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# TABLE OF CONTENTS

## VOLUME 1

<b>Bandwidth Extension for Microphone Arrays</b> .....	1
<i>Benjamin Bernschütz</i>	
<b>Period Deviation Tolerance Templates: A Novel Approach to Evaluation and Specification of Self-synchronizing Audio Converter</b> .....	11
<i>Francis Legray, Thierry Heeb, Sebastien Genevey, Hugo Kuo</i>	
<b>Distance Learning Strategies for Sound Recording Technology</b> .....	23
<i>Duncan Williams</i>	
<b>Further Testing and Newer Methods in Evaluating Amplifiers for Induced Phase and Frequency Modulation via Tones, Amplitude Modulated Signals, and Pulsed Waveforms</b> .....	28
<i>Ronald Quan</i>	
<b>Modeling the Large Signal Behavior of Micro-Speakers</b> .....	58
<i>Wolfgang Klippel</i>	
<b>Overview and Comparison of and Guide to Audio Measurement Methods</b> .....	68
<i>Gregor Schmidle, Danilo Zanatta</i>	
<b>Consistency of Balance Preferences in Three Musical Genres</b> .....	76
<i>Richard King, Brett Leonard, Grzegorz Sikora</i>	
<b>Drum Pattern Humanisation Using a Recursive Bayesian Framework</b> .....	82
<i>Ryan Stables, Cham Athwal, Rob Cade</i>	
<b>On the Study of Frontal-emitter Headphone to Improve 3D Audio Playback</b> .....	89
<i>Kaushik Sunder, Ee-Leng Tan, Woon-Seng Gan</i>	
<b>Investigating the Benefit of Silicon Carbide for a Class D Power Stage</b> .....	99
<i>Verena Grifone Fuchs, Carsten Wegner, Sebastian Neuser, Dietmar Ehrhardt</i>	
<b>Real-time Implementation of Glasberg &amp; Moore's Loudness Model for Time-Varying Sounds</b> .....	107
<i>Elvira Burdiel, Lasse Vetter, Andrew J. R. Simpson, Michael J. Terrell, Andrew McPherson, Mark Sandler</i>	
<b>Frequency-Domain Implementation of Time-Varying FIR Filters</b> .....	116
<i>Earl Vickers</i>	
<b>Flexible Acoustic Transducer from Dielectric-Compound Elastomer Film</b> .....	125
<i>Takehiro Sugimoto, Kazuho Ono, Akio Ando, Hiroyuki Okubo, Kentaro Nakamura</i>	
<b>Coupling Lumped and Boundary Element Methods Using Superposition</b> .....	132
<i>Joerg Panzer</i>	
<b>Spherical Sound Source for Acoustic Measurements</b> .....	141
<i>Plamen Valtchev, Dimitar Dimitrov, Rumén Artarski</i>	
<b>A Tilt Filter in a Servo Loop</b> .....	153
<i>John Lazzaro, John Wawrzynek</i>	
<b>Does Understanding of Test Items Help or Hinder Subjective Assessment of Basic Audio Quality?</b> .....	167
<i>Nadja Schinkel-Bielefeld, Netaya Lotze, Frederik Nagel</i>	
<b>A Model for Rendering Stereo Signals in the ITD-Range of Hearing</b> .....	178
<i>Siegfried Linkwitz</i>	
<b>Estimating a Signal from a Magnitude Spectrogram Via Convex Optimization</b> .....	191
<i>Dennis L. Sun, Julius O. Smith III</i>	
<b>A Downmix Approach with Acoustic Shadow, Low Frequency Effects and Loudness Control</b> .....	199
<i>Regis Faria, José Augusto Mannis</i>	
<b>Drum Synthesis via Low-Frequency Parametric Modes and Altered Residuals</b> .....	208
<i>Haiying Xia, Julius O. Smith</i>	
<b>Perception of Time-varying signals: Timbre and Phonetic JND of diphthong</b> .....	216
<i>S. Arthi, T.V. Sreenivas</i>	
<b>Comparison of Filter Bank Design Algorithms for use in Low Delay Audio Coding</b> .....	224
