sqrt{s} \). This sheet was generated by rotating \( w \) by 180°. This should be compared to the right panel of Fig. 1.24 which shows \(-\sqrt{-s}\), and to the upper sheet of Fig. 1.22. Note how the panel on the right of Fig. 1.24 matches the right half of \( s \) (purple = -90°, yellow/green = +90°) while the middle panel above comes from the left side of \( s \) (green to purple). The center sheet is green at -180°, and purple at +180°, which matches the right sheet at ±180°respectively. (i.e., \( e^{\pi j \sqrt{s}} \)). Right: the lower sheet of Fig. 1.22.

1.24 Here we use the Cartesian coordinates from the domain \( s = \sigma + \omega j \) to the range \( w(\sigma, \omega) = u + v j \). Left: Mapping: \( w(s) = s^2 \). Right: Mapping of the principal branch from \(-s\) to \( w(s) = -\sqrt{-s} \) (i.e., the rotated inverse of \( s^2 \)). Here the branch cut is along the positive \( \sigma \) axis.

1.25 Colorized plot of two LT pairs: Left: \( \sqrt{\pi/s} \leftrightarrow u(t)/\sqrt{t} \). Right: \( \sqrt{s^2 + 1} \leftrightarrow \delta(t) + \frac{1}{2} J_1(t) u(t) \).

1.26 Colorized plot of the of \( 1/(s^2 + 1) \leftrightarrow J_0(t) u(t) \).

1.27 Figure defining the analytic open surface \( S \), having boundary \( B \).

1.28 Experimental setup showing a large pipe on the left terminating the wall containing a small hole with a balloon, shown in green. At time \( t = 0 \) the balloon is pricked and a pressure pulse is created. The baffle on the left is meant to represent a semi∞ long tube having a large radius compared to the horn input diameter \( 2a \), such that the acoustic admittance looking to the left \((A/\rho_0 c_0 + A \to \infty)\), is very large compared to the horn’s throat admittance (Eq. 5.30). At time \( T \) the outbound pressure pulse \( p(r, T) = \delta(t - T/c_0)/r \) has reached radius \( T/c_0 \). Here \( r = x + r_o \) where \( x \) is the location of the source at the throat of the horn and \( r \) is measured from the vertex.
1.29 Throat acoustical resistance $r_A$ and acoustical reactance $x_A$, frequency characteristics of infinite parabolic, conical, exponential, hyperbolic and cylindrical horns having a throat area of 1 square centimeter. Note how the “critical” frequency (defined here as the frequency where the reactive and real parts of the radiation impedance are equal) of the horn reduces dramatically with the type of horn. For the uniform horn, the reactive component is zero, so there is no cutoff frequency. For the parabolic horn (1), the cutoff is around 3 kHz. For the conical horn (2), the cutoff is at 0.6 [kHz]. For the exponential horn the critical frequency is around 0.18 [kHz], which is 16 times smaller than for the parabolic horn. For each horn the cross-sectional area of the parabolic, conical exponential and hyperbolic horns is defined as 100 square centimeters at a distance of 100 centimeters from the throat (Olson, 1947, p. 101).

1.30 Derivation of Gauss’s Law: The divergence of the velocity $\nabla \cdot \mathbf{u}$, within $\delta x$, shown as the filled shaded region, is integrated over the closed volume. Then the divergence theorem is applied, transforming the integral to a surface integral normal to the surface, resulting in the difference of the two volume velocities $\delta \nu = \nu(x + \delta x) - \nu(x) = [\mathbf{u}(x + \delta x) \cdot \mathbf{A}(x + \delta x) - \mathbf{u}(x) \cdot \mathbf{A}(x)]$, as the limit of $\delta x \to 0$. Because the ends are in the direction of $\mathbf{u}$, the contribution to the volume velocity from the ends is zero.

1.31 The integral definition of the curl is similar to that of the divergence (Greenberg, 1988, p. 774), except that 1) the integration is over the plane tangent to the surface, defined as the cross product of $\hat{n}$, the normal to the surface, crossed with a current density of $\mathbf{n} \times \mathbf{H} \text{[Amps/m}^2\text{]}$, and 2) the surface is open, giving it a boundary $\mathcal{B}$, along the open edge. As with the divergence, which lead to Gauss’ Law, this definition leads to a second fundamental theorem of vector calculus: Stokes’ Law (aka the Curl theorem). There has been a serious problem in the past, since an open area does not have any enclosed volume $||d\mathcal{S}||$ (Greenberg, 1988, p. 814). Greenberg (1988, p. 824) try’s to resolve this, but perhaps has not resolved it. To solve it here we take the limit over the open surface $||d\mathcal{S}||$.

