

Parthasaarathy Sudarsanam

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EDUCATION

Tampere University

Doctoral Researcher

Tampere, Finland

Sep. 2023 – Present

Tampere University

Master of Science in Machine Learning

Tampere, Finland

Aug. 2020 – Jan 2023

Anna University, Madras Institute of Technology

Bachelor of Engineering in Electronics and Communication

Chennai, India

Aug. 2013 – May 2017

WORK EXPERIENCE

Graduate research assistant

Tampere University

June 2021 – Aug 2023

Tampere, Finland

- Involved in data collection and organization of sound event localization and detection challenge in DCASE 2022.
- Developed machine learning models and dataset for audio question answering.

Senior software engineer - Deep learning

Sony India Software Center

Aug 2017 – July 2020

Bangalore, India

- Research engineer in auditory intelligence, worked in collaboration with the Audio Technology Department of Sony Corporation, Tokyo.
- Research domains include speech source separation, voice conversion, and music source separation.

RECENT PUBLICATIONS

- K. Shimada, A. Politis, P. Sudarsanam et al., **STARSS23: An audio-visual dataset of spatial recordings of real scenes with spatiotemporal annotations of sound events**, in Advances in Neural Information Processing Systems (NeurIPS), 2024.*
- P. Sudarsanam, and T. Virtanen, **Attention-Based Methods For Audio Question Answering**, in 30th European Signal Processing Conference (EUSIPCO), 2023.*
- S. Lipping, P. Sudarsanam, K. Drossos, and T. Virtanen, **Clotho-AQA: A Crowdsourced Dataset for Audio Question Answering**, in 30th European Signal Processing Conference (EUSIPCO), 2022.*
- P. Sudarsanam, A. Politis, and K. Drossos, **Assessment of self-attention on learned features for sound event localization and detection**, in 6th workshop on Detection and Classification of Acoustic Scenes and Events (DCASE), 2021.*
- N. Takahashi, M. K. Singh, S. Basak, P. Sudarsanam, Y. Mitsufuji, and S. Ganapathy, **Improving Voice Separation by Incorporating End-to-end Speech Recognition**, in 45th International Conference on Acoustics, Speech, & Signal Processing (ICASSP), 2020.*
- N. Takahashi, P. Sudarsanam, N. Goswami, and Y. Mitsufuji, **Recursive speech separation for unknown number of speakers**, in Interspeech, 2019.*

TECHNICAL SKILLS

Languages: Python, C/C++, Matlab.

Frameworks: PyTorch, Tensorflow, Keras.

Developer Tools: Git, Pycharm, Visual studio.

PROJECTS

Smart Camera using AI

- Developed a prototype of an intelligent camera that analyzes scenes and clicks photos automatically.
- Pre-trained object detection and image captioning models were used for scene analysis.
- It can automatically detect and click birthday parties, playing with pets, marriages, etc.

Selective active noise cancellation using deep learning

- Developed a prototype for noise-canceling headphones to selectively cancel ambient noise sources while allowing important sounds like sirens, fire alarms, etc, to pass through.
- The audio detection model was trained on Google's Audioset dataset.