Stefania Cecchi

List of Publications by Year in descending order

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Version: 2024-02-01

713332 759055 73 640 12 21 h-index citations g-index papers 75 75 75 312 docs citations times ranked citing authors all docs

#	Article	IF	CITATIONS
1	Comparison of Feature Extraction Methods for Sound-Based Classification of Honey Bee Activity. IEEE/ACM Transactions on Audio Speech and Language Processing, 2022, 30, 112-122.	4.0	18
2	Sensors Characterization for a Calibration-Free Connected Smart Insole for Healthy Ageing. Lecture Notes of the Institute for Computer Sciences, Social-Informatics and Telecommunications Engineering, 2021, , 35-54.	0.2	1
3	A Swept-Sine-Type Single Measurement to Estimate Intermodulation Distortion in a Dynamic Range of Audio Signal Amplitudes. IEEE Transactions on Instrumentation and Measurement, 2021, 70, 1-11.	2.4	3
4	A Low-Complexity Linear-Phase Graphic Audio Equalizer Based on IFIR Filters. IEEE Signal Processing Letters, 2021, 28, 429-433.	2.1	5
5	A Room Impulse Response Measurement Method Robust Towards Nonlinearities Based on Orthogonal Periodic Sequences. IEEE/ACM Transactions on Audio Speech and Language Processing, 2021, 29, 3104-3117.	4.0	8
6	An Efficient Active Noise Control System with Online Secondary Path Estimation for Snoring Reduction. , $2021, \dots$		4
7	A Swept-Sine Pulse Compression Procedure for an Effective Measurement of Intermodulation Distortion. IEEE Transactions on Instrumentation and Measurement, 2020, 69, 1708-1719.	2.4	6
8	On the Importance of the Sound Emitted by Honey Bee Hives. Veterinary Sciences, 2020, 7, 168.	0.6	46
9	Orthogonal Periodic Sequences for the Identification of Functional Link Polynomial Filters. IEEE Transactions on Signal Processing, 2020, 68, 5308-5321.	3.2	12
10	Robust Room Impulse Response Measurement Using Perfect Periodic Sequences for Wiener Nonlinear Filters. Electronics (Switzerland), 2020, 9, 1793.	1.8	9
11	A Smart Sensor-Based Measurement System for Advanced Bee Hive Monitoring. Sensors, 2020, 20, 2726.	2.1	60
12	Identification Issues Associated with the Use of Wearable Accelerometers in Lifelogging. Lecture Notes in Computer Science, 2020, , 338-351.	1.0	2
13	Review on Electric Vehicles Exterior Noise Generation and Evaluation. , 2020, , .		1
14	Features Extraction Applied to the Analysis of the Sounds Emitted by Honey Bees in a Beehive., 2019,,.		16
15	Introducing the Orthogonal Periodic Sequences for the Identification of Functional Link Polynomial Filters. , 2019, , .		3
16	Audio-based Identification of Beehive States. , 2019, , .		35
17	Nonlinear system identification using Wiener basis functions and multiple-variance perfect sequences. Signal Processing, 2019, 160, 137-149.	2.1	16
18	On Room Impulse Response Measurement Using Orthogonal Periodic Sequences. , 2019, , .		3

#	Article	IF	Citations
19	Digital Signal Processing for Audio Applications: Then, Now and the Future. , 2019, , 37-51.		O
20	Identification of Volterra Models of Tube Audio Devices using Multiple-Variance Method. AES: Journal of the Audio Engineering Society, 2018, 66, 823-838.	0.8	19
21	Room Response Equalization—A Review. Applied Sciences (Switzerland), 2018, 8, 16.	1.3	45
22	Orthogonal LIP Nonlinear Filters. , 2018, , 15-46.		7
23	Multichannel acoustic echo cancellation exploiting effective fundamental frequency estimation. Speech Communication, 2017, 86, 97-106.	1.6	7
24	Multivariance nonlinear system identification using Wiener basis functions and perfect sequences. , 2017, , .		2
25	Investigation on Audio Algorithms Architecture for Stereo Portable Devices. AES: Journal of the Audio Engineering Society, 2016, 64, 75-88.	0.8	5
26	Robust Room Impulse Response Measurement Using Perfect Sequences for Legendre Nonlinear Filters. IEEE/ACM Transactions on Audio Speech and Language Processing, 2016, 24, 1969-1982.	4.0	11
27	Perfect periodic sequences for nonlinear Wiener filters. , 2016, , .		8
28	Combining evolution strategies and neural network procedures for compression driver design. , 2016, , .		0
29	Multichannel Double-Talk Detector based on Fundamental Frequency Estimation. IEEE Signal Processing Letters, 2016, 23, 94-97.	2.1	8
30	A Hybrid Approach for Reverberation Simulation. IEICE Transactions on Fundamentals of Electronics, Communications and Computer Sciences, 2015, E98.A, 2101-2108.	0.2	0
31	An interactive optimization procedure for stereophonic acoustic echo cancellation systems. , 2015, , .		1
32	Room impulse response estimation using perfect sequences for Legendre nonlinear filters. , 2015, , .		4
33	An adaptive multichannel identification system for room response equalization. , 2015, , .		2
34	Real-time implementation and performance evaluation of digital control for loudspeakers line arrays. Applied Acoustics, 2015, 97, 121-132.	1.7	0
35	A voice activity detection algorithm for multichannel acoustic echo cancellation exploiting fundamental frequency estimation. , $2015, , .$		2
36	Investigation on crosstalk cancellation introducing loudspeakers directivity in HRTFs approximation. , $2015, , .$		0