1.32 Left: von Helmholtz portrait taken from the English translation of his 1858 paper “On integrals of the hydrodynamic equations that correspond to Vortex motions” (in German) (Helmholtz, 1978). Right: Gustav Kirchhoff

1.33 Figure showing the four possible classifications of scalar and vector potential fields, rotational/irrotational, compressible/incompressible. Rotational fields are generated by on the vector potential (e.g., $\mathbf{A}(x,t)$), while compressible fields are generated by the scalar potentials (e.g., voltage $\phi(x,t)$, fluid density $\rho(x,t)$, temperature $T(x,t)$).

1.34 A solenoid is a uniform coil of wire. When a current is passed through the wire, a uniform magnetic field intensity $\mathbf{H}$ is created. From a properties point of view, this coil is indistinguishable from a permanent bar magnet, having north and south poles. Depending on the direction of the current, one end of a finite solenoidal coil is the north pole of the magnet, and the other end is the south pole. The uniform field inside the coil is called solenoidal, a confusing synonym for irrotational. (Figure from Wikipedia.)

1.35 The variables of Maxwell’s equations have names (e.g., $E$, $F$, $M$, $I$) and units (in square brackets [SI Units]). The units are required to gain a full understanding of each of the four variable and their corresponding equation. For example, Eq. EF has units [V/m]. When you integrate $E$ from $x = a$, $b$, you obtain the voltage difference between those two points. The speed of light in vacuum is $c = 3 \times 10^8 = 1/\sqrt{\mu_0/\varepsilon_0}$ [m/s], and the characteristic resistance of light $r_o = 377 = \sqrt{\mu_0/\varepsilon_0}$ [Ω] (i.e., ohms). The dot over a vector is shorthand for the partial with respect to time (i.e., $\mathbf{B} = \partial \mathbf{B}/\partial t$).
LIST OF FIGURES

260
2.1

The Euclidean Algorithm recursively subtracts n from m until the remainder m − kn is either
less than n or zero. Note that this is similar to mod(m,n). The GCD recursively computes mod,
then swaps m, n so that n < M , and repeats, until it finds the GCD. For the case depicted here the
value of k that renders the remainder less than n is k = 6. If one more step were taken (k = 7)
the remainder would become negative. By linear interpolation we can find that m − kn = 0 when
k = f loorm/n. For this example k = 6.5, thus k = 6. thus 6 = f loor(m/n) < n. This is
nonlinear arithmetic, which is the key to the GCD. . . . . . . . . . . . . . . . . . . . . . . 168

2.2

This transmission line (TL) is known as a low-pass filter wave-filter (Campbell, 1922), composed
of series inductors having an impedance Zk = sL and shunt admittance Yk = sC . The characterp
√
√
istic resistance of the TL is r0 = Zk /Yk and the wave velocity is κ = Zk Yk = s/ LC = s/c.
√
Thus the wave velocity is c = 1/ LC . The length ∆ [m] defines the physical size of each cell.
For long wavelengths the wave-filter acts as a pure delay. But when the frequency increases above
a cutoff frequency fc > c/∆, the wavelength λ = c/f is less than the size of a section ∆, and the
system response becomes very high-order low-pass filter. . . . . . . . . . . . . . . . . . . . 169

2.3

Derivation of Euclid’s formula for the Pythagorean triplets [a, b, c], based on a composition
of a line, having a rational slope t = p/q ∈ N, and a circle c2 = a2 + b2 , [a, b, c] ∈ N. This
analysis is attributed to Diophantus (Di·o·phan′ ·tus) (250 CE), and today such equations
are called Diophantine (Di·o·phan′ ·tine) equations. PTs have applications in architecture
and scheduling, and many other practical problems. . . . . . . . . . . . . . . . . . . . 171

2.4

Derivation of general solution to Pell’s equation, using a chord/tangent composition. . . . 174

2.5

Properties of the Fibonacci numbers (Stillwell, 2010, p. 28). . . . . . . . . . . . . 176

