

EDUCATION

•Fudan University

Bachelor in Software Engineering. Advisor: Prof. Bihuan Chen

September 2019 - June 2024

Shanghai, China

•The Chinese University of Hong Kong, Shenzhen

PhD in Computer Science. Advisor: Prof. Zhizheng Wu

Research Topics: Audio Enhancement, Music Generation.

September 2024 - Present

Shenzhen, China

SELECTED PUBLICATIONS

Full list at [Google Scholar](#), *Equal Contribution.

- **Junan Zhang**, Jing Yang, Zihao Fang, Yuancheng Wang, Zehua Zhang, Zhuo Wang, Fan Fan, Zhizheng Wu. ANYENHANCE: A Unified Generative Model with Prompt-Guidance and Self-Critic for Voice Enhancement *Accepted*. *IEEE Transactions on Audio, Speech, and Language Processing (TASLP)*, 2025.
- **Junan Zhang**, Xueyao Zhang, Jing Yang, Yuancheng Wang, Fan Fan, Zhizheng Wu. Multi-Metric Preference Alignment for Generative Speech Restoration *Accepted*. *AAAI Conference on Artificial Intelligence (AAAI)*, 2026.
- **Junan Zhang***, Yunjia Zhang*, Xueyao Zhang, Zhizheng Wu. ANYACCOMP: Generalizable Accompaniment Generation via Quantized Melodic Bottleneck *Under review*. *IEEE International Conference on Acoustics, Speech, and Signal Processing (ICASSP)*, 2026.
- **Junan Zhang***, Yunjia Zhang*, Shaohan Jiang*, Jing Yang, Fan Fan, Zhizheng Wu. SINGVERSE: A Diverse, Real-World Benchmark for Singing Voice Enhancement Evaluation *Under review*. *IEEE International Conference on Acoustics, Speech, and Signal Processing (ICASSP)*, 2026.
- Xueyao Zhang, **Junan Zhang**, Yuancheng Wang, Chaoren Wang, Yuanzhe Chen, Dongya Jia, Zhuo Chen, Zhizheng Wu. Vevo2: Bridging Controllable Speech and Singing Voice Generation via Unified Prosody Learning *Under review*. *IEEE Transactions on Audio, Speech, and Language Processing (TASLP)*, 2025.
- Yicheng Gu, **Junan Zhang**, Chaoren Wang, Jerry Li, Zhizheng Wu, Lauri Juvela. Aliasing Free Neural Audio Synthesis *Under review*. *IEEE Transactions on Audio, Speech, and Language Processing (TASLP)*, 2025.
- Yuancheng Wang, Jiachen Zheng, **Junan Zhang**, Xueyao Zhang, Huan Liao, Zhizheng Wu. Metis: A Foundation Speech Generation Model with Masked Generative Pre-training *Accepted*. *Annual Conference on Neural Information Processing Systems (NeurIPS)*, 2025.
- Yicheng Gu*, Chaoren Wang*, **Junan Zhang***, Xueyao Zhang, Zihao Fang, Haorui He, Zhizheng Wu. SingNet: Towards a Large-Scale, Diverse, and In-the-Wild Singing Voice Dataset *Technical Report*, 2024.
- Junyan Ye*, Baichuan Zhou*, Zilong Huang*, **Junan Zhang***, Tianyi Bai, Hengrui Kang, Jun He, Honglin Lin, Zihao Wang, Tong Wu, Zhizheng Wu, Yiping Chen, Dahua Lin, Conghui He, Weijia Li. LOKI: A Comprehensive Synthetic Data Detection Benchmark using Large Multimodal Models *Accepted (Spotlight)*. *International Conference on Learning Representations (ICLR)*, 2025.

RESEARCH EXPERIENCE

•CCF Advanced Audio Technology Competition 2025

Lead Organizer (Track 1: Speech Restoration)

June 2025 - August 2025

Shenzhen, China

- Spearheaded the complete technical design and execution of the Speech Restoration track for CCF Advanced Audio Technology Competition (AATC 2025) hosted by the China Computer Federation (CCF).
- Engineered and open-sourced the official baseline system based on the ANYENHANCE model, providing a robust and competitive starting point for participating teams.
- Defined the competition's entire technical architecture, from problem formulation and data curation to the comprehensive evaluation protocol (incl. DNSMOS, PESQ, complexity constraints), attracting over 100 registered teams.

•CUHKSZ Speech Lab

PhD Research Assistant, supervised by Prof. Zhizheng Wu

July 2024 - Present

Shenzhen, China

- Led the development of ANYENHANCE, a unified generative model for enhancing both speech and singing voices, capable of handling tasks such as denoising, dereverberation, declipping, super-resolution, and target speaker extraction simultaneously without fine-tuning.
- Introduced a prompt-guidance mechanism for in-context learning, both improving enhancement performance and enabling target speaker extraction without architectural changes. And a self-critic mechanism into enhancing masked generative process.
- Conducted extensive experiments demonstrating that ANYENHANCE outperforms existing methods in both objective metrics and subjective listening tests across various enhancement tasks.
- Published a first-author paper to TASLP 2025.

•Shanghai AI Lab

Undergraduate Research Intern

January 2024 - June 2024

Shanghai, China

- Co-developed a comprehensive benchmark for detecting synthetic data via large multimodal models.
- Conducted extensive research on synthetic datasets and open/closed-source models across four domains: speech, singing, audio, and music. Collected and generated 2,280 real/synthetic parallel audios, curated corresponding QA datasets for benchmarking.
- Evaluated 7 audio multimodal large models, including Qwen-Audio, SALMONN, AnyGPT, and Gemini-1.5-Flash.
- Discovered that current audio multimodal large models struggle to distinguish synthetic audio from real audio (all ~50% accuracy), with this benchmark forming the foundation of the LOKI project, accepted by ICLR 2025 (Spotlight).

•Software Engineering Lab, Fudan University

Undergraduate Research Assistant, supervised by Prof. Bihuan Chen

May 2022 - June 2023

Shanghai, China

- Led a project leveraging pre-trained language models (BERT, RoBERTa) and malicious behavior sequence modeling to detect malicious packages in PyPI/NPM, outperforming state-of-the-art methods by 10.8% in precision and 7.3% in recall.
- Discovered 1,482 novel malicious packages in the wild (683 PyPI, 799 NPM), leading to 707 official acknowledgements from the PyPI and NPM security teams.
- Published the complete methodology and findings as a first-author paper in TOSEM 2024.

PROJECT EXPERIENCE

•Amphion Development

Core Contributor

November 2023 - Present

- Acted as a core contributor to Amphion, an audio generation toolkit with over 9,000 GitHub stars.
- Led the development and integration of multiple key modules, including Speech Enhancement (SE), Singing Voice Conversion (SVC), and Text-to-Music (TTM) generation.
- Contributed two original, novel models to the framework: ANYENHANCE for unified voice enhancement and ANYACCOMP for accompaniment generation.
- Implemented and integrated a VITS-based model for singing voice conversion.

HONORS & AWARDS

•Merit Fellowship for PhD Students, The Chinese University of Hong Kong, Shenzhen September 2024

•Scholarship for Outstanding Students, Fudan University (Top 20%) November 2023, November 2022

SKILLS

Programming Languages: Python, Bash, LaTeX

Developer Tools: Linux, Git

Framework: PyTorch, Hugging Face, Transformers

Language: Chinese(Native), English(Fluent, TOEFL 102(S 23))

Music Production: Semi-Professional, over 2,000,000 streams on Netease Cloud Music