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TABLE OF CONTENTS

Volume 1

Advanced Passive Speaker Protection	1
<i>Scott Dorsey</i>	
A Platform for Audiovisual Telepresence Using Model- and Data-based Wave-field Synthesis	7
<i>Gregor Heinrich, Michael Leitner, Christoph Jung, Fabian Logemann, Volker Hahn</i>	
Clandestine Wireless Development During WWII	21
<i>Jon Paul</i>	
Detecting Changes in Audio Signals by Digital Differencing	40
<i>Bill Waslo</i>	
A Switchmode Power Supply Suitable for Audio Power Amplifiers	49
<i>Jay Gordon</i>	
Simple Arbitrary IIRs	59
<i>Richard Lee</i>	
Absolute Threshold of Coherence of Position Perception between Auditory and Visual Sources for Dialog	69
<i>Roberto Muñoz Soto, Manuel Recuero López, Diego Durán Blanc, Manuel Gazzo</i>	
Individual Subjective Preferences for the Relationship between SPL and Different Cinema Shot Sizes	75
<i>Roberto Muñoz Soto, Manuel Recuero López, Diego Durán Blanc, Manuel Gazzo</i>	
Testing Loudness Models - Real vs. Artificial Content	82
<i>James Johnston</i>	
Target Modes in Moving Assemblies of Compression Drivers and Other Speakers	88
<i>Fernando Bolaños, Pablo Seoane</i>	
Bend Radius	109
<i>Stephen Lampen, Carl Dole, Shulamite Wan</i>	
Sound Extraction of Delackered Records	123
<i>Ottar Johnsen , Frédéric Bapst, Lionel Seydoux</i>	
A Piano Sound Database for Testing Automatic Transcription Records	130
<i>.Luis Ortiz-Berenguer, Elena Blanco-Martin, Alberto Alvarez-Fernandez, Jose A. Blas-Moncalvillo, Francisco J. Casajus-Quiros</i>	
Classification of Musical Genres Using Audio Waveform Descriptors in MPEG-7	133
<i>Nemin Osmanovic</i>	
A Framework for a Near-Optimal Excitation Based Rate-Distortion Algorithm for Audio Coding	146
<i>Mikka Vilermo</i>	
Analysis of Design Parameters for Crosstalk Cancellation Filters Applied to Different Loudspeaker Configurations	150
<i>Yesenia Lacouture Parodi</i>	
An Investigation of 2D Multizone Surround Sound Systems	167
<i>Mark Poletti</i>	
An Improved Distortion Measure for Audio Coding and a Corresponding Two-Layered Trellis Approach for its Optimization	176
<i>Vinay Melkote, Kenneth Rose</i>	
A Digital Model of the Echoplex Tape Delay	190
<i>Steinunn Arnardottir, Jonathan S. Abel, Julius O. Smith</i>	

