Audio Effects: Theory, Implementation and Application explores digital audio effects relevant to audio signal processing and music informatics. It supplies fundamental background information on digital signal processing, focusing on audio-specific aspects that constitute the building block on which audio effects are developed. The text integrates theory and practice, relating technical implementation to musical implications. It can be used to gain an understanding of the operation of existing audio effects or to create new ones. In addition to delivering detailed coverage of common (and unusual) audio effects, the book discusses current digital audio standards, most notably VST and AudioUnit. Source code is provided in C/C++ and implemented as audio effect plug-ins with accompanying sound samples. Each section of the book includes study questions, anecdotes from the history of music technology, and examples that offer valuable real-world insight, making this an ideal resource for researchers and for students moving directly into industry.

The textbook provides both profound technological knowledge and a comprehensive treatment of essential topics in music processing and music Page 1/25

information retrieval (MIR). Including numerous examples, figures, and exercises, this book is suited for students, lecturers, and researchers working in audio engineering, signal processing, computer science, digital humanities, and musicology. The book consists of eight chapters. The first two cover foundations of music representations and the Fourier transform concepts used throughout the book. Each of the subsequent chapters starts with a general description of a concrete music processing task and then discusses in a mathematically rigorous way essential techniques and algorithms applicable to a wide range of analysis. classification, and retrieval problems. By mixing theory and practice, the book's goal is to offer detailed technological insights and a deep understanding of music processing applications. As a substantial extension, the textbook's second edition introduces the FMP (fundamentals of music processing) notebooks, which provide additional audio-visual material and Python code examples that implement all computational approaches step by step. Using Jupyter notebooks and open-source web applications, the FMP notebooks yield an interactive framework that allows students to experiment with their music examples, explore the effect of parameter settings, and understand the computed results by suitable visualizations and sonifications. The FMP notebooks are available from the author's institutional web page at the International Audio Laboratories Erlangen.

(Technical Reference). More than simply the book of the award-winning DVD set, Art & Science of Sound Recording, the Book takes legendary engineer, producer, and artist Alan Parsons' approaches to sound recording to the next level. In book form, Parsons has the space to include more technical background information, more detailed diagrams, plus a complete set of course notes on each of the 24 topics, from "The Brief History of Recording" to the now-classic "Dealing with Disasters." Written with the DVD's coproducer, musician, and author Julian Colbeck, ASSR, the Book offers readers a classic "big picture" view of modern recording technology in conjunction with an almost encyclopedic list of specific techniques, processes, and equipment. For all its heft and authority authored by a man trained at London's famed Abbey Road studios in the 1970s ASSR, the Book is also written in plain English and is packed with priceless anecdotes from Alan Parsons' own career working with the Beatles, Pink Floyd, and countless others. Not just informative, but also highly entertaining and inspirational, ASSR, the Book is the perfect platform on which to build expertise in the art and science of sound recording.

With Computational Thinking in Sound, veteran educators Gena R. Greher and Jesse M. Heines provide the first book ever written for music fundamentals educators that is devoted specifically to music, sound, and technology. Using a Page 3/25

student-centered approach that emphasizes project-based experiences, the book provides music educators with multiple strategies to explore, create, and solve problems with music and technology in equal parts. It also provides examples of hands-on activities that encourage students, alone and in groups, to explore the basic principles that underlie today's music technology and freely available multimedia creation tools. Computational Thinking in Sound is an effective tool for educators to introduce students to the complex process of computational thinking in the context of the creative arts through the more accessible medium of music.

Content-based multimedia retrieval is a challenging research field with many unsolved problems. This monograph details concepts and algorithms for robust and efficient information retrieval of two different types of multimedia data: waveform-based music data and human motion data. It first examines several approaches in music information retrieval, in particular general strategies as well as efficient algorithms. The book then introduces a general and unified framework for motion analysis, retrieval, and classification, highlighting the design of suitable features, the notion of similarity used to compare data streams, and data organization.