3.1

The Schwarz inequality is related to the shortest distance (length of a line) between the ends of the
p
two vectors. ||U || = (U · U ) as the dot product of that vector with itself. . . . . . . . . . . . 186

3.2

Depiction of a train consisting of cars, treated as a mass M and linkages, treated as springs of stiffness K or compliance C = 1/K . Below it is the electrical equivalent circuit, for comparison. The
mass is modeled as an inductor and the springs as capacitors to ground. The velocity is analogous
to a current and the force fn (t) to the voltage vn (t). . . . . . . . . . . . . . . . . . . . . . 193

3.3

On the left is the Bessel function J0 (πz), which is similar to cos(πz), except the zeros are distorted
away from s = 0 by a small amount, due to the cylindrical geometry. On the right is the related
(1)
Hankel function H0 (w2 ). The Bessel and Hankel functions are solution to the wave equation in
cylindrical coordinates with different boundary conditions. The zeros in the function are the places
where the pinned boundary condition is satisfied (where the string is restricted, by the boundary,
(1)
from moving). The Hankel function H0 (πz/2) has a branch cut, and a zero z0,1 at π ∗ z0,1 /2 =
−1.5 − 0.1. . . . . . . . . . . . . . . . . . . . . . . . . . . . . . . . . . . . . . . 197

3.4

On the left is the function w(z) = tan(z). On the right is the inverse s = tan−1 (w). Of special
interest is zviz atan(i*Z) (i/2)*log((1+Z)./(1-Z)). . . . . . . . . . . . . . . 197

3.5

Plot of w(s) = 1−e−s1 ln π1 (note π1 = 2), which is factor ζ1 (s) of Eq. 3.34. Here w(s) has poles
where esn ln 2 = 1, namely where ωn ln 2 = n2π , as seen in the colorized–map (s = σ + ω is the
Laplace frequency [rad]). . . . . . . . . . . . . . . . . . . . . . . . . . . . . . . . . . 202

3.6

This feedback network is described by a time-domain difference equation with delay Tp = ln πk ,
has an all-pole transfer function, given by Eq. 3.39. Physically this delay corresponds to a delay of
Tp [s]. . . . . . . . . . . . . . . . . . . . . . . . . . . . . . . . . . . . . . . . . . 202

3.7

Angle of Riemann Zeta function ∠ζ(z) as a function of complex z. Red ⇒ ∠ζ(z) < ±π/2 . . . . 205


3.8 A schematic representation of a 2-port ABCD electro-mechanic system using Hunt parameters $Z_e(s), z_m(s)$, and $T(s)$: electrical impedance, mechanical impedances, and transduction coefficient (Hunt, 1952; Kim and Allen, 2013). Also $V(f), I(f), F(f)$, and $U(f)$ are the frequency domain voltage, current, force, and velocity respectively. Notice how the matrix method ‘factors’ the 2-port model into three $2 \times 2$ matrices. This allows one to separate the physical modeling from the algebra. It is a standard impedance convention that the flows $I(f), U(f)$ are always defined into the port. Thus it is necessary to apply a negative sign on the velocity $-U(f)$ so that it has an outward flow, to feed the next cell with an inward flow. Replace $\Phi$ with $V$. .......................... 206

5.1 Real and imaginary parts of the radiation admittance $Y_{in}/\mathcal{Y}$ (upper) and impedance (lower), as a function of frequency, for a infinite section of horn, as computed from Eq. 5.7. The real and imaginary parts of the impedance are very similar to the results of Olson (1947) shown in Fig. 1.29. The real and imaginary parts of the admittance are much simpler to characterize: the imaginary part is a mass (i.e., is $\propto 1/s$) while the real part $\rightarrow 1$ above the critical frequency (where the real and imaginary parts are equal) at $\approx 0.3$ [kHz]. .... 219

5.2 Derivation of horn equation using Gauss’s Law: The divergence of the velocity $\nabla \cdot \mathbf{u}$, within $\delta x$, shown as the filled shaded region, is integrated over the enclosed volume. Next the divergence theorem is applied, transforming the integral to a surface integral normal to the surface of propagation. This results in the difference of the two volume velocities $\delta \nu = \nu(x + \delta x) - \nu(x) = [\mathbf{u}(x + \delta x) \cdot \mathbf{A}(x + \delta x) - \mathbf{u}(x) \cdot \mathbf{A}(x)], .... 220

