

FOURTH EDITION

Advanced
Digital Signal
Processing
and
Noise
Reduction

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 WILEY

Contents

Preface	xix
Acknowledgements	xxiii
Symbols	xxv
Abbreviations	xxix
1 Introduction	1
1.1 Signals, Noise and Information	1
1.2 Signal Processing Methods	3
1.2.1 <i>Transform-Based Signal Processing</i>	3
1.2.2 <i>Source-Filter Model-Based Signal Processing</i>	5
1.2.3 <i>Bayesian Statistical Model-Based Signal Processing</i>	5
1.2.4 <i>Neural Networks</i>	6
1.3 Applications of Digital Signal Processing	6
1.3.1 <i>Digital Watermarking</i>	6
1.3.2 <i>Bio-medical, MIMO, Signal Processing</i>	8
1.3.3 <i>Echo Cancellation</i>	10
1.3.4 <i>Adaptive Noise Cancellation</i>	12
1.3.5 <i>Adaptive Noise Reduction</i>	12
1.3.6 <i>Blind Channel Equalisation</i>	13
1.3.7 <i>Signal Classification and Pattern Recognition</i>	13
1.3.8 <i>Linear Prediction Modelling of Speech</i>	15
1.3.9 <i>Digital Coding of Audio Signals</i>	16
1.3.10 <i>Detection of Signals in Noise</i>	17
1.3.11 <i>Directional Reception of Waves: Beam-forming</i>	18
1.3.12 <i>Space-Time Signal Processing</i>	20
1.3.13 <i>Dolby Noise Reduction</i>	20
1.3.14 <i>Radar Signal Processing: Doppler Frequency Shift</i>	21
1.4 A Review of Sampling and Quantisation	22
1.4.1 <i>Advantages of Digital Format</i>	24
1.4.2 <i>Digital Signals Stored and Transmitted in Analogue Format</i>	25
1.4.3 <i>The Effect of Digitisation on Signal Bandwidth</i>	25
1.4.4 <i>Sampling a Continuous-Time Signal</i>	25
1.4.5 <i>Aliasing Distortion</i>	27
1.4.6 <i>Nyquist Sampling Theorem</i>	27

1.4.7	<i>Quantisation</i>	28
1.4.8	<i>Non-Linear Quantisation, Companding</i>	30
1.5	Summary	32
	Bibliography	32
2	Noise and Distortion	35
2.1	Introduction	35
2.1.1	<i>Different Classes of Noise Sources and Distortions</i>	36
2.1.2	<i>Different Classes and Spectral/Temporal Shapes of Noise</i>	37
2.2	White Noise	37
2.2.1	<i>Band-Limited White Noise</i>	38
2.3	Coloured Noise; Pink Noise and Brown Noise	39
2.4	Impulsive and Click Noise	39
2.5	Transient Noise Pulses	41
2.6	Thermal Noise	41
2.7	Shot Noise	42
2.8	Flicker (I/f) Noise	43
2.9	Burst Noise	44
2.10	Electromagnetic (Radio) Noise	45
2.10.1	<i>Natural Sources of Radiation of Electromagnetic Noise</i>	45
2.10.2	<i>Man-made Sources of Radiation of Electromagnetic Noise</i>	45
2.11	Channel Distortions	46
2.12	Echo and Multi-path Reflections	47
2.13	Modelling Noise	47
2.13.1	<i>Frequency Analysis and Characterisation of Noise</i>	47
2.13.2	<i>Additive White Gaussian Noise Model (AWGN)</i>	48
2.13.3	<i>Hidden Markov Model and Gaussian Mixture Models for Noise</i>	49
	Bibliography	50
3	Information Theory and Probability Models	51
3.1	Introduction: Probability and Information Models	52
3.2	Random Processes	53
3.2.1	<i>Information-bearing Random Signals vs Deterministic Signals</i>	53
3.2.2	<i>Pseudo-Random Number Generators (PRNG)</i>	55
3.2.3	<i>Stochastic and Random Processes</i>	56
3.2.4	<i>The Space of Variations of a Random Process</i>	56
3.3	Probability Models of Random Signals	57
3.3.1	<i>Probability as a Numerical Mapping of Belief</i>	57
3.3.2	<i>The Choice of One and Zero as the Limits of Probability</i>	57
3.3.3	<i>Discrete, Continuous and Finite-State Probability Models</i>	58
3.3.4	<i>Random Variables and Random Processes</i>	58
3.3.5	<i>Probability and Random Variables – The Space and Subspaces of a Variable</i>	58
3.3.6	<i>Probability Mass Function – Discrete Random Variables</i>	60
3.3.7	<i>Bayes' Rule</i>	60
3.3.8	<i>Probability Density Function – Continuous Random Variables</i>	61
3.3.9	<i>Probability Density Functions of Continuous Random Processes</i>	62
3.3.10	<i>Histograms – Models of Probability</i>	63
3.4	Information Models	64
3.4.1	<i>Entropy: A Measure of Information and Uncertainty</i>	65
3.4.2	<i>Mutual Information</i>	68

