# Li-Wei Chen

□ +1 412 636 3999
 ☑ lwchen.jeff@gmail.com
 ☑ github.com/b04901014
 ☑ Google Scholar

## **Research Interests**

- Speech processing, especially voice conversion (VC), text-to-speech synthesis (TTS), self-supervised representation learning (SSL), and speech emotion recognition (SER)
- o NLP tasks including natural language understanding, language modeling and task-oriented conversation
- O Design machine learning systems for the above tasks or improve existing algorithms

## Education

2022–Present **Ph.D. in Language and Information Technology**, *Carnegie Mellon University*, Pittsburgh Sept.

2020–2022 Sept. Aug. **Master of Language Technologies**, *Carnegie Mellon University*, Pittsburgh *Overall GPA – 4.23/4.33* 

2015–2020 Sept. Jan. Bachelor of Electronic Engineering, National Taiwan University, Taipei Overall GPA – 4.10/4.3 – 3.96/4.0

## **Research Experience**

2022–Present <sub>Sept.</sub>	<ul> <li>Ph.D. Student, Carnegie Mellon University, Pittsburgh</li> <li>Advisor: Alexander Rudnicky, Shinji Watanabe</li> <li>Developed an auto-regressive TTS system for real-world spontaneous speech that outperforms existing methods in terms of intelligibility, naturalness, and diversity</li> <li>Proposed a unified system for one-shot voice conversion (VC) on pitch, rhythm, and speaker attributes of speech, outperforming existing works</li> </ul>
2020–2022 Sept. Aug.	<ul> <li>Master's Student, Carnegie Mellon University, Pittsburgh</li> <li>Advisor: Alexander Rudnicky</li> <li>Investigated the importance of temporal context and different features for speech emotion recognition</li> <li>Improved the performance on speech emotion recognition by a large margin by exploring different fine-tuning techniques of self-supervised pretrained models</li> <li>Designed a novel architecture with cross-modality attention mechanism to realize fine-grained style control on the transformer-based text-to-speech synthesis (TTS)</li> </ul>
2018–2018 Aug. Sept.	<ul> <li>Research Intern, Taiwan AI Labs, Taipei</li> <li>Responsible for the TTS part of a smart speaker product, using Tacotron2 for speech synthesize</li> <li>Implemented parallel WaveNet as vocoder replacing WaveNet to significantly reduce latency</li> </ul>
2017–2019 Sept. Sept.	<ul> <li>Undergraduate Student, National Taiwan University, Taipei</li> <li>Advisor: Hung-Yi Lee, Lin-Shan Lee</li> <li>Developed an end-to-end spoken term detection system with attention-based CNN</li> <li>Proposed a model using GANs to transform impaired speech into normal one while preserving linguistic content</li> <li>Achieved better results in terms of Mean Opinion Score compared to existing models</li> <li>Competition Experience</li> </ul>
2021—2022 May. May.	<ul> <li>Alexa Prize Taskbot Competition, Carnegie Mellon University, Pittsburgh</li> <li>Advisor: Alexander Rudnicky</li> <li>Technical team leader of a 8 member team, mainly responsible for the software engineering of the dialogue logic</li> <li>Our system advanced to the semi-finals, see our paper (in publications) for the details of our system</li> </ul>
2021–2021 Jan. May	<ul> <li>Zero Resource Speech Challenge 2021, Carnegie Mellon University, Pittsburgh</li> <li>Advisor: Alexander Rudnicky, Shinji Watanabe</li> <li>Collaborated with Yahoo! JAPAN to develop a algorithm on learning unsupervised speech representation, achieving top result in one of the metrics</li> </ul>

• Paper accepted to Interspeech (see in publications)

## Publications

- [1] L. Chen, S. Watanabe, A. Rudnicky, "A Unified One-Shot Prosody and Speaker Conversion System with Self-Supervised Discrete Speech Units", arXiv preprint arXiv:2211.06535, 2022. (Accepted to ICASSP 2023)
- [2] L. Chen, A. Rudnicky, "Exploring Wav2vec 2.0 fine-tuning for improved speech emotion recognition", arXiv preprint arXiv:2110.06309, 2021. (Accepted to ICASSP 2023)
- [3] P. Wu, L. Chen, C. Cho, S. Watanabe, L. Goldstein, A. Black, G. Anumanchipalli, "Speaker-Independent Acoustic-to-Articulatory Speech Inversion", arXiv preprint arXiv:2302.06774, 2023. (Accepted to ICASSP 2023)
- [4] L. Chen, A. Rudnicky, S. Watanabe, "A Vector Quantized Approach for Text to Speech Synthesis on Real-World Spontaneous Speech", arXiv preprint arXiv:2302.04215, 2023. (Accepted to AAAI 2023 Main Track)
- [5] L. Chen, A. Rudnicky, "Fine-grained style control in Transformer-based Text-to-speech Synthesis", in Proceedings of ICASSP 2022, 2022.
- [6] L. Chen, T. Chi, D. Shah, C. Gomes, J. Bao, K. Ganesan, P. Joshi, S. Kumar, D. Naik, J. Hagerty, A. Rudnicky, "Tartan: A taskbot that assists with recipes and do-it-yourself projects", Alexa Prize TaskBot Challenge Proceedings, 2022.
- [7] Y. Xia\*, L. Chen\*, A. Rudnicky, R. M. Stern, "Temporal Context in Speech Emotion Recognition", in Proceedings of the Interspeech 2021, 2021. (\* Equal contribution)
- [8] T. Maekaku, X. Chang, Y. Fujita, L. Chen, S. Watanabe, A. Rudnicky, "Speech Representation Learning Combining Conformer CPC with Deep Cluster for the ZeroSpeech Challenge 2021", in Proceedings of the Interspeech 2021, 2021.
- [9] L. Chen, H. Lee, and Y. Tsao, "Generative Adversarial Networks for Unpaired Voice Transformation on Impaired Speech", in *Proceedings of the Interspeech 2019*, 2019.

2022–2022 Jan. May	Teaching Assistant, Carnegie Mellon University, Pittsburgh <ul> <li>Course: Multimodal Machine Learning (Course number 11-777, 2022 Spring by Yonatan Bisk)</li> <li>Course Website: https://yonatanbisk.com/teaching/mmml-s22/</li> <li>Graded and advised each team regarding their final project, responsible for teams that involves speech modality in their project</li> </ul>
2018–2018 Jan. July	<ul> <li>Teaching Assistant, National Taiwan University, Taipei</li> <li>Course: Machine Learning and Having it Deep and Structured (2018 Spring by Hung-Yi Lee)</li> <li>Course Website: https://speech.ee.ntu.edu.tw/~hylee/mlds/2018-spring.php</li> <li>Designed and graded assignments of Reinforcement Learning, including Policy Gradient, Deep Q-learning Networks, Actor-Critic, and other improved algorithms of them</li> </ul>

## Computer Skills

Teaching Experience

- Basic JAVASCRIPT, HTML, MATLAB
- Intermediate C#, LTFX, Linux, Bash, Java
  - Advanced PYTHON, C++

    - Others O Speech Toolkits: Kaldi, World, HTK, SPTK • Machine Learning Libraries: Pytorch, Tensorflow, Keras

2/2

Language Proficiency

Native Mandarin Chinese, Taiwanese Hokkien

GRE General Q: 170, V: 161, AWA: 4.0

TOEFL iBT 106 (30/29/22/25)