D.1 This summarizes the solution of Pell’s equation due to the Pythagoreans using matrix recursion, for the case of N=3. The integer solutions are shown in on the right. Not that $x_n/y_n \rightarrow \sqrt{3}$, in agreement with the Euclidean algorithm. The Matlab program for generating this data is PellSol3.m. It seem likely that $\beta_0$ could be absorbed in the starting solution, and then be removed from the generating function, other than as the known factor $\beta_0^n$. .......................... 251
List of Tables

1.1 Three streams followed from Pythagorean theorem: Number Systems (Stream 1), Geometry (Stream 2) and Infinity (Stream 3) ........................................................................... 24
1.2 The fundamental theorems of mathematics ........................................................................... 35
1.3 Comparison between Euclidean geometry and analytic geometry. I am uncertain about the items in the third column. ................................................................. 72
1.4 The following table provides a brief table of simple Fourier Transforms. Note $a > 0 \in \mathbb{R}$ has units [rad/s]. To flag this necessary condition, we use $|a|$ to assure this condition will be met. The other constant $T_0 \in \mathbb{R}$ [s] has no restrictions, other than being real. Complex constants may not appear as the argument to a delta function, since complex numbers do not have the order property. ................................................................. 89
1.5 The following table provides a brief table of Laplace Transforms of $f(t), \delta(t), u(t), \text{rect}(t), T_0, p, e, \in \mathbb{R}$ and $F(s), G(s), s, a \in \mathbb{C}$. Given a Laplace transform ($\mathcal{L}$) pair $f(t) \leftrightarrow F(s)$, the frequency domain will always be upper-case [e.g. $F(s)$] and the time domain lower case [$f(t)$] and causal (i.e., $f(t < 0) = 0$). An extended table of transforms is given in Table 1.6 on page 95. ................................................................. 93
1.6 Functional relationships between Laplace Transforms. ......................................................... 95
1.7 The three vector operators manipulate scalar and vector fields, as indicated here. The gradient converts scalar fields into vector fields. The divergence eats vector fields and outputs scalar fields. Finally the curl maps vector fields into vector fields. It is helpful to have a name for second order operators (e.g., DoG, GoD: mnemonics defined in Sect. 1.5.10, p. 150). ......................................................... 122
1.8 Summary of the fundamental theorems of integral calculus, each of which deals with integration. There are at least two main theorems related to scalar calculus, and three more for vector calculus. ................................................................. 149
3.1 The following table provides an extended table of Laplace Transforms. $J_0, K_1$ are Bessel functions of the first and second kind. ................................................................. 199
3.2 There are several ways of indicating the quasi-static (QS) approximation. For network theory there is only one lattice constant $a$, which must be much less than the wavelength (wavelength constraint). These three constraints are not equivalent when the object may be a larger structured medium, spanning many wavelengths, but with a cell structure size much less than the wavelength. For example, each cell could be a Helmholtz resonator, or an electromagnetic transducer (i.e., an earphone). ................................................................. 211
5.1 **Table of horns and their properties for 1, 2 and 3 dimensions, and the exponential horn.**

In this table the horn’s range variable is \( x \), having area \( A(x) \), radius \( r_o(x) = \sqrt{A(x)/\pi} \).

\( F(x) \) is the coefficient on \( P_x \) (Eq. 5.18, p. 220), \( \kappa(s) \equiv s/c_o \), where \( c_o \) is the speed of sound. A dimensionless range variable is defined as \( x \equiv (\xi - \xi_o)/(L - \xi_o) \), with \( \xi \) the linear distance along the horn axis, from \( x = \xi_o \) to \( L \) corresponding to \( x = 0 \) to 1. The horn’s primitive solutions are \( P^\pm(x, \omega) \leftrightarrow \varphi^\pm(x, t) \). When \( \pm \) is indicated, the outbound solution corresponds to the negative sign. The last column is the radiation admittance normalized by \( A(x)/\rho c_o \) where \( s = \sigma + \omega j \) is the Laplace frequency.

