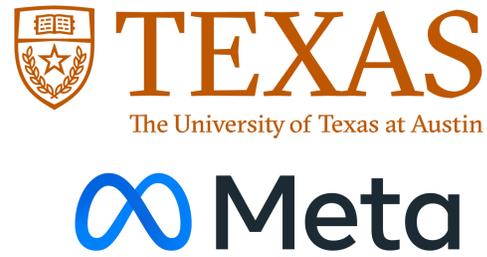


Continual Learning for On-Device Speech Recognition using Disentangled Conformers



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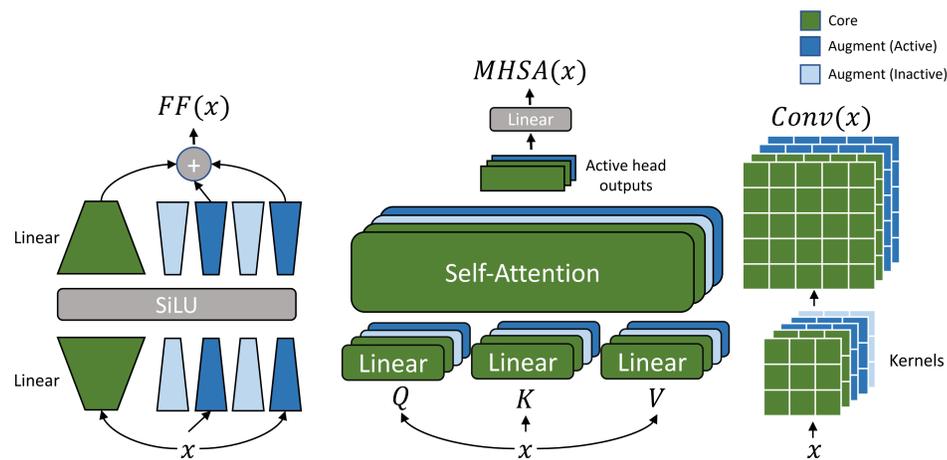
1. Overview

ASR models deployed in households encounter ever-changing speaker distributions. Given a base ASR model (trained on a general-purpose dataset), we would like to build and evaluate models that can **continually** adapt as new speaker-specific data is received, in an **efficient** manner (for on-device adaptation). Our contributions are two-fold:

Evaluation: Our **LibriContinual** ASR benchmark

Modelling: Our **DisConformer** model with **NetAug** for Base ASR training and **DisentangledCL** for Continual Learning

3. DisConformer



DisConformer splits the parameters of the FFN, Self-Attention and Conv modules of the Conformer into **core** and **augment** parameters.

1. Base ASR Training with NetAug

Pass inputs through just core (term 1) as well as core + random subset of augment (term 2)

$$\tilde{W}_{\text{aug}} \subseteq_R W_{\text{aug}}$$

$$L(\mathcal{M}, x, y) = \text{CTC}(\mathcal{M}(W_{\text{core}}, x), y) + \alpha \text{CTC}(\mathcal{M}([W_{\text{core}}, \tilde{W}_{\text{aug}}], x), y)$$

2. Continual Learning with LibriContinual

Freeze core, finetune only a fixed, small, random subset of augment

$$W'_{\text{aug}} \subseteq_R W_{\text{aug}}$$

$$L(\mathcal{M}, x, y) = \text{CTC}(\mathcal{M}([W_{\text{core}}, W'_{\text{aug}}], x), y)$$

Use W_{core} for general-purpose and $[W_{\text{core}}, W'_{\text{aug}}]$ for speaker-specific ASR

2. LibriContinual & Evaluation Metrics

What is it?

Data Source: 118 diff. speakers reading LibriVox books; transcripts generated by wav2vec2.0
Data Splits: Train: 10m, 30m, 1h, 2h, 5h, 10h ; Val: ~3.13h ; Test: ~2.66h for every speaker
 Increasingly-sized train data simulates continual interaction

Evaluation Framework

- Base ASR Training:** Train a base ASR model M on a general-purpose dataset (Librispeech)
- Continual Learning:** Given a continual learning algorithm A , run it on the base ASR model using the LibriContinual train set of every speaker s to obtain 118 different ASR models $M^{(s)}$

Evaluation Metrics

- #Params:** # Avg. trainable parameters modified by the CL algorithm A (proxy for **efficiency**)
- LibriContinual WER:** Median WER of model $M^{(s)}$ on its respective speaker s test set
- Librispeech WER:** Median WER of model $M^{(s)}$ on Librispeech; tests catastrophic forgetting

4. Key Results

DisCo-* models disentangle each module type individually. **Base-* baselines** are DisCo-* models with just the **core**

1. NetAug trains better base ASR models

Model	LibriSpeech		LibriContinual	
	test-c	test-o	val	test
Base-FF	4.02	10.16	7.92	8.36
DisCo-FF	3.75	9.82	7.41	7.82
Base-Att	3.42	8.54	6.40	6.76
DisCo-Att	3.29	8.22	6.08	6.34
Base-Conv	3.50	8.62	6.88	7.22
DisCo-Conv	3.28	8.19	6.66	6.94

Metric: Word Error Rate

- DisCL outperforms CL baselines on Librispeech**
- DisCL outperforms parameter-matched CL baselines, and even performs as well as fully-finetuned baselines on LibriContinual**

5. Conclusion & Future Work

LibriContinual reveals that current base ASR models underperform on speaker-specific data and current baseline CL algorithms are parameter-inefficient and catastrophically forget general-purpose data; on the other hand, our **DisConformer** with **NetAug** and **DisCL** is parameter-efficient and has high performance across the board! We invite future work on continual learning in absence of labelled data, multi-speaker adaptation, and more!

Full-FT: Fully finetune the model
KD: Full-FT + KL divergence(current model $M^{(s)}$, init model M)
***-Eff:** Efficient versions that only finetune top 1-2 layers

All models finetuned on 1hr split, decoded using 4-gram LM (for other settings, see paper!)