<i>Stephan Preihs, Thomas Krause, Jorn Ostermann</i>	
<b>High-Level Semantic Metadata for the Control of Multitrack Adaptive Digital Audio Effects</b> .....	232
<i>Thomas Wilmering, Gyorgy Fazekas, Mark Sandler</i>	
<b>Improved Methods for Generating Focused Sources Using Circular Arrays</b> .....	243
<i>M. Poletti</i>	
<b>Subjective Evaluation of Personalized Equalization Curves in Music</b> .....	253
<i>Weidong Shen, Tiffany Chua, Kelly Reavis, Hongmei Xia, Duo Zhang, Gerald A. Maguire, David Franklin, Vincent Liu, Wei Hou, Hung Tran</i>	

<b>On the use of Dynamically Varied Loudspeaker Spacing in Wave Field Synthesis .....</b>	<b>261</b>
<i>Rishabh Ranjan, Woon-Seng Gan</i>	
<b>Thresholds for the Discrimination of Tonal and Narrowband Noise Bursts .....</b>	<b>271</b>
<i>Armin Taghipour, Bernd Edler, Masoumeh Amirpour, Jurgen Herre</i>	
<b>Loudspeaker Localization Based on Audio Watermarking .....</b>	<b>278</b>
<i>Florian Kolbeck, Giovanni Del Galdo, Iwona Sobieraj, Tobias Bliem</i>	
<b>Low-frequency Temporal Accuracy of Smallroom Sound Reproduction.....</b>	<b>287</b>
<i>Adam J. Hill, Malcolm O.J. Hawksford</i>	
<b>Limitations of Point-Source Sub-woofer Array Models for Live Sound .....</b>	<b>300</b>
<i>Ambrose Thompson, Josebaitor Luzarraga Iturrioz, Phil Anthony</i>	
<b>Efficiency Optimization of Class G Amplifiers: Impact of the Input Signals.....</b>	<b>315</b>
<i>Patrice Russo, Gael Pillonnet, Nacer Abouchi, Sophie Taupin, Frederic Goutti</i>	
<b>A Low-Voltage Low-Power Output Stage for Class-G Headphone Amplifiers.....</b>	<b>323</b>
<i>Alexandre Hufpenus</i>	
<b>Simulating Autophony with Auralized Oral-binaural Room Impulse Responses .....</b>	<b>335</b>
<i>Manuj Yadav, Luis Miranda, Densil Cabrera, William L. Martens</i>	
<b>Objective and Subjective Evaluations of Digital Audio Workstation Summing .....</b>	<b>343</b>
<i>Brett Leonard, Scott Levine, Padraig Buttner-Schnirer</i>	
<b>Service Discovery Using Open Sound Control .....</b>	<b>348</b>
<i>Andrew Eales, Richard Foss</i>	
<b>A Viewer-Centered Revision of Audiovisual Content Classifiers .....</b>	<b>355</b>
<i>Katrien De Moor, Ulrich Reiter</i>	
<b>Comparing Separation Quality of Nonnegative Matrix Factorization and Nonnegative Matrix Factor 2D Deconvolution in Audio Source Separation Tasks .....</b>	<b>363</b>
<i>Julian M. Becker, Volker Gnann</i>	
<b>Balanced Phase Equalization; IIR Filters with Independent Frequency Response and Identical Phase Response .....</b>	<b>371</b>
<i>Peter Eastty</i>	
<b>Utilizing Instantaneous Direct-to-Reverberant Ratio in Parametric Spatial Audio Coding .....</b>	<b>380</b>
<i>Mikko-Ville Laitinen, Ville Pulkki</i>	
<b>Audio Signal Decorrelation Based on Reciprocal-Maximal Length Sequence Filters and Its Applications to Spatial Sound.....</b>	<b>390</b>
<i>Bo-sun Xie, Bei Shi, Ning Xiang</i>	
<b>Approximation of Dynamic Convolution Exploiting Principal Component Analysis: Objective and Subjective Quality Evaluation .....</b>	<b>396</b>
<i>Andrea Primavera, Stefania Cecchi, Laura Romoli, Michele Gasparini, Francesco Piazza</i>	
<b>Kinect Application for a Wave Field Synthesis Based Reproduction System.....</b>	<b>405</b>
