

# Ruchao Fan

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## Research Interests

Speech Processing, Self-supervised and Unsupervised Learning, Domain Adaptation, Data augmentation, End-to-end Speech Recognition (Non-autoregressive transformer, CTC, RNN-T), Children's ASR

## Education

**University of California, Los Angeles (UCLA)**

*Ph.D. in Electrical and Computer Engineering*

**Los Angeles, U.S.A.**

*Sept. 2019 - Present*

**Beijing University of Posts and Telecommunications (BUPT)**

*M.S. in Information and Communication Engineering*

*B.Eng. in Communication Engineering*

**Beijing, China**

*Sept. 2016 - Jun. 2019*

*Sept. 2012 - Jun. 2016*

## Professional Experience

**UCLA - Speech Processing and Auditory Perception Lab**

*Graduate Student Researcher - Advisor: Prof. Abeer Alwan*

Research on children's ASR as a low-resource task

- Develop data augmentation and model pre-training methods to improve children's ASR
- Develop non-autoregressive models for child ASR to improve decoding efficiency

**Los Angeles, U.S.A.**

*Sept. 2019 - Present*

**Microsoft Corporation. - MSR Redmond**

*Research Intern - Mentors: Dr. Yiming Wang, Yashesh Gaur, and Jinyu Li*

Research on self-supervised learning algorithms

- Develop a novel CTC training mechanism to solve the misalignment problem.
- Proposed methods consistently improve the performance compared to CE training.

**Redmond, U.S.A.**

*June 2022 - Sept. 2022*

**Microsoft Corporation. - MSR Redmond**

*Research Intern - Mentors: Dr. Guoli Ye and Dr. Jinyu Li*

Research on spell correction for end-to-end speech recognition

- Investigated methods of non-autoregressive spell correction for a transformer-transducer
- Proposed a sampling-based error detection method for spell correction

**Redmond, U.S.A.**

*June 2021 - Sept. 2021*

**PAII Inc. - US Research Lab**

*Research Scientist Intern - Mentors: Dr. Wei Chu and Dr. Peng Chang*

Research on non-autoregressive transformers for end-to-end speech recognition

- Used CTC alignment as extra information for token-level acoustic embedding extraction
- Proposed an error-based sampling method during inference to improve performance

**Palo Alto, U.S.A.**

*June 2020 - Sept. 2020*

**Sogou Inc. - Voice Interaction Technology Center**

*Research Intern - Mentors: Dr. Pan Zhou and Dr. Wei Chen*

Research on attention-based encoder-decoder (AED) end-to-end speech recognition

- Proposed an online AED with 3.5% relative WER degradation compared to an offline AED
- Improved the performance of speech-transformer with parallel-schedule sampling and relative positional encoding

**Beijing, China**

*Apr. 2018 - Aug. 2019*

## Publications

- [16] **R. Fan**, W. Chu, P. Chang, A. Alwan, "A CTC Alignment-based Non-autoregressive Transformer for End-to-end Automatic Speech Recognition", Accepted to Transactions on Audio, Speech and Language Processing.
- [15] **R. Fan**, Y. Wang, Y. Gaur, and J. Li, "CTCBERT: Advancing Hidden-unit BERT with CTC Objectives", Accepted to ICASSP 2023.
- [14] **R. Fan**, G. Ye, Y. Gaur, and J. Li, "Acoustic-aware Non-autoregressive Spell Correction with Mask Sample Decoding", arXiv preprint arXiv:2210.08665.
- [13] **R. Fan**, Y. Zhu, J. Wang, and A. Alwan, "Towards Better Domain Adaptation for Self-Supervised Models: A Case Study of Child ASR," in IEEE Journal of Selected Topics in Signal Processing, vol. 16, no. 6, pp. 1242-1252, Oct. 2022, doi: 10.1109/JSTSP.2022.3200910..
- [12] **R. Fan**, and A. Alwan, "DRAFT: A Novel Framework to Reduce Domain Shifting in Self-supervised Learning and Its Application to Children's ASR", Proc. Interspeech 2022, pp. 4900-4904.
- [11] A. Johnson, **R. Fan**, R. Morris, and A. Alwan, "LPC AUGMENT: An LPC-Based ASR Data Augmentation Algorithm for Low and Zero-Resource Children's Dialects," in ICASSP 2022, IEEE, pp. 8577--8581.
- [10] Y. Zhu, **R. Fan**, and A. Alwan, "Towards Better Meta-Initialization with Task Augmentation for Kindergarten-aged Speech Recognition," in ICASSP 2022, IEEE, pp. 8582--8586.
- [9] G. Yeung, **R. Fan**, and A. Alwan, "Fundamental frequency feature warping for frequency normalization and data augmentation in child automatic speech recognition," Speech Communication, 2021, doi: <https://doi.org/10.1016/j.specom.2021.08.002>.
- [8] **R. Fan**, W. Chu, P. Chang, J. Xiao and A. Alwan, "An Improved Single Step Non-autoregressive Transformer for Automatic Speech Recognition," Proc. Interspeech 2021, pp. 3715-3719.
- [7] J. Wang, Y. Zhu, **R. Fan**, W. Chu and A. Alwan, "Low Resource German ASR with Untranscribed Data Spoken by Non-native Children-INTERSPEECH 2021 Shared Task SPAPL System," Proc. Interspeech 2021, pp. 1279-1283.
- [6] **R. Fan**, W. Chu, P. Chang, and J. Xiao, "CASS-NAT: CTC Alignment-based Single Step Non-autoregressive Transformer for Speech Recognition," in ICASSP 2021, IEEE, pp. 5889-5893.
- [5] **R. Fan**, A. Afshan and A. Alwan, "BI-APC: Bidirectional Autoregressive Predictive Coding for Unsupervised Pre-training and its Application to Children's ASR," in ICASSP 2021, IEEE, pp. 7023-7027.
- [4] G. Yeung, **R. Fan**, and A. Alwan, "Fundamental Frequency Feature Normalization and Data Augmentation for Child Speech Recognition," in ICASSP 2021, IEEE, pp. 6993-6997.
- [3] V. Ravi, **R. Fan**, A. Afshan, H. Lu, and A. Alwan, "Exploring the use of an unsupervised autoregressive model as a shared encoder for text-dependent speaker verification," Proc. Interspeech 2020, pp. 766-770, 2020.
- [2] **R. Fan**, P. Zhou, W. Chen, J. Jia, and G. Liu, "An Online Attention-Based Model for Speech Recognition," in Proc. Interspeech 2019, 2019, pp. 4390-4394.
- [1] **R. Fan** and G. Liu, "CNN-based audio front end processing on speech recognition," International Conference on Audio, Language and Image Processing. IEEE, 2018, pp. 349-354.

## Skills and Coursework

### Computer Languages and Open-source Framework

- Python, C/C++, Shell and Matlab, Pytorch, Tensorflow, Kaldi, Espnet, Fairseq;

### Coursework

- ECE: Matrix Analysis, Digital Speech Processing, Advanced Topics in Speech Processing, Linear Programming; CS: Natural Language Generation, Algorithmic Machine Learning
- Overall GPA: 3.93/4.0

## Honors and Awards

- UCLA-Amazon Science Hub Fellowship, honored as Amazon Fellow (2021-2022)
- One-year research funding from PAII Inc. (2020-2021)
- A first-year graduate fellowship for the ECE Ph.D. program at UCLA (2019-2020)
- Huawei Enterprise Scholarship (2017)
- Graduate Academic Scholarship (2016-2018)
- Undergraduate National Inspirational Scholarship (2013-2015)