Digital Audio Theory: A Practical Guide bridges the fundamental concepts and Page 4/25

equations of digital audio with their real-world implementation in an accessible introduction, with dozens of programming examples and projects. Starting with digital audio conversion, then sequeing into filtering, and finally real-time spectral processing, Digital Audio Theory introduces the uninitiated reader to signal processing principles and techniques used in audio effects and virtual instruments that are found in digital audio workstations. Every chapter includes programming snippets for the reader to hear, explore, and experiment with digital audio concepts. Practical projects challenge the reader, providing hands-on experience in designing real-time audio effects, building FIR and IIR filters, applying noise reduction and feedback control, measuring impulse responses, software synthesis, and much more. Music technologists, recording engineers, and students of these fields will welcome Bennett's approach, which targets readers with a background in music, sound, and recording. This guide is suitable for all levels of knowledge in mathematics, signals and systems, and linear circuits. Code for the programming examples and accompanying videos made by the author can be found on the companion website, DigitalAudioTheory.com. Fundamentals of Music ProcessingUsing Python and Jupyter NotebooksSpringer Nature

Fundamentals of Image, Audio, and Video Processing Using MATLAB® Page 5/25

introduces the concepts and principles of media processing and its applications in pattern recognition by adopting a hands-on approach using program implementations. The book covers the tools and techniques for reading, modifying, and writing image, audio, and video files using the data analysis and visualization tool MATLAB®. Key Features: Covers fundamental concepts of image, audio, and video processing Demonstrates the use of MATLAB® on solving problems on media processing Discusses important features of Image Processing Toolbox, Audio System Toolbox, and Computer Vision Toolbox MATLAB® codes are provided as answers to specific problems Illustrates the use of Simulink for audio and video processing Handles processing techniques in both the Spatio-Temporal domain and Frequency domain This is a perfect companion for graduate and post-graduate students studying courses on image processing, speech and language processing, signal processing, video object detection and tracking, and related multimedia technologies, with a focus on practical implementations using programming constructs and skill developments. It will also appeal to researchers in the field of pattern recognition, computer vision and content-based retrieval, and for students of MATLAB® courses dealing with media processing, statistical analysis, and data visualization. Dr. Ranjan Parekh, PhD (Engineering), is Professor at the School of Education

Technology, Jadavpur University, Calcutta, India, and is involved with teaching subjects related to Graphics and Multimedia at the post-graduate level. His research interest includes multimedia information processing, pattern recognition, and computer vision.

Counterpoint manuals have long been central to the music education of composers, historians, and theorists. In this book a conductor uses counterpoint exercises to aid musicians in becoming sensitive to the fundamental ingredients of good music making. 16 45-minute lectures at the college level. An introduction to the instruments of the orchestra, timbre, dynamics, meter, pitch, mode, the major/minor key system, melody, harmony, modulation. Includes excerpts from classical, jazz, and popular music recordings; and live piano demonstrations.

Speech and audio processing has undergone a revolution in preceding decades that has accelerated in the last few years generating game-changing technologies such as truly successful speech recognition systems; a goal that had remained out of reach until very recently. This book gives the reader a comprehensive overview of such contemporary speech and audio processing techniques with an emphasis on practical implementations and illustrations using MATLAB code. Core concepts are firstly covered giving an introduction to the physics of audio and vibration together with their representations using complex numbers, Z transforms and frequency analysis transforms such as the FFT. Later chapters give a description of the human auditory

