

HAO SHI

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RESEARCH INTERESTS

Automatic Speech Recognition:

- Robust ASR (Under noise conditions, with front-end)
- Knowledge distillation for ASR (Model compression)
- Accents-based ASR (Accent adaptation)

Speech Enhancement:

- Spectrograms fusion (Complementary between different systems, frequency-domain models)
- Front-end for robust ASR (Improve both human hearing and recognition performance)

Speech Separation:

- Blind source separation (for ASR)

EDUCATION

Ph.D. in Computer Science and Technology , Kyoto University, Kyoto, Japan Department of Intelligence Science and Technology, Graduate School of Informatics ◦ Supervisor: Prof. Tatsuya Kawahara	Apr. 2021 – Present.
Master in Computer Science and Technology , Tianjin University, Tianjin, China College of Intelligence and Computing ◦ Supervisor: Prof. Longbiao Wang	Sep. 2018 – Jan. 2021
B.Sc. in Computer Science and Technology , Southwest Jiaotong University, Sichuan, China The School of Information Science and Technology	Sep. 2014 – Jun. 2018

Research Internships

Research Assistant , at Tianjin University	Aug. 2021 – Jan. 2022
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LANGUAGE SKILL

- * Chinese (native)
- * English (fluent)

PUBLICATIONS

Conference Papers (Reviewed Paper, [First Author](#)):

- Hao Shi**, Yuchun Shu, Longbiao Wang, Jianwu Dang, Tatsuya Kawahara, "Fusing Multiple Bandwidth Spectrograms for Improving Speech Enhancement," in Proc. of APSIPA ASC, 2022, pp. 1935–1940.
- Hao Shi**, Longbiao Wang, Sheng Li, Jianwu Dang, 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, Tatsuya Kawahara, "Monaural speech enhancement based on spectrogram decomposition for convolutional neural network-sensitive feature extraction," in Proc. of Interspeech, 2022, pp. 221–225.
- Hao Shi**, Longbiao Wang, Sheng Li, Cunhang Fan, Jianwu Dang, Tatsuya Kawahara, "Spectrograms Fusion-based End-to-end Robust Automatic Speech Recognition," in Proc. of APSIPA ASC, 2021, pp. 438–442.
- Hao Shi**, Longbiao Wang, Meng Ge, Sheng Li, Jianwu Dang, "Spectrograms Fusion with Minimum Difference Masks Estimation for Monaural Speech Dereverberation," in Proc. of ICASSP, 2020, pp. 7544–7548.
- Hao Shi**, Longbiao Wang, Sheng Li, Chenchen Ding, Meng Ge, Nan Li, Jianwu Dang, Hiroshi Seki, "Singing Voice Extraction with Attention based Spectrograms Fusion," in Proc. of Interspeech, 2020, pp. 2412–2416.

Conference Papers (Reviewed Paper, [Corresponding Author](#)):

- Tongtong Song, Qiang Xu, Meng Ge, Longbiao Wang, **Hao Shi**, Yongjie Lv, Yuqin Lin, and Jianwu Dang, "Language-specific Characteristic Assistance for Code-switching Speech Recognition," in Proc. of Interspeech, 2022, pp. 3924–3928.

2. Qiang Xu, Tongtong Song, Longbiao Wang, **Hao Shi**, 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.

Conference Papers (Reviewed Paper, Joint first author, equal contribution):

1. Luya Qiang, **Hao Shi**, 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.
2. Haoran Yin, **Hao Shi**, 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.

Conference Papers (Reviewed Paper, Co-Author):

1. Yanbing Yang, **Hao Shi**, 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. xx–xx.
2. Meng Ge, Longbiao Wang, Nan Li, **Hao Shi**, Jianwu Dang, Xiangang Li, "Environment-dependent attention-driven recurrent convolutional neural network for robust speech enhancement," in Proc. of Interspeech, 2019, pp. 3153–3157.

MANUSCRIPTS

Conference Papers (First Author):

1. **Hao Shi**, Masato Mimura, Longbiao Wang, Jianwu Dang, Tatsuya Kawahara, "Time-domain Speech Enhancement Assisted by Multi-resolution Frequency Encoder And Decoder," (Under review, ICASSP 2023).

Conference Papers (Corresponding Author):

1. Qiang Xu, Tongtong Song, Longbiao Wang, **Hao Shi**, Yuqin Lin, Qiang Yu, Jianwu Dang, "Inter-layer Soft Knowledge Distillation for Speech Recognition," (Under review, ICASSP 2023).
2. Tongtong Song, Qiang Xu, Haoyu Lu, Longbiao Wang, **Hao Shi**, Yuqin Lin, Yanbing Yang, Jianwu Dang, "Monolingual Recognizers Fusion for Code-switching Speech Recognition," (Under review, ICASSP 2023).

REVIEWER

Lead Reviewer:

1. ICASSP 2023

Secondary Reviewer:

1. ICASSP 2022
2. Interspeech 2022
3. Interspeech 2021
4. APSIPA ASC 2020
5. ICASSP 2020