#	Article	IF	CITATIONS
37	An optimization procedure for a nonlinear system identification approach based on cubic splines. , 2015, , .		O
38	Intelligent Acoustic Interfaces With Multisensor Acquisition for Immersive Reproduction. IEEE Transactions on Multimedia, 2015, 17, 1262-1272.	5.2	10
39	Legendre nonlinear filters. Signal Processing, 2015, 109, 84-94.	2.1	40
40	Advanced intelligent acoustic interfaces for multichannel audio reproduction. , 2014, , .		2
41	A novel decorrelation approach for multichannel system identification. , 2014, , .		8
42	Introducing Legendre nonlinear filters. , 2014, , .		15
43	An efficient implementation of acoustic crosstalk cancellation for 3D audio rendering. , 2014, , .		3
44	Advanced audio spatializer combined with a multipoint equalization system. , 2014, , .		0
45	A low latency implementation of a non-uniform partitioned convolution algorithm for room acoustic simulation. Signal, Image and Video Processing, 2014, 8, 985-994.	1.7	4
46	A multichannel and multiple position adaptive room response equalizer in warped domain: Real-time implementation and performance evaluation. Applied Acoustics, 2014, 82, 28-37.	1.7	12
47	An efficient time varying hybrid reverberator for room acoustic simulation. , 2014, , .		1
48	Objective and Subjective Investigation on a Novel Method for Digital Reverberator Parameters Estimation. IEEE/ACM Transactions on Audio Speech and Language Processing, 2014, 22, 441-452.	4.0	5
49	Adaptive Feedback Active Noise Control for Yacht Environments. IEEE Transactions on Control Systems Technology, 2014, 22, 737-744.	3.2	11
50	Evaluation of a channel decorrelation approach for stereo acoustic echo cancellation., 2013,,.		0
51	Identification of Hammerstein model using cubic splines and FIR filtering. , 2013, , .		8
52	A Combined Approach for Channel Decorrelation in Stereo Acoustic Echo Cancellation Exploiting Time-Varying Frequency Shifting. IEEE Signal Processing Letters, 2013, 20, 717-720.	2.1	10
53	IIR filter approximation of an innovative digital audio equalizer. , 2013, , .		3
54	Hybrid Reverberator Using Multiple Impulse Responses for Audio Rendering Improvement. , 2013, , .		1

#	Article	IF	CITATIONS
55	An efficient DSP implementation of a dynamic convolution approach using principal component analysis. , $2012, , .$		1
56	An efficient DSP-based implementation of a fast convolution approach with non uniform partitioning, , 2012, , .		4
57	Low-complexity implementation of a real-time decorrelation algorithm for stereophonic acoustic echo cancellation. Signal Processing, 2012, 92, 2668-2675.	2.1	15
58	Multiple Position Room Response Equalization in Frequency Domain. IEEE Transactions on Audio Speech and Language Processing, 2012, 20, 122-135.	3.8	19
59	A Mixed Decorrelation Approach for Stereo Acoustic Echo Cancellation Based on the Estimation of the Fundamental Frequency. IEEE Transactions on Audio Speech and Language Processing, 2012, 20, 690-698.	3.8	25
60	In Car Audio. , 2012, , 155-192.		0
61	A Combined Psychoacoustic Approach for Stereo Acoustic Echo Cancellation. IEEE Transactions on Audio Speech and Language Processing, 2011, 19, 1530-1539.	3 . 8	18
62	A theoretical analysis of a buffer frame size conversion algorithm for audio applications ensuring minimum latency. Signal, Image and Video Processing, 2011, 5, 185-190.	1.7	0
63	A novel approach to channel decorrelation for stereo Acoustic Echo Cancellation based on missing fundamental theory. , 2010, , .		8
64	Advanced CIS architecture and algorithms for enhanced in-car audio listening. , 2009, , .		2
65	A Subband Approach to Wave Domain Adaptive Filtering. , 2008, , .		4
66	Real-time implementation of Wave Field Synthesis for sound reproduction systems. , 2008, , .		3
67	A Novel Approach to Active Noise Control Based on Wave Domain Adaptive Filtering. , 2007, , .		11
68	A New Approach to Digital Audio Equalization. , 2007, , .		5
69	A New Approach to Bass Enhancement Based on Prony's Method. , 2007, , .		9
70	Echo State Networks for Real-Time Audio Applications. Lecture Notes in Computer Science, 2007, , 731-740.	1.0	6
71	Gaussianization Based Approach for Post-Nonlinear Underdetermined BSS with Delays. Lecture Notes in Computer Science, 2007, , 783-792.	1.0	0
72	Comparing Performances of Different Multiband Adaptive Architectures for Noise Reduction., 2006,,.		1

ARTICLE IF CITATIONS

Performance Evaluation of Adaptive Algorithms for Wave Field Analysis/Synthesis Using Sound Field

Simulations., 0, , . 3