Digital Signal Processing

Fourth Edition

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Contents

1 P	reface xvii	ан Айтар Хамар Та р	3.3	Rational
			3.4	Inversio
1.1	Signals, Systems, and Signal Processing	2	3.5	Analysis Time-Inv
1.2	Classification of Signals	6		z-Domai
1.3	The Concept of Frequency in	12	3.6	Summar
	Continuous-Time and			Problem
	Discrete-Time Signals		- 231) 	
1.4	Analog-to-Digital and	19	4 Fi	requency
	Digital-to-Analog Conversion		4.1	Frequen
1.5	Summary and References	37		Continu
	Problems	37	4.2	Frequen Discrete
2 D	iscrete-Time Signals and	40	4.3	Frequen
S	ystems			Time-Do
21	Discrete-Time Signals	41		Properti
		50	4.4	Properti
L .L	Discrete-Time Systems	32		Transfor
2.3	Analysis of Discrete-Time	68	AE	Signais
	Linear Time-Invariant Systems	00	4.3	Summar
2.4	Discrete-Time Systems Described by Difference	88	. .	Problem
75		100	Э н 11	requency
<i>4</i>)	Discrete-Time Systems	100	an a	r system
26	Correlation of Discrete-Time	115	5.1	Frequen
	Signals			
2.7	Summary and References	127	5.2	Frequen
	Problems	128		Systems
			5.3	Linear T
3 т	he z-Transform and Its	141	an a	as Frequ
A	pplication to the Analysis of		5.4	Deconvo
L .		ي و و	5.5	Summar
3.1	The z-Transform	141		Problem
3.2	Properties of the z-Transform	151		

.3	Rational z-Transforms	165
.4	Inversion of the z-Transform	175
.5	Analysis of Linear Time-Invariant Systems in the z-Domain	189
.6	Summary and References	200
	Problems	202
F	requency Analysis of Signals	209
.1	Frequency Analysis of	210
.2	Frequency Analysis of Discrete-Time Signals	226
.3	Frequency-Domain and Time-Domain Signal	253
.4	Properties Properties of the Fourier Transform for Discrete-Time	255
.5	Signals Summary and References	275
	Problems	276
Fi	requency-Domain Analysis of FL Systems	281
.1	Frequency-Domain Characteristics of Linear Time-Invariant Systems	281
.2	Frequency Response of LTI Systems	295
.3	Linear Time-Invariant Systems as Frequency-Selective Filters	302
.4	Inverse Systems and Deconvolution	326
.5	Summary and References	339
	Problems	340

vi	Contents
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6 Sampling and Reconstruction 354 of Signals

- 6.1 Ideal Sampling and 354 Reconstruction of Continuous-Time Signals
- **6.2 Discrete-Time Processing of** 365 **Continuous-Time Signals**
- 6.3 Analog-to-Digital and 371 Digital-to-Analog Converters
- 6.4 Sampling and Reconstruction 380 of Continuous-Time Bandpass Signals
- 6.5 Sampling of Discrete-Time 397 Signals
- 6.6 Oversampling A/D and D/A 403 Converters
- 6.7Summary and References410Problems410

7 The Discrete Fourier 416 Transform: Its Properties and Applications

- 7.1 Frequency-Domain Sampling: 416 The Discrete Fourier Transform
- **7.2 Properties of the DFT** 431
- 7.3 Linear Filtering Methods Based 447 on the DFT
- 7.4 Frequency Analysis of Signals 455 Using the DFT
- 7.5 The Discrete Cosine 462 Transform
- 7.6Summary and References468Problems469
- 8 Efficient Computation of the 476 DFT: Fast Fourier Transform Algorithms
 - 8.1 Efficient Computation of the 476 DFT: FFT Algorithms
 8.2 Applications of FFT 503
 - 8.2 Applications of FFT Algorithms

8.3	A Linear Filtering Approach to Computation of the DFT	507
8.4	Quantization Effects in the Computation of the DFT	514
8.5	Summary and References	520
	Problems	521
9 In Di	nplementation of iscrete-Time Systems	525
9.1	Structures for the Realization of Discrete-Time Systems	525
9.2	Structures for FIR Systems	527
9.3	Structures for IIR Systems	544
9.4	Representation of Numbers	563
9.5	Quantization of Filter Coefficients	575
9.6	Round-Off Effects in Digital Filters	586
9.7	Summary and References	602
	Problems	603
30 -		299
IV D	esign of Digital Filters	012
10.1	General Considerations	612
10.2	Design of FIR Filters	618
10.3	Design of IIR Filters From Analog Filters	659
10.4	Frequency Transformations	688
10.5	Summary and References	692
	Problems	693
41 .		701
	rocessing	/ 74
11.1	Introduction	705
11.2	Decimation by a Factor D	709
11.3	Interpolation by a Factor I	714
11.4	Sampling Rate Conversion by	716

a Rational Factor I/D 11.5 Implementation of Sampling 720 Rate Conversion

Contents Vii

11.6	Multistage Implementation of Sampling Rate Conversion	729	12.7	Wien Predi
11.7	Sampling Rate Conversion of Bandpass Signals	733	12.8	Sum
11.8	Sampling Rate Conversion by an Arbitrary Factor	735		FIOD
11.9	Applications of Multirate	738	13 A	daptiv
	Signal Processing		13.1	Appli
11.10	Digital Filter Banks	744		Filter
11.11	Two-Channel Quadrature Mirror Filter Bank	752	13.2	Adap Filter
11.12	M-Channel QMF Bank	761	13.3	Adap
11.13	Summary and References	767		Filter
	Problems	767	13.4	Adap Filter
12 LI 0	near Prediction and ptimum Linear Filters	774	13.5	Sumr Prob
12.1	Random Signals, Correlation Functions, and Power Spectra	774	A R	andor
12.2	Innovations Representation of a Stationary Random Process	785	Вт	ables oeffic
12.3	Forward and Backward Linear Prediction	789	L	inear-
12.4	Solution of the Normal Equations	797	i destrada R Optenana di	efere
12.5	Properties of the Linear Prediction-Error Filters	806	A	nswe
12.6	AR Lattice and ARMA Lattice-Ladder Filters	809		ndex

12.7	Wiener Filters for Filtering and Prediction	814		
12.8	Summary and References	824		
	Problems	825		
3	Adaptive Filters	830		
13.1	Applications of Adaptive Filters	830		
13.2	Adaptive Direct-Form FIR Filters—The LMS Algorithm	852		
13.3	Adaptive Direct-Form Filters—RLS Algorithms	866		
13.4	Adaptive Lattice-Ladder Filters	877		
13.5	Summary and References	904		
	Problems	905		
	Random Number Generators	909		
	Tables of Transition Coefficients for the Design of Linear-Phase FIR Filters	915		
	References and Bibliography			
	Answers to Selected Problems			

943