

Simple and Effective Unsupervised Speech Synthesis Alexander H. Liu^{1*}

Introduction

Background

- Supervised methods have achieved great success in text-to-speech (TTS) synthesis [1].
- However, the success relied heavily on the amount and the quality of labeled training data [2].

Our goal

Speech synthesis without labeled training data.

Key idea

Leveraging recent advance in unsupervised ASR to create pseudo-labeled training data.

Setup Experiment conducted on LJSpeech[5], a benchmark dataset with about 24 hours of read English speech from single female speaker. Text transcription is not used for unsupervised model. (Multi-speaker result on LibriTTS available in paper.)

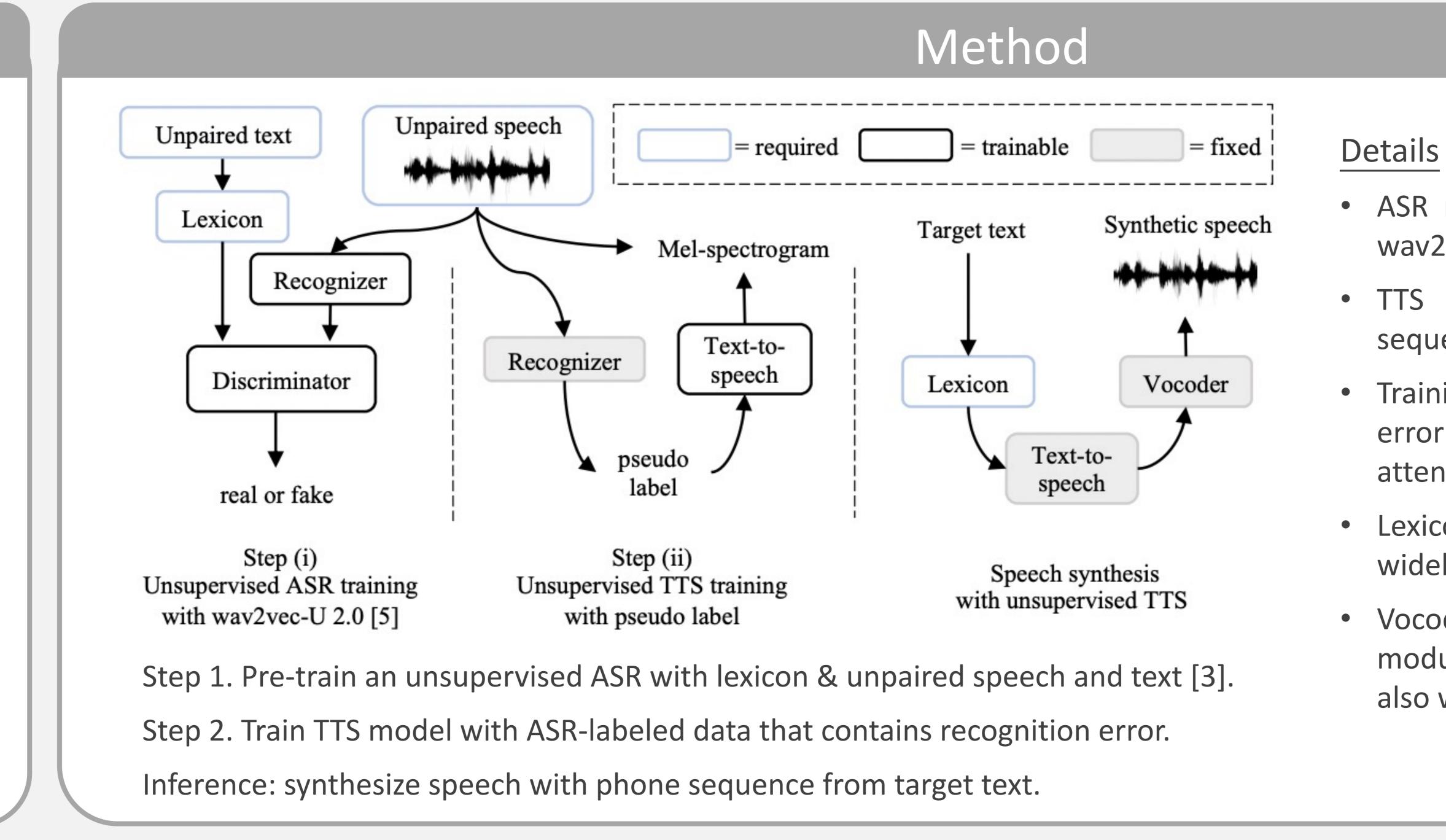
Highlights

Preference Test (human evaluation)

Mean Opinion Score (human evaluation)

	Preference or	Metho		
	Naturalness	Intelligibility		
Unsupervised	50.2%	54.0%	Natura Superv	
	Demo p (sample	bage es & code)	Unsup	

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Results

• The quality of synthetic speech from unsupervised model matches supervised method in human evaluation. Unsupervised model have slightly worse intelligibility when measured by machines.

• The TTS performs better with raw text input despite learning from imperfect pseudo-labeled data.

Method	Input phone error rate suffered during training	MOS		Method	Source of input phone sequence for synthesize	Word error rate (%)
Natural Supervised	- 0%	4.05 ± 0.07 3.94 ± 0.08 3.91 ± 0.08		Natural Supervised	- text [†]	18.0 19.2
Unsupervised	6.97%		Unsupervised	text [†] ASR transcription [‡]	21.7 22.0	

Intelligibility Test (commercial ASR evaluation)

Key contribution

First unsupervised TTS: with simple and effective method, we show that training TTS without human-labeled data is feasible.

Future directions

- End-to-end training

[1] Natural TTS synthesis by conditioning Wavenet on Mel Spectrogram predictions, Shen et al., 2021 [2] Semi-supervised training for improving data efficiency in end-to-end speech synthesis,

- Chung et al., 2017
- attention, Tachibana et al., 2018

[5] The LJ speech dataset, Ito et al., 2017 Acknowledgement We thank Tomoki Hayashi and Erica Cooper for their advice on TTS training and evaluation. This research was supported in part by the MIT-IBM Watson AI Lab.

Meta Al

• ASR model: 2-layer CNN with input wav2vec2.0 feature.

• TTS model: 12-layer sequence-tosequence Transformer.

Training objective: L2 reconstruction error on Mel-spectrogram + guided attention loss [4].

• Lexicons defined by linguistics are widely used in existing TTS systems.

Vocoder (spectrogram-to-waveform module) is trained with speech only, also widely used in TTS systems.

Conclusion

Generalize to low-resource languages where unsupervised methods are preferred.

Reference

[3] Towards end-to-end unsupervised speech recognition, *Liu et al., 2022*

[4] Efficiently trainable text-to-speech system based on deep convolutional networks with guided