system and the fundamentals of psychoacoustics. Insights, results, and analyses given in these chapters are subsequently used as the basis of understanding of the middle section of the book covering: wideband audio compression (MP3 audio etc.), speech recognition and speech coding. The final chapter covers musical synthesis and applications describing methods such as (and giving MATLAB examples of) AM, FM and ring modulation techniques. This chapter gives a final example of the use of timefrequency modification to implement a so-called phase vocoder for time stretching (in MATLAB). Features A comprehensive overview of contemporary speech and audio processing techniques from perceptual and physical acoustic models to a thorough background in relevant digital signal processing techniques together with an exploration of speech and audio applications. A carefully paced progression of complexity of the described methods; building, in many cases, from first principles. Speech and wideband audio coding together with a description of associated standardised codecs (e.g. MP3, AAC and GSM). Speech recognition: Feature extraction (e.g. MFCC features), Hidden Markov Models (HMMs) and deep learning techniques such as Long Short-Time Memory (LSTM) methods. Book and computer-based problems at the end of each chapter. Contains numerous real-world examples backed up by many MATLAB functions and code.

A Solid Introduction to Sound and Vibration: No Formal Background NeededThis Second Edition of Fundamentals of Sound and Vibration covers the physical,

mathematical and technical foundations of sound and vibration at audio frequencies. It presents Acoustics, vibration, and the associated signal processing at a level suitable for graduate stude

Learn the basics of recording, processing, and mixing audio using Reason software, the robust digital audio workstation and musical toolkit used by artists, producers, and sound designers worldwide. Audio Production Basics with Reason Software will guide you every step of the way. The exercises in this book are designed to be completed using the low-cost Reason Intro edition, allowing you to get hands-on practice and easily experience the world of Reason software. Everything discussed in this book translates fully to the standard edition of Reason software, as well as to the expansive Reason Suite edition. With this book and the included online media files, you'll get working experience using Reason, covering everything from setting up your computer to the fundamentals of audio production, including: Basic digital audio workstation operations and audio hardware options Principles of sound production and microphone use Essential Reason concepts and operations MIDI fundamentals for playing and recording virtual instruments Managing devices and routing signals in Reason's unique rack interface Using automation to create dynamic changes to audio Mixing your project and exporting your final mixed track Reason Intro is affordable, easy, and fun. And everything you learn here will apply when you are ready to move on to more advanced versions of Reason. Take the first step now, with Audio Production Basics with Reason

#### Software.

This book introduces a subject that will be new to many: sonic arts. The application of sound to other media (such as film or video) is well known and the idea of sound as a medium in its own right (such as radio) is also widely accepted. However, the idea that sound could also be a distinct art form by itself is less well established and often misunderstood. The Fundamentals of Sonic Art & Sound Design introduces, describes and begins the process of defining this new subject and to provide a starting point for anyone who has an interest in the creative uses of sound. The book explores the worlds of sonic art and sound design through their history and development, and looks at the present state of these extraordinarily diverse genres through the works and words of established artists and through an examination of the wide range of practices that currently come under the heading of sonic arts. The technologies that are used and the impact that they have upon the work are also discussed. Additionally, The Fundamentals of Sonic Art & Sound Design considers new and radical approaches to sound recording, performance, installation works and exhibitions and visits the worlds of the sonic artist and the sound designer.

This book provides a comprehensive introduction to the theory and practice of spherical microphone arrays. It is written for graduate students, researchers and engineers who work with spherical microphone arrays in a wide range of applications. The first two chapters provide the reader with the necessary mathematical and physical background,

including an introduction to the spherical Fourier transform and the formulation of plane-wave sound fields in the spherical harmonic domain. The third chapter covers the theory of spatial sampling, employed when selecting the positions of microphones to sample sound pressure functions in space. Subsequent chapters present various spherical array configurations, including the popular rigid-sphere-based configuration. Beamforming (spatial filtering) in the spherical harmonics domain, including axis-symmetric beamforming, and the performance measures of directivity index and white noise gain are introduced, and a range of optimal beamformers for spherical arrays, including beamformers that achieve maximum directivity and maximum robustness, and the Dolph-Chebyshev beamformer are developed. The final chapter discusses more advanced beamformers, such as MVDR and LCMV, which are tailored to the measured sound field.

