KaitongZheng 郑凯桐

Tel: 15959520089 Email: a593170280@gmail.com Age: 27

Nationality: China Education: Master

Personal website: https://kaitongzheng.github.io/

Research Interests: Deep learning and signal processing for audio applications

Education

> Xiamen University

College of ocean and earth sciences Marine Technology Bachelor 2015.9-2019.7

Rank: Top 1% in Class GPA:3.59

> Institute of Acoustics, Chinese Academy of Sciences

Key laboratory of noise and vibration control Signal Processing Master 2019.9-2022.7

Advisor: Chengshi Zheng, Xiaodong Li GPA:3.57

Career

► IFlytek, Suzhou (Junior audio algorithm engineer)

2022.7-Now

Publications

Zheng K, Zheng C, Sang J, et al. Noise-robust blind reverberation time estimation using noise-aware time–frequency masking [J]. Measurement, 2022, 192: 110901.

Zheng K, Meng R, Zheng C, et al. EmotionBox: A music-element-driven emotional music generation system based on music psychology [J]. Frontiers in Psychology, 2022: 5189.

Chen Y, **Zheng K**, Fang X, et al. QMCR: A Q-learning-based multi-hop cooperative routing protocol for underwater acoustic sensor networks[J]. China Communications, 2021, 18(8): 224-236. (as an undergraduate student)

Zheng K, Zheng C, Sang J, et al. A room impulse response generator and its application on blind reverberation time estimation [J]. Applied Acoustics (Chinese Version) 2022:1-12

Projects

> Road noise control (RNC) in automotive

2022.7-Now

Goal: To reduce or eliminate unwanted road noise inside vehicles.

Methods: Implementation of multi-channel Filtered-x Least Mean Squares (FxLMS) algorithms and their variants in both time and frequency domains. This involves secondary path identification and resampling algorithms.

> Noise-dependent music gain compensation in automotive

2022.7-2024.7

Goal: To maintain a consistent listening experience for in-car audio systems, even in noisy environments. Methods: Implementation of multi-channel echo cancellation, noise estimation, and volume mapping algorithms to compensate for media volume in noisy music scenarios.

Blind estimation of room acoustic parameters based on deep learning (Master thesis) 2020.10-2022.7 Goal: To accurately estimate room acoustic parameters (e.g., reverberation time) from noisy and reverberant

speech signals without requiring prior knowledge of the room or signal.

Methods: Utilization of deep learning models to analyze audio signals and extract relevant acoustic features.

Emotional music generation using Recurrent Neural Network

2020.5-2020.10

Goal: To create music that evokes specific emotions using deep learning models.

Methods: Employment of recurrent neural networks to generate music sequences with desired emotional characteristics.

Research on room equalization methods

2020.1-2020.5

Goal: To improve the sound quality in a room by compensating for frequency response variations caused by multi-path effects.

Methods: Application of sound field equalization techniques to counteract the negative impacts of room acoustics.

Skill

- > Proficient in acoustic signal processing, adaptive filtering, beamforming, and deep learning.
- Familiar with basic in-car audio, spatial audio, speech front-end, and active noise cancellation algorithms.
- > Proficient in MATLAB and Python, and PyTorch framework.
- Familiar with C language, VST plugin development, and basic acoustic experiments.
- ➤ IELTS score of 7 (L8 R7.5 W6.5 S6).
- > Skilled in guitar playing, with a good understanding of music theory and a strong sense of music.

Patents

Zheng C, **Zheng K**, Sang J, et al. Noise robust blind reverberation time estimation method based on deep neural network. CN: CN114255780A, 2022-03-29.

Zheng K, Xia L, Hu M, et al. Noise-following gain method and device, vehicle-mounted system, electronic equipment and storage medium. CN: CN115862657B, 2023-07-28.

Zheng K, Xia L, Hu M, et al. Volume compensation method and device for in-vehicle media source and vehicle. CN: CN115938389B, 2023-07-28.