3.4.3	<i>Entropy Coding – Variable Length Codes</i>	69
3.4.4	<i>Huffman Coding</i>	70
3.5	Stationary and Non-Stationary Random Processes	73
3.5.1	<i>Strict-Sense Stationary Processes</i>	75
3.5.2	<i>Wide-Sense Stationary Processes</i>	75
3.5.3	<i>Non-Stationary Processes</i>	76
3.6	Statistics (Expected Values) of a Random Process	76
3.6.1	<i>Central Moments</i>	77
3.6.1.1	<i>Cumulants</i>	77
3.6.2	<i>The Mean (or Average) Value</i>	77
3.6.3	<i>Correlation, Similarity and Dependency</i>	78
3.6.4	<i>Autocovariance</i>	81
3.6.5	<i>Power Spectral Density</i>	81
3.6.6	<i>Joint Statistical Averages of Two Random Processes</i>	83
3.6.7	<i>Cross-Correlation and Cross-Covariance</i>	83
3.6.8	<i>Cross-Power Spectral Density and Coherence</i>	84
3.6.9	<i>Ergodic Processes and Time-Averaged Statistics</i>	85
3.6.10	<i>Mean-Ergodic Processes</i>	85
3.6.11	<i>Correlation-Ergodic Processes</i>	86
3.7	Some Useful Practical Classes of Random Processes	87
3.7.1	<i>Gaussian (Normal) Process</i>	87
3.7.2	<i>Multivariate Gaussian Process</i>	88
3.7.3	<i>Gaussian Mixture Process</i>	89
3.7.4	<i>Binary-State Gaussian Process</i>	90
3.7.5	<i>Poisson Process – Counting Process</i>	91
3.7.6	<i>Shot Noise</i>	92
3.7.7	<i>Poisson–Gaussian Model for Clutters and Impulsive Noise</i>	93
3.7.8	<i>Markov Processes</i>	94
3.7.9	<i>Markov Chain Processes</i>	95
3.7.10	<i>Homogeneous and Inhomogeneous Markov Chains</i>	96
3.7.11	<i>Gamma Probability Distribution</i>	96
3.7.12	<i>Rayleigh Probability Distribution</i>	97
3.7.13	<i>Chi Distribution</i>	97
3.7.14	<i>Laplacian Probability Distribution</i>	98
3.8	Transformation of a Random Process	98
3.8.1	<i>Monotonic Transformation of Random Processes</i>	99
3.8.2	<i>Many-to-One Mapping of Random Signals</i>	101
3.9	Search Engines: Citation Ranking	103
3.9.1	<i>Citation Ranking in Web Page Rank Calculation</i>	104
3.10	Summary	104
	Bibliography	105
4	Bayesian Inference	107
4.1	Bayesian Estimation Theory: Basic Definitions	108
4.1.1	<i>Bayes' Theorem</i>	109
4.1.2	<i>Elements of Bayesian Inference</i>	109
4.1.3	<i>Dynamic and Probability Models in Estimation</i>	110
4.1.4	<i>Parameter Space and Signal Space</i>	111
4.1.5	<i>Parameter Estimation and Signal Restoration</i>	111
4.1.6	<i>Performance Measures and Desirable Properties of Estimators</i>	112
4.1.7	<i>Prior and Posterior Spaces and Distributions</i>	114

