Contents

Preface — V

1 Basic Concepts and Principles of Signals and Systems — 1
1.1 Introduction — 1
1.2 Basic Concepts of Signals and Systems — 1
1.2.1 Basic Concept of Signals — 1
1.2.2 The Classification of Signals — 3
1.2.3 Classical Signals and Their Properties — 5
1.2.4 The Operation of Signals — 15
1.2.5 The Basic Concept of Systems — 18
1.2.6 Classification of Systems and Their Properties — 19
1.2.7 Basic Analysis Methods for Systems — 22
1.3 Time Domain Analysis of LTI System and Convolution — 23
1.3.1 The Basic Concept on LTI System — 23
1.3.2 The Time Domain Analysis for Continuous-Time LTI Systems — 23
1.3.3 Convolution of Continuous-Time LTI Systems — 24
1.3.4 The Time Domain Analysis for Discrete-Time LTI Systems — 27
1.3.5 Convolution of Discrete-Time LTI Systems — 28
1.4 Basic Properties of LTI Systems — 31
1.4.1 System with and Without Memory — 31
1.4.2 Invertibility and Inverse Systems — 31
1.4.3 Causality and Causal Systems — 32
1.4.4 Stability and Stable Systems — 32
Exercises — 33

2 Fourier Theory and Frequency Analysis of Signals and Systems — 35
2.1 Introduction — 35
2.1.1 Frequency Analysis of Signals and Systems and a Summary on Fourier Theory — 35
2.1.2 The Development of Fourier Theory — 35
2.1.3 Classification of FS and FT — 36
2.2 FS Representation of Continuous-Time Periodic Signals — 37
2.2.1 Continuous-Time Periodic Signal and Its FS Representation — 37
2.2.2 Properties of FS — 41
2.2.3 Other Forms of FS — 42
2.3 FS Representation of Discrete-Time Periodic Signals — 44
2.3.1 Discrete-Time Periodic Signals and FS — 44
2.3.2 Properties of DFS — 46
2.4 FT of Continuous-Time Signals — 46

Bibliografische Informationen
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2.4.1 From FS to FT — 46
2.4.2 FT of Continuous-Time Signals — 49
2.4.3 Properties of FT and Basic FT Pairs — 53
2.5 Discrete-Time Fourier Transform — 56
2.5.1 Discrete-Time Fourier Transform — 56
2.5.2 Properties of DTFT and Basic DTFT Pairs — 59
2.5.3 The Duality in FS and FT — 59
2.6 Frequency Domain Analysis of Signals and Systems — 64
2.6.1 The Spectral Representation of Signals — 64
2.6.2 The Frequency Analysis of Linear Time-Invariant systems — 66
2.6.3 Bode Plots — 72
2.6.4 Conditions for Distortionless Transmission and Physical Realization of Systems — 73
Exercises — 76

3 The Complex Frequency Domain Analysis of Signals and Systems with Laplace Transform and *z*-Transform — 80
3.1 Introduction — 80
3.2 The Laplace Transform — 81
3.2.1 The Definition of the Laplace Transform — 81
3.2.2 Properties of the ROC for the Laplace Transform — 83
3.2.3 The Inverse Laplace Transform — 84
3.2.4 Properties of the Laplace Transform and Basic Laplace Transform Pairs — 87
3.3 The Analysis of Continuous-Time Signals and Systems in Complex Frequency Domain — 88
3.3.1 The Laplace Transform of Differential Equations and System Functions — 89
3.3.2 Causality and Stability Analysis of LTI Systems — 91
3.3.3 Unilateral Laplace Transform and Its Applications — 92
3.4 The *z*-Transform — 95
3.4.1 The Definition of the *z*-Transform — 95
3.4.2 The Properties of ROC for the *z*-Transform — 98
3.4.3 The Inverse *z*-Transform — 99
3.4.4 Properties of the *z*-Transform and Basic *z*-Transform Pairs — 102
3.5 The Analysis of Discrete-Time Signals and Systems in Complex Frequency Domain — 103
3.5.1 The *z*-Transform of Difference Equations and System Functions — 104
3.5.2 Causality and Stability Analysis of LTI Systems — 106
3.5.3 The Block Diagram Representation for Discrete-Time Systems — 107
3.5.4 Unilateral *z*-Transform and Its Applications — 110
Exercises — 112

