

## VIET ANH TRINH

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### Education

- 2016 - Present    Ph.D. in Computer Science, The Graduate Center, The City University of New York (CUNY), US
- Research interests: Machine Learning, Speech and Language Processing and Artificial Intelligence
  - Advisor: Professor Michael I Mandel
- 2003 - 2008    B.S. in Electronics and Telecommunications, Hanoi University of Science and Technology, Viet Nam

### Technical Skills

Python, Matlab, C, C++ , PHP, Java, Visual Basic, R, MySQL, HTML  
PyTorch, Tensorflow, Keras, Kaldi, NLTK, Moses, Message Passing Interface

### Publications

- Conference    V. A. Trinh, B. McFee, and M. I. Mandel, "Bubble cooperative networks for identifying important speech cues," in *Proceedings of Interspeech*, 2018.
- A. R. Syed, V. A. Trinh, and M. I. Mandel, "Concatenative resynthesis with improved training signals for speech enhancement," in *Proceedings of Interspeech*, 2018.

### Research experience

- 2019    Directly comparing the listening strategies of humans and machines
- Compared human listeners with automatic speech recognitions (ASR) in terms of their strategies for recognizing speech in noise
  - ASR (Kaldi): time-delay neural network - long short-term memory networks (TDNN-LSTM) and Gaussian mixture model - hidden Markov models (GMM-HMM) acoustic models
- 2019    Grammatical Error Correction
- Implemented a multilayer convolutional encoder-decoder neural network model and augmented the available training data, using the approach of minimally-augmented grammatical error correction.
- 2018    Bubble cooperative networks for identifying important speech cues
- Developed a network consisting of a generator (LSTM) and a discriminator (LSTM) to identify important time-frequency regions of speech
  - The predicted masks show patterns that are similar to analyses derived from human listening tests, but with better generalization and less context-dependence than previous approaches

- 2017      Concatenative analysis-by-synthesis
- Utilized pitch and intensity information to improve the performance of a feed-forward neural network unit-selection in a concatenative speech synthesizer system. This system aims to produce a high-quality clean speech from noisy speech for the task of source separation and speech enhancement
- 2016      Multi-channel speech enhancement
- Reviewed literature and deployed a baseline method, which estimates noise covariance matrix for the beamforming to improve far-field speech recognition

## Work experience

- 2019      **Amazon**, Applied Scientist Intern, US
- New project - Non-Disclosure Agreement
- 2016 - Present      **Research Foundation - CUNY**, Research Assistant, US
- Worked with my advisor in three projects: Bubble cooperative networks, multi-channel speech enhancement and concatenative analysis-by-synthesis
- 2011 - 2016      **Texas Instruments(TI)** Technical Business Development Engineer, Vietnam
- Managed TI North Vietnam sale and increased revenue by 250% in 2012, 27% in 2013, 69% in 2014, 150% in 2015 and 30% in 2016
  - Conducted bi-weekly review with distributors: Avnet, Arrow, SS, WT and WPI to achieve sale targets
  - Recommended TI solutions and products (integrated circuit) to build electronic devices: smart phone, telecom base station, set top box, smart home devices, car GPS tracking and toy robots
  - Received reward letter from TI Asia President for achievement in 2016
- 2008 - 2011      **Viettel Technologies**, Technical Team Leader, Vietnam
- Led team to build up and propose video conferencing and network solutions to customers

## Related Courses

Theoretical deep learning (Princeton University), artificial intelligence, machine learning, deep learning, natural language processing, speech and audio understanding, algorithms, stochastic optimization, graph based analysis of social networks, big data and data mining.

## Service

Subreviewer: International Conference on Acoustics, Speech and Signal Processing (ICASSP) 2018 and 2020, International Conference on Learning Representations (ICLR) 2019, International Speech Communication Association (Interspeech) 2019, Neural Information Processing Systems (NeurIPS) 2018, Association for the Advancement of Artificial Intelligence (AAAI) 2018.