<i>Michele Gasparini, Stefania Cecchi, Laura Romoli, Andrea Primavera, Paolo Peretti, Francesco Piazza</i>	
<b>Optimized Implementation of an Innovative Digital Audio Equalizer.....</b>	<b>413</b>
<i>Marco Virgulti, Stefania Cecchi, Andrea Primavera, Laura Romoli, Emanuele Ciavattini, Ferruccio Bettarelli, Francesco Piazza</i>	
<b>The Effect of Acoustic Environment on Reverberation Level Preference .....</b>	<b>421</b>
<i>Brett Leonard, Richard King, Grzegorz Sikora</i>	
<b>Low Complexity Transient Detection in Audio Coding Using an Image Edge Detection Approach.....</b>	<b>426</b>
<i>Julien Capobianco, Gregory Pallone, Laurent Daudet</i>	
<b>Digitally Driven Speaker System Using Direct Digital Spread Spectrum Technology to Reduce EMI Noise.....</b>	<b>436</b>
<i>Masayuki Yashiro, Mitsuhiro Iwaide, Akira Yasuda, Michitaka Yoshino, Kazuyuki Yokota, Yugo Moriyasu, Kenji Sakuda, Fumiaki Nakashima</i>	
<b>Analysis on Multiple Scattering Between the Rigid-spherical Microphone Array and Nearby Surface in Sound Field Recording.....</b>	<b>442</b>
<i>Guangzheng Yu, Bosun Xie, Yu Liu</i>	
<b>Multi-track Mixing Using a Model of Loudness and Partial Loudness .....</b>	<b>448</b>
<i>Dominic Ward, Joshua D. Reiss, Cham Athwal</i>	
<b>Long Distance Induction Drive Loud Hailer Characterization .....</b>	<b>457</b>
<i>Marshall Buck, David Graebener, Ron Sauro</i>	
<b>Scalable Binaural Synthesis on Mobile Devices .....</b>	<b>466</b>
<i>Christian Sander, Frank Wefers, Dieter Leckschat</i>	
<b>Computer Aided Estimation of the Athenian Agora Aulos Scales Based on Physical Modeling .....</b>	<b>474</b>
<i>Arati Andreopoulou, Agnieszka Roginska</i>	

<b>A Simple and Efficient Method for Real-time Computation and Transformation of Spherical Harmonic Based Sound Fields</b> .....	484
<i>Robert E. Davis, D. Fraser Clark</i>	
<b>Speech Separation with Microphone Arrays Using the Mean Shift Algorithm</b> .....	494
<i>David Ayllon, Roberto Gil-Pita, Manuel Rosa-Zurera</i>	
<b>The Radiation Characteristics of a Horizontally Asymmetrical Waveguide That Utilizes a Continuous Arc Diffraction Slot</b> .....	500
<i>Soichiro Hayashi, Akira Mochimaru, Paul Fidlin</i>	
<b>Higher Order Loudspeakers for Improved Surround Sound Reproduction in Rooms</b> .....	513
<i>M. Poletti, Terence Betlehem, Thushara Abhayapala</i>	
<b>Switching/Linear Hybrid Audio Power Amplifiers for Domestic Applications, Part 2: The Class-B+D Amplifier</b> .....	526
<i>Harry Dymond, Phil Mellor</i>	
<b>A Method for Reproducing Frontal Sound Field of 22.2 Multichannel Sound utilizing a Loudspeaker Array Frame</b> .....	543
<i>Hiroyuki Okubo, Takehiro Sugimoto, Satoshi Oishi, Akio Ando</i>	
<b>A Mixing Matrix Estimation Method for Blind Source Separation of Underdetermined Audio Mixture</b> .....	551
<i>Mingu Lee, Koeng-Mo Sung</i>	
<b>A Computational Acoustic Model of the Coupled Interior Architecture of Ancient Chavin</b> .....	561
<i>Regina E. Collecchia, Miriam A. Kolar, Jonathan Abel</i>	
<b>Direct-Diffuse Decomposition of Multichannel Signals Using a System of Pairwise Correlations</b> .....	573
<i>Jeffrey Thompson, Brandon Smith, Aaron Warner, Jean-Marc Jot</i>	
