# HiFi-GAN: Generative Adversarial Networks for Effi cient and High Fidelity Speech Synthesis

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Youngwon Choi
대무의연구소

풀잎스쿨 Hands-on TTS

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## 1. Introduction

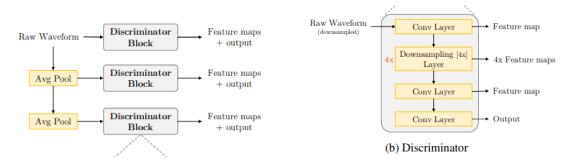
#### Introduction

#### □ Previous work

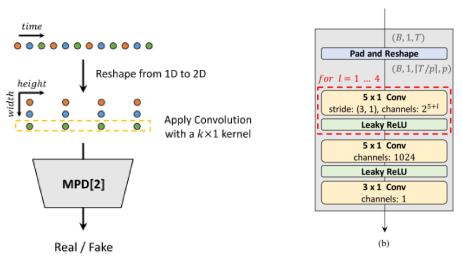
- Several recent work on speech synthesis have employed generative adversarial networks (GANs) to produce raw waveforms.
- Although such methods improve the sampling efficiency and memory usage, their sample quality
  has not yet reached that of autoregressive and flow-based generative models.
- Despite of the sophisticated GANs (MelGAN, Parallel WaveGAN, etc., ), there is still a gap in sample quality between the GAN models and AR or flow-based models.
- □ Authors propose HiFi-GAN, which achieves both higher computational efficiency and sample quality than AR or flow-based models.
  - As speech audio consists of sinusoidal signals with various periods, modeling the periodic patterns matters to generate realistic speech audio. Therefore, we propose a discriminator which consists of small sub-discriminators, each of which obtains only a specific periodic parts of raw waveforms

# 2. HifiGAN

- □ HiFi-GAN consist of one generator and two discriminators.
  - Discriminators: multi-scale and multi-period discriminator
  - Multi-scale discriminator



- Multi-period discriminator





#### □ Generator

- The generator is a fully convolutional neural network.
- It uses a mel-spectrogram as input and upsamples it through transposed convolutions until the length of the output sequence matches the temporal resolution of raw waveforms
- Multi-Receptive Field Fusion (MRF)
  - Different kernel sizes and dilation rates are selected for each residual block to form diverse receptive field patterns.
  - MRF module returns the sum of outputs from multiple residual blocks.

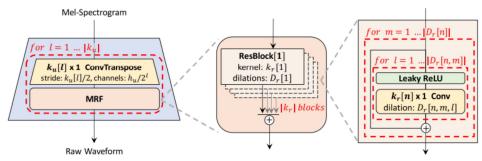
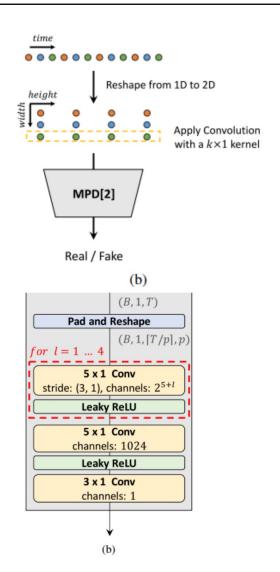
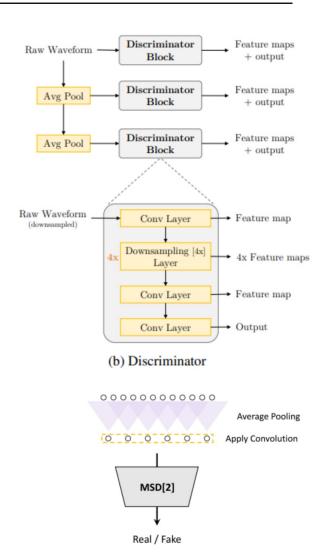


Figure 1: The generator upsamples mel-spectrograms up to  $|k_u|$  times to match the temporal resolution of raw waveforms. A MRF module adds features from  $|k_r|$  residual blocks of different kernel sizes and dilation rates. Lastly, the n-th residual block with kernel size  $k_r[n]$  and dilation rates  $D_r[n]$  in a MRF module is depicted.

- □ Multi-period discriminator
  - MPD is a mixture of sub-discriminators, each of which only accepts equally spaced samples of an input audio; the space is given as period p.
    - period p: [2, 3, 5, 7, 11]
    - The sub-discriminators are designed to capture different implicit structures from each other by looking at different parts of an input audio
  - We first reshape 1D raw audio of length T into 2D data of height T /p and width p and then apply 2D convolutions to the reshaped data.
  - In every convolutional layer of MPD, we restrict the kernel size in the width axis to be 1 to process the periodic samples independently.
  - By reshaping the input audio into 2D data instead of sampling periodic signals of audio, gradients from MPD can be delivered to all time steps of the input audio.