4 Discretization of Continuous-Time Signals and Reconstruction of Discrete-Time Signals — 115
4.1 Introduction — 115
4.2 Sampling and the Sampling Theorem for Continuous-Time Signals — 116
4.2.1 Ideal Sampling Based on the Unit Impulse Train and the Sampling Theorem — 116
4.2.2 Sampling with a Zero-Order Hold for Continuous-Time Signals — 121
4.3 The Interpolation and Fitting for Discrete-Time Signals — 123
4.3.1 The Interpolation for Discrete-Time Signals — 123
4.3.2 The Fitting for Discrete-Time Signals — 128
4.3.3 Error Analysis of Interpolation and Fitting — 131
Exercises — 133

5 Discrete Fourier Transform and Fast Fourier Transform — 135
5.1 Introduction — 135
5.2 Discrete Fourier Transform (DFT) — 135
5.2.1 A Brief Review of Fourier Analysis — 135
5.2.2 From Discrete Fourier Series to Discrete Fourier Transform — 137
5.2.3 Properties of the Discrete Fourier Transform — 141
5.3 Some Topics in the Theory of DFT — 147
5.3.1 Frequency Aliasing — 147
5.3.2 Spectrum Leakage — 148
5.3.3 Fence Effect — 150
5.3.4 Frequency Resolution and Parameter Selection in DFT — 150
5.3.5 Zero-padding — 154
5.3.6 Time Width and Bandwidth — 155
5.4 A Brief Introduction to the Two-Dimensional Fourier Transform — 157
5.4.1 Commonly Used Two-Dimensional Discrete Sequences — 157
5.4.2 The Definitions of Two-Dimensional *Z*-Transform and Discrete-Time Fourier Transform — 158
5.4.3 Properties of Two-Dimensional Discrete-Time Fourier Transform — 159
5.4.4 Two-Dimensional Discrete Fourier Transform — 160
5.4.5 Applications of Two-Dimensional Discrete Fourier Transform — 161
5.5 The Fast Fourier Transform (FFT) — 163
5.5.1 The appearance of FFT — 163
5.5.2 Problems for Calculating DFT Directly and Possible Improvement — 163
5.5.3 Decimation-in-Time Radix-2 FFT Algorithm — 165
5.5.4 Decimation-in-Frequency Radix-2 FFT Algorithm — 171
5.5.5 Chirp *Z*-Transform (CZT) — 174

5.6 Applications of FFT — 178
5.6.1 Calculating the Linear Convolution Based on FFT — 178
5.6.2 Calculating the Linear Correlation Based on FFT — 181
Exercises — 182

6 Digital Filter and Digital Filter Design — 184
6.1 Introduction — 184
6.1.1 Classification of Digital Filters — 184
6.1.2 Design of Digital Filters — 185
6.2 Structure of Digital Filters — 186
6.2.1 Representation with Difference Equations — 186
6.2.2 Representation with System Functions — 187
6.2.3 Representation with System Block Diagrams and Signal Flow Graphs — 187
6.3 IIR Digital Filter — 188
6.3.1 The Structure of Direct Form I — 189
6.3.2 The Structure of Direct Form II — 189
6.3.3 Cascade Structure — 190
6.3.4 Parallel Structure — 191
6.4 FIR Digital Filter — 192
6.4.1 Transverse Structure — 193
6.4.2 Cascade Structure — 193
6.4.3 Frequency Sampling Structure — 194
6.4.4 Fast Convolution Structure — 196
6.4.5 The Linear Phase FIR Digital Filter and the Minimum Phase Structure — 197
6.5 Lattice Structure of Digital Filters — 200
6.5.1 All-Zero Lattice FIR Systems — 200
6.5.2 All-Pole Lattice IIR Systems — 203
6.5.3 Pole–Zero Lattice IIR Systems — 204
6.6 IIR Digital Filter Design — 205
6.6.1 Technical Requirement of Filters and Design Brief for Analog Filters — 205
6.6.2 Digital Filter Design Based on the Analog Filter — 209
6.6.3 Digital Filter Design by the Impulse Invariance Procedure — 209
6.6.4 Digital Filter Design by the Bilinear Transform — 213
6.6.5 The Design Idea for High-Pass, Band-Pass, and Band-Stop Digital Filters — 216
6.6.6 IIR Digital Filter Design with MATLAB Programming — 217
6.7 FIR Digital Filter Design — 219
6.7.1 FIR Digital Filter Design by Windowing — 219

