Chin-HuiLee

List of Publications by Citations

Source: https://exaly.com/author-pdf/3301645/chin-huilee-publications-by-citations.pdf

Version: 2024-04-20

This document has been generated based on the publications and citations recorded by exaly.com. For the latest version of this publication list, visit the link given above.

The third column is the impact factor (IF) of the journal, and the fourth column is the number of citations of the article.

2,288 46 91 21 g-index h-index citations papers 3,169 113 3.5 5.43 L-index avg, IF ext. citations ext. papers

#	Paper	IF	Citations
91	A Regression Approach to Speech Enhancement Based on Deep Neural Networks. <i>IEEE/ACM Transactions on Audio Speech and Language Processing</i> , 2015 , 23, 7-19	3.6	509
90	An Experimental Study on Speech Enhancement Based on Deep Neural Networks. <i>IEEE Signal Processing Letters</i> , 2014 , 21, 65-68	3.2	407
89	Evaluation of sliding window correlation performance for characterizing dynamic functional connectivity and brain states. <i>Neurolmage</i> , 2016 , 133, 111-128	7.9	148
88	Developments and directions in speech recognition and understanding, Part 1 [DSP Education]. <i>IEEE Signal Processing Magazine</i> , 2009 , 26, 75-80	9.4	121
87	A Vector Space Modeling Approach to Spoken Language Identification. <i>IEEE Transactions on Audio Speech and Language Processing</i> , 2007 , 15, 271-284		109
86	A deep learning approach to automatic teeth detection and numbering based on object detection in dental periapical films. <i>Scientific Reports</i> , 2019 , 9, 3840	4.9	75
85	A Regression Approach to Single-Channel Speech Separation Via High-Resolution Deep Neural Networks. <i>IEEE/ACM Transactions on Audio Speech and Language Processing</i> , 2016 , 24, 1424-1437	3.6	53
84	Hermitian Polynomial for Speaker Adaptation of Connectionist Speech Recognition Systems. <i>IEEE Transactions on Audio Speech and Language Processing</i> , 2013 , 21, 2152-2161		39
83	A Reverberation-Time-Aware Approach to Speech Dereverberation Based on Deep Neural Networks. <i>IEEE/ACM Transactions on Audio Speech and Language Processing</i> , 2017 , 25, 102-111	3.6	39
82	Convolutional-Recurrent Neural Networks for Speech Enhancement 2018,		39
81	On Mean Absolute Error for Deep Neural Network Based Vector-to-Vector Regression. <i>IEEE Signal Processing Letters</i> , 2020 , 27, 1485-1489	3.2	37
8o	Deep Learning-Based Noise Reduction Approach to Improve Speech Intelligibility for Cochlear Implant Recipients. <i>Ear and Hearing</i> , 2018 , 39, 795-809	3.4	33
79	An End-to-End Deep Learning Approach to Simultaneous Speech Dereverberation and Acoustic Modeling for Robust Speech Recognition. <i>IEEE Journal on Selected Topics in Signal Processing</i> , 2017 , 11, 1289-1300	7.5	31
78	An Information-Extraction Approach to Speech Processing: Analysis, Detection, Verification, and Recognition. <i>Proceedings of the IEEE</i> , 2013 , 101, 1089-1115	14.3	29
77	Boosting attribute and phone estimation accuracies with deep neural networks for detection-based speech recognition 2012 ,		29
76	Speech Enhancement Based on TeacherBtudent Deep Learning Using Improved Speech Presence Probability for Noise-Robust Speech Recognition. <i>IEEE/ACM Transactions on Audio Speech and Language Processing</i> , 2019 , 27, 2080-2091	3.6	28
75	Speech separation based on improved deep neural networks with dual outputs of speech features for both target and interfering speakers 2014 ,		27