4.2 Bayesian Estimation	117
4.2.1 <i>Maximum A Posteriori Estimation</i>	117
4.2.2 <i>Maximum-Likelihood (ML) Estimation</i>	118
4.2.3 <i>Minimum Mean Square Error Estimation</i>	121
4.2.4 <i>Minimum Mean Absolute Value of Error Estimation</i>	122
4.2.5 <i>Equivalence of the MAP, ML, MMSE and MAVE Estimates for Gaussian Processes with Uniform Distributed Parameters</i>	123
4.2.6 <i>Influence of the Prior on Estimation Bias and Variance</i>	123
4.2.7 <i>Relative Importance of the Prior and the Observation</i>	126
4.3 Expectation-Maximisation (EM) Method	128
4.3.1 <i>Complete and Incomplete Data</i>	128
4.3.2 <i>Maximisation of Expectation of the Likelihood Function</i>	129
4.3.3 <i>Derivation and Convergence of the EM Algorithm</i>	130
4.4 Cramer–Rao Bound on the Minimum Estimator Variance	131
4.4.1 <i>Cramer–Rao Bound for Random Parameters</i>	133
4.4.2 <i>Cramer–Rao Bound for a Vector Parameter</i>	133
4.5 Design of Gaussian Mixture Models (GMMs)	134
4.5.1 <i>EM Estimation of Gaussian Mixture Model</i>	134
4.6 Bayesian Classification	136
4.6.1 <i>Binary Classification</i>	137
4.6.2 <i>Classification Error</i>	139
4.6.3 <i>Bayesian Classification of Discrete-Valued Parameters</i>	139
4.6.4 <i>Maximum A Posteriori Classification</i>	140
4.6.5 <i>Maximum-Likelihood Classification</i>	140
4.6.6 <i>Minimum Mean Square Error Classification</i>	140
4.6.7 <i>Bayesian Classification of Finite State Processes</i>	141
4.6.8 <i>Bayesian Estimation of the Most Likely State Sequence</i>	142
4.7 Modelling the Space of a Random Process	143
4.7.1 <i>Vector Quantisation of a Random Process</i>	143
4.7.2 <i>Vector Quantisation using Gaussian Models of Clusters</i>	143
4.7.3 <i>Design of a Vector Quantiser: K-Means Clustering</i>	144
4.8 Summary	145
Bibliography	146
5 Hidden Markov Models	147
5.1 Statistical Models for Non-Stationary Processes	147
5.2 Hidden Markov Models	149
5.2.1 <i>Comparison of Markov and Hidden Markov Models</i>	149
5.2.1.1 Observable-State Markov Process	149
5.2.1.2 Hidden-State Markov Process	149
5.2.2 <i>A Physical Interpretation: HMMs of Speech</i>	151
5.2.3 <i>Hidden Markov Model as a Bayesian Model</i>	152
5.2.4 <i>Parameters of a Hidden Markov Model</i>	152
5.2.5 <i>State Observation Probability Models</i>	153
5.2.6 <i>State Transition Probabilities</i>	154
5.2.7 <i>State-Time Trellis Diagram</i>	154
5.3 Training Hidden Markov Models	155
5.3.1 <i>Forward–Backward Probability Computation</i>	156
5.3.2 <i>Baum–Welch Model Re-estimation</i>	157
5.3.3 <i>Training HMMs with Discrete Density Observation Models</i>	158

5.3.4	<i>HMMs with Continuous Density Observation Models</i>	159
5.3.5	<i>HMMs with Gaussian Mixture pdfs</i>	160
5.4	Decoding Signals Using Hidden Markov Models	161
5.4.1	<i>Viterbi Decoding Algorithm</i>	162
5.4.1.1	<i>Viterbi Algorithm</i>	163
5.5	<i>HMMs in DNA and Protein Sequences</i>	164
5.6	HMMs for Modelling Speech and Noise	165
5.6.1	<i>Modelling Speech</i>	165
5.6.2	<i>HMM-Based Estimation of Signals in Noise</i>	166
5.6.3	<i>Signal and Noise Model Combination and Decomposition</i>	167
5.6.4	<i>Hidden Markov Model Combination</i>	168
5.6.5	<i>Decomposition of State Sequences of Signal and Noise</i>	169
5.6.6	<i>HMM-Based Wiener Filters</i>	169
5.6.7	<i>Modelling Noise Characteristics</i>	170
5.7	Summary	171
	Bibliography	171
6	Least Square Error Wiener-Kolmogorov Filters	173
6.1	Least Square Error Estimation: Wiener-Kolmogorov Filter	173
6.1.1	<i>Derivation of Wiener Filter Equation</i>	174
6.1.2	<i>Calculation of Autocorrelation of Input and Cross-Correlation of Input and Desired Signals</i>	177
6.2	Block-Data Formulation of the Wiener Filter	178
6.2.1	<i>QR Decomposition of the Least Square Error Equation</i>	179
6.3	Interpretation of Wiener Filter as Projection in Vector Space	179
6.4	Analysis of the Least Mean Square Error Signal	181
6.5	Formulation of Wiener Filters in the Frequency Domain	182
6.6	Some Applications of Wiener Filters	183
6.6.1	<i>Wiener Filter for Additive Noise Reduction</i>	183
6.6.2	<i>Wiener Filter and Separability of Signal and Noise</i>	185
6.6.3	<i>The Square-Root Wiener Filter</i>	186
6.6.4	<i>Wiener Channel Equaliser</i>	187
6.6.5	<i>Time-Alignment of Signals in Multi-channel/Multi-sensor Systems</i>	187
6.7	Implementation of Wiener Filters	188
6.7.1	<i>Choice of Wiener Filter Order</i>	189
6.7.2	<i>Improvements to Wiener Filters</i>	190
6.8	Summary	191
	Bibliography	191
7	Adaptive Filters: Kalman, RLS, LMS	193
7.1	Introduction	194
7.2	State-Space Kalman Filters	195
7.2.1	<i>Derivation of Kalman Filter Algorithm</i>	197
7.2.2	<i>Recursive Bayesian Formulation of Kalman Filter</i>	200
7.2.3	<i>Markovian Property of Kalman Filter</i>	201
7.2.4	<i>Comparison of Kalman filter and hidden Markov model</i>	202
7.2.5	<i>Comparison of Kalman and Wiener Filters</i>	202
7.3	Extended Kalman Filter (EKF)	206
7.4	Unscented Kalman Filter (UKF)	208
7.5	Sample Adaptive Filters – LMS, RLS	211