This extraordinarily comprehensive text, requiring no special background, discusses the nature of sound waves, musical instruments, musical notation, acoustic materials, elements of sound reproduction systems, and electronic music. Includes 376 figures. A CREATIVE APPROACH TO MUSIC FUNDAMENTALS is a reader-friendly, creative text that focuses on music fundamentals through written and aural exercises. In addition, the text strives to teach students how to create music through learning rhythm, melody, scales, intervals, and triads. Important Notice: Media content referenced within the product description or the product text may

not be available in the ebook version.

Comprehensive and accessible, this foundational text surveys general principles of sound, musical scales, characteristics of instruments, mechanical and electronic recording devices, and many other topics. More than 300 illustrations plus questions, problems, and projects.

(Yamaha Products). Sound reinforcement is the use of audio amplification systems. This book is the first and only book of its kind to cover all aspects of designing and using such systems for public address and musical performance. The book features information on both the audio theory involved and the practical applications of that theory, explaining everything from microphones to loudspeakers. This revised edition features almost 40 new pages and is even easier to follow with the addition of an index and a simplified page and chapter numbering system. New topics covered include: MIDI, Synchronization, and an Appendix on Logarithms. 416 Pages.

Provides an introduction to the nature, synthesis and transformation of sound which forms the basis of digital sound processing for music and multimedia. Background information in computer techniques is included so that you can write computer algorithms to realise new processes central to your own musical and sound processing ideas. Finally, material is inlcuded to explain the way in which

people contribute to the development of new kinds of performance and composition systems. Key features of the book include: · Contents structured into free-standing parts for easy navigation · `Flow lines' to suggest alternative paths through the book, depending on the primary interest of the reader. Practical examples are contained on a supporting website. Digital Sound Processing can be used by anyone, whether from an audio engineering, musical or music technology perspective. Digital sound processing in its various spheres - music technology, studio systems and multimedia - are witnessing the dawning of a new age. The opportunities for involvement in the expansion and development of sound transformation, musical performance and composition are unprecedented. The supporting website (www.york.ac.uk/inst/mustech/dspmm.htm) contains working examples of computer techniques, music synthesis and sound processing.

This concise book builds upon the foundational concepts of MIDI, synthesis, and sampled waveforms. It also covers key factors regarding the data footprint optimization work process, streaming versus captive digital audio new media assets, digital audio programming and publishing platforms, and why data footprint optimization is important for modern day new media content development and distribution. Digital Audio Editing Fundamentals is a new media

mini-book covering concepts central to digital audio editing using the Audacity open source software package which also apply to all of the professional audio editing packages. The book gets more advanced as chapters progress, and covers key concepts for new media producers such as how to maximize audio quality and which digital audio new media formats are best for use with Kindle, Android Studio, Java, JavaFX, iOS, Blackberry, Tizen, Firefox OS, Chrome OS, Opera OS, Ubuntu Touch and HTML5. You will learn: Industry terminology involved in digital audio editing, synthesis, sampling, analysis and processing The work process which comprises a fundamental digital audio editing, analysis, and effects pipeline The foundational audio waveform sampling concepts that are behind modern digital audio publishing How to install, and utilize, the professional, open source Audacity digital audio editing software Concepts behind digital audio sample resolution and sampling frequency and how to select settings How to select the best digital audio data codec and format for your digital audio content application How to go about data footprint optimization, to ascertain which audio formats give the best results Using digital audio assets in computer programming languages and content publishing platforms This book covers the fundamentals of music. If you want to 'make music', there are definite things that need to be learned. There are many books on the subject

of music. What makes this book different? The data is presented in a simple, straightforward and concise manner along with practice routines that will result in real ability to read music, play songs, improvise and compose! This book applies to any musician at any level of expertise regardless of what instrument or style of music he/she plays. It is invaluable as a reference book for private lessons or as a textbook for classroom study.

"This book offers an introduction to digital signal processing (DSP) with an emphasis on audio signals and computer music ... This book is designed for both technically and musically inclined readers alike--folks with a common goal of exploring digital signal processing"--Cover, p. [4].