74	SNR-Based Progressive Learning of Deep Neural Network for Speech Enhancement		26
73	Approximate Test Risk Bound Minimization Through Soft Margin Estimation. <i>IEEE Transactions on Audio Speech and Language Processing</i> , 2007 , 15, 2393-2404		25
72	A Gender Mixture Detection Approach to Unsupervised Single-Channel Speech Separation Based on Deep Neural Networks. <i>IEEE/ACM Transactions on Audio Speech and Language Processing</i> , 2017 , 25, 1535-1546	3.6	23
71	Joint training of front-end and back-end deep neural networks for robust speech recognition 2015,		22
70	Improving non-native mispronunciation detection and enriching diagnostic feedback with DNN-based speech attribute modeling 2016 ,		20
69	Towards bottom-up continuous phone recognition 2007,		20
68	Speech Recognition Using Long-Span Temporal Patterns in a Deep Network Model. <i>IEEE Signal Processing Letters</i> , 2013 , 20, 201-204	3.2	19
67	Toward a detector-based universal phone recognizer. <i>Proceedings of the IEEE International Conference on Acoustics, Speech, and Signal Processing</i> , 2008 ,	1.6	19
66	A maximal figure-of-merit learning approach to maximizing mean average precision with deep neural network based classifiers 2014 ,		17
65	A Theory on Deep Neural Network Based Vector-to-Vector Regression With an Illustration of Its Expressive Power in Speech Enhancement. <i>IEEE/ACM Transactions on Audio Speech and Language Processing</i> , 2019 , 27, 1932-1943	3.6	14
64	i-Vector Modeling of Speech Attributes for Automatic Foreign Accent Recognition. <i>IEEE/ACM Transactions on Audio Speech and Language Processing</i> , 2016 , 24, 29-41	3.6	14
63	A unified DNN approach to speaker-dependent simultaneous speech enhancement and speech separation in low SNR environments. <i>Speech Communication</i> , 2017 , 95, 28-39	2.8	12
62	A new approach to utterance verification based on neighborhood information in model space. <i>IEEE Transactions on Speech and Audio Processing</i> , 2003 , 11, 425-434		12
61	Analyzing Upper Bounds on Mean Absolute Errors for Deep Neural Network-Based Vector-to-Vector Regression. <i>IEEE Transactions on Signal Processing</i> , 2020 , 68, 3411-3422	4.8	11
60	An iterative mask estimation approach to deep learning based multi-channel speech recognition. <i>Speech Communication</i> , 2019 , 106, 31-43	2.8	11
59	A Hybrid Approach to Combining Conventional and Deep Learning Techniques for Single-Channel Speech Enhancement and Recognition 2018 ,		11
58	Cross-language transfer learning for deep neural network based speech enhancement 2014,		10
57	Global variance equalization for improving deep neural network based speech enhancement 2014,		10

56	Decentralizing Feature Extraction with Quantum Convolutional Neural Network for Automatic Speech Recognition 2021 ,		10
55	Using Generalized Gaussian Distributions to Improve Regression Error Modeling for Deep Learning-Based Speech Enhancement. <i>IEEE/ACM Transactions on Audio Speech and Language Processing</i> , 2019 , 27, 1919-1931	3.6	9
54	A detection-based approach to broadcast news video story segmentation 2009 ,		9
53	A study on target feature activation and normalization and their impacts on the performance of DNN based speech dereverberation systems 2016 ,		9
52	Tensor-To-Vector Regression for Multi-Channel Speech Enhancement Based on Tensor-Train Network 2020 ,		8
51	Image region annotation based on segmentation and semantic correlation analysis. <i>IET Image Processing</i> , 2018 , 12, 1331-1337	1.7	8
50	Hierarchical Bayesian combination of plug-in maximum a posteriori decoders in deep neural networks-based speech recognition and speaker adaptation. <i>Pattern Recognition Letters</i> , 2017 , 98, 1-7	4.7	8
49	On frequency dependencies of sliding window correlation 2015 ,		8
48	A study on model-based error rate estimation for automatic speech recognition. <i>IEEE Transactions on Speech and Audio Processing</i> , 2003 , 11, 581-589		8
47	Improving Mandarin Tone Recognition Based on DNN by Combining Acoustic and Articulatory Features Using Extended Recognition Networks. <i>Journal of Signal Processing Systems</i> , 2018 , 90, 1077-10	o đ 7 4	7
46	A transfer learning and progressive stacking approach to reducing deep model sizes with an application to speech enhancement 2017 ,		7
45	Bayesian Unsupervised Batch and Online Speaker Adaptation of Activation Function Parameters in Deep Models for Automatic Speech Recognition. <i>IEEE/ACM Transactions on Audio Speech and Language Processing</i> , 2017 , 25, 64-75	3.6	7
44	Unsupervised anchor shot detection using multi-modal spectral clustering. <i>Proceedings of the IEEE International Conference on Acoustics, Speech, and Signal Processing</i> , 2008 ,	1.6	7
43	Approximate Test Risk Minimization Through Soft Margin Estimation 2007,		7
42	Unsupervised single-channel speech separation via deep neural network for different gender mixtures 2016 ,		7
41	A Probabilistic Framework for Representing Dialog Systems and Entropy-Based Dialog Management Through Dynamic Stochastic State Evolution. <i>IEEE/ACM Transactions on Audio Speech</i> and Language Processing, 2015 , 23, 2026-2035	3.6	6
40	Boosting of Maximal Figure of Merit Classifiers for Automatic Image Annotation 2007,		6
39	A Cross-Entropy-Guided Measure (CEGM) for Assessing Speech Recognition Performance and Optimizing DNN-Based Speech Enhancement. <i>IEEE/ACM Transactions on Audio Speech and Language Processing</i> , 2021 , 29, 106-117	3.6	6