7.6	Recursive Least Square (RLS) Adaptive Filters	213
7.6.1	<i>Matrix Inversion Lemma</i>	214
7.6.2	<i>Recursive Time-update of Filter Coefficients</i>	215
7.7	The Steepest-Descent Method	217
7.7.1	<i>Convergence Rate</i>	219
7.7.2	<i>Vector-Valued Adaptation Step Size</i>	220
7.8	Least Mean Squared Error (LMS) Filter	220
7.8.1	<i>Leaky LMS Algorithm</i>	220
7.8.2	<i>Normalised LMS Algorithm</i>	221
7.8.2.1	Derivation of the Normalised LMS Algorithm	221
7.8.2.2	Steady-State Error in LMS	222
7.9	Summary	223
	Bibliography	224
8	Linear Prediction Models	227
8.1	Linear Prediction Coding	227
8.1.1	<i>Predictability, Information and Bandwidth</i>	228
8.1.2	<i>Applications of LP Model in Speech Processing</i>	229
8.1.3	<i>Time-Domain Description of LP Models</i>	229
8.1.4	<i>Frequency Response of LP Model and its Poles</i>	230
8.1.5	<i>Calculation of Linear Predictor Coefficients</i>	232
8.1.6	<i>Effect of Estimation of Correlation Function on LP Model Solution</i>	233
8.1.7	<i>The Inverse Filter: Spectral Whitening, De-correlation</i>	234
8.1.8	<i>The Prediction Error Signal</i>	235
8.2	Forward, Backward and Lattice Predictors	236
8.2.1	<i>Augmented Equations for Forward and Backward Predictors</i>	238
8.2.2	<i>Levinson–Durbin Recursive Solution</i>	238
8.2.2.1	Levinson–Durbin Algorithm	240
8.2.3	<i>Lattice Predictors</i>	240
8.2.4	<i>Alternative Formulations of Least Square Error Prediction</i>	241
8.2.4.1	Burg's Method	241
8.2.5	<i>Simultaneous Minimisation of the Backward and Forward Prediction Errors</i>	242
8.2.6	<i>Predictor Model Order Selection</i>	242
8.3	Short-Term and Long-Term Predictors	243
8.4	MAP Estimation of Predictor Coefficients	245
8.4.1	<i>Probability Density Function of Predictor Output</i>	245
8.4.2	<i>Using the Prior pdf of the Predictor Coefficients</i>	246
8.5	Formant-Tracking LP Models	247
8.6	Sub-Band Linear Prediction Model	248
8.7	Signal Restoration Using Linear Prediction Models	249
8.7.1	<i>Frequency-Domain Signal Restoration Using Prediction Models</i>	251
8.7.2	<i>Implementation of Sub-Band Linear Prediction Wiener Filters</i>	253
8.8	Summary	254
	Bibliography	254
9	Eigenvalue Analysis and Principal Component Analysis	257
9.1	Introduction – Linear Systems and Eigen Analysis	257
9.1.1	<i>A Geometric Interpretation of Eigenvalues and Eigenvectors</i>	258
9.2	Eigen Vectors and Eigenvalues	261
9.2.1	<i>Matrix Spectral Theorem</i>	263
9.2.2	<i>Computation of Eigenvalues and Eigen Vectors</i>	263

