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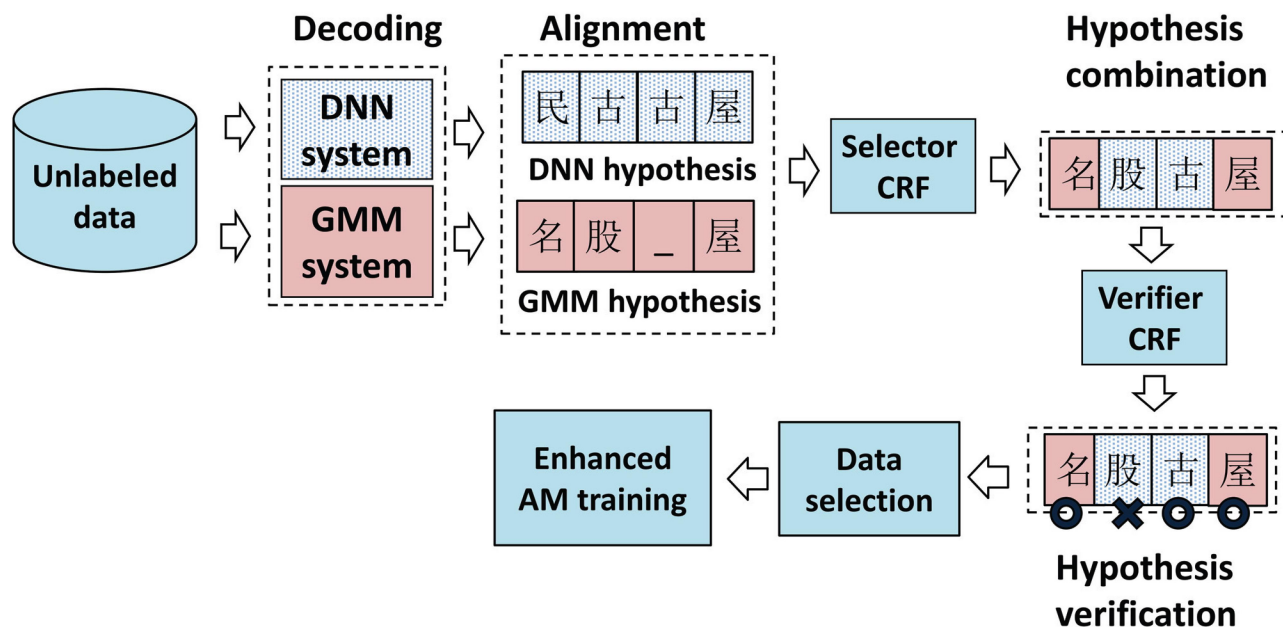
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Flowchart of the proposed CRF-based hypothesis combination and data selection method. For more, see "Semi-Supervised Acoustic Model Training by Discriminative Data Selection From Multiple ASR Systems' Hypotheses" by S. Li *et al.*, p. 1524.

For the September 2016 issue, see p. 1477 for Table of Contents
For the October 2016 issue, see p. 1677 for Table of Contents



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REGULAR PAPERS

Estimation of Room Acoustic Parameters: The ACE Challenge	1681
. <i>J. Eaton, N. D. Gaubitch, A. H. Moore, and P. A. Naylor</i>	
Efficient Implementation of Global Variance Compensation for Parametric Speech Synthesis	1694
. <i>T. Nose</i>	
Generative Modeling of Pseudo-Whisper for Robust Whispered Speech Recognition	1705
. <i>S. Ghaffarzadegan, H. Bořil, and J. H. L. Hansen</i>	
A Generalized Nonnegative Tensor Factorization Approach for Distant Speech Recognition With Distributed Microphones	1721
. <i>S. Mirsamadi, and J. H. L. Hansen</i>	
Adaptive Filtered-x Algorithms for Room Equalization Based on Block-Based Combination Schemes	1732
. <i>L. Fuster, M. de Diego, L. A. Azpicueta-Ruiz, and M. Ferrer</i>	

(Contents Continued on Page 1678)

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(Contents Continued from Page 1677)

Variational Bayesian Inference for Source Separation and Robust Feature Extraction	<i>K. Adiloğlu, and E. Vincent</i>	1746
Auditory Model-Based Dynamic Compression Controlled by Subband Instantaneous Frequency and Speech Presence Probability Estimates	<i>S. Kortlang, G. Grimm, V. Hohmann, B. Kollmeier, and S. D. Ewert</i>	1759
Differentiable Pooling for Unsupervised Acoustic Model Adaptation	<i>P. Swietojanski, and S. Renals</i>	1773
Optimal Microphone Array Observation for Clear Recording of Distant Sound Sources	<i>K. Niwa, Y. Hioka, and K. Kobayashi</i>	1785
Spherical Harmonic Signal Covariance and Sound Field Diffuseness	<i>N. Epain, and C. T. Jin</i>	1796
Near and Far Field Speech-in-Noise Intelligibility Improvements Based on a Time-Frequency Energy Reallocation Approach	<i>T.-C. Zorilă, Y. Stylianou, T. Ishihara, and M. Akamine</i>	1808
Similar Word Model for Unfrequent Word Enhancement in Speech Recognition	<i>X. Ma, D. Wang, and J. Tejedor</i>	1819
Summarizing Meeting Transcripts Based on Functional Segmentation	<i>M. H. Bokaei, H. Sameti, and Y. Liu</i>	1831
Abstractive Cross-Language Summarization via Translation Model Enhanced Predicate Argument Structure Fusing	<i>J. Zhang, Y. Zhou, and C. Zong</i>	1842
A Morphological Model for Simulating Acoustic Scenes and Its Application to Sound Event Detection	<i>G. Lafay, M. Lagrange, M. Rossignol, E. Benetos, and A. Roebel</i>	1854
Parallel Reference Speaker Weighting for Kinematic-Independent Acoustic-to-Articulatory Inversion	<i>A. Ji, M. T. Johnson, and J. J. Berry</i>	1865

EDICS—Editor’s Information Classification Scheme		1876
Information for Authors		1878
