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## EDITORIAL

Inaugural Editorial Innovations in an Era of Ubiquitous Audio, Speech, and Language Processing . . . . . *D. Hakkani-Tür* 5

## REGULAR PAPERS

A New Ratio Mask Representation for CASA-Based Speech Enhancement . . . . .	<i>F. Bao and W. H. Abdulla</i>	7
Complex ISNMF: A Phase-Aware Model for Monaural Audio Source Separation . . . . .	<i>P. Magron and T. Virtanen</i>	20
Gaussian Modeling-Based Multichannel Audio Source Separation Exploiting Generic Source Spectral Model . . . . .	<i>T. T. H. Duong, N. Q. K. Duong, P. C. Nguyen, and C. Q. Nguyen</i>	32
Decentralized Two-Channel Active Noise Control for Single Frequency by Shaping Matrix Eigenvalues . . . . .	<i>G. Zhang, J. Tao, X. Qiu, and I. Burnett</i>	44
Two-Stage Deep Learning for Noisy-Reverberant Speech Enhancement . . . . .	<i>Y. Zhao, Z.-Q. Wang, and D. Wang</i>	53
Phase-Aware Speech Enhancement Based on Deep Neural Networks . . . . .	<i>N. Zheng and X.-L. Zhang</i>	63
Evolution-Strategy-Based Automation of System Development for High-Performance Speech Recognition . . . . .	<i>T. Moriya, T. Tanaka, T. Shinozaki, S. Watanabe, and K. Duh</i>	77
Semantic Speech Retrieval With a Visually Grounded Model of Untranscribed Speech . . . . .	<i>H. Kamper, G. Shakhnarovich, and K. Livescu</i>	89
Model-Based Speech Enhancement for Intelligibility Improvement in Binaural Hearing Aids . . . . .	<i>M. S. Kavalekalam, J. K. Nielsen, J. B. Boldt, and M. G. Christensen</i>	99
Glottal Inverse Filtering Using Probabilistic Weighted Linear Prediction . . . . .	<i>A. R. MV and P. K. Ghosh</i>	114



---

Two-Stage Monaural Source Separation in Reverberant Room Environments Using Deep Neural Networks . . . . .	<i>Y. Sun, W. Wang, J. Chambers, and S. M. Naqvi</i>	125
Toward Fail-Safe Speaker Recognition: Trial-Based Calibration With a Reject Option . . . . .	<i>L. Ferrer, M. K. Nandwana, M. McLaren, D. Castan, and A. Lawson</i>	140
Asymmetric Coding for Rate-Constrained Noise Reduction in Binaural Hearing Aids . . . . .	<i>J. Amini, R. C. Hendriks, R. Heusdens, M. Guo, and J. Jensen</i>	154
Global Inference for Aspect and Opinion Terms Co-Extraction Based on Multi-Task Neural Networks . . . . .	<i>J. Yu, J. Jiang, and R. Xia</i>	168
Robust Speaker Localization Guided by Deep Learning-Based Time-Frequency Masking . . . . .	<i>Z.-Q. Wang, X. Zhang, and D. Wang</i>	178
Gated Residual Networks With Dilated Convolutions for Monaural Speech Enhancement . . . . .	<i>K. Tan, J. Chen, and D. Wang</i>	189
Phonology-Augmented Statistical Framework for Machine Transliteration Using Limited Linguistic Resources . . . . .	<i>G. H. Ngo, M. Nguyen, and N. F. Chen</i>	199
Unsupervised Detection of Anomalous Sound Based on Deep Learning and the Neyman–Pearson Lemma . . . . .	<i>Y. Koizumi, S. Saito, H. Uematsu, Y. Kawachi, and N. Harada</i>	212
A Bayesian Hierarchical Model for Speech Enhancement With Time-Varying Audio Channel . . . . .	<i>Y. Laufer and S. Gannot</i>	225

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ERRATA		
Erratum for “Nonlinear Audio Systems Identification Through Audio Input Gaussianization” . . . . .		240

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Information for Authors . . . . .		

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