6.7.2 The Concept of Window Function and Some Commonly Used Windows — 223
6.7.3 FIR Digital Filter Design with the Frequency Sampling Method — 224
6.7.4 Some Simple Digital Filters — 229
Exercises — 233

7 Finite-Precision Numerical Effects in Digital Signal Processing — 236
7.1 Introduction — 236
7.2 Quantization Effects in Analog-to-Digital (A/D) Conversion — 236
7.2.1 The Basic Concept and the Principle of A/D Conversion — 236
7.2.2 Quantization Effects in A/D Conversion and Error Analysis — 237
7.3 Quantization Effects of Digital Filters — 239
7.3.1 Coefficient Quantization Effects of IIR Digital Filter Coefficients — 239
7.3.2 Coefficient Quantization Effects of FIR Digital Filter Coefficients — 242
7.4 Quantization Effects in the Operation of Digital Filters — 243
7.4.1 The Limit Cycle Oscillation in IIR Digital Filters — 243
7.4.2 Finite Word-Length Effects in Multiplication Operation of IIR Digital Filters — 244
7.4.3 Finite Word-Length Effects in Multiplication Operation of FIR Digital Filters — 245
7.5 Finite Word-Length Effects in Discrete Fourier Transform — 245
Exercises — 247

8 Data Error Analysis and Signal Preprocessing — 249
8.1 Introduction — 249
8.2 The Basic Concept and the Principle of Errors — 249
8.2.1 The Basic Concept of Errors — 249
8.2.2 Stochastic Error — 250
8.2.3 Systematic Error — 252
8.2.4 Gross Error — 253
8.2.5 Combination of Errors — 254
8.2.6 Distribution of Errors — 255
8.3 The Assessment and Estimation of Measurement Uncertainty — 257
8.3.1 The Basic Concept of Measurement Uncertainty — 257
8.3.2 The Assessment of Standard Uncertainty — 258
8.3.3 The Combination of Measurement Uncertainty — 259
8.4 The Least Squares (LS) Method for Data Processing — 259
8.4.1 The Principle of the Least Squares Method — 260
8.4.2 Normal Equation: The Basic Method of the LS Processing — 263
8.4.3 The Estimation Accuracy of the LS Processing — 267

8.4.4 The LS Processing for the Combined Measurement — 269
8.5 The Regression Analysis — 270
8.5.1 Univariate Linear Regression — 271
8.5.2 Univariate Nonlinear Regression — 273
8.5.3 Multiple Linear Regression — 275
8.6 Removal of Trends and Outliers in Signals — 277
8.6.1 Removal of Trends in Signals — 277
8.6.2 Recognition and Processing of Outlines in Signals — 279
8.7 Data Processing Examples of Temperature Measurements — 282
8.7.1 Temperature and Its Measurement — 282
8.7.2 An Engineering Example of Temperature Measurement with Platinum Resistance — 282
8.7.3 Data Analysis and Processing of the Temperature Measurement — 283
Exercises — 284