9.3	Principal Component Analysis (PCA)	264
9.3.1	<i>Computation of PCA</i>	265
9.3.2	<i>PCA Analysis of Images: Eigen-Image Representation</i>	265
9.3.3	<i>PCA Analysis of Speech in White Noise</i>	266
9.4	Summary	269
	Bibliography	270
10	Power Spectrum Analysis	271
10.1	Power Spectrum and Correlation	271
10.2	Fourier Series: Representation of Periodic Signals	272
10.2.1	<i>The Properties of Fourier's Sinusoidal Basis Functions</i>	272
10.2.2	<i>The Basis Functions of Fourier Series</i>	273
10.2.3	<i>Fourier Series Coefficients</i>	274
10.3	Fourier Transform: Representation of Non-periodic Signals	274
10.3.1	<i>Discrete Fourier Transform</i>	276
10.3.2	<i>Frequency-Time Resolutions: The Uncertainty Principle</i>	277
10.3.3	<i>Energy-Spectral Density and Power-Spectral Density</i>	278
10.4	Non-Parametric Power Spectrum Estimation	279
10.4.1	<i>The Mean and Variance of Periodograms</i>	279
10.4.2	<i>Averaging Periodograms (Bartlett Method)</i>	280
10.4.3	<i>Welch Method: Averaging Periodograms from Overlapped and Windowed Segments</i>	280
10.4.4	<i>Blackman-Tukey Method</i>	282
10.4.5	<i>Power Spectrum Estimation from Autocorrelation of Overlapped Segments</i>	282
10.5	Model-Based Power Spectrum Estimation	283
10.5.1	<i>Maximum-Entropy Spectral Estimation</i>	283
10.5.2	<i>Autoregressive Power Spectrum Estimation</i>	285
10.5.3	<i>Moving-Average Power Spectrum Estimation</i>	286
10.5.4	<i>Autoregressive Moving-Average Power Spectrum Estimation</i>	286
10.6	High-Resolution Spectral Estimation Based on Subspace Eigen-Analysis	287
10.6.1	<i>Pisarenko Harmonic Decomposition</i>	287
10.6.2	<i>Multiple Signal Classification (MUSIC) Spectral Estimation</i>	289
10.6.3	<i>Estimation of Signal Parameters via Rotational Invariance Techniques (ESPRIT)</i>	291
10.7	Summary	293
	Bibliography	293
11	Interpolation – Replacement of Lost Samples	295
11.1	Introduction	295
11.1.1	<i>Ideal Interpolation of a Sampled Signal</i>	296
11.1.2	<i>Digital Interpolation by a Factor of I</i>	297
11.1.3	<i>Interpolation of a Sequence of Lost Samples</i>	299
11.1.4	<i>The Factors That Affect Interpolation Accuracy</i>	300
11.2	Polynomial Interpolation	301
11.2.1	<i>Lagrange Polynomial Interpolation</i>	302
11.2.2	<i>Newton Polynomial Interpolation</i>	303
11.2.3	<i>Hermite Polynomial Interpolation</i>	304
11.2.4	<i>Cubic Spline Interpolation</i>	305
11.3	Model-Based Interpolation	306
11.3.1	<i>Maximum A Posteriori Interpolation</i>	307

<i>11.3.2 Least Square Error Autoregressive Interpolation</i>	308
<i>11.3.3 Interpolation Based on a Short-Term Prediction Model</i>	309
<i>11.3.4 Interpolation Based on Long-Term and Short-term Correlations</i>	312
<i>11.3.5 LSAR Interpolation Error</i>	314
<i>11.3.6 Interpolation in Frequency-Time Domain</i>	316
<i>11.3.7 Interpolation Using Adaptive Code Books</i>	317
<i>11.3.8 Interpolation Through Signal Substitution</i>	318
<i>11.3.9 LP-HNM Model based Interpolation</i>	318
11.4 Summary	319
Bibliography	319
12 Signal Enhancement via Spectral Amplitude Estimation	321
12.1 Introduction	321
<i>12.1.1 Spectral Representation of Noisy Signals</i>	322
<i>12.1.2 Vector Representation of Spectrum of Noisy Signals</i>	323
12.2 Spectral Subtraction	324
<i>12.2.1 Power Spectrum Subtraction</i>	325
<i>12.2.2 Magnitude Spectrum Subtraction</i>	326
<i>12.2.3 Spectral Subtraction Filter: Relation to Wiener Filters</i>	326
<i>12.2.4 Processing Distortions</i>	327
<i>12.2.5 Effect of Spectral Subtraction on Signal Distribution</i>	328
<i>12.2.6 Reducing the Noise Variance</i>	329
<i>12.2.7 Filtering Out the Processing Distortions</i>	329
<i>12.2.8 Non-Linear Spectral Subtraction</i>	330
<i>12.2.9 Implementation of Spectral Subtraction</i>	332
12.3 Bayesian MMSE Spectral Amplitude Estimation	333
12.4 Estimation of Signal to Noise Ratios	335
12.5 Application to Speech Restoration and Recognition	336
12.6 Summary	338
Bibliography	338
13 Impulsive Noise: Modelling, Detection and Removal	341
13.1 Impulsive Noise	341
<i>13.1.1 Definition of a Theoretical Impulse Function</i>	341
<i>13.1.2 The Shape of a Real Impulse in a Communication System</i>	342
<i>13.1.3 The Response of a Communication System to an Impulse</i>	343
<i>13.1.4 The Choice of Time or Frequency Domain for Processing of Signals Degraded by Impulsive Noise</i>	343
13.2 Autocorrelation and Power Spectrum of Impulsive Noise	344
13.3 Probability Models of Impulsive Noise	345
<i>13.3.1 Bernoulli-Gaussian Model of Impulsive Noise</i>	346
<i>13.3.2 Poisson-Gaussian Model of Impulsive Noise</i>	346
<i>13.3.3 A Binary-State Model of Impulsive Noise</i>	347
<i>13.3.4 Hidden Markov Model of Impulsive and Burst Noise</i>	348
13.4 Impulsive Noise Contamination, Signal to Impulsive Noise Ratio	349
13.5 Median Filters for Removal of Impulsive Noise	350
13.6 Impulsive Noise Removal Using Linear Prediction Models	351
<i>13.6.1 Impulsive Noise Detection</i>	352
<i>13.6.2 Analysis of Improvement in Noise Detectability</i>	353
<i>13.6.3 Two-Sided Predictor for Impulsive Noise Detection</i>	355
<i>13.6.4 Interpolation of Discarded Samples</i>	355

