1. Record Nr. UNINA9910876973703321

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Titolo	Digital Signal Processing : Theory and Practice
Pubbl/distr/stampa	Newark : , : John Wiley & Sons, Incorporated, , 2024 ©2024
ISBN	1-394-18269-4 1-394-18267-8
Edizione	[10th ed.]
Descrizione fisica	1 online resource (397 pages)
Altri autori (Persone)	EngelBenjamin A
Lingua di pubblicazione	Inglese
Formato	Materiale a stampa
Livello bibliografico	Monografia
Nota di contenuto	Cover -- Title Page -- Copyright -- Contents -- Foreword (Historical Perspective) -- Preface -- Introduction -- Chapter 1 Signal Digitizing - Sampling and Coding -- 1.1 Fourier Analysis -- 1.1.1 Fourier Series Expansion of a Periodic Function -- 1.1.2 Fourier Transform of a Function -- 1.2 Distributions -- 1.2.1 Definition -- 1.2.2 Differentiation of Distributions -- 1.2.2.1 The Fourier Transform of a Distribution -- 1.3 Some Commonly Studied Signals -- 1.3.1 Deterministic Signals -- 1.3.2 Random Signals -- 1.3.3 Gaussian Signals -- 1.3.3.1 Peak Factor of a Random Signal -- 1.4 The Norms of a Function -- 1.5 Sampling -- 1.6 Frequency Sampling -- 1.7 The Sampling Theorem -- 1.8 Sampling of Sinusoidal and Random Signals -- 1.8.1 Sinusoidal Signals -- 1.8.2 Discrete Random Signals -- 1.8.3 Discrete Noise Generation -- 1.9 Quantization -- 1.10 The Coding Dynamic Range -- 1.11 Nonlinear Coding with the 13segment Alaw -- 1.12 Optimal Coding -- 1.13 Quantity of Information and Channel Capacity -- 1.14 Binary Representations -- Exercises -- References -- Chapter 2 The Discrete Fourier Transform -- 2.1 Definition and Properties of the Discrete Fourier Transform -- 2.2 Fast Fourier Transform (FFT) -- 2.2.1 Decimationintime Fast Fourier Transform -- 2.2.2 Decimationinfrequency Fast Fourier Transform -- 2.2.3 Radix4 FFT Algorithm -- 2.2.4 Splitradix FFT Algorithm -- 2.3 Degradation Arising from Wordlength Limitation Effects -- 2.4 Calculation of a

Spectrum Using the DFT -- 2.4.1 The Filtering Function of the DFT -- 2.4.2 Spectral Resolution -- 2.5 Fast Convolution -- 2.6 Calculations of a DFT Using Convolution -- 2.7 Implementation -- Exercises -- References -- Chapter 3 Other Fast Algorithms for the FFT -- 3.1 Kronecker Product of Matrices -- 3.2 Factorizing the Matrix of a DecimationinFrequency Algorithm -- 3.3 Partial Transforms.
3.3.1 Transform of Real Data and Odd DFT -- 3.3.2 The Oddtime Odd frequency DFT -- 3.3.3 Sine and Cosine Transforms -- 3.3.4 The Two dimensional DCT -- 3.4 Lapped Transform -- 3.5 Other Fast Algorithms -- 3.6 Binary Fourier Transform - Hadamard -- 3.7 NumberTheoretic Transforms -- Exercises -- References -- Chapter 4 TimeInvariant Discrete Linear Systems -- 4.1 Definition and Properties -- 4.2 The ZTransform -- 4.3 Energy and Power of Discrete Signals -- 4.4 Filtering of Random Signals -- 4.5 Systems Defined by Difference Equations -- 4.6 State Variable Analysis -- Exercises -- References -- Chapter 5 Finite Impulse Response (FIR) Filters -- 5.1 FIR Filters -- 5.2 Practical Transfer Functions and Linear Phase Filters -- 5.3 Calculation of Coefficients by Fourier Series Expansion for Frequency Specifications -- 5.4 Calculation of Coefficients by the LeastSquares Method -- 5.5 Calculation of Coefficient by Discrete Fourier Transform -- 5.6 Calculation of Coefficients by Chebyshev Approximation -- 5.7 Relationships Between the Number of Coefficients and the Filter Characteristic -- 5.8 RaisedCosine Transition Filter -- 5.9 Structures for Implementing FIR Filters -- 5.10 Limitation of the Number of Bits for Coefficients -- 5.11 Z-Transfer Function of an FIR Filter -- 5.12 MinimumPhase Filters -- 5.13 Design of Filters with a Large Number of Coefficients -- 5.14 TwoDimensional FIR Filters -- 5.15 Coefficients of TwoDimensional FIR Filters by the LeastSquares Method -- Exercises -- References -- Chapter 6 Infinite Impulse Response (IIR) Filter Sections -- 6.1 FirstOrder Section -- 6.2 Purely Recursive Second Order Section -- 6.3 General SecondOrder Section -- 6.4 Structures for Implementation -- 6.5 Coefficient Wordlength Limitation -- 6.6 Internal Data Wordlength Limitation -- 6.7 Stability and Limit Cycles -- Exercises -- References.
Chapter 7 Infinite Impulse Response Filters -- 7.1 General Expressions for the Properties of IIR Filters -- 7.2 Direct Calculations of the Coefficients Using Model Functions -- 7.2.1 Impulse Invariance -- 7.2.2 Bilinear Transform -- 7.2.2.1 Butterworth Filters -- 7.2.2.2 Elliptic Filters -- 7.2.2.3 Calculating any Filter by Transformation of a Lowpass Filter -- 7.2.3 Iterative Techniques for Calculating IIR Filter with Frequency -- 7.2.3.1 Minimizing the Mean Square Error -- 7.2.3.2 Chebyshev Approximation -- 7.2.4 Filters Based on Spheroidal Sequences -- 7.2.5 Structures Representing the Transfer Function -- 7.2.6 Limiting the Coefficient Wordlength -- 7.2.7 RoundOff Noise -- 7.2.8 Comparison of IIR and FIR Filters -- Exercises -- References -- Chapter 8 Digital Ladder Filters -- 8.1 Properties of TwoPort Circuits -- 8.2 Simulated Ladder Filters -- 8.3 SwitchedCapacitor Filters -- 8.4 Lattice Filters -- 8.5 Comparison Elements -- Exercises -- References -- Chapter 9 Complex Signals - Quadrature Filters - Interpolators -- 9.1 The Fourier Transform of a Real and Causal Set -- 9.2 Analytic Signals -- 9.3 Calculating the Coefficients of an FIR Quadrature Filter -- 9.4 Recursive 90° Phase Shifters -- 9.5 Single SideBand Modulation -- 9.6 MinimumPhase Filters -- 9.7 Differentiator -- 9.8 Interpolation Using FIR Filters -- 9.9 Lagrange Interpolation -- 9.10 Interpolation by Blocks - Splines -- 9.11 Interpolations and Signal Restoration -- 9.12 Conclusion -- Exercises -- References -- Chapter 10 Multirate Filtering -- 10.1 Decimation and ZTransform -- 10.2 Decomposition of a Low Pass FIR Filter -- 10.3 HalfBand FIR Filters -- 10.4 Decomposition with

