

Mobile PresenTra:

NICT Fast Neural Text-To-Speech System on Smartphones With Incremental Inference of MS-FC-HiFi-GAN for Low-Latency Synthesis



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Press release

- 21-language, fast and high-fidelity neural text-to-speech
- Synthesizing one second of speech at high speed in only 0.1 seconds using a CPU
- Fast synthesis with a latency of 0.5 seconds on a smartphone without network connection



<https://www.nict.go.jp/en/press/2024/07/26-1.html>

Hi-Fi-CAPTAIN corpus

High-fidelity and high-capacity conversational speech synthesis corpus developed by NICT

- 1 female and 1 male (English): 14K utts (parallel: 13K)
- 1 female and 1 male (Japanese): 19K utts (parallel: 18.5K)

ESPnet2-TTS recipe for JETS-based E2E-TTS



<https://ast-astrec.nict.go.jp/en/release/hi-fi-captain/>

1. Introduction

- Background
 - Fast and High-fidelity neural text-to-speech (TTS) can be realized
 - NICT developed 21-language neural TTS models and implemented them on VoiceTra

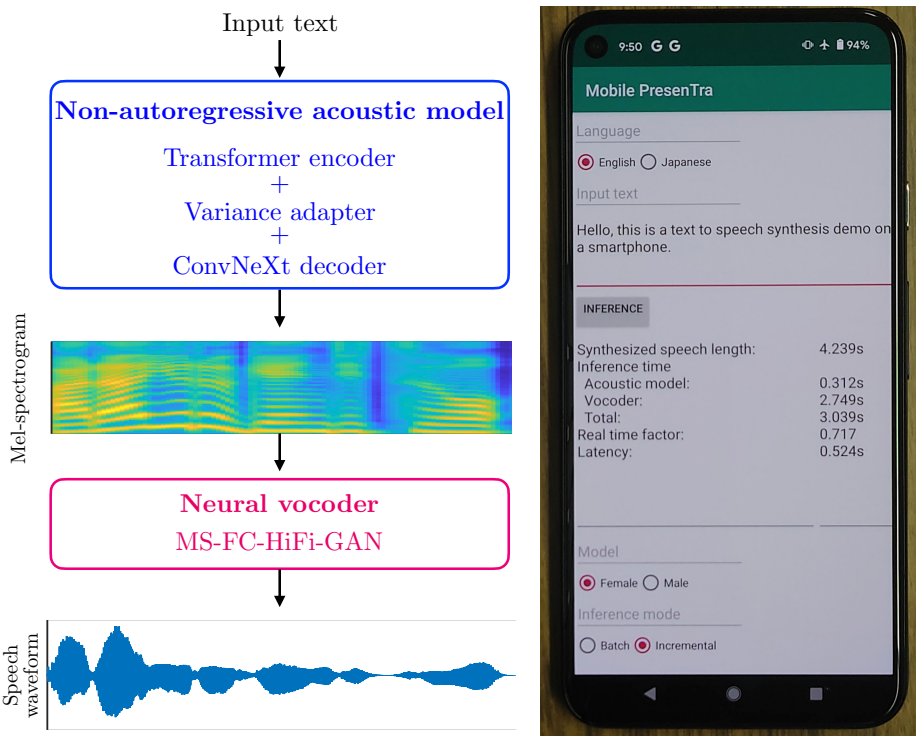


<https://voicetra.nict.go.jp/en/index.html>

- Typical neural TTS systems are run on servers and require network connectivity
- Purpose
 - Fast and High-fidelity neural TTS models working on edge devices are required

2. Prototyped Mobile PresenTra

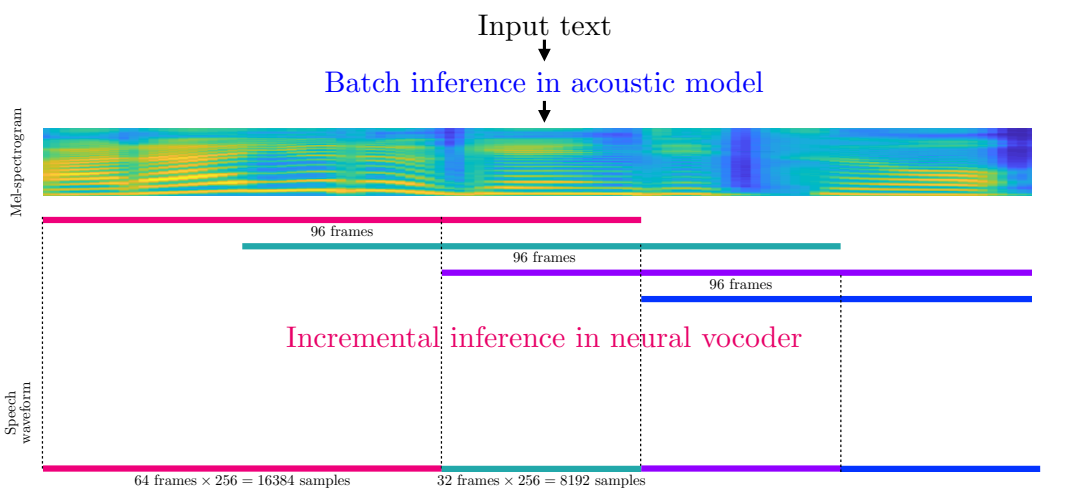
- Fast and high-fidelity neural TTS model on smartphones @ 24 kHz
 - Trained on PyTorch and implemented to smartphones using LibTorch and C++
 - Inference time, real-time factor and latency can be displayed