13.7	Robust Parameter Estimation	355
13.8	Restoration of Archived Gramophone Records	357
13.9	Summary	358
	Bibliography	358
14	Transient Noise Pulses	359
14.1	Transient Noise Waveforms	359
14.2	Transient Noise Pulse Models	361
14.2.1	<i>Noise Pulse Templates</i>	361
14.2.2	<i>Autoregressive Model of Transient Noise Pulses</i>	362
14.2.3	<i>Hidden Markov Model of a Noise Pulse Process</i>	363
14.3	Detection of Noise Pulses	364
14.3.1	<i>Matched Filter for Noise Pulse Detection</i>	364
14.3.2	<i>Noise Detection Based on Inverse Filtering</i>	365
14.3.3	<i>Noise Detection Based on HMM</i>	365
14.4	Removal of Noise Pulse Distortions	366
14.4.1	<i>Adaptive Subtraction of Noise Pulses</i>	366
14.4.2	<i>AR-based Restoration of Signals Distorted by Noise Pulses</i>	367
14.5	Summary	369
	Bibliography	369
15	Echo Cancellation	371
15.1	Introduction: Acoustic and Hybrid Echo	371
15.2	Echo Return Time: The Sources of Delay in Communication Networks	373
15.2.1	<i>Transmission link (electromagnetic wave propagation) delay</i>	374
15.2.2	<i>Speech coding/decoding delay</i>	374
15.2.3	<i>Network processing delay</i>	374
15.2.4	<i>De-Jitter delay</i>	375
15.2.5	<i>Acoustic echo delay</i>	375
15.3	Telephone Line Hybrid Echo	375
15.3.1	<i>Echo Return Loss</i>	376
15.4	Hybrid (Telephone Line) Echo Suppression	377
15.5	Adaptive Echo Cancellation	377
15.5.1	<i>Echo Canceller Adaptation Methods</i>	379
15.5.2	<i>Convergence of Line Echo Canceller</i>	380
15.5.3	<i>Echo Cancellation for Digital Data Transmission</i>	380
15.6	Acoustic Echo	381
15.7	Sub-Band Acoustic Echo Cancellation	384
15.8	Echo Cancellation with Linear Prediction Pre-whitening	385
15.9	Multi-Input Multi-Output Echo Cancellation	386
15.9.1	<i>Stereophonic Echo Cancellation Systems</i>	386
15.9.2	<i>Non-uniqueness Problem in MIMO Echo Channel Identification</i>	387
15.9.3	<i>MIMO In-Cabin Communication Systems</i>	388
15.10	Summary	389
	Bibliography	389
16	Channel Equalisation and Blind Deconvolution	391
16.1	Introduction	391
16.1.1	<i>The Ideal Inverse Channel Filter</i>	392
16.1.2	<i>Equalisation Error, Convolutional Noise</i>	393
16.1.3	<i>Blind Equalisation</i>	394