<b>Generating Matrix Coefficients for Feedback Delay Networks Using Genetic Algorithm</b> .....	588
<i>Michael Chemistruck, Kyle Marcolini, Will Pirkle</i>	
<b>Spatial Sound Design Tool for 22.2 Channel 3D Audio Productions, with Height</b> .....	594
<i>Wieslaw Woszczyk, Brett Leonard, David Benson</i>	
<b>Envelope-Based Spatial Parameter Estimation in Directional Audio Coding</b> .....	604
<i>Michael Kratschmer, Oliver Thiergart, Ville Pulkki</i>	

## VOLUME 2

<b>Preferred Spatial Post-Processing of Popular Stereophonic Music for Headphone Reproduction</b> .....	613
<i>Ella Manor, William L. Martens, Densil Cabrera</i>	
<b>The Influence of 2D and 3D Video Playback on the Perceived Quality of Spatial Audio Rendering for Headphones</b> .....	627
<i>Amir Iljazovic, Florian Leschka, Bernhard Neugebauer, Jan Plogsties</i>	
<b>The Dynamic Redistribution of Spectral Energies for Upmixing and Re-animation of Recorded Audio</b> .....	637
<i>Christopher J. Keyes</i>	
<b>A New Method of Multi-Channel Surround Sound Coding Utilizing Head Dynamic Cue</b> .....	646
<i>Qinghua Ye, Lingling Zhang, Hefei Yang, Xiaodong Li</i>	
<b>Midrange Resonant Scattering in Loudspeakers</b> .....	653
<i>Juha Backman</i>	
<b>Temporal Coherence-based Howling Detection for Speech Applications</b> .....	667
<i>Chengshi Zheng, Hao Liu, Renhua Peng, Xiaodong Li</i>	
<b>Study of the Interaction Between Radiating Systems in a Coaxial Loudspeaker</b> .....	675
<i>A. Espí, W.A. Cárdenas, J. Martínez, J. Ramis, J. Carbajo</i>	
<b>HRIR Database with Measured Actual Source Direction Data</b> .....	686
<i>Javier Gomez Bolanos, Ville Pulkki</i>	
<b>Potential of Non-uniformly Partitioned Convolution with Freely Adaptable FFT Sizes</b> .....	694
<i>Frank Wefers, Michael Vorländer</i>	
<b>Wave Field Synthesis with Primary Source Correction: Theory, Simulation Results, and Comparison to Earlier Approaches</b> .....	702
<i>Florian Volk, Hugo Fastl</i>	
<b>Headphone Virtualisation: Improved Localisation and Externalisation of Nonindividualised HRTFs by Cluster Analysis</b> .....	722
<i>Robert P. Tame, Daniele Barchiesi, Anssi Klapuri</i>	
<b>Measurement of Harmonic Distortion Audibility Using a Simplified Psychoacoustic Model</b> .....	728
<i>Steve Temme, Pascal Brunet, Parastoo Qarabaqi</i>	

<b>Procedural Audio Modeling for Particle-based Environmental Effects</b> .....	737
<i>Charles Verron, George Drettakis</i>	
<b>DReaM: A Novel System for Joint Source Separation and Multi-track Coding</b> .....	749
<i>Sylvain Marchand, Roland Badeau, Cleo Baras, Laurent Daudet, Dominique Fourer, Laurent Girin, Stanislaw Gorlow, Antoine Liutkus, Jonathan Pinel, Gael Richard, Nicolas Sturmel, Shuhua Zang</i>	
<b>Experiments of Sound Field Reproduction Inside Aircraft Cabin Mock-up</b> .....	759
<i>Philippe-Aubert Gauthier, Cedric Camier, F.-A. Lebel, Y. Pasco, Alain Berry</i>	
<b>New Strategy to Minimize Dead-Time Distortion in DCI-NPC Power Amplifiers Using COE-Error Injection</b> .....	777
<i>T. Resano, V. Sala, J.L. Romeral, Jose Manuel Moreno</i>	
<b>Evaluation of <math>t_{tr}</math> Distorting Effects Reduction in DCI-NPC Multilevel Power Amplifiers by Using SiC Diodes and MOSFET Technologies</b> .....	785