9 Fundamentals of Random Signal Processing — 286
9.1 Introduction — 286
9.2 The Concept of Random Variable and Its Properties — 286
9.2.1 The Concept of Random Variable — 286
9.2.2 The Distribution of Random Variables — 287
9.2.3 Numerical Characteristics of Random Variables — 290
9.2.4 Characteristic Functions of Random Variables — 293
9.3 Random Processes and Random Signals — 294
9.3.1 The Statistical Distribution of Random Processes and Random Signals — 294
9.3.2 The Stationary Random Process — 297
9.3.3 Ergodicity — 298
9.3.4 Power Spectral Density of Random Signals — 299
9.3.5 Nonstationary Random Signals — 300
9.4 Commonly Used Random Signals and Random Noises — 301
9.4.1 Gaussian (Normal) Distributed Random Signals — 301
9.4.2 White Noise and Band-Limited White Noise — 302
9.4.3 The Gauss–Markov Process — 303
9.4.4 Other Common Noises — 304
9.4.5 The Generation of Random Signals and Noises — 305
9.5 Linear Systems with Random Inputs — 307
9.5.1 The Output of the Linear System and Its Probability Distribution — 307
9.5.2 Numerical Characteristics of the Linear System — 307
9.5.3 Equivalent Noise Bandwidth of System and Bandwidth of Random Signal — 310

9.6 Classical Analysis of Random Signals — 311
9.6.1 The PDF of Common Random Signals — 311
9.6.2 The Calculation of the Numerical Characteristics for Random Signals — 314
9.7 Analysis for Random Signals with Modern Parameter Models — 315
9.7.1 Wold Decomposition Theorem for Random Signals — 315
9.7.2 Linear Parameter Models for Stationary Random Signals — 316
9.7.3 The Estimation of AR Parameters — 317
9.7.4 The Order Determination of the AR Model — 321
Exercises — 322

10 Correlation Estimation and Power Spectral Density (PSD) Estimation of Random Signals — 325
10.1 Introduction — 325
10.1.1 Basic Tasks of Parameter Estimation — 325
10.1.2 Evaluation Criteria of Parameter Estimation — 326
10.2 Correlation Functions and Power Spectral Density Functions — 327
10.2.1 Correlation Functions — 327
10.2.2 Power Spectral Density Functions — 330
10.3 Estimation for Correlation Sequences — 330
10.3.1 Unbiased Estimation for Autocorrelation Sequences — 330
10.3.2 Biased Estimation for Autocorrelation Sequences — 332
10.3.3 Fast Estimation for Autocorrelation Sequences — 334
10.4 Classical Methods for PSD Estimation — 335
10.4.1 The General Development of PSD Estimation — 335
10.4.2 The Periodogram Method — 337
10.4.3 The Performance of Periodogram for PSD Estimation — 340
10.4.4 Methods for Improving the Performance of Periodogram — 343
10.5 Modern Methods for PSD Estimation — 347
10.5.1 Drawbacks of Classical PSD Estimation Methods — 347
10.5.2 The PSD Estimation with AR Model — 348
10.5.3 The PSD Estimation with Maximum Entropy Method — 351
10.5.4 The PSD Estimation with MA and ARMA Models — 355
10.5.5 The PSD Estimation with Minimum Variance Method — 356
10.5.6 Pisarenko Harmonic Decomposition — 358
10.5.7 The PSD Estimation Based on the Matrix Eigenvector Decomposition — 359
10.5.8 A Comparison of Various PSD Estimation Methods — 363
10.6 The Cepstrum Analysis of Signals — 365
10.6.1 The Concept of the Cepstrum — 365
10.6.2 Homomorphic Filtering and Applications of the Cepstrum — 366
10.7 Applications of PSD Estimation and Analysis — 370

10.7.1 Examples of PSD Estimation in Signal Analysis — 370
10.7.2 Examples of PSD Estimation in Medical Auxiliary Diagnosis — 372
Exercises — 374