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A.1 **Double-bold notation for the types of numbers.** (#) is a page number. Symbol with an exponent denote the dimensionality. Thus \( \mathbb{R}^2 \) represents the real plane. An exponent of 0 denotes point, e.g., \( j \in \mathbb{C}^0 \).
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Index

F: See fractional numbers, 27
∇: See gradient, 128
N: See counting numbers, 25, 229
P: See primes, 26
Q: See rational numbers, 27
T(s): see transmission matrix, 81
Y: See characteristic admittance, 141
Z: See integers, 26
Z: See characteristic impedance, 141
κ, See propagation function, 139
⊥: See perp p. 230, 26
ζ(s) function: See zeta function, 200
viz.m, see: plots, colorized, 197
ceil(c), 44
cov(A, B), 59, 70
deco v(A, B), 60
fix(c), 44
floor(c), 44
floor(x), 42
mod(x, y), 44
poly(R), 59
polyder(), 59
polyval(A, x), 59
polyval(P, X), 70
rem(x, y), 44
residue(N, D), 59, 70
root(P), 70
root, 66
round(c), 44
round(x), 42
roundb(c), 44
1–port, 97
2–port, 97
abacus, 25, 160
ABCD: see transmission matrix, 81
abscissa, see domain, 111
admittance: see impedance, 82
analytic, 35, 37, 53, 60, 61, 63, 65
analytic functions, 99
analytic geometry, 71, 75
analytic series, 60
Bézout’s theorem, 183
Bardeen, 65
Beranek, 162
Bernoulli, Daniel, 54
Bernoulli, Jakob, 20
Bernoulli, Johann, 20, 98
Bethe, 65
Bháskara II, 47
bi-harmonic, 102
bilinear form, 70
bilinear transformation, 81, 84, 191, 192, 208, 238
Boas, 162
Bode, 162
Bombelli, 18, 22, 31, 98, 99
Brahmagupta, 18, 25, 47, 49, 160
branch cut, 107, 109, 196
branch cuts, 108
branch points, 108
Brillouin, 54, 65, 140, 162
Brillouin zones, 104, 179
Brune Condition, 105, 135
Brune impedance, 59, 102, 104, 105, 112, 115, 134, 135
Cantor, 28
Cardano, 55
Cauchy, 36, 54, 65, 83, 107
Cauchy integral formula, 114
Cauchy integral theorem, 114
Cauchy residue theorem, 114
Cauchy’s formula, 113, 149
Cauchy’s theorem, 113, 149
Cauchy-Riemann conditions, 101
causality, 37, 87, 97, 206, 207
CFA, 42
characteristic admittance, 141
INDEX

Chebyshev, 36
cord–tangent methods, 82, 83, 163, 170, 174
codomain, 64
colorized plots: see “plots, colorized”, 202
complex analytic, 99, 106, 107, 109, 196
complex analytic functions, 83, 84, 86, 100, 107
complex analytic functions: see analytic, 64
complex analytic: see analytic, 64
complex numbers, 28
Complex numbers, History of, 31
complex numbers: polar representation, 29
composition, 25, 36, 47, 69, 71, 72, 75, 79, 81, 83, 84, 183, 184
composition, matrix, 51
compressible, 153
conjugate variables, 81, 113, 138, 187
conservation of energy, 104
conservation of energy: see energy, 16
convolution, 66, 67, 90, 116
coprime, 26, 27, 163, 164, 167
counting numbers, 25
cubic, 55
Curl $\nabla \times ()$, 37, 128, 150, 226
d'Alembert, 15, 20, 23, 53, 54, 66, 98, 103, 132, 133, 141, 157, 221
deconvolution, 26, 60, 69
degrees of freedom, 151
Descartes, 18, 22, 23, 66, 98, 177
determinant, 76, 78, 79, 123, 234
DFT, 88
Dicke, 162
Diophantus, 17, 18, 46
dispersion relation, 104
dispersive wave propagation, 53
Divergence $\nabla \cdot ()$, 37, 128, 150, 226
division with rounding, 42
DoF: see degrees of freedom, 151
domain, 64
domain-coloring: see “plots, colorized”, 86
dot product, 67
driving-point impedance, 206
DTFT, 88
eigen analysis of reversible systems, 246
eigen-analysis, 49
eigen-function expansion, 175
eigen-values of $d/dt$, 117
eigen-values of a 2x2 matrix, 245
eigen-values, physical interpretation, 124
eigen-vectors of a 2x2 matrix, 245
Einstein, 54, 98
elimination, 71
emf, 156
encryption, 15
energy, 16, 52, 113, 187, 208, 209
Energy, Conservation, 59
entropy, 40
epicycle, 52
Feynman, 128, 161, 162
FFT, 88
Fibonacci sequence, 49
Fourier Series, 88
Fourier transform, 87
Fourier Transform Table, 89
fractional numbers, 27
frequency domain, 37, 53, 87, 106, 107
Fry, 162
FTC, 149
FTCC, 149
FTVC, 126