HalfBand Filters -- 10.5 Digital Filtering by Polyphase Network -- 10.6 Multirate Filtering with IIR Elements -- 10.7 Filter Banks Using Polyphase Networks and DFT -- 10.8 Conclusion -- Exercises. References -- Chapter 11 QMF Filters and Wavelets -- 11.1 Decomposition into Two SubBands and Reconstruction -- 11.2 QMF Filters -- 11.3 Perfect Decomposition and Reconstruction -- 11.4 Wavelets -- 11.5 Lattice Structures -- Exercises -- References -- Chapter 12 Filter Banks -- 12.1 Decomposition and Reconstruction -- 12.2 Analyzing the Elements of the Polyphase Network -- 12.3 Determining the Inverse Functions -- 12.4 Banks of PseudoQMF Filters -- 12.5 Determining the Coefficients of the Prototype Filter -- 12.6 Realizing a Bank of Real Filters -- Exercises -- References -- Chapter 13 Signal Analysis and Modeling -- 13.1 Autocorrelation and Intercorrelation -- 13.2 Correlogram Spectral Analysis -- 13.3 Single Frequency Estimation -- 13.4 Correlation Matrix -- 13.5 Modeling -- 13.6 Linear Prediction -- 13.7 Predictor Structures -- 13.7.1 Sensor Networks - Antenna Processing -- 13.8 Multiple Sources - MIMO -- 13.9 Conclusion -- Appendix: Estimation Bounds -- Exercises -- References -- Chapter 14 Adaptive Filtering -- 14.1 Principle of Adaptive Filtering -- 14.2 Convergence Conditions -- 14.3 Time Constant -- 14.4 Residual Error -- 14.5 Complexity Parameters -- 14.6 Normalized Algorithms and Sign Algorithms -- 14.7 Adaptive FIR Filtering in Cascade Form -- 14.8 Adaptive IIR Filtering -- 14.9 Conclusion -- Exercises -- References -- Chapter 15 Neural Networks -- 15.1 Classification -- 15.2 Multilayer Perceptron -- 15.3 The Backpropagation Algorithm -- 15.4 Examples of Application -- 15.5 Convolution Neural Networks -- 15.6 Recurrent/Recursive Neural Networks -- 15.7 Neural Network and Signal Processing -- 15.8 On Activation Functions -- 15.9 Conclusion -- Exercises -- References -- Chapter 16 ErrorCorrecting Codes -- 16.1 Reed-Solomon Codes -- 16.1.1 Predictable Signals -- 16.1.2 Reed-Solomon Codes in the Frequency Domain.
16.1.3 Reed-Solomon Codes in the Time Domain -- 16.1.4 Computing in a Finite Field -- 16.1.5 Performance of Reed-Solomon Codes -- 16.2 Convolutional Codes -- 16.2.1 Channel Capacity -- 16.2.2 Approaching the Capacity Limit -- 16.2.3 A Simple Convolutional Code -- 16.2.4 Coding Gain and Error Probability -- 16.2.5 Decoding and Output Signals -- 16.2.6 Recursive Systematic Coding (RSC) -- 16.2.7 Principle of Turbo Codes -- 16.2.8 TrellisCoded Modulations -- 16.3 Conclusion -- Exercises -- References -- Chapter 17 Applications -- 17.1 Frequency Detection -- 17.2 Phaselocked Loop -- 17.3 Differential Coding of Speech -- 17.4 Coding of Sound -- 17.5 Echo Cancelation -- 17.5.1 Data Echo Canceller -- 17.5.1.1 Twowire Line -- 17.5.2 Acoustic Echo Canceler -- 17.6 Television Image Processing -- 17.7 Multicarrier Transmission - OFDM -- 17.8 Mobile Radiocommunications -- References -- Exercises: Solutions and Hints -- Index -- EULA.