<i>16.1.4 Minimum- and Maximum-Phase Channels</i>	396
<i>16.1.5 Wiener Equaliser</i>	396
16.2 Blind Equalisation Using Channel Input Power Spectrum	398
<i>16.2.1 Homomorphic Equalisation</i>	398
<i>16.2.2 Homomorphic Equalisation Using a Bank of High-Pass Filters</i>	400
16.3 Equalisation Based on Linear Prediction Models	400
<i>16.3.1 Blind Equalisation Through Model Factorisation</i>	401
16.4 Bayesian Blind Deconvolution and Equalisation	402
<i>16.4.1 Conditional Mean Channel Estimation</i>	403
<i>16.4.2 Maximum-Likelihood Channel Estimation</i>	403
<i>16.4.3 Maximum A Posteriori Channel Estimation</i>	404
<i>16.4.4 Channel Equalisation Based on Hidden Markov Models</i>	404
<i>16.4.5 MAP Channel Estimate Based on HMMs</i>	406
<i>16.4.6 Implementations of HMM-Based Deconvolution</i>	407
16.5 Blind Equalisation for Digital Communication Channels	409
<i>16.5.1 LMS Blind Equalisation</i>	410
<i>16.5.2 Equalisation of a Binary Digital Channel</i>	413
16.6 Equalisation Based on Higher-Order Statistics	414
<i>16.6.1 Higher-Order Moments, Cumulants and Spectra</i>	414
<i>16.6.1.1 Cumulants</i>	415
<i>16.6.1.2 Higher-Order Spectra</i>	416
<i>16.6.2 Higher-Order Spectra of Linear Time-Invariant Systems</i>	416
<i>16.6.3 Blind Equalisation Based on Higher-Order Cepstra</i>	417
<i>16.6.3.1 Bi-Cepstrum</i>	418
<i>16.6.3.2 Tri-Cepstrum</i>	419
<i>16.6.3.3 Calculation of Equaliser Coefficients from the Tri-cepstrum</i>	420
16.7 Summary	420
Bibliography	421
17 Speech Enhancement: Noise Reduction, Bandwidth Extension and Packet Replacement	423
17.1 An Overview of Speech Enhancement in Noise	424
17.2 Single-Input Speech Enhancement Methods	425
<i>17.2.1 Elements of Single-Input Speech Enhancement</i>	425
<i>17.2.1.1 Segmentation and Windowing of Speech Signals</i>	426
<i>17.2.1.2 Spectral Representation of Speech and Noise</i>	426
<i>17.2.1.3 Linear Prediction Model Representation of Speech and Noise</i>	426
<i>17.2.1.4 Inter-Frame and Intra-Frame Correlations</i>	427
<i>17.2.1.5 Speech Estimation Module</i>	427
<i>17.2.1.6 Probability Models of Speech and Noise</i>	427
<i>17.2.1.7 Cost of Error Functions in Speech Estimation</i>	428
<i>17.2.2 Wiener Filter for De-noising Speech</i>	428
<i>17.2.2.1 Wiener Filter Based on Linear Prediction Models</i>	429
<i>17.2.2.2 HMM-Based Wiener Filters</i>	429
<i>17.2.3 Spectral Subtraction of Noise</i>	430
<i>17.2.3.1 Spectral Subtraction Using LP Model Frequency Response</i>	431
<i>17.2.4 Bayesian MMSE Speech Enhancement</i>	432
<i>17.2.5 Kalman Filter for Speech Enhancement</i>	432
<i>17.2.5.1 Kalman State-Space Equations of Signal and Noise Models</i>	433