Measurements of Spaciousness for Stereophonic Music	198
<i>Andy Sarroff, Juan P. Bello</i>	
Effective Car Audio System Enabling Individual Signal Processing Operations of Coincident Multiple Audio Sources through Single Digital Audio Interface Line	210
<i>Chul-Jae Yoo, In-Sik Ryu</i>	
Separation of Singing Voice from Music Accompaniment with Unvoiced Sounds Reconstruction for Monaural Recordings	217
<i>Chao-Ling Hsu , Jyh-Shing Roger Jang , Te-Lu Tsai</i>	
Music Annotation and Retrieval System using Anti-models	223
<i>Zhi-Sheng Chen, Jia-Min Zen, Jyh-Shing Roger Jang</i>	
Efficient Delay Interpolation for Wave Field Synthesis	230
<i>Andreas Franck, Karlheinz Brandenburg, Ulf Richter</i>	
An Initial Validation of Individualised Crosstalk Cancellation Filters for Binaural Perceptual Experiments	247
<i>Alastair H Moore, Anthony I Tew, Rozem Nicol</i>	
Assessing the Acoustic Performance and Potential Intelligibility of Assistive Audio Systems for Hard of Hearing and Other Users	258
<i>Peter Mapp</i>	
Digital Equalization of Automotive Sound Systems Employing Spectral Smoothed Fir Filters	272
<i>Marco Binelli, Angelo Farina</i>	
The Effects of Lossy Audio Encoding on Onset Detection Tasks	281
<i>Kurt Jacobson, Matthew Davies, Mark Sandler</i>	
A New Methodology for the Acoustic Design Of Compression Driver Phase Plugs with Radial Channels	289
<i>Mark Dodd, Jack Oclew-Brown,</i>	
A Hybrid Time and Frequency Domain Audio Pitch Shifting Algorithm	305
<i>Nicolas Juillerat, Stefan Mueller Arisona, Simon Schubiger-Banz</i>	
Spatialized Additive Synthesis of Environmental Sounds	315
<i>Charles Verron, Mitsuko Aramaki, Richard Kronland-Martinet, Gregory Pallone</i>	
Factors Affecting Perception of Audio-video Synchronisation in Television	325
<i>Andrew Mason, Richard Salmon</i>	
Determination and Correction of Individual Channel Time Offsets for Signals Involved in an Audio Mixture	339
<i>Enrique Perez Gonzalez, Joshua Reiss</i>	
QESTRAL (Part 1): Quality Evaluation of Spatial Transmission and Reproduction using an Artificial Listener	349
<i>Francis Rumsey, Slawomir Zielinski, Philip Jackson, Martin Dewhurst, Robert Conetta, Sunish George, Søren Bech, David Meares</i>	
QESTRAL (Part 2): Calibrating the Qestral Model Using Listening Test Data	357
<i>R. Conetta, F. Rumsey, S. Zielinski, P.J.B. Jackson, M. Dewhurst, S. Bech, D. Meare, S. George</i>	
Automatic Verification of Large Sound Reinforcement Systems Using Models of Loudspeaker Performance Data	375
<i>Klas Dalbjörn, Johan Berg</i>	
On the Minimum-Phase Nature of Head-Related Transfer Functions	383
<i>Juhan Nam, Miriam A. Kolar, Jonathan S. Abel</i>	
A Colored Noise Suppressor Using Lattice Filter with Correlation Controlled Algorithm	391
<i>Arata Kawamura, Youji Iguni</i>	

Applications of Algorithmically Generated Digital Audio for Web-Based Sonic Measure Ear Training	397
<i>Christopher Ariza</i>	
Accurate IIR Equalization to an Arbitrary Frequency Response, with Low Delay and Low Noise Real-Time Adjustment	402
<i>Peter Eastty</i>	
Quantifying the Effect of Room Response on Automatic Speech Recognition Systems	418
<i>Jeremy Anderson, John Harris</i>	
Preliminary Results of Calculation of a Sound Field Distribution for the Design of a Sound Field Effector Using a 2-way Loudspeaker Array with Pseudorandom Configuration	423
<i>Yoshihiro Iijima, Kaoru Ashihara, Shogo Kiryu</i>	
A Parametric Instrument Codec for Very Low Bitrates	427
<i>Mirko Arnold, Gerald Schuller</i>	
Non-linear Loudspeaker Unit Modelling	434
<i>Bo Rohde Pedersen, Finn T. Agerkvist</i>	
Audio Bandwidth Extension by Frequency Scaling of Sinusoidal Partial	443
<i>Tomasz ?ernicki, Maciej Bartkowiak</i>	
Audibility of High Q-factor All-pass Components in Head-related Transfer Functions	450
<i>Daniela Toledo, Henrik Møller</i>	
Exploiting Human Spatial Resolution in Surround Sound Decoder Design	463
<i>David Moore, Jonathan Wakefield</i>	
Improvements to a Spherical Binaural Capture Model for Objective Measurement of Spatial Impression with Consideration of Head Movements	472
<i>Chungeun Kim, Russell Mason, Tim Brookes</i>	
Implementation of a Generic Algorithm on Various Automotive Platforms	496
<i>Thomas Esnaultand Jean-Michel Raczinsk</i>	
On the Optimization of Enhanced Cascade	504
<i>Dimitri Danyuk</i>	
Design and Implementation of a Sound Field Effector Using a Loudspeaker Array	512
<i>Seigo Hayashi, Tomoaki Tanno, Toru Kamekawa, Kaoru Ashihara, ShogoKiryu</i>	
A Method of Capacity Increase for Time-Domain Audio Watermarking Based on Low-Frequency Amplitude Modification	516
<i>Harumi Murata , Akio Ogihara , Motoi Iwata, Akira Shiozaki</i>	
A Super-wide-range Microphone with Cardioid Directivity	522
<i>Kazuho Ono, Takehiro Sugimoto, Akio Ando, Tomohiro Nomura, Yutaka Chiba, Keishi Imanaga</i>	
An Investigation of Audio Balance for Elderly Listeners using Loudness as the Main Parameter	531
<i>Tomoyasu Komori, Tohru Takagi, Koichi Kurozumi, Kazuhiro Murakawa</i>	
An Optimized Pair-wise Constant Power Panning Algorithm for Stable Lateral Sound Imagery in the 5.1 Reproduction System	538
<i>Kim Sungyoung, Ikeda Masahiro, Takahashi Akio</i>	
An Active Load and Test Method for Evaluating the Efficiency of Audio Power Amplifiers	545
<i>Harry Dymond, Phil Mellor</i>	
Imperfections and Possible Advances in Analogue Summing Amplifier Design	557
<i>Milan Kovinic, Dragan Drincic, Sasha Jankovic</i>	
Distributed Mechanical Parameters Describing Vibration and Sound Radiation of Loudspeaker Drive Units	561
<i>Wolfgang Klippel, Joachim Schlechter</i>	