<i>V. Sala, T. Resano, L. Romeral, M. Moreno</i>	
<b>An Autonomous System for Multi-track Stereo Pan Positioning</b> .....	792
<i>Stuart Mansbridge, Saoirse Finn, Joshua D. Reiss</i>	
<b>Identification and Evaluation of Target Curves for Headphones</b> .....	804
<i>Felix Fleischmann, Andreas Silzle, Jan Plogsties</i>	
<b>Interactive 3D Audio: Enhancing Awareness of Details in Immersive Soundscapes?</b> .....	815
<i>Mikkel N. Schmidt, Stephen Schwartz, Jan Larsen</i>	
<b>Localization of a virtual point source within the listening area for Wave Field Synthesis</b> .....	822
<i>Hagen Wierstorf, Alexander Raake, Sascha Spors</i>	
<b>Knowledge Representation Issues in Audio Related Metadata Model Design</b> .....	831
<i>Gyorgy Fazekas, Mark Sandler</i>	
<b>Audio Latency Masking in Music Telepresence Using Artificial Reverberation</b> .....	843
<i>Ren Gang, Samarth H. Shivaswamy, Stephen Roessner, Akshay Rao, Dave Headlam, Mark Bocko</i>	
<b>Teaching Audio Processing Software Development in the Web Browser</b> .....	849
<i>Matthias Birkenstock, Jorn Loviscach</i>	
<b>A Study on Correlation Between Tempo and Mood of Music</b> .....	855
<i>Magdalena Plewa, Bozena Kostek</i>	
<b>Parametric Coding of Piano Signals</b> .....	863
<i>Michael Schnabel, Benjamin Schubert, Gerald Schuller</i>	
<b>Subjective Selection of Head-Related Transfer Functions (HRTFs) based on Spectral Coloration and Interaural Time Differences (ITD) Cues</b> .....	872
<i>Kyla McMullen, Agnieszka Roginska, Gregory H. Wakefield</i>	
<b>The Relationship between Perception and Measurement of Headphone Sound Quality</b> .....	881
<i>Sean E. Olive, Todd Welti</i>	
<b>Searching Impulse Response libraries using Room Acoustic Descriptors</b> .....	898
<i>David Benson, Wieslaw Woszczyk</i>	
<b>Employing Supercomputing Cluster to the Acoustic Noise Map Creation</b> .....	909
<i>Andrzej Czyzewski, Jozef Kotus, Maciej Szczodrak, Bozena Kostek</i>	
<b>Head-Tracking Techniques for Virtual Acoustics Applications</b> .....	916
<i>Wolfgang Hess</i>	
<b>A Method for Enhancement of Background Sounds in Forensic Audio Recordings</b> .....	931
<i>Robert C. Maher</i>	
<b>Subjective Assessments of Higher Order Ambisonic Sound Systems in Varying Acoustical Conditions</b> .....	939
<i>Andrew J. Horsburgh, Robert E. Davis, Martyn Moffat, D. Fraser Clark</i>	
<b>Matching Artificial Reverb Settings to Unknown Room Recordings: A Recommendation System for Reverb Plugins</b> .....	956
<i>Nils Peters, Jaeyoung Choi, Howard Lei</i>	
<b>An Indirect Study of Compliance and Damping in Linear Array Transducers</b> .....	965
<i>Richard Little</i>	
<b>Multimodal Information Management: Evaluation of Auditory and Haptic Cues for NextGen Communication Displays</b> .....	974
<i>Durand R. Begault, Rachel M. Bittner, Mark R. Anderson</i>	
<b>Automatic Speaker Delay Adjustment System Using Wireless Audio Capability of Zigbee Networks</b> .....	982
<i>Jaeho Choi, Myoung woo Nam, Kyogu Lee</i>	
<b>Transient Room Acoustics Using a 2.5 Dimensional Approach</b> .....	988
<i>Patrick Macey</i>	
<b>Prototype Spatial Auditory Display for Remote Planetary Exploration</b> .....	994
