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TABLE OF CONTENTS

Acoustic Echo Cancellation Based on Independent Component Analysis and Integrated Residual Echo Enhancement	1
<i>T. S. Wada, B. H. Juang</i>	
Applications of Signal Analysis Using Autoregressive Models for Amplitude Modulation	5
<i>S. Ganapathy, S. Thomas, P. Motlicek, H. Hermansky</i>	
Coherent Signals Direction-of-Arrival Estimation Using a Spherical Microphone Array: Frequency Smoothing Approach	9
<i>D. Khaykin, B. Rafaely</i>	
Unifying Semantic and Content-Based Approaches for Retrieval of Environmental Sounds	13
<i>G. Wichern, H. Thornburg, A. Spanias</i>	
Dolph-Chebyshev Radial Filter for the Near-Field Spherical Microphone Array	17
<i>E. Fisher, B. Rafaely</i>	
Source Enumeration of Speech Mixtures Using Pitch Harmonics	21
<i>K. D. Gilbert, K. L. Payton</i>	
On Robustness of Multi-Channel Minimum Mean-Squared Error Estimators Under Super-Gaussian Priors	25
<i>R. C. Hendriks, R. Heusdens, J. Jensen</i>	
Statistical Models for Speech Dereverberation	29
<i>T. Yoshioka, H. Kameoka, T. Nakatani, H. G. Okuno</i>	
Acoustic Coupling in Multi-Dimensional Finite Difference Schemes for Physically Modeled Voice Synthesis	33
<i>M. Speed, D. Murphy, D. M. Howard</i>	
ITU-T G.719: A new Low-Complexity Full-Band (20KHZ) Audio Coding Standard for High-Quality Conversational Applications	37
<i>M. Xie, P. Chu, A. Taleb, M. Briand</i>	
Soundfield Rendering with Loudspeaker Arrays Through Multiple Beam Shaping	41
<i>F. Antonacci, A. Calatroni, A. Canclini, A. Galbiati, A. Sarti, S. Tubaro</i>	
A Wiener-Based Implementation of Equalization-Cancellation Pre-Processing for Binaural Speech Intelligibility Prediction	45
<i>N. N. Ellaham, C. Giguere, W. Gueaieb</i>	
HRTF Interpolation in the Wavelet Transform Domain	49
<i>J. C. B. Torres, M. R. Petraglia</i>	
Acoustic Topic Model for Audio Information Retrieval	53
<i>S. Kim, S. Narayanan, S. Sundaram</i>	
A Perceptually Enhanced Scalable-To-Lossless Audio Coding Scheme and a Trellis-Based Approach for its Optimization	57
<i>E. Ravelli, V. Melkote, K. Rose</i>	
Generalized State Coherence Transform for Multidimensional Localization of Multiple Sources	61
<i>F. Nesta, M. Omologo</i>	
Improved a Priori SNR Estimation with Plication in Log-MMSE Speech Estimation	65
<i>N. Hoglund, S. Nordholm</i>	
Spectral HRTF Enhancement for Improved Vertical-Polar Auditory Localization	69
<i>D. S. Brungart, G. D. Romigh</i>	
On the Application of the LCMV Beamformer to Speech Enhancement	73
<i>E. A. P. Habets, J. Benesty, S. Gannor, P. A. Naylor, I. Cohen</i>	
Panoramic Recording and Reproduction of Multichannel Audio Using A Circular Microphone Array	77
<i>H. Hacıhabiboglu, Z. Cvetkovic</i>	
A Zone of Quiet Based Approach to Integrated Active Noise Control and Noise Reduction in Hearing Aids	81
<i>R. Serizel, M. Moonen, J. Wouters, S. H. Jensen</i>	
Towards a Musical Beat Emphasis Function	85
<i>M. E. P. Davies, M. D. Plumbley, D. Eck</i>	
A Spatio-Temporal Power Method for Time-Domain Multi-Channel Speech Enhancement	89
<i>M. Gupta, S. Angrignon, C. Forrester, S. Simmons, S. C. Douglas</i>	
Guided Harmonic Sinusoid Estimation in a Multi-Pitch Environment	93
<i>C. Smit, D. P. W. Ellis</i>	

Domain Decomposition Method for the Digital Waveguide Mesh	97
<i>M. Kim, G. P. Scavone</i>	
Feature Selection for Room Volume Identification From Room Impulse Response	101
<i>N. R. Shabtai, Y. Zigel, B. Rafaely</i>	
Diffuseness Estimation Using Temporal Variation of Intensity Vectors	105
<i>J. Ahonen, V. Pulkki</i>	
A Novel Framework for Recognizing Phonemes of Singing Voice in Polyphonic Music	109
<i>H. Fujihara, M. Goto, H. G. Okuno</i>	
Automatic Gain and Fader Control For Live Mixing	113
<i>E. P. Gonzalez, J. Reiss</i>	