Ewald, 65
Exam I, 51
Exam II, 97
Exam III, 118
Exam IV (Final), 160
extended complex plane, 83
factoring, 55
Fermat, 18, 22, 23, 47, 66, 98, 177
Feshbach, 162
Euclid, 27, 40
Euclid’s formula, 17, 18, 25, 32, 46, 170, 171
Euclidean algorithm, 40, 41, 164, 253
Euclidean length, 73
Euler, 20, 22, 23, 32, 54, 83, 98, 177, 200
Euler’s equation, 131
Euler’s functional zeta equation, 203

INDEX
function, circular, 64
function, inverse, 64
function, periodic, 64
functions, causal, 95
functions, periodic, 95
functions: \( \Gamma(s) \), 204
Fundamental theorem of algebra, 37, 66, 69, 129
Fundamental theorem of arithmetic, 27, 31, 34, 35, 37
Fundamental theorem of calculus, 37, 100
Fundamental theorem of complex calculus, 37, 83, 98, 100
Fundamental theorem of vector calculus, 37, 150
Fundamental theorems, 34, 35, 37, 149
Fundamental Theorems, vectors, 98
Galileo, 16, 22, 52, 65, 98, 120
Gamma function, 204
Gauss, 36, 54, 98, 107, 164
Gauss-Lucas Theorem, 59
Gaussian elimination, 77, 78
GCD, 42
gcd, 40, 41, 253
geometric series, 61
Gibbs phenomenon, 90, 92
Gradient \( \nabla() \), 37, 128, 150, 226
gravity, 16, 52, 75, 144
Green, 17, 36, 162
Greens Theorem, 102
gyro-scope, 148
Halley, 9
Heaviside, 17, 54, 98
Heisenberg, 65
Helmholtz, 17, 36, 54, 98, 150, 162, 227
Helmholtz portrait, 152
Helmholtz’ Theorem, 126
Helmholtz’s decomposition, 150
Helmholtz’s decomposition, see Fundamental theorem of vector calculus, 150
higher order modes, 227
Hilbert, 98
Humpty Dumpty, 17
Hunt, 162
Huygens, 98, 131
Huygens principle, 133
IEEE-754, 27, 32, 55, 59
impedance, 71, 81, 82, 91, 97, 102–104, 113–115
impedance, see Brune, 70
impulse response, 238
incompressible, 154
integers, 26
integers, utility of, 31
internet security, 15, 33, 165
intersection, 71
intersection of curves, 36, 70, 72, 75–77, 170, 172, 174, 177, 182, 184, 185, 188–190
intersection of sets, 72
intersection point, 76
irrotational, 148, 153, 154
Kepler, 52, 98
Kirchhoff, 36, 54, 98, 227
Kirchhoff’s Law, 129
Kirchhoff’s laws, 187
Kirchhoff’s portrait, 152
Lagrange, 47, 98
Lamb, 162
Laplace, 98
Laplace frequency, 81, 87, 91
Laplace transform, 35–37, 87, 106
Laplace Transform Table, 93, 199
Laplacian \( \nabla^2 \), 37, 113, 123, 128, 134, 150
lattice vector, 179
laws of gravity, 52
Leibniz, 36
length, Euclidean, 73
Leonardo da Vinci, 22
Lighthill, 162
linear equations, 66
linearity, 97, 206, 207
Lord Kelvin, 17
Möbius, 98
Möbius transformation, 81, 102, 192
Möbius, see bilinear, 71
Mason, Warren P., 227
Maxwell, 17, 36, 54, 98
Maxwell’s equations, 24, 37, 98, 154, 155, 158, 227
meromorphic, 108
Mersenne, 15, 22, 98, 120
INDEX

modulo, 237
Montgomery, 162
Morse, 162
music, 15–18, 21, 33, 36, 52, 54, 161

Needham, Joseph, 17
network postulates, 53, 91, 95, 97, 106, 205
Newton, 16, 18, 22, 23, 47, 52, 54, 65, 83, 98, 131, 161, 177, 183, 187
Newton’s laws, 129
Newton’s laws of motion, 52
Newton’s method for finding roots, 56
nonlinear algebra, 95
norm, 73