38	A keyword-aware grammar framework for LVCSR-based spoken keyword search 2015,		5
37	Ensemble speaker and speaking environment modeling approach with advanced online estimation process 2009 ,		5
36	Language Recognition Based on Score Distribution Feature Vectors and Discriminative Classifier Fusion 2006 ,		5
35	Improving Mispronunciation Detection of Mandarin Tones for Non-Native Learners With Soft-Target Tone Labels and BLSTM-Based Deep Tone Models. <i>IEEE/ACM Transactions on Audio Speech and Language Processing</i> , 2019 , 27, 2012-2024	3.6	4
34	Preference Music Ratings Prediction Using Tokenization and Minimum Classification Error Training. <i>IEEE Transactions on Audio Speech and Language Processing</i> , 2011 , 19, 2294-2303		4
33	Experimental studies on continuous speech recognition using neural architectures with <code>Edaptivell</code> hidden activation functions 2010,		4
32	A phonetic feature based lattice rescoring approach to LVCSR 2009,		4
31	Optimizing the Performance of Spoken Language Recognition With Discriminative Training. <i>IEEE Transactions on Audio Speech and Language Processing</i> , 2008 , 16, 1642-1653		4
30	Information Fusion in Attention Networks Using Adaptive and Multi-Level Factorized Bilinear Pooling for Audio-Visual Emotion Recognition. <i>IEEE/ACM Transactions on Audio Speech and Language Processing</i> , 2021 , 29, 2617-2629	3.6	4
29	DNN Training Based on Classic Gain Function for Single-channel Speech Enhancement and Recognition 2019 ,		3
28	A Cross-Task Transfer Learning Approach to Adapting Deep Speech Enhancement Models to Unseen Background Noise Using Paired Senone Classifiers 2020 ,		3
27	A Multi-Target SNR-Progressive Learning Approach to Regression Based Speech Enhancement. <i>IEEE/ACM Transactions on Audio Speech and Language Processing</i> , 2020 , 28, 1608-1619	3.6	3
26	Improving Deep Neural Network Based Speech Synthesis through Contextual Feature Parametrization and Multi-Task Learning. <i>Journal of Signal Processing Systems</i> , 2018 , 90, 1025-1037	1.4	3
25	Automatic image region annotation through segmentation based visual semantic analysis and discriminative classification 2016 ,		3
24	Detection-based accented speech recognition using articulatory features 2011 ,		3
23	A study on cross-language knowledge integration in Mandarin LVCSR 2012 ,		3
22	A Comparison of Single- and Multi-Objective Programming Approaches to Problems with Multiple Design Objectives. <i>Journal of Signal Processing Systems</i> , 2010 , 61, 39-50	1.4	3
21	A Flexible Classifier Design Framework Based on Multiobjective Programming. <i>IEEE Transactions on Audio Speech and Language Processing</i> , 2008 , 16, 779-789		3

20	Two extensions to ensemble speaker and speaking environment modeling for robust automatic speech recognition 2007 ,		3
19	Improving Mandarin Tone Mispronunciation Detection for Non-Native Learners with Soft-Target Tone Labels and BLSTM-Based Deep Models 2018 ,		3
18	A unified deep modeling approach to simultaneous speech dereverberation and recognition for the reverb challenge 2017 ,		2
17	Reliable Accent-Specific Unit Generation With Discriminative Dynamic Gaussian Mixture Selection for Multi-Accent Chinese Speech Recognition. <i>IEEE Transactions on Audio Speech and Language Processing</i> , 2013 , 21, 2073-2084		2
16	Multiple time resolution analysis of speech signal using MCE training with application to speech recognition 2009 ,		2
15	An efficient gradient computation approach to discriminative fusion optimization in semantic concept detection 2008 ,		2
14	An Iterative Constrained Optimization Approach to Classifier Design		2
13	Using tone-based extended recognition network to detect non-native Mandarin tone mispronunciations 2016 ,		2
12	Two-Stage Enhancement of Noisy and Reverberant Microphone Array Speech for Automatic Speech Recognition Systems Trained with Only Clean Speech 2018 ,		2
11	Correlating subword articulation with lip shapes for embedding aware audio-visual speech enhancement. <i>Neural Networks</i> , 2021 , 143, 171-182	9.1	2
10	Improving Audio-visual Speech Recognition Performance with Cross-modal Student-teacher Training 2019 ,		1
9	Performance Analysis for Tensor-Train Decomposition to Deep Neural Network Based Vector-to-Vector Regression 2020 ,		1
8	A Keyword-Aware Language Modeling Approach to Spoken Keyword Search. <i>Journal of Signal Processing Systems</i> , 2016 , 82, 197-206	1.4	1
7	A single-ensemble-based hybrid approach to clutter rejection combining bilinear Hankel with regression. <i>Journal of Medical Ultrasonics (2001)</i> , 2013 , 40, 99-105	1.4	1
6	Model-based margin estimation for hidden Markov model learning and generalisation. <i>IET Signal Processing</i> , 2013 , 7, 704-709	1.7	1
5	MAP estimation of online mapping parameters in ensemble speaker and speaking environment modeling 2009 ,		1
4	A study on soft margin estimation for LVCSR 2007 ,		1
3	A study on knowledge source integration for candidate rescoring in automatic speech recognition		1

A Two-Stage Approach to Device-Robust Acoustic Scene Classification **2021**,

Learning auxiliary categorical information for speech synthesis based on deep and recurrent neural networks **2016**,

1

1