When Speech and Audio Signal Processing published in 1999, it stood out from its competition in its breadth of coverage and accessible, intutiont-based style. This book was aimed atindividual students and engineers excited about the broad span ofaudio processing and curious to understand the available techniques. Since then, with the advent of the iPod in 2001, the field of digital audio and music has exploded, leading to amuch greater interest in the technical aspects of audioprocessing. This Second Edition will update and revise the original book to augment it with new material describing both the enabling technologies of digital music distribution (most significantly the MP3) and

a range of exciting new research areas in automatic musiccontent processing (such as automatic transcription, musicsimilarity, etc.) that have emerged in the past five years, drivenby the digital music revolution. New chapter topics include: Psychoacoustic Audio Coding, describing MP3 and relatedaudio coding schemes based on psychoacoustic masking ofquantization noise Music Transcription, including automatically derivingnotes, beats, and chords from music signals. Music Information Retrieval, primarily focusing onaudio-based genre classification, artist/style identification, and similarity estimation. Audio Source Separation, including multi-microphonebeamforming, blind source separation, and the perception-inspiredtechniques usually referred to as Computational Auditory SceneAnalysis (CASA).

Design and implement video game sound from beginning to end with this hands-on course in game audio. Music and sound effects speak to players on a deep level, and this book will show you how to design and implement powerful, interactive sound that measurably improves gameplay. If you are a sound designer or composer and want to do more than just create audio elements and hand them over to someone else for insertion into the game, this book is for you. You'll understand the game development process and implement vital audio experiences-not just create music loops or one-off sound effects. The Game Audio Tutorial isn't just a book-you also get a powerful website (www.thegameaudiotutorial.com) With the proliferation of digital audio distribution over digital media, audio content analysis is

fast becoming a requirement for designers of intelligent signal-adaptive audio processing systems. Written by a well-known expert in the field, this book provides quick access to different analysis algorithms and allows comparison between different approaches to the same task, making it useful for newcomers to audio signal processing and industry experts alike. A review of relevant fundamentals in audio signal processing, psychoacoustics, and music theory, as well as downloadable MATLAB files are also included. Please visit the companion website: www.AudioContentAnalysis.org

This book contains everything you need to know about the business of Music Production, Creativity, Songwriting, Audio Technology, Releasing your own music, plus more. Authored by Music Industry Veteran and Associate Professor, Tony Dofat.

A fully updated second edition of the excellent Digital Audio Signal Processing Well established in the consumer electronics industry, Digital Audio Signal Processing (DASP) techniques are used in audio CD, computer music and multi-media components. In addition, the applications afforded by this versatile technology now range from real-time signal processing to room simulation. Digital Audio Signal Processing, Second Edition covers the latest signal processing algorithms for audio processing. Every chapter has been completely revised with an easy to understand introduction into the basics and exercises have been included for self testing. Additional Matlab files and Java Applets have been provided on an accompanying website, which support the book by easy to access application examples. Key features include: A thoroughly updated and revised second edition of the popular Digital Audio Signal Processing, a comprehensive coverage of the topic as whole Provides basic principles and fundamentals for Quantization, Filters, Dynamic Range Control, Room Simulation,

Sampling Rate Conversion, and Audio Coding Includes detailed accounts of studio technology, digital transmission systems, storage media and audio components for home entertainment Contains precise algorithm description and applications Provides a full account of the techniques of DASP showing their theoretical foundations and practical solutions Includes updated computer-based exercises, an accompanying website, and features Web-based Interactive JAVA-Applets for audio processing This essential guide to digital audio signal processing will serve as an invaluable reference to audio engineering professionals, R&D engineers, researchers in consumer electronics industries and academia, and Hardware and Software developers in IT companies. Advanced students studying multi-media courses will also find this guide of interest.

