Zhongweiyang Xu

https://xzwy.github.io/alanweiyang.github.io/

EDUCATION

• University of Illinois Urbana-Champaign

MS/PhD Student in Electrical and Computer Engineering; GPA: 4.00

Champaign, IL

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Aug. 2021 – May 2026

• University of Illinois Urbana-Champaign

Bachelor of Science in Computer Engineering; GPA: 3.97

Champaign, IL *Aug.* 2018 – *May.* 2021

RESEARCH OVERVIEW AND INTEREST

- Current: My research interest is mostly involved with the combination of multimedia signal processing and machine learning in general. I have worked on speech separation/enhancement a lot and also have decent knowledge about other speech related technologies like speech recognition, voice conversion, speech synthesizing and audio-visual related topics.
- Future: I plan to step more into speech technologies and also involve other sensors like IMU Motion sensors into multimedia applications. I also intend to design more efficient models which are realistic in real-world applications like hearing aids, video conferencing, and augmented reality.

Industry Experience

• Tencent Ethereal Lab Research Scientist Intern Shenzhen, China

May 2021 - Aug. 2021

• KiNN: Develop a Knowledge-inspired Nerual Netork (KiNN) for real-time computationally efficient ultra-lightweight model for speech enhancement. This model is designed for deployment in the Tencent Meeting's (most popular video conferecing app in China) speech AI codec.

RESEARCH EXPERIENCE

• UIUC SyNRG Lab

Champaign, IL

Graduate Research Assistant Advised by Prof. Romit Roy Choudhury

May 2021 - current

- Learning to Sparate Voices by Spatial Regions (First Author Paper Accepted to ICML2022):
 - 1. Propose a spatial **region-wise** separation mechanism for **binaural** speech applications like audio augmented reality and hearing aids.
 - 2. Use traditional clustering as a teacher model to allow **self-supervised** training for binaural spatial separation, so that personalized model (with personalized HRTF encoded) can be trained simply from personal recordings.
 - 3. Our model shows superior results comparing with using a general model.

• UIUC Statistical Speech Technology Group Lab

Champaign, IL

Advised by Prof. Mark Hasegawa-Johnson

May 2021 - Feb 2022

• Dual-path Attention is All You Need for Audio-Visual Speech Extraction (in preparation for ICASSP2023):

Propose a dual-path multi-modal attention model for audio-visual speech extraction for better multi-modal fusion. Our model is able to achieve 17+ dB for SI-SNRi on LRS3 dataset (in the wild) for 2-5 number of speakers and achieves SOTA performance which exceeds all previous work by a large margin (> 3dB).

Teaching Experience

• UIUC CS/ECE374 SP2021: Introduction to Algorithms & Models of Computation

PUBLICATIONS

• ACM ICML2022: Zhongweiyang Xu, Romit Roy Choudhury, "Learning to Sparate Voices by Spatial Regions"

PROJECTS

- Linux Kernel Project: Develop an OS Kernel for the OS Class.
- BeatDance on FPGA: Develop a GuitarHero-like game on a Cyclone-V FPGA.
- Two Factor Authentication using Earphones: Use personal earphone's production imperfection as a source for 2FA.
- RISC-V CPU: Use SystemVerilog to develop a RISC-V CPU with cache and pipeline.
- Feature Visualization Survey and Experiment: Write a survey about feature visualization techniques for vision neural networks and also gives several propositions regarding how to interpret features for audio-related models. Experiment on simple visualization of audio separation networks.

SKILLS

- Languages Python, C, C++, SystemVerilog
- Coursework and Skills Machine Learning, Deep Learning, Multimedia Signal Processing, Random Process, Generative AI, Computer Vision, NLP, Optimization, Adaptive Signal Processing, Information Theory, Wireless Sensing, Digital System Design, Operating System.