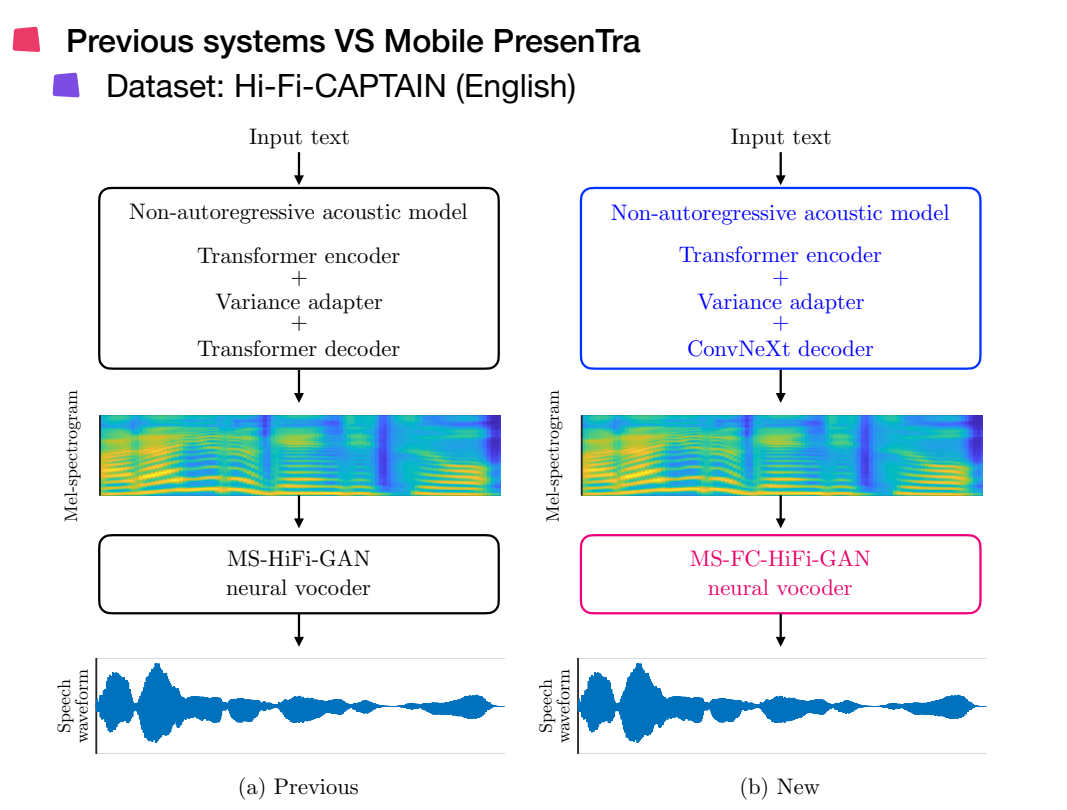
- ConvNext-TTS & ConvNext-VC (Okamoto+ ICASSP 2024)
 - E2E-S2S-TTS and VC models with ConvNeXt-based encoder and decoder instead of Transformer-based ones
 - Improving inference speed while keeping synthesis quality
- MS-FC-HiFi-GAN (Yamashita+ IEEE Access 2024)
 - HiFi-GAN-based neural vocoder with linear layer-based upsampling
 - Improving inference speed while keeping synthesis quality

3. Incremental inference for low-latency synthesis

- Incremental inference applied only to neural vocoder
 - Acoustic model: Incremental inference increases synthesis error -> batch inference
 - Neural vocoder: Incremental inference without performance degradation



4. Results of evaluations



	Previous system	Mobile PresenTra	Original
Acoustic model:	Transformer encoder	Transformer encoder	
	Transformer decoder	ConvNeXt decoder	
Neural vocoder:	MS-HiFi-GAN	MS-FC-HiFi-GAN	
UTMOS	4.39	4.43	4.45
RTF for batch inference on server	0.2	0.08	
RTF for batch inference on smartphone	0.85	0.30	
Latency for incremental inference on server	0.35 s	0.17 s	
Latency for incremental inference on smartphone	1.13 s	0.47 s	