An Evaluation of Pre-Processing Algorithms for Rhythmic Pattern Analysis	581
<i>Matthias Gruhne, Christian Dittmar, Daniel Gaertner , Gerald Schuller</i>	
Graceful Degradation for Digital Radio Mondiale (DRM)	589
<i>Ferenc Kraemer, Gerald Schuller</i>	
A Psycho-Acoustic Measurement and ABR for Sound Signals in the Frequency Range between 10 kHz and 24 kHz	596
<i>Mizuki Omata, Kaoru Ashihara, Motoki Koubori, Yoshitaka Moriya, Masaki Kyoso, Shogo Kiryu</i>	
Magnetic Development: Magneto-Optical Indicator Film Imaging v. Ferrofluids	601
<i>Jonathan Broyles</i>	
Advanced Audio Algorithms for a Real Automotive Digital Audio System	613
<i>Stefania Cecchi, Lorenzo Palestini, Paolo Peretti , Emanuele Moretti , Francesco Piazza , Ariano, Lattanzi , Ferruccio Bettarelli</i>	
Real Time Implementation of an ESPRIT-based Bass Enhancement Algorithm	621
<i>Lorenzo Palestini, Emanuele Moretti, Paolo Peretti, Stefania Cecchi, Laura Romoli, Francesco Piazza</i>	
Ambience Sound Recording Utilizing Dual MS (Mid-Side) Microphone Systems Based Upon Frequency Dependent Spatial Cross Correlation (FSCC) [Part-2 : Acquisition of On-stage Sounds]	629
<i>Teruo Muraoka, Takahiro Miura, Tohru Ifukube,</i>	
Robustness Issues in Multi-View Audio Coding	641
<i>Mauri Väänänen</i>	
Methods and Limitations of Line Source Simulation	648
<i>Stefan Feistel, Ambrose Thompson, Wolfgang Ahnert</i>	
Loudspeaker Directivity Improvement Using Low Pass and All Pass Filters	698
<i>Charles Hughes</i>	
An Audio Reproduction Grand Challenge: Design a System to Sonic Boom an Entire House	709
<i>Victor W. Sparrow , Steven L. Garrett</i>	
Is My Decoder Ambisonic?	719
<i>Aaron J. Heller, Richard Lee, Eric M. Benjamin</i>	