Ohm’s Law, 81, 129, 187
Ohm’s law, 81, 82, 97
Olson, 162
ordinate, 111
ordinate, see range, 111

partial fraction expansion, 63, 110, 112, 135, 137
passive, 97, 206, 208
Pauli, 65
Pell’s equation, 46, 47, 175
Pipes, 162
Plimpton-322, 47
plots, colorized, 85, 86, 109–111, 116, 197, 201, 202
PNT: see prime number theorem, 26
Poincare, 54
point at infinity, 83, 84
poles, 92, 110, 135, 238
poles, simple, 112
polynomial, long division, 69
positive-definite, 208
positive-real, 208
postulates, 53, 91, 95, 205
potential, 120, 125, 145
Power, 104
power, 81, 104, 113, 187
power series, 60
prime, 33, 34, 40, 161, 164, 165, 200
prime number theorem, 26, 34
primes, 26
primes, utility of, 165
Principia, 9, 53, 98, 99

propagation function, 140, 179, 215
Prucell, 162
Pythagoras, 15
Pythagorean comma, 161
Pythagorean theorem, 25
Pythagorean triplets, 18
Pythagoreans, 21, 22, 27, 31, 32, 46, 47, 49, 174
Pythagoreans, fate, 22
quadratic, 67
quadratic equation, solution, 55, 68
quartic, 55
quasi-statics, 97, 135, 157–159, 206, 210, 227
quintic, 55
Ramo, 162
rational function, 70
rational numbers, 27
Rayleigh, 17, 54, 98, 162, 227
Rayleigh reciprocity, 97, 206
reflectance, 71, 134
region of convergence: see RoC, 60, 62
residue, 63, 108, 110, 112, 114, 135
Residue Theorem, 115
Residue theorem, 113, 149
residue, impedance, 112
reversibility, 97, 206, 210
reversible system, eigen analysis, 246
RHP, 102
Riemann, 54, 98, 107, 117
Riemann sheet, 107, 109, 196
Riemann sphere, 83
right half plane, see RHP, 102
RoC, 60–63, 88, 99, 102, 107
Rolle’s Theorem, 59
roots, 55
roots of polynomials., 30
scalar field, 120
scalar product, 73–75, 87, 153
Schwarz, 185, 186
Schwarz inequality, 73
See energy, 100
set theory, 27, 28
set theory symbols defined, 230
Shannon entropy, 40
sieve, 38, 39, 200
signal processing, digital, 49
signals, 87
Slater, 162
solenoidal, see irrotational, 150, 154
Sommerfeld, 54, 65, 107, 134, 140, 158, 162
spacial-invariant, 206
spatial invariant, 210
spatial-invariant, 97
speed of light, 155, 156
speed of sound, 16, 53, 131, 139, 158, 159
Stokes, 17, 36, 162
stream 1, 24, 25, 34
stream 2, 24, 34, 36, 37, 52
stream 3, 24, 34, 36, 37, 49
stream 3a, 98
stream 3b, 119
Sturm-Liouville, 131, 141
symmetry, 87, 97
systems, 87
Taylor series, 61–63, 107
Taylor series formula, 101
Thévenin, 156
Thévenin parameters, 156, 207
The fundamental theorem of arithmetic, 34
thermal conductivity, 127, 134
time domain, 87
time invariant, 209
time-invariant, 97, 206
top, 148
Train problem, 193
transmission line, 210
Transmission line problem, 193
transmission matrix, 79, 81, 82, 96, 136, 142, 191, 193, 206, 207, 217, 225
trap door, 165
vector cross product, see vector product, 75
vector identity: CoC, 153
vector identity: CoG, 152
vector identity: DoC, 150, 152
vector identity: DoG, 122, 150
vector identity: GoD, 122, 150
vector identity: Laplacian, 153
vector inner product, see scalar product, 75
vector product, 74, 75, 153
vector triple product, 75
vector wave equation, 179
vector, three, 75
vector, unit, 75
vectors, 72
visco-thermal losses, 227
wave equation, 52, 131, 156, 179
wave number, 53, 79, 104, 179, 227
wave number with losses, 227
wave number, complex, 185
wavelength, 53, 104
z-transform, 49, 88
Zeno’s paradox, 98
zeros, 92, 238
zeta function, Euler, 37, 200
zeta function, Riemann, 37, 200
zviz.m, see: “plots, colorized”, 86