This book contains a complete and accurate mathematical treatment of the sounds of music with an emphasis on musical timbre. The book spans the range from tutorial introduction to advanced research and application to speculative assessment of its various techniques. All the contributors use a generalized additive sine wave model for describing musical timbre which gives a conceptual unity, but is of sufficient utility to be adapted to many different tasks. Principles of Musical Acoustics focuses on the basic principles in the science and technology of music. Musical examples and specific musical instruments demonstrate the principles. The book begins with a study of vibrations and waves, in that order. These topics constitute the basic physical properties of sound, one of two pillars supporting the science of musical acoustics. The second pillar is the human element, the physiological and psychological aspects of acoustical science. The perceptual topics include loudness, pitch, tone color, and localization of sound. With these two pillars in place, it is possible to go in a variety of

directions. The book treats in turn, the topics of room acoustics, audio both analog and digital, broadcasting, and speech. It ends with chapters on the traditional musical instruments, organized by family. The mathematical level of this book assumes that the reader is familiar with elementary algebra. Trigonometric functions, logarithms and powers also appear in the book, but computational techniques are included as these concepts are introduced, and there is further technical help in appendices.

Fundamentals of Signal Processing for Sound and Vibration Engineers is based on Joe Hammond's many years of teaching experience at the Institute of Sound and Vibration Research, University of Southampton. Whilst the applications presented emphasise sound and vibration, the book focusses on the basic essentials of signal processing that ensures its appeal as a reference text to students and practitioners in all areas of mechanical, automotive, aerospace and civil engineering. Offers an excellent introduction to signal processing for students and professionals in the sound and vibration engineering field. Split into two parts, covering deterministic signals then random signals, and offering a clear explanation of their theory and application together with appropriate MATLAB examples. Provides an excellent study tool for those new to the field of signal processing. Integrates topics within continuous, discrete, deterministic and random signals to facilitate better understanding of the topic as a whole. Illustrated with MATLAB examples, some using 'real' measured data, as well as fifty MATLAB codes on an accompanying website.

Starting with essential maths, fundamentals of signals and systems, and classical concepts of DSP, this book presents, from an application-oriented perspective, modern concepts and methods of DSP including machine learning for audio acoustics and engineering. Content

highlights include but are not limited to room acoustic parameter measurements, filter design, codecs, machine learning for audio pattern recognition and machine audition, spatial audio, array technologies and hearing aids. Some research outcomes are fed into book as worked examples. As a research informed text, the book attempts to present DSP and machine learning from a new and more relevant angle to acousticians and audio engineers. Some MATLAB® codes or frameworks of algorithms are given as downloads available on the CRC Press website. Suggested exploration and mini project ideas are given for "proof of concept" type of exercises and directions for further study and investigation. The book is intended for researchers, professionals, and senior year students in the field of audio acoustics. This open access book provides a concise explanation of the fundamentals and background of the surround sound recording and playback technology Ambisonics. It equips readers with the psychoacoustical, signal processing, acoustical, and mathematical knowledge needed to understand the inner workings of modern processing utilities, special equipment for recording, manipulation, and reproduction in the higher-order Ambisonic format. The book comes with various practical examples based on free software tools and open scientific data for reproducible research. The book's introductory section offers a perspective on Ambisonics spanning from the origins of coincident recordings in the 1930s to the Ambisonic concepts of the 1970s, as well as classical ways of applying Ambisonics in first-order coincident sound scene recording and

reproduction that have been practiced since the 1980s. As, from time to time, the underlying mathematics become quite involved, but should be comprehensive without sacrificing readability, the book includes an extensive mathematical appendix. The book offers readers a deeper understanding of Ambisonic technologies, and will especially benefit scientists, audio-system and audio-recording engineers. In the advanced sections of the book, fundamentals and modern techniques as higher-order Ambisonic decoding, 3D audio effects, and higher-order recording are explained. Those techniques are shown to be suitable to supply audience areas ranging from studio-sized to hundreds of listeners, or headphone-based playback, regardless whether it is live, interactive, or studio-produced 3D audio material.