Volume 2

Wave Field Synthesis: Practical Implementation and Application to Sound Beam Digital Pointing	741
<i>Paolo Peretti, Laura Romoli, Lorenzo Palestini, Stefania Cecchi, Francesco Piazza</i>	
An Unintrusive Objective Model for Predicting the Sensation of Envelopment Arising from Surround Sound Recordings	753
<i>Sunish George, Slawomir Zielinski, Francis Rumsey, Robert Conetta, Martin Dewhirst, Philip Jackson, David Meares, Søren Bech</i>	
On the Necessary Delay for the Design of Causal and Stable Recursive Inverse Filters for Loudspeaker Equalization	762
<i>Avelino Marques, Diamantino Freitas</i>	
Quantifying the Strategy Taken by a Pair of Ensemble Hand-clappers Under the Influence of Delay	770
<i>Nima Darabi, Peter Svensson, Snorre Farnør</i>	
“Roughometer”: Realtime Roughness Calculation and Profiling	777
<i>Juli’an Villegas, Michael Cohen</i>	
Quality Improvement of Very Low Bitrate HE-ACC Using Linear Prediction Module	787
<i>Gun-Woo Lee, Jae-sung Lee, Young-Cheol Park, Dae-Hee Youn</i>	
Research on a Measuring Method of Headphones and Earphones Using HATS	795
<i>Kiyofumi Inanaga, Takeshi Hara, Gunnar Rasmussen, Yasuhiro Riko</i>	

A Framework for Producing Rich Musical Metadata in Creative Music Production	805
<i>Gyorgy Fazekas, Yves Raimond, Mark Sandler,</i>	
The Incongruency Advantage for Sounds in Natural Scenes	813
<i>Brian Gygi, Valeriy Shafiro</i>	
SoundTorch: Quick Browsing in Large Audio Collections	819
<i>Sebastian Heise, Michael Hlatky, Jorn Loviscach</i>	
An Objective Method of Measuring Subjective Click-and-Pop Performance for Audio Amplifiers	827
<i>Kymerly Christman Schmidt</i>	
Highly Focused Sound Beamforming Algorithm Using Loudspeaker Array System	837
<i>Yoomi Hur, Seong-Woo Kim, Young-cheol Park, Dae Hee Youn</i>	
A New Audio Post-production Tool for Speech Dereverberation	843
<i>Keisuke Kinoshita, Tomohiro Nakatani, Masato Miyoshi, Toshiyuki Kubota</i>	
Apparatus Comparison for the Characterization of Spaces	848
<i>Adam Kestian, Agnieszka Roginska</i>	
Determining Manufacture Variation in Loudspeakers Through Measurement of Thiele/Small Parameters	854
<i>Scott Laurin, Karl Reichard</i>	
About Phase Optimization in Multitone Excitations	858
<i>Delphine Bard, Vincent Meyer</i>	
Viscous Friction and Temperature Stability of the Mid-high Frequency Loudspeaker	862
<i>Ivan Djurek, Antonio Petosic, Danijel Djurek</i>	
Calorimetric Evaluation of Intrinsic Friction in the Loudspeaker Membrane	866
<i>Antonio Petošić, Ivan Djurek, Danijel Djurek</i>	
Loudness Descriptors to Characterize Programs and Music Tracks	870
<i>Esben Skovenborg, Thomas Lund</i>	
Super-directive Loudspeaker Array for the Generation of Personal Sound Zone	879
<i>Jung-Woo Choi, Youngtae Kim, Sangchul Ko, Jungho Kim</i>	
Quantitative and Qualitative Evaluations for Tv Advertisements Relative to the Adjacent Programs	889
<i>Eiichi Miyasaka, Akiko Kimura</i>	
An Ironless Low Frequency Subwoofer Functioning Under Its Resonance Frequency	895
<i>Benoit Merit, Guy Lemarquand, Bernard Nemoff</i>	
An Efficient Method to Generate Particle Sounds in Wave Field Synthesis	899
<i>Michael Beckinger, Sandra Brix</i>	
Low-Power Implementation of a Subband Acoustic Echo Canceller for Portable Devices	906
<i>Julie Johnson, David Hermann, John Wdowiak, Edward Chau, Hamid Sheikhzadeh</i>	
Obtaining Binaural Room Impulse Responses from B-Format Impulse Responses	912
<i>Fritz Menzer, Christof Faller</i>	
Audibility of Phase Response Differences in a Stereo Playback System. Part 2: Narrow-band Stimuli in Headphones and Loudspeakers	920
<i>Sylvain Choisel, Geoff Martin</i>	
Microphone Front-Ends for Spatial Audio Coders	928
<i>Christof Faller</i>	
Time Variance of the Suspension Nonlinearity	938
<i>Finn Agerkvist, Bo R. Pedersen</i>	

