# Cheng-chieh Yeh

National Taiwan University Speech Processing Lab

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Education

National Taiwan University (NTU), MS in Data Science Group of Communication Engineering	
GPA: 4.30/4.30	2017 - 2019
Advisors: Prof. Lin-shan Lee and Prof. Hung-yi Lee	
Thesis: Multi-target Rhythm-flexible Voice Conversion without Parallel Data (Link)	
National Taiwan University (NTU), BS in Computer Science	
$GPA \ (Last \ 60) : \ 3.73/4.30$	2013 - 2017
Selected Course (A+, all graduate level):	
Introduction to Digital Speech Decogging, Natural Language Decogging, Machine Learning Foundations	

Introduction to Digital Speech Processing, Natural Language Processing, Machine Learning Foundations, Machine Learning, Machine Learning and Having it Deep and Structured, Applied Deep Learning

# **Research Experiences**

## Non-parallel Voice Conversion

NTU Speech Processing Lab

- Dealing with many-to-many speaker pairs conversion with only one model trained, as compared to conventional approaches with many one-to-one speaker pair models, (Publication [1], Award [c])
- Considering the flexibility of rhythmic patterns in voice conversion between speakers with very different speaking rates, (Publication [2])
- One-shot unsupervised voice conversion by separating speaker and content into two embeddings using Instance Normalization (IN) and Adaptive Instance Normalization (adaIN), (Publication [4])

### Text-to-Speech

NTU Speech Processing Lab

- o Github open-sourced project with re-implementation of Google's end-to-end TTS system (GST Tacotron)
- Proposed an end-to-end TTS for low-resource languages by transfer learning, including a learnable Phonetic Transformation Network to map the input representations between different languages, (Publication [3])
- Trying to gain more controllability over unsupervised emotion and style modeling in TTS

### Audio2Text Abstractive Machine Summarization

NTU Speech Processing Lab Undergraduate Research Project

• Built a system which takes audio news stories and generates the headlines, including data collection, and proposing an abstractive machine summarization approach with reduced performance gap caused by ASR errors (Award [d])

# **Working Experiences**

### **Research Internship**

Siri Understanding, Apple Inc.

- Speaker modeling in end-to-end neural speech synthesis, seeking better unseen speaker adaptation results
- Try out different strategies on improving the stability and scalability of end-to-end systems

# **Teaching Experiences**

#### **Teaching Assistant**

• Digital Speech Processing Undergraduate Research Project (Instructor: Lin-shan, Lee), Machine Learning (Instructor: Hung-yi, Lee), Machine Learning and Having it Deep and Structured (Instructor: Hung-yi, Lee), Applied Deep Learning (Instructor: Yun-Nung, Chen)

### Working Group Leader for a National Grand Challenge "Talk to AI"

• A Chinese spoken language understanding competition held by the Ministry of Science and Technology of Taiwan

• Leading a working group of graduate students offering technical support to the nation-wide Kaggle competition for undergraduate and high school students, including collecting data, sharing fundamental techniques and tools, preparing baselines for grading, etc

2017 - present

July 2019 - October 2019

June 2017 - September 2017

2017 - present

# 2015 - 2017

2015 - 2017

# **Publications**

- Multi-target Voice Conversion without Parallel Data by Adversarially Learning Disentangled Audio Representations, J.-C. Chou, C.-C. Yeh, H.-Y. Lee, L.-S. Lee, Interspeech 2018 (selected as one of the 12 BEST STUDENT PAPER FINALIST, award [c]) (Paper link)
- [2] Rhythm-Flexible Voice Conversion without Parallel Data Using Cycle-GAN over Phoneme Posteriorgram Sequences, C.-C. Yeh, P.-C. Hsu, J.-C. Chou, H.-Y. Lee, L.-S. Lee, SLT 2018 (Paper link)
- [3] End-to-end Text-to-speech for Low-resource Languages by Cross-Lingual Transfer Learning, Tao Tu, Yuan-Jui Chen, C.-C. Yeh, H.-Y. Lee, *Interspeech 2019* (Paper link)
- [4] One-shot Voice Conversion by Separating Speaker and Content Representations with Instance Normalization, J.-C. Chou, C.-C. Yeh, H.-Y. Lee, Interspeech 2019 (Paper link)

# **Awards and Grants**

- [a] 2019 The Phi Tau Phi Scholastic Honorable member
- [b] 2018 National Taiwan University Advanced Speech Technologies Scholarship
  o Grants (\$16,000 USD) for students who excel in speech related technologies and researches
- [c] 2018 Interspeech Best Student Paper Finalist: Non-parallel Voice Conversion (Publication [1]) (Paper link)
- [d] 2017 NTU CSIE Undergraduate Research Competition: Audio2Text Summarization (Poster link)
  - Departmental **FIRST PRIZE**, also Appier second prize, Viscovery third prize, Leopard Mobile first prize