- □ Multi-scale discriminator
  - The architecture of MSD is drawn from that of MelGAN.
  - Because each sub-discriminator in MPD only accepts disjoint samples, we add MSD to consecutively evaluate the audio sequence.
  - MSD is a mixture of three sub-discriminators operating on different input scales:
    - raw audio, ×2 average-pooled audio, and ×4 average-pooled audio.
    - Note that MPD operates on disjoint samples of raw waveforms, whereas MSD operates on smoothed waveforms.



#### □ Training loss terms

- − *G*: Generator, *D*: Discriminator
- -x: ground truth audio, s: mel-spectrogram of the ground truth audio
- Final Loss = GAN Loss + Mel-Spectrogram loss + Feature matching loss

#### ☐ GAN Loss

- For training stability, the objectives follow the least-squares GAN (LSGAN). (1 for real, 0 for fake)
- Discriminator

$$\mathcal{L}_{Adv}(D;G) = \mathbb{E}_{(x,s)} \left[ (D(x) - 1)^2 + \left( D(G(s)) \right)^2 \right]$$

Generator

$$\mathcal{L}_{Adv}(G;D) = \mathbb{E}_{(x,s)} \left[ \left( D(G(s)) - 1 \right)^2 \right]$$

#### □ Mel Spectrogram loss

- Reconstruction loss
- The mel-spectrogram loss is the L1 distance between the mel-spectrogram of a waveform synthesized by the generator and that of a ground truth waveform.

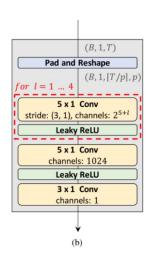
$$\mathcal{L}_{Mel}(G) = \mathbb{E}_{(x,s)}[\|\phi(x) - \phi(G(s))\|_{1}]$$

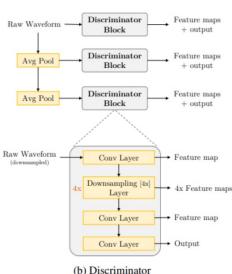
- The mel-spectrogram loss helps the generator to synthesize a realistic waveform corresponding to an input condition, and also stabilizes the adversarial training process from the early stages.
- The mel-spectrogram loss can be expected to have the effect of focusing more on improving the perceptual quality due to the characteristics of the human auditory system.

#### □ Feature matching loss

- The feature matching loss is a learned similarity metric measured by the difference in features of the discriminator between a ground truth sample and a generated sample.
- This objective minimizes the L1 distance between discriminator feature maps of reals and synthetic speech.

$$\mathcal{L}_{FM}(G; D) = \mathbb{E}_{(x,s)} \left[ \sum_{i=1}^{T} \frac{1}{N_i} \| D^i(x) - D^i(G(s)) \|_1 \right]$$





#### ☐ Final Loss

— Final Loss = GAN Loss + Mel-Spectrogram loss + Feature matching loss

$$\mathcal{L}_{G} = \mathcal{L}_{Adv}(G; D) + \lambda_{fm} \cdot \mathcal{L}_{FM}(G; D) + \lambda_{mel} \cdot \mathcal{L}_{Mel}(G)$$

$$\mathcal{L}_D = \mathcal{L}_{Adv}(D;G)$$

Where  $\lambda_{fm}=2$ ,  $\lambda_{mel}=45$ 

- $\Box$  The three variations of the generator V1, V2 and V3:
  - V1:  $h_u$  = 512,  $k_r$  = [3, 7, 11],  $k_u$  = [16, 16, 4, 4],  $D_r$  =[[1,1], [3,1], [5,1]] x 3
  - V2 : The small version of V1,  $h_u$  = 128.
  - V3:  $h_u$  = 256,  $k_r$  = [3, 5, 7],  $k_u$  = [16, 16, 8],  $D_r$  =[[1], [2]], [[2], [6]], [[3], [12]]
- □ 기타 Experiment configuration 은 생략하겠습니다.

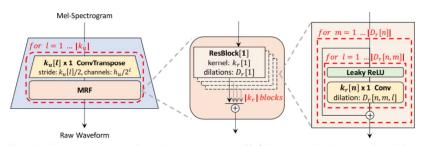


Figure 1: The generator upsamples mel-spectrograms up to  $|k_u|$  times to match the temporal resolution of raw waveforms. A MRF module adds features from  $|k_r|$  residual blocks of different kernel sizes and dilation rates. Lastly, the n-th residual block with kernel size  $k_r[n]$  and dilation rates  $D_r[n]$  in a MRF module is depicted.