Perfect Sequence LMS for Rapid Acquisition of Continuous-Azimuth Head Related Impulse Responses	117
<i>C. Antweiler, G. Enzner</i>	
Improving Midi-Audio Alignment with Acoustic Features	121
<i>J. Devaney, M. I. Mandel, D. P. W. Ellis</i>	
A Phase-Based Dual Microphone Method to Count and Locate Audio Sources in Reverberant Rooms	125
<i>Z. E. Chami, A. Guerin, A. Pham, C. Serviere</i>	
Synthesis of Guitar by Digital Waveguides: Modeling the Plectrum in the Physical Interaction of the Player with the Instrument	129
<i>F. Germain, G. Evangelista</i>	
Spatial Covariance Models for Under-Determined Reverberant Audio Source Separation	133
<i>N. Q. K. Duong, E. Vincent, R. Gribonval</i>	
Controlling a Spatialized Environmental Sound Synthesizer	137
<i>C. Verron, G. Pallone, M. Aramaki, R. Kronland-Martinet</i>	
Blind Alignment of Asynchronously Recorded Signals for Distributed Microphone Array	141
<i>N. Ono, H. Kohno, N. Ito, S. Sagayama</i>	
The Ideal Interaural Parameter Mask: A Bound on Binaural Separation Systems	145
<i>M. I. Mandel, D. P. W. Ellis</i>	
Note Detection with Dynamic Bayesian Networks as a Postanalysis Step for NMF-Based Multiple Pitch Estimation Techniques	149
<i>S. A. Raczynski, N. Ono, S. Sagayama</i>	
Unsupervised Single -Channel Source Separation Using Bayesian NMF	153
<i>O. Dikmen, A. T. Cemgil</i>	
Realization of Arbitrary Filters in the STFT Domain	157
<i>M. M. Goodwin</i>	
Finding Similar Acoustic Events Using Matching Pursuit and Locality-Sensitive Hashing	161
<i>C. Cotton, D. P. W. Ellis</i>	
Estimating Pressure at Eardrum with Pressure-Velocity Measurement From Ear Canal Entrance	165
<i>M. Hiipakka, M. Karjalainen, V. Pulkki</i>	
IIR QMF-Bank Design For Speech and Audio Subband Coding	169
<i>H. W. Lollmann, M. Hildenbrand, B. Geiser, P. Vary</i>	
Theoretical and Practical Comparisons of the Reassignment Method and the Derivative Method for the Estimation of the Frequency Slope	173
<i>B. Hamilton, P. Depalle, S. Marchand</i>	
Multi-Voice Polyphonic Music Transcription Using Eigeninstruments	177
<i>G. Grindlay, D. P. W. Ellis</i>	
An Investigation of Discrete-State Discriminant Approaches to Single_Sensor Source Separation	181
<i>V. Emiya, E. Vincent, R. Gribonval</i>	
Artifacts in the Sound Field of a Moving Sound Source Reconstructed from a Microphone Array Recording	185
<i>J. Ahrens, S. Spors</i>	
A Probabilistic Speaker Clustering for DOA-Based Diarization	189
<i>K. Ishiguro, T. Yamada, S. Araki, T. Nakatani</i>	
Single-Microphone Wind Noise Reduction by Adaptive Postfiltering	193
<i>E. Nemer, W. Leblanc</i>	
Temporal Quantization of Spatial Information Using Directional Clustering for Multichannel Audio Coding	197
<i>S. Miyabe, K. Masatoki, H. Saruwatari, K. Shikano, T. Nomura</i>	
Parsimonious Sound Field Synthesis Using Compressive Sampling	201
<i>G. N. Lilis, D. Angelosante, G. B. Giannakis</i>	
Regularized HRTF Fitting Using Spherical Harmonics	205
<i>D. N. Zotkin, R. Duraiswami, N. A. Gumerov</i>	

Gain Adaptation Based on Signal-to-Noise Ratio for Noise Suppression	209
<i>D. N. Parikh, S. Ravindran, D. V. Anderson</i>	
Multichannel Voice Activity Detection with Spherically Invariant Sparse Distributions	213
<i>B. Lee, T. Kalker</i>	
Multizone 2D Soundfield Reproduction Via Spatial Band Stop Filters	217
<i>Y. J. Wu, T. D. Abhayapala</i>	
On Optimal Beamforming for Noise Reduction and Interference Rejection	221
<i>M. Souden, J. Benesty, S. Affes</i>	
Towards Co-Channel Speaker Separation by 2-D Demodulation of Spectrograms	225
<i>T. T. Wang, T. F. Quatieri</i>	
Factorial Scaled Hidden Markov Model for Polyphonic Audio Representation and Source Separation	229
<i>A. Ozerov, C. Fevotte, M. Charbit</i>	
Sinewave Parameter Estimation Using the Fast Fan-Chirp Transform	233
<i>R. B. Dunn, T. F. Quatieri, N. Malyska</i>	