11 The Optimal Filtering for Random Signals — 377
11.1 Introduction — 377
11.1.1 The Classical Filter and the Statistical Optimal Filter — 377
11.1.2 Two Important Statistical Optimal Filters — 377
11.2 The Principle of Wiener Filter — 379
11.2.1 Causal Wiener Filter — 379
11.2.2 The Solution of the Wiener–Hopf Equation — 381
11.3 Wiener Predictor — 390
11.3.1 Causal Wiener Predictor — 390
11.3.2 *N*-Step Pure Predictor — 392
11.3.3 One-Step Linear Wiener Predictor — 393
11.4 A Brief Introduction to Kalman Filter — 394
11.4.1 The Principle of Kalman Filter — 394
11.4.2 The Analysis of Kalman Filter — 397
11.4.3 The Calculation of Kalman Filter — 398
Exercises — 401

12 Adaptive Filtering — 404
12.1 Introduction — 404
12.2 Structure of Transversal Adaptive Filter and Random Gradient Method — 404
12.2.1 The Structure and Performance Function of Transversal Adaptive Filter — 405
12.2.2 The Search on the Quadratic Performance Surface — 408
12.3 The Least Mean Square Algorithm of Adaptive Filter — 414
12.3.1 Least Mean Square Algorithm — 414
12.3.2 Performance Analysis of LMS Algorithm — 416
12.3.3 Improved LMS Algorithms — 426
12.3.4 Practical Considerations — 432
12.4 The Recursive Least Square Algorithm of Adaptive Filters — 437
12.4.1 The Principle of the Linear LS Algorithm — 437
12.4.2 RLS Adaptive Filter — 439
12.4.3 Practical Considerations — 442
12.5 The Main Structures of Adaptive Filters — 442
12.5.1 Adaptive Noise Cancellation and Its Applications — 442
12.5.2 Adaptive Line Enhancement and Its Applications — 447
12.5.3 Adaptive System Identification and Its Applications — 451
Exercises — 452

13 Higher-Order and Fractional Lower-Order Statistics — 455
13.1 Higher-Order Statistics — 455
13.1.1 Overview of Higher-Order Statistics — 455
13.1.2 The Characteristic Function — 455
13.1.3 Statistical Definitions of Moments and Cumulants — 456
13.1.4 The Properties of Higher-Order Cumulant — 460
13.1.5 Estimation of Higher-Order Cumulant — 461
13.1.6 Higher-Order Spectra and Higher-Order Spectra Estimation — 463
13.2 Application of HOS for the Classification of Lung Sounds — 476
13.3 Alpha-Stable Distribution and Fractional Lower-Order Statistics — 477
13.3.1 Introduction — 477
13.3.2 General Description — 481
13.3.3 Fractional Lower- Order Statistics — 482
13.3.4 General Properties — 485
13.3.5 Characteristic Exponent Estimation of $S\alpha S$ — 488
13.3.6 Multivariate Alpha-Stable Random Variables — 490
13.3.7 Generation of Random Samples — 490
13.4 Application of Fractional Lower-Order Statistics to Time Delay Estimation — 492
Exercises — 498

14 Introduction of Modern Signal Processing — 500
14.1 Time–Frequency Analysis — 500
14.1.1 Fundamental Concept of Time–Frequency Analysis — 500
14.1.2 Short-Time Fourier Analysis — 502
14.1.3 Gabor Expansion and Gabor Transform — 509
14.1.4 Cohen's Class Time–Frequency Distribution — 517
14.1.5 The Application of Time–Frequency Distribution in Electrocardiogram Signal Processing — 526
14.2 The Basic of Wavelets Analysis — 529
14.2.1 Overview — 529
14.2.2 Continuous Wavelet Transform — 531
14.2.3 Discrete Wavelet Transform — 535
14.2.4 Multiscale (Resolution) Analysis — 538
14.2.5 Implementation Technology of Wavelet Transform — 548
14.2.6 Common Wavelet Basis Functions — 551
14.2.7 Application of Wavelet Analysis in Signal Denoising — 552
14.3 Hilbert–Huang Transformation — 556
14.3.1 Introduction — 556
14.3.2 The Basic Concepts and Theory of Hilbert–Huang Transform — 557

14.3.3 Intrinsic Mode Function — **561**
14.3.4 Hilbert Spectrum and Marginal Hilbert Spectrum — **565**
14.3.5 Applications — **568**
Exercises — **571**

Bibliography — 573

Index — 579