 $h_u$ : hidden dimension

 $k_r$ : kernel size of standard convolution

 $k_u$ : kernel size of transposed convolution

 $D_r$ : dilation rates



#### □ MOS Test

Table 1: Comparison of the MOS and the synthesis speed. Speed of n kHz means that the model can generate  $n \times 1000$  raw audio samples per second. The numbers in () mean the speed compared to real-time.

Model	MOS (CI)	Speed on CPU (kHz)	Speed on GPU (kHz)	# Param (M)
Ground Truth	$4.45~(\pm 0.06)$	_	-	_
WaveNet (MoL) WaveGlow MelGAN	4.02 (±0.08) 3.81 (±0.08) 3.79 (±0.09)	4.72 (×0.21) 145.52 (×6.59)	0.07 (×0.003) 501 (×22.75) 14,238 (×645.73)	24.73 87.73 4.26
HiFi-GAN $V1$ HiFi-GAN $V2$ HiFi-GAN $V3$	<b>4.36</b> (±0.07) 4.23 (±0.07) 4.05 (±0.08)	31.74 (×1.43) 214.97 (×9.74) <b>296.38</b> (× <b>13.44</b> )	3,701 (×167.86) 16,863 (×764.80) <b>26,169</b> (× <b>1,186.80</b> )	13.92 <b>0.92</b> 1.46

#### □ Ablation Study

Model	MOS (CI)
Ground Truth	4.57 (±0.04)
Baseline (HiFi-GAN V3)	4.10 (±0.05)
w/o MPD w/o MSD w/o MRF w/o Mel-Spectrogram Loss MPD p=[2,4,8,16,32]	2.28 (±0.09) 3.74 (±0.05) 3.92 (±0.05) 3.25 (±0.05) 3.90 (±0.05)
MelGAN MelGAN with MPD	$2.88~(\pm 0.08)$ $3.35~(\pm 0.07)$

☐ Generalization to Unseen Speakers

Table 3: Quality comparison of synthesized utterances for unseen speakers.

Model	MOS (CI)
Ground Truth	$3.79 (\pm 0.07)$
WaveNet (MoL)	$3.52 (\pm 0.08)$
WaveGlow	$3.52 (\pm 0.08)$
MelGAN	$3.50 (\pm 0.08)$
HiFi-GAN $V1$	3.77 (±0.07)
HiFi-GAN $V2$	3.69 (±0.07)
HiFi-GAN $V3$	3.61 (±0.07)

#### ☐ End to End speech synthesis®

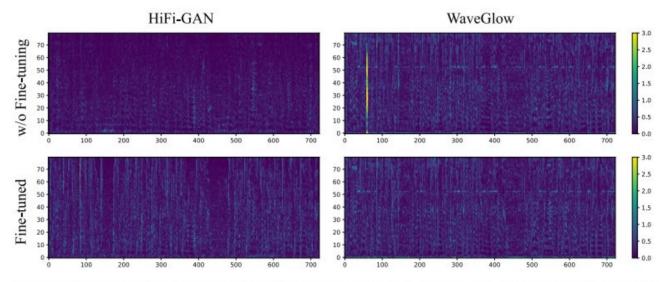


Figure 3: Pixel-wise difference in the mel-spectrogram domain between generated waveforms and a mel-spectrogram from Tacotron2. Before fine-tuning, HiFi-GAN generates waveforms corresponding to input conditions accurately. After fine-tuning, the error of the mel-spectrogram level increased, but the perceptual quality increased.

Table 4: Quality comparison for end-to-end speech synthesis.

Model	MOS (CI)
Ground Truth	4.23 (±0.07)
WaveGlow (w/o fine-tuning)	3.69 (±0.08)
HiFi-GAN V1 (w/o fine-tuning) HiFi-GAN V2 (w/o fine-tuning) HiFi-GAN V3 (w/o fine-tuning)	$3.91 (\pm 0.08)$ $3.88 (\pm 0.08)$ $3.89 (\pm 0.08)$
WaveGlow (find-tuned)	3.66 (±0.08)
HiFi-GAN V1 (find-tuned) HiFi-GAN V2 (find-tuned) HiFi-GAN V3 (find-tuned)	<b>4.18</b> (±0.08) 4.12 (±0.07) 4.02 (±0.08)

# Thank you for listening! Q&A