## INDEX

Acoustic Channel Modelling, 4 Acoustic Echo Cancellation, 4, 32 Acoustic Interface, 12 Adaptive Algorithm, 42 Adaptive Combination of Filters, 6, 146, 158 Adaptive Filter, 41 Adaptive Noise Canceller, 161 Affine Combination of Filters, 149 Affine Projection Algorithm, 57, 62, 69, 162 All-Zero Kernel, 178 Artificial Neural Network, 42, 102, 111 Black Box, 110 Block-Based Collaborative FLAF, 183 Blocking Matrix, 161 Channel Diversity, 28

Channel Invertibility, 27

Chebyschev Expansion, 121 Clipping Nonlinearity, 99, 101, 135, 186 Collaborative Adaptive Filtering, 146, 178 Collaborative Filtering Architecture, 178 Collaborative FLAF, 181 Combined Adaptive Filtering, 146, 162 Combined Adaptive Noise Canceller, 161 Computational Intelligence, 19, 42 Convex Combination of Filters, 149, 162, 180 Cover's Theorem, 114 Delay-and-Sum Beamformer, 161 Distant-Talking, vii Distributed Microphone Array, 12 Double Talk, 30, 91

Double Talk Detector, 30, 91 Dynamic Adaptive Filter, 44, 45 Dynamic Nonlinearity, 123 Echo Return Loss Enhancement, 35 EG $\pm$  Algorithm, 62 Enclosure Vibration Nonlinearity, 31, 98 Energy Conservation Relation, 126, 130 Excess Mean Square Error, 168 Far-end Signal, 33 Filter-by-Filter Combined Architecture, 163 FIR Filter, 25 Forgetting Factor, 54 Frequency Selectivity, 26 Functional Link Adaptive Filter, 110, 112, 114 Functional Link Expansion Order, 121 Functional Link Memory Order, 126 Functional Links, 112 General Properties, 66 Generalized Sidelobe Canceller, 158, 161 Grey Box, 110 Hammerstein Nonlinear Model, 111, 112 Hands-free Speech Communications, vii Hardware Limitations, 29 Human-Machine Interaction, 14

IIR Filter, 25 Immersive Communications, vii, 20, 25, 33, 193 Improved PAPA, 70 Improved PNLMS Algorithm, 61, 69, 70 Improved Proportionate NLMS Algorithm, 82 Instantaneous Nonlinearity, 123 Intelligent Acoustic Interface, 3, 14, 193 Intelligent Circuit, 42 Intelligent Switching Circuit, 179 Intermodulation Distortion, 36

Kalman Gain, 57

Learning Algorithm, 42 Least Mean Square Algorithm, 52 Least Perturbation Property, 66 Legendre Expansion, 122 Linear Adaptive Filter, 44 Loudspeaker Nonlinearity, 31, 36, 98, 100, 101, 106

Machine Listening, 22 Mean Square Error, 47 Minimum Mean Square Error, 49 Minimum-Phase System, 27 Mixing Parameter, 149, 163 Multi-Stage Combined Architecture, 165 Multipah Delay Spread, 26 Natural Gradient Adaptation, 67 Near-end Signal, 33 Nonlinear Acoustic Echo Cancellation, 98 Nonlinear Adaptive Filter, 44 Nonlinear System Identification, 112 Nonlinear Transformation, 99 Nonlinearity, 31, 36, 98, 100, 110, 180, 183 Nonstationarity, 30 Normalized Least Mean Square Algorithm, 30, 53, 60, 68, 82 Normalized Misalignment, 37, 188 Normalized Natural Gradient Algorithms, 68 Organized Microphone Array, 12 PNLMS++ Algorithm, 61 Projection Order, 58, 64 Proportionality Factor, 70 Proportionate Adaptive Algorithm, 60, 63 Proportionate Adaptive Algorithms, 28, 59, 69 Proportionate APA, 62, 70, 83 Proportionate Block APA, 72, 85 Proportionate NLMS Algorithm, 61, 69 Recursive Least Squares Algorithm,

30, 54 Regularization Parameter, 54 Reverberation Time, 27 Riccati Equation, 57 Shift-Invariance, 24 Sparse AIR, 28, 60 Spatial Sound Reproduction, 22 Spline Function, 99, 105, 111 Split FLAF, 134, 183, 186 Static Adaptive Filter, 44 Steepest Descent Algorithm, 49 Step Size Parameter, 49, 51, 53 Stochastic Gradient Algorithm, 51 Supervised Adaptive Filter, 43 System-by-System Combined Architecture, 163

Time-Varying System, 25 Trigonometric Expansion, 120

Under-Modelling, 30, 63, 74, 89 Uniform Linear Array, 174 Unsupervised Adaptive Filter, 43

Variable Step Size Parameter, 63, 73 Volterra Filter, 5, 99, 101, 111, 139 VSS Proportionate Algorithms, 73, 88

White Box, 110 Wiener Filter, 45 Wiener Nonlinear Model, 111 Wiener-Hopf Equation, 48