Head-related Transfer Function Reconstruction from Sparse Measurements Considering a Priori Knowledge from Database Analysis: a Pattern Recognition Approach	947
<i>Pierre Guillon, Rozenn Nicol</i>	
A Study of the Creep Effect in Loudspeaker Suspension	963
<i>Finn Agerkvist, Knud Thorborg, Carsten Tinggaard</i>	
Line Arrays with Controllable Directional Characteristics – Theory and Practice	973
<i>Laurie Fincham, Peter Brown</i>	
Harmonic Sinusoidal+Noise Modeling of Audio based on Multiple F0 Estimation	988
<i>Maciej Bartkowiak, Tomasz ?ernicki</i>	
File System Tricks for Audio Production	995
<i>Michael Hlatky, Sebastian Heise, J orn Loviscach</i>	
QESTRAL (Part 3): System and Metrics for Spatial Quality Prediction	1005
<i>Martin Dewhirst, Robert Conetta, Francis Rumsey, Philip Jackson, Slawomir Zielinski, Sunish George, Søren Bech, David Meares</i>	
MPEG-4 Enhanced Low Delay AAC - A New Standard for High Quality Communication	1014
<i>Markus Schnell, Markus Schmidt, Manuel Jander, Tobias Albert, Ralf Geiger, Vesa Ruoppila, Per Ekstrand, Manfred Lutzky, Bernhard Grill</i>	
Head-related Transfer Function Customization by Frequency Scaling and Rotation Shift Based on a New Morphological Matching Method	1028
<i>Pierre Guillon, Thomas Guignard , Rozenn Nicol</i>	
Loudspeaker Production Variance	1042
<i>Steve Hutt, Laurie Fincham,</i>	
Speech Enhancement of Movie Sound	1052
<i>Christian Uhle, Oliver Hellmuth, Jan Weigel</i>	
STFT-Domain Estimation of Subband Correlations	1067
<i>Michael Goodwin</i>	
The Use of Delay Control for Stereophonic Audio Rendering Based on VBAP	1079
<i>Dongil Hyun, Tacksung Choi, Daehee Youn, Seokpil Lee, Youngcheol Park</i>	
QESTRAL (Part 4): Test Signals, Combining Metrics and the Prediction of Overall Spatial Quality	1085
<i>Martin Dewhirst, Robert Conetta, Francis Rumsey, Philip Jackson, Slawomir Zielinski, Sunish George, Søren Bech, David Meares</i>	
SMART-I^2 “Spatial Multi-user Audio-visual Real-Time Interactive Interface”	1093
<i>Marc R’ebillat, Etienne Corteel, Brian F.G. Katz</i>	
Accuracy Issues in Finite Element Simulation of Loudspeakers	1109
<i>Patrick Macey</i>	
Can One Perform Quasi-anechoic Loudspeaker Measurements in Normal Rooms?	1115
<i>John Vanderkooy, Stanley P. Lipshitz</i>	
Confirmation of Chaos in a Loudspeaker System Using Time Series Analysis	1126
<i>Joshua D. Reiss, Ivan Djurek, Antonio Petosic, Danijel Djurek</i>	
Extraction of Electric Network Frequency Signals from Recordings Made in a Controlled Magnetic Field	1143
<i>Richard W. Sanders, Peter S. Popolo</i>	
Spatial Audio Scene Coding	1151
<i>Michael M. Goodwin, Jean-Marc Jot</i>	
Reproduction of Virtual Sound Sources Moving at Supersonic Speeds in Wave Field Synthesis	1159
<i>Jens Ahrens, Sascha Spors</i>	
Near-Field Compensation for HRTF Processing	1170
<i>David Romblo, Bryan Cook</i>	