<i>17.2.6 Speech Enhancement Using LP-HNM Model</i>	435
17.2.6.1 Overview of LP-HNM Enhancement System	436
17.2.6.2 Formant Estimation from Noisy Speech	437
17.2.6.3 Initial-Cleaning of Noisy Speech	437
17.2.6.4 Formant Tracking	437
17.2.6.5 Harmonic Plus Noise Model (HNM) of Speech Excitation	438
17.2.6.6 Fundamental Frequency Estimation	439
17.2.6.7 Estimation of Amplitudes Harmonics of HNM	439
17.2.6.8 Estimation of Noise Component of HNM	440
17.2.6.9 Kalman Smoothing of Trajectories of Formants and Harmonics	440
17.3 Speech Bandwidth Extension—Spectral Extrapolation	442
<i>17.3.1 LP-HNM Model of Speech</i>	443
<i>17.3.2 Extrapolation of Spectral Envelope of LP Model</i>	444
17.3.2.1 Phase Estimation	445
17.3.2.2 Codebook Mapping of the Gain	445
<i>17.3.3 Extrapolation of Spectrum of Excitation of LP Model</i>	446
17.3.3.1 Sensitivity to Pitch	446
17.4 Interpolation of Lost Speech Segments—Packet Loss Concealment	447
<i>17.4.1 Phase Prediction</i>	450
<i>17.4.2 Codebook Mapping</i>	452
17.4.2.1 Evaluation of LP-HNM Interpolation	453
17.5 Multi-Input Speech Enhancement Methods	455
<i>17.5.1 Beam-forming with Microphone Arrays</i>	457
17.5.1.1 Spatial Configuration of Array and The Direction of Reception	458
17.5.1.2 Directional of Arrival (DoA) and Time of Arrival (ToA)	459
17.5.1.3 Steering the Array Direction: Equalisation of the ToAs at the Sensors	459
17.5.1.4 The Frequency Response of a Delay-Sum Beamformer	460
17.6 Speech Distortion Measurements	462
<i>17.6.1 Signal-to-Noise Ratio – SNR</i>	462
<i>17.6.2 Segmental Signal to Noise Ratio – SNR_{seg}</i>	462
<i>17.6.3 Itakura–Saito Distance – ISD</i>	463
<i>17.6.4 Harmonicity Distance – HD</i>	463
<i>17.6.5 Diagnostic Rhyme Test – DRT</i>	463
<i>17.6.6 Mean Opinion Score – MOS</i>	464
<i>17.6.7 Perceptual Evaluation of Speech Quality – PESQ</i>	464
Bibliography	464
18 Multiple-Input Multiple-Output Systems, Independent Component Analysis	467
18.1 Introduction	467
18.2 A note on comparison of beam-forming arrays and ICA	469
18.3 MIMO Signal Propagation and Mixing Models	469
<i>18.3.1 Instantaneous Mixing Models</i>	469
<i>18.3.2 Anechoic, Delay and Attenuation, Mixing Models</i>	470
<i>18.3.3 Convolutional Mixing Models</i>	471
18.4 Independent Component Analysis	472
<i>18.4.1 A Note on Orthogonal, Orthonormal and Independent</i>	473
<i>18.4.2 Statement of ICA Problem</i>	474
<i>18.4.3 Basic Assumptions in Independent Component Analysis</i>	475
<i>18.4.4 The Limitations of Independent Component Analysis</i>	475

<i>18.4.5 Why a mixture of two Gaussian signals cannot be separated?</i>	476
<i>18.4.6 The Difference Between Independent and Uncorrelated</i>	476
<i>18.4.7 Independence Measures; Entropy and Mutual Information</i>	477
18.4.7.1 Differential Entropy	477
18.4.7.2 Maximum Value of Differential Entropy	477
18.4.7.3 Mutual Information	478
18.4.7.4 The Effect of a Linear Transformation on Mutual Information	479
18.4.7.5 Non-Gaussianity as a Measure of Independence	480
18.4.7.6 Negentropy: A measure of Non-Gaussianity and Independence	480
18.4.7.7 Fourth Order Moments – Kurtosis	481
18.4.7.8 Kurtosis-based Contrast Functions – Approximations to Entropic Contrast	481
<i>18.4.8 Super-Gaussian and Sub-Gaussian Distributions</i>	482
<i>18.4.9 Fast-ICA Methods</i>	482
18.4.9.1 Gradient search optimisation method	483
18.4.9.2 Newton optimisation method	483
<i>18.4.10 Fixed-point Fast ICA</i>	483
<i>18.4.11 Contrast Functions and Influence Functions</i>	484
<i>18.4.12 ICA Based on Kurtosis Maximization – Projection Pursuit Gradient Ascent</i>	485
<i>18.4.13 Jade Algorithm – Iterative Diagonalisation of Cumulant Matrices</i>	487
18.5 Summary	490
Bibliography	490
19 Signal Processing in Mobile Communication	491
19.1 Introduction to Cellular Communication	491
<i>19.1.1 A Brief History of Radio Communication</i>	492
<i>19.1.2 Cellular Mobile Phone Concept</i>	493
<i>19.1.3 Outline of a Cellular Communication System</i>	494
19.2 Communication Signal Processing in Mobile Systems	497
19.3 Capacity, Noise, and Spectral Efficiency	498
<i>19.3.1 Spectral Efficiency in Mobile Communication Systems</i>	500
19.4 Multi-path and Fading in Mobile Communication	500
<i>19.4.1 Multi-path Propagation of Electromagnetic Signals</i>	501
<i>19.4.2 Rake Receivers for Multi-path Signals</i>	502
<i>19.4.3 Signal Fading in Mobile Communication Systems</i>	502
<i>19.4.4 Large-Scale Signal Fading</i>	504
<i>19.4.5 Small-Scale Fast Signal Fading</i>	504
19.5 Smart Antennas – Space-Time Signal Processing	505
<i>19.5.1 Switched and Adaptive Smart Antennas</i>	506
<i>19.5.2 Space-Time Signal Processing – Diversity Schemes</i>	506
19.6 Summary	508
Bibliography	508
Index	509