(Berklee Guide). Understanding Audio explores the fundamentals of audio and acoustics that impact every stage of the music recording process. Whether you are a musician setting up your first Pro Tools project studio, or you are a seasoned recording engineer or producer eager to find a reference that fills in the gaps in your understanding of audio, this book is for you. Understanding Audio will enable you to develop a thorough understanding of the underlying principles of sound, and take some of the mystery and guesswork out of how equipment setup affects the quality of your recordings. Projects at the end of each chapter

will assist you in applying these principles to your own recording environment.

Learn about: \* Basic and advanced audio theory \* Cables and studio wiring \*

Recording studio and console signal flow \* Digital and analog audio \* Studio and listening room acoustics \* Psychoacoustics \* "In the Studio" insights, relating audio principles to real recording situations

How is it possible to take advantage of the processing power of a computer when creating music? This book guides you through the first steps in learning Computer Assisted Composition. The text does not assume any previous experience in Computer Music. It explains how musical ideas can be processed and generated with the help of a computer. You will discover: -How to formalize music so a computer can process it. -The musical effect of computing processes. -Some basic techniques as well as some more advanced concepts. Free software that run on any standard computer makes Computer Assisted Composition more available than ever before. This book gives you the basic knowledge of established techniques for how to use computers in music composition.

Introduction to Audio Analysis serves as a standalone introduction to audio analysis, providing theoretical background to many state-of-the-art techniques. It covers the essential theory necessary to develop audio engineering applications,

but also uses programming techniques, notably MATLAB®, to take a more applied approach to the topic. Basic theory and reproducible experiments are combined to demonstrate theoretical concepts from a practical point of view and provide a solid foundation in the field of audio analysis. Audio feature extraction, audio classification, audio segmentation, and music information retrieval are all addressed in detail, along with material on basic audio processing and frequency domain representations and filtering. Throughout the text, reproducible MATLAB® examples are accompanied by theoretical descriptions, illustrating how concepts and equations can be applied to the development of audio analysis systems and components. A blend of reproducible MATLAB® code and essential theory provides enable the reader to delve into the world of audio signals and develop real-world audio applications in various domains. Practical approach to signal processing: The first book to focus on audio analysis from a signal processing perspective, demonstrating practical implementation alongside theoretical concepts Bridge the gap between theory and practice: The authors demonstrate how to apply equations to real-life code examples and resources. giving you the technical skills to develop real-world applications Library of MATLAB code: The book is accompanied by a well-documented library of MATLAB functions and reproducible experiments

This textbook provides both profound technological knowledge and a comprehensive treatment of essential topics in music processing and music information retrieval. Including numerous examples, figures, and exercises, this book is suited for students, lecturers, and researchers working in audio engineering, computer science, multimedia, and musicology. The book consists of eight chapters. The first two cover foundations of music representations and the Fourier transform?concepts that are then used throughout the book. In the subsequent chapters, concrete music processing tasks serve as a starting point. Each of these chapters is organized in a similar fashion and starts with a general description of the music processing scenario at hand before integrating it into a wider context. It then discusses?in a mathematically rigorous way?important techniques and algorithms that are generally applicable to a wide range of analysis, classification, and retrieval problems. At the same time, the techniques are directly applied to a specific music processing task. By mixing theory and practice, the book?s goal is to offer detailed technological insights as well as a deep understanding of music processing applications. Each chapter ends with a section that includes links to the research literature, suggestions for further reading, a list of references, and exercises. The chapters are organized in a modular fashion, thus offering lecturers and readers many ways to choose.

rearrange or supplement the material. Accordingly, selected chapters or individual sections can easily be integrated into courses on general multimedia, information science, signal processing, music informatics, or the digital humanities.

This book teaches the basics of recording, editing, mixing, and processing audio and MIDI using Logic software. It also provides plenty of power tips to take you beyond the basics and unleash the true power of using Logic Pro X as a creative tool.

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