<i>Elizabeth M. Wenzel, Martine Godfroy, Joel D. Miller</i>	

<b>Optimal Configurations for Subwoofers in Rooms Considering Seat to Seat Variation and Low Frequency Efficiency</b> .....	1004
<i>Todd Welti</i>	
<b>Efficient Primary-Ambient Decomposition Algorithm for Audio Upmix</b> .....	1041
<i>Yong-Hyun Baek, Se-Woon Jeon, Young-Cheol Park, S.-P. Lee</i>	
<b>Calibration of Soundfield Microphones using the Diffuse-Field Response</b> .....	1048
<i>Aaron J. Heller, Eric Benjamin</i>	
<b>Low Frequency Noise Reduction by Synchronous Averaging under Asynchronous Measurement System in Real Sound Field</b> .....	1057
<i>Takuma Suzuki, Hiroshi Koide, Akihiko Shoji, Kouichi Tsuchiya, Tomohiko Endo, Shokichiro Hino</i>	
<b>Flexilink: A Unified Low Latency Network Architecture for Multichannel Live Audio</b> .....	1066
<i>Yonghao Wang, John Grant, Jeremy Foss</i>	
<b>Distance-Based Automatic Gain Control with Continuous Proximity-Effect Compensation</b> .....	1073
<i>Walter Etter</i>	
<b>Automatic Mode Estimation of Persian Musical Signals</b> .....	1084
<i>Peyman Heydarian, Lewis Jones, Allan Seago</i>	
<b>Blind Upmixing for Height and Wide Channels Based on Image Source Method</b> .....	1088
<i>Sunwoong Choi, Dong-il Hyun, Young-cheol Park, S. Lee, Dae Hee Youn</i>	
<b>On Accommodating Pitch Variation in Long Term Prediction of Speech and Vocals in Audio Coding</b> .....	1095
<i>Tejaswi Nanjundaswamy, Kenneth Rose</i>	
<b>Subband Comfort Noise Insertion for an Acoustic Echo Suppressor</b> .....	1104
<i>Guangji Shi, Changxue Ma</i>	
<b>Hong Kong Film Score Production: A Hollywood Informed Approach</b> .....	1109
<i>Robert Jay Ellis-Geiger</i>	
<b>Luciverb: Iterated Convolution for the Impatient</b> .....	1123
<i>Jonathan Abel, Michael J. Wilson</i>	
<b>Predicting the Fluctuation Strength of the Output of a Spatial Chorus Effects Processor</b> .....	1133
<i>William L. Martens, Robert W. Taylor, Luis Miranda</i>	
<b>A Second-order Soundfield Microphone with Improved Polar Pattern Shape</b> .....	1144
<i>Eric Benjamin</i>	
<b>Investigation into Electric Vehicles Exterior Noise Generation</b> .....	1152
<i>Stefania Cecchi, Andrea Primavera, Laura Romoli, Francesco Piazza, Ferruccio Bettarelli, Ariano Lattanzi</i>	
<b>Measurement and Analysis of the Spectral Directivity of an Electric Guitar Amplifier: Vertical Plane</b> .....	1162
<i>Agnieszka Roginska, Justin Mathew, Andrew Madden, Jim Anderson, Alex U. Case</i>	
<b>Simulating an Asymmetrically Saturated Nonlinearity Using an LNLNL Cascade</b> .....	1172
<i>Keun Sup Lee, Jonathan Abel</i>	
<b>Coefficient Interpolation for the Max Mathews Phasor Filter</b> .....	1180
<i>Dana Massie</i>	
<b>Aspects of Microphone Array Source Separation Performance</b> .....	1188
<i>Bjoern Erlach, Rob Bullen, Jonathan Abel</i>	
<b>A New Algorithm for Generating Realistic Three-dimensional Reverberation Based on Image Sound Source Distribution in Consideration of Room Shape Complexity</b> .....	1194
<i>Toshiki Hanyu</i>	
<b>On the Study of Ionic Microphone</b> .....	1203
<i>Hiroshi Akino, Hirofumi Shimokawa, Tadashi Kikutani, Jackie Green</i>	
<b>Author Index</b>	