Aswin Shanmugam Subramanian

Research Scientist, MERL

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EDUCATION

2016-Present Doctor of Philosophy, Electrical & Computer Engineering

Johns Hopkins University, USA. Advisor: Dr. Shinji Watanabe.

2016-2017 $\,$ Master of Science, Electrical & Computer Engineering, GPA: 3.7/4

Johns Hopkins University, USA. CeDiD: 1857-C2C7-A8N8

2012-2015 Master of Science (by Research), Computer Science & Engineering

Indian Institute of Technology Madras, India.

Advisor: Dr. Hema A. Murthy.

2008-2012 **Bachelor of Technology**, Information Technology

SSN College of Engineering, Anna University, India.

EXPERIENCE

September 2021 Research Scientist, Mitsubishi Electric Research Laboratories

- Present **Q** Cambridge, MA, USA

• Working on total transcription and source localization as a member of speech & audio team.

Fall 2017 - Research Assistant, Johns Hopkins University, Advisor: Prof. Shinji Watanabe

- Fall 2020 devised a paradigm called *directional ASR* that can train source localization with ASR objectives [1]. Contributed to speech enhancement functionality in ESPnet [3].
- Spring 2020 led speech separation efforts for the JHU team in CHiME-6 challenge and our team finished second in "diarization+ASR" track. Collaborated with SJTU on joint dereverberation & beamforming [4].
- Fall 2018 & Spring 2019 worked on speech dereverberation in collaboration with Yahoo Japan [7]. Devised a technique to estimate speech enhancement hyper-parameters using ASR objectives [6].
- Spring 2018 participated in CHiME-5 challenge as a member of the JHU-Hitachi team and contributed to the neural denoising module. Our team was placed in top 2.
- Fall 2017 developed a novel single channel speech enhancement method [8], and contributed to make Kaldi CHiME-4 recipe state-of-art [9].

Summer 2020 & **NLP Research Intern**, *Tencent AI Lab*, Mentors: Dr. Chao Weng & Dr. Dong Yu Summer 2019 **♥** Bellevue, WA, USA

- o 2020 devised DNN approaches for multi-source localization and showed its effectiveness in cutting down ASR word error rates by about a factor of two [2].
- o 2019 implemented a novel target speech extraction method with end-to-end speech recognition objectives [4].

Summer 2018 Research Intern, NTT Communication Sciences Lab, Mentor: Dr. Marc Delcroix

- ♥ Kyoto, Japan
- Experimented with speaker and environment adaptation techniques for end-to-end noise robust speech recognition.

July 2015 - Network Software Engineer, Intel

August 2016 Pangalore, India

o IPv6 module for Axxia network accelerators and automation of performance benchmarks.

June 2012 - July **Project Associate**, *IIT Madras*, , Advisor: Prof. Hema A. Murthy

2015 **Q** Chennai, India

- Member of the TTS consortium that developed a common framework for HMM based speech synthesis of 13 Indian languages.
- Participated in Blizzard Challenge 2014 & 2015.
- Developed an automatic segmentation tool for the TTS consortium [10].

SKILLS

ProgrammingPython, C, C++, Java, C#, Shell, Perl,ToolkitsKaldi, ESPnet, HTK, HTS, Festival,
Chainer, Pytorch

TEACHING

Fall '17 & '18 Course Assistant, Digital Signal Processing, Johns Hopkins University

Spring '18 - '20 Course Assistant, Information Extraction from Speech and Text, Johns Hopkins University

KEY PUBLICATIONS[†]

- 1. Aswin Shanmugam Subramanian, Chao Weng, Shinji Watanabe, Meng Yu, Yong Xu, Shi-Xiong Zhang, and Dong Yu, "Directional ASR: A New Paradigm for E2E Multi-Speaker Speech Recognition with Source Localization," accepted in *IEEE ICASSP 2021*.
- 2. Aswin Shanmugam Subramanian, Chao Weng, Shinji Watanabe, Meng Yu, and Dong Yu, "Deep Learning based Multi-Source Localization with Source Splitting and its Effectiveness in Multi-Talker Speech Recognition," arXiv:2102.07955.
- Chenda Li, Jing Shi, Wangyou Zhang, <u>Aswin Shanmugam Subramanian</u>, Xuankai Chang, Naoyuki Kamo, Moto Hira, Tomoki Hayashi, Christoph Boeddeker, Zhuo Chen, and Shinji Watanabe, "ESPNET-SE: End-to-End Speech Enhancement and Separation Toolkit Designed for ASR Integration," in Proc. of *IEEE SLT 2021*, pp. 785–792.
- 4. Wangyou Zhang, Aswin Shanmugam Subramanian, Xuankai Chang, Shinji Watanabe, and Yanmin Qian, "End-to-End Far-Field Speech Recognition with Unified Dereverberation and Beamforming", in Proc. of *ISCA INTERSPEECH 2020*, pp. 324-328.
- 5. Aswin Shanmugam Subramanian, Chao Weng, Meng Yu, Shi-Xiong Zhang, Yong Xu, Shinji Watanabe, and Dong Yu, "Far-Field Location Guided Target Speech Extraction using End-to-End Speech Recognition Objectives," in Proc. of *IEEE ICASSP 2020*, pp. 7299-7303.
- Aswin Shanmugam Subramanian, Xiaofei Wang, Murali Karthick Baskar, Shinji Watanabe, Toru Taniguchi, Dung Tran, and Yuya Fujita, "Speech Enhancement Using End-to-End Speech Recognition Objectives," in Proc. of *IEEE WASPAA 2019*, pp. 229–233.
- 7. Toru Taniguchi, Aswin Shanmugam Subramanian, Xiaofei Wang, Dung Tran, Yuya Fujita, and Shinji Watanabe, "Generalized Weighted-Prediction-Error Dereverberation with Varying Source Priors for Reverberant Speech Recognition," in Proc. of *IEEE WASPAA 2019*, pp. 288–292.
- 8. Aswin Shanmugam Subramanian, Szu-Jui Chen, and Shinji Watanabe, "Student-Teacher Learning for BLSTM Mask-based Speech Enhancement," in Proc. of *ISCA INTERSPEECH 2018*, pp. 3249–3253.
- 9. Szu-Jui Chen, <u>Aswin Shanmugam Subramanian</u>, Hainan Xu, and Shinji Watanabe, "**Building state-of-the-art distant speech recognition using the CHiME-4 challenge with a setup of speech enhancement baseline**," in Proc. of *ISCA INTERSPEECH 2018*, pp. 1571–1575.
- 10. S Aswin Shanmugam, and Hema Murthy, "A Hybrid Approach to Segmentation of Speech Using Group Delay Processing and HMM Based Embedded Reestimation," in Proc. of ISCA INTERSPEECH 2014, pp. 1648–1652.

For full list of publications please visit https://sas91.github.io/publication