HAO SHI

PhD Candidate, at Speech and Audio Processing Lab, Kyoto University

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Personal Website: https://hshi-speech.github.io

RESEARCH INTERESTS

Automatic Speech Recognition:

- Noise-robust
- Adaptation
- Multi-speaker
- Knowledge distillation

Speech Enhancement:

- Front-end for robust ASR
- Integration of complementary systems
- Probabilistic Model
- Multi-model

Speech Separation:

- Target speaker extraction
- Blind source separation

EDUCATION

Ph.D. in Informatics, Kyoto University, Kyoto, Japan

2021 - Present

- o Department of Intelligence Science and Technology, Graduate School of Informatics
- o Supervisor: Prof. Tatsuya Kawahara

Master in Computer Science and Technology, Tianjin University, Tianjin, China

2018 - 2021

- College of Intelligence and Computing
- o Supervisor: Prof. Longbiao Wang

Working Experiences

Research Fellow, at Kyoto University

04. 2024 - . Present

o Mentor: Tatsuya Kawahara

Research Intern, at NTT (CS Lab @ Keihanna)

 $08.\ 2023 - 09.\ 2023$

o Mentor: Naoyuki Kamo, Shoko Araki, Tomohiro Nakatani, Marc Delcroix

Research Intern, at Sony (R&D @ Osaki)

01. 2023 – 03. 2023

o Mentor: Kazuki Shimada, Shusuke Takahashi

Research Assistant, at Tianjin University (Cognitive Computing and Application Lab.)

08. 2021 - 01. 2022

o Mentor: Longbiao Wang, Jianwu Dang

HONORS

Fellowship, awarded by Japan Science and Technology Agency (JST)

04.2022 - 03.2024

• The establishment of university fellowships towards the creation of science and technology innovation.

LANGUAGE SKILL

- o Chinese (native)
- o English (fluent)

PUBLICATIONS

Conference Papers (Reviewed Paper):

- <u>Hao Shi</u>, Naoyuki Kamo, Marc Delcroix, Tomohiro Nakatani, and Shoko Araki, "Ensemble Inference for Diffusion Model-based Speech Enhancement", in Proc. of HSCMA, 2024, pp. xx-xx.
- Hao Shi, Kazuki Shimada, Masato Hirano, Takashi Shibuya, Yuichiro Koyama, Zhi Zhong, Shusuke Takahashi, Tatsuya Kawahara, and Yuki Mitsufuji, "Diffusion-Based Speech Enhancement with Joint Generative and Predictive Decoders", in Proc. of ICASSP, 2024, pp. 12951–12955.
- Yuan Gao, <u>Hao Shi</u>, Chenhui Chu, and Tatsuya Kawahara, "Enhancing Two-stage Finetuning for Speech Emotion Recognition Using Adapters", in Proc. of ICASSP, 2024, pp. 11316–11320.
- <u>Hao Shi</u>, and Tatsuya Kawahara, "Investigation of Adapter for Automatic Speech Recognition in Noisy Environment", in SIG Technical Reports, 2023, pp. 1–6.
- Zhi Zhong, <u>Hao Shi</u>, Masato Hirano, Kazuki Shimada, Kazuya Tateishi, Takashi Shibuya, Shusuke Takahashi, and Yuki Mitsufuji, "Extending Audio Masked Autoencoders Toward Audio Restoration", in Proc. of WASPAA, 2023, pp. 1–5.
- <u>Hao Shi</u>, Masato Mimura, Longbiao Wang, Jianwu Dang, and Tatsuya Kawahara, "Time-domain Speech Enhancement Assisted by Multi-resolution Frequency Encoder And Decoder," in Proc. of ICASSP, 2023, pp. 1–5.
- Yanbing Yang, <u>Hao Shi</u>, Yuqin Lin, Meng Ge, Longbiao Wang, Qingzhi Hou and Jianwu Dang, "Adaptive Attention Network with Domain Adversarial Training for Multi-Accent Speech Recognition," in Proc. of ISCSLP, 2022, pp. 6–10.
- <u>Hao Shi</u>, Yuchun Shu, Longbiao Wang, Jianwu Dang, and Tatsuya Kawahara, "Fusing Multiple Bandwidth Spectrograms for Improving Speech Enhancement," in Proc. of APSIPA ASC, 2022, pp. 1935–1940.
- <u>Hao Shi</u>, Longbiao Wang, Sheng Li, Jianwu Dang, and Tatsuya Kawahara, "Subband-Based Spectrogram Fusion for Speech Enhancement by Combining Mapping and Masking Approaches," in Proc. of APSIPA ASC, 2022, pp. 286–292.
- Hao Shi, Longbiao Wang, Sheng Li, Jianwu Dang, and Tatsuya Kawahara, "Monaural speech enhancement based on spectrogram decomposition for convolutional neural network-sensitive feature extraction," in Proc. of Interspeech, 2022, pp. 221–225.
- Tongtong Song, Qiang Xu, Meng Ge, Longbiao Wang, <u>Hao Shi</u>, Yongjie Lv, Yuqin Lin, and Jianwu Dang, "Language-specific Characteristic Assistance for Code-switching Speech Recognition," in Proc. of Interspeech, 2022, pp. 3924–3928. (Corresponding Author)
- Qiang Xu, Tongtong Song, Longbiao Wang, <u>Hao Shi</u>, Yuqin Lin, Yongjie Lv, Meng Ge, Qiang Yu, and Jianwu Dang, "Self-Distillation Based on High-level Information Supervision for Compressing End-to-End ASR Model," in Proc. of Interspeech, 2022, pp. 1716–1720. (Corresponding Author)
- <u>Hao Shi</u>, Longbiao Wang, Sheng Li, Cunhang Fan, Jianwu Dang, and Tatsuya Kawahara, "Spectrograms Fusion-based End-to-end Robust Automatic Speech Recognition," in Proc. of APSIPA ASC, 2021, pp. 438–442.
- Luya Qiang, <u>Hao Shi</u>, Meng Ge, Haoran Yin, Nan Li, Longbiao Wang, Sheng Li, and Jianwu Dang, "Speech Dereverberation Based on Scale-aware Mean Square Error Loss," in Proc. of ICONIP, 2021, pp. 55–63. (Joint first author, equal contribution)
- Haoran Yin, <u>Hao Shi</u>, Longbiao Wang, Luya Qiang, Sheng Li, Meng Ge, Gaoyan Zhang, and Jianwu Dang, "Simultaneous Progressive Filtering-based Monaural Speech Enhancement," in Proc. of ICONIP, 2021, pp. 213–221. (Joint first author, equal contribution)
- <u>Hao Shi</u>, Longbiao Wang, Meng Ge, Sheng Li, and Jianwu Dang, "Spectrograms Fusion with Minimum Difference Masks Estimation for Monaural Speech Dereverberation," in Proc. of ICASSP, 2020, pp. 7544-7548.
- <u>Hao Shi</u>, Longbiao Wang, Sheng Li, Chenchen Ding, Meng Ge, Nan Li, Jianwu Dang, and Hiroshi Seki, "Singing Voice Extraction with Attention based Spectrograms Fusion," in Proc. of Interspeech, 2020, pp. 2412-2416.
- Meng Ge, Longbiao Wang, Nan Li, <u>Hao Shi</u>, Jianwu Dang, and Xiangang Li, "Environment-dependent attention-driven recurrent convolutional neural network for robust speech enhancement," in Proc. of Interspeech, 2019, pp. 3153-3157.