Parametric AM/FM Decomposition for Speech and Audio Coding	237
<i>T. Backstrom, S. Disch</i>	
Acoustic Reflection Path Tracing Using A Highly Directional Loudspeaker	241
<i>S. Tervo, J. Patynen, T. Lokki</i>	
Fast Bayesian NMF Algorithms Enforcing Harmonicity and Temporal Continuity in Polyphonic Music Transcription	245
<i>N. Bertin, R. Badeau, E. Vincent</i>	
3D-Continuous-Azimuth Acquisition of Head-Related Impulse Responses Using Multi-Channel Adaptive Filtering	249
<i>G. Enzner</i>	
Robust Audio Precompensation with Probabilistic Modeling of Transfer Function Variability	253
<i>L. J. Brannmark</i>	
Nested Microphone Array Processing for Parameter Estimation in Directional Audio Coding	257
<i>G. D. Galdo, O. Thiergart, F. Kuech</i>	
Robust Spherical Microphone Array Beamforming with Multi-Beam-Multi-Null Steering, and Sidelobe Control	261
<i>H. Sun, S. Yan, U. P. Svensson</i>	
Variable Control of the Pre-Response Error in Mixed Phase Audio Precompensation	265
<i>L. J. Brannmark, A. Ahlen</i>	
Wave Field Analysis Using Multiple RADII Measurements	269
<i>A. Kuntz, R. Rabenstein</i>	
Coherent Spectral Estimation for a Robust Solution of the Permutation Problem	273
<i>F. Nesta, T. S. Wada, B. H. Juang</i>	
On the Non-Uniqueness Problem and the Semi-Blind Source Separation	277
<i>F. Nesta, T. S. Wada, S. Miyabe, B. H. Juang</i>	
Stochastic Particle Filtering: A Fast SRP-PHAT Single Source Localization Algorithm	281
<i>H. Do, H. F. Silverman</i>	
An Auditory-Based Transform for Audio Signal Processing	285
<i>Q. Li</i>	
Separation By "Humming": User-Guided Sound Extraction From Monophonic Mixtures	289
<i>P. Smaragdīs, G. I. Mysore</i>	
Semi-Blind Disjoint Non-Negative Matrix Factorization for Extracting Target Source From Single Channel Noisy Mixture	293
<i>S. Y. Jeong, K. Kim, J. H. Jeong, K. C. Oh</i>	
Computing Predominant Local Periodicity Information in Music Recordings	297
<i>P. Grosche, M. Muller</i>	
Binaural Reproduction for Directional Audio Coding	301
<i>M. V. Laitinen, V. Pulkki</i>	
Chord Recognition Using Measures of Fit, Chord Templates and Filtering Methods	305
<i>L. Oudre, Y. Grenier, C. Fevotte</i>	
Adaptive FIR Filters With Automatic Length Optimization by Monitoring a Normalized Combination Scheme	309
<i>M. Zeller, L. A. Azpicueta-Ruiz, W. Kellermann</i>	
Enhancement of Speech Intelligibility Using Transients Extracted by Wavelet Packets	313
<i>D. M. Rasetshwane, C. C. Li</i>	
Improving Separation of Harmonic Sources with Iterative Estimation of Spatial Cues	317
<i>J. Han, B. Pardo</i>	

Sound Texture Synthesis Via Filter Statistics	321
<i>J. H. McDermott, A. J. Oxenham, E. P. Simoncelli</i>	
A Nonlocally Weighted Soft-Constrained Natural Gradient Algorithm for Blind Separation of Reverberant Speech	325
<i>J. Xin, M. Yu, Y. Qi, H. I. Yang, F. G. Zeng</i>	
Multiple Sound Sources Tracking Method Based on Subspace Tracking	329
<i>N. Ohwada, K. Suyama</i>	
Maximum Directivity Beamformer for Spherical-Aperture Microphones	333
<i>M. Agmon, B. Rafaely, J. Tabrikian</i>	
Two-Stage Binaural Speech Enhancement with Wiener Filter Based on Equalization-Cancellation Model	337
<i>J. Li, S. Sakamoto, S. Hongo, M. Akagi, Y. Suzuki</i>	
An Overall Optimization Method for Arbitrary Sample Rate Converters Based on Integer Rate SRC and Lagrange Interpolation	341
<i>A. Franck, K. Brandenburg</i>	
Coding of Spatio-Temporal Audio Spectra Using Tree-Structured Directional Filterbanks	345
<i>F. Pinto, M. Vetterli</i>	
Dynamic Impulse Response Model For Nonlinear Acoustic System and its Application to Acoustic Echo Canceller	349
<i>S. Saito, A. Nakagawa, Y. Haneda</i>	
Polyphonic Music Transcription Employing Max-Margin Classification of Spectrographic Features	353
<i>R. Gang, M. F. Bocko, D. Headlam, J. Lundberg</i>	
Author Index	