A Comparison of Wave Field Synthesis and Higher-Order Ambisonics with Respect to Physical Properties and Spatial Sampling	1176
<i>Jens Ahrens, Sascha Spors</i>	
Ambisonic Loudspeaker Arrays	1193
<i>Eric Benjamin</i>	
Forensic Voice Identification Utilizing Digitally Extracted Speech Characteristics	1214
<i>Jeff M. Smith, Richard W. Sanders</i>	
Optimal Modal Spacing and Density for Critical Listening	1222
<i>Bruno Fazenda, Matthew Wankling</i>	
A Perceptual Model Based Speech Enhancement Algorithm	1236
<i>Rongshan Yu</i>	
The Illusion of Continuity Revisited on Filling Gaps in the Saxophone Sound	1248
<i>Piotr Kleczkowski</i>	
Predicting Perceived Off-center Sound Degradation in Surround Loudspeaker Setups for Various Multichannel Microphone Techniques	1254
<i>Nils Peters, Bruno L. Giordano, Sungyoung Kim, Jonas Braasch, Stephen McAdams</i>	
Application of Multichannel Impulse Response Measurement to Automotive Audio	1266
<i>Michael J. Strauß, Diemer de Vries</i>	
Optimum Placement for Small Desktop/PC Loudspeakers	1272
<i>Vladimir E. Filevski</i>	
Estimating the Transfer Function from Air Conduction Recording to One's Own Hearing	1277
<i>Sook Young Won, Jonathan Berger</i>	
A Digital Reverberator Modeled after the Scattering of Acoustic Waves by Trees in a Forest	1284
<i>Kyle Spratt, Jonathan S. Abel</i>	
An Implementation of MPEG-4 ALS Standard Compliant Decoder on ARM Core CPUs	1294
<i>Noboru Harada, Takehiro Moriya, Yutaka Kamamoto</i>	
Stereo ACC Real-time Audio Communication	1300
<i>Aníbal Ferreira, Filipe Abreu, Deepen Sinha</i>	
Methods for Identification of the Tuning System in Audio Musical Signals	1309
<i>Peyman Heydarian, Lewis Jones, Allan Seago</i>	
Surround System Based on Three Dimensional Sound Field Reconstruction	1315
<i>Filippo M. Fazi, Philip A. Nelson, Jens E. N. Christensen, Jeongil Seo</i>	
Rapid Learning of Subjective Preference in Equalization	1336
<i>Andrew T. Sabin, Bryan Pardo</i>	
A Method for Estimating Interaural Time Difference for Binaural Synthesis	1343
<i>Juhan Nam, Jonathan S. Abel, Julius O. Smith II</i>	
In Situ Determination of Acoustic Absorption Coefficients	1350
<i>Scott Mallais</i>	
Multi-Channel Low Frequency Room Simulation with Properly Modeled Source Terms – Multiple Equalization Comparison	1360
<i>Ryan J. Matheson</i>	
Efficient Detection of Exact Redundancies in Audio Signals	1368
<i>José R. Zapata G. and Ricardo A. Garcia</i>	
Parametric Interpolation of Gaps in Audio Signals	1375
<i>Alexey Lukin, Jeremy Todd</i>	
Constrained-Optimized Sound Beamforming of Speaker-array System	1381
<i>Myung-suk Song, Soonho Baek, Seok-pil Lee, Hong-goo Kang</i>	

The Influence of Acoustic Environment on the Threshold of Audibility of Loudspeaker Resonances	1388
<i>Shelley Uprichard , Sylvain Choisel</i>	
Reverberation Echo Density Psychoacoustics	1399
<i>Patty Huang, Jonathan S. Abel, Hiroko Terasawa, Jonathan Berger</i>	
Low Latency Convolution In One Dimension Via Two Dimensional Convolutions	1409
<i>Jeffrey R. Hurchalla</i>	
Two-Channel Matrix Surround Encoding for Flexible Interactive 3-D Audio Reproduction	1420
<i>Jean-Marc Jot</i>	
Mechanical Properties of Ferrofluids in Loudspeakers	1433
<i>Guy Lemarquand, Romain Ravaud, Valerie Lemarquand and Claude Depollier</i>	
Phantom Powering the Modern Condenser Microphone: A Practical Look at Conditions for Optimized Performance	1437
<i>Mark Zaim, Tadashi Kikutani, Jackie Green</i>	
Aging and Sound Perception: Desirable Characteristics of Entertainment Audio for the Elderly	1451
<i>Hannes Müsch</i>	

Author Index