

TEXT-TO-SPEECH SYNTHESIS USING STFT SPECTRA BASED ON LOW-/MULTI-RESOLUTION GENERATIVE ADVERSARIAL NETWORKS

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1. SYNOPSIS

In Text-To-Speech (TTS) synthesis w/ Short-Term Fourier Transform (STFT) spectra:

- (1) Proposes Generative Adversarial Networks (GANs)^[1]-based training algorithm using low-/multi-resolution spectra.
- (2) Demonstrates that the proposed algorithm using GANs w/ low-resolution spectra improves synthetic speech quality.

2. CONVENTIONAL ALGORITHMS

DNN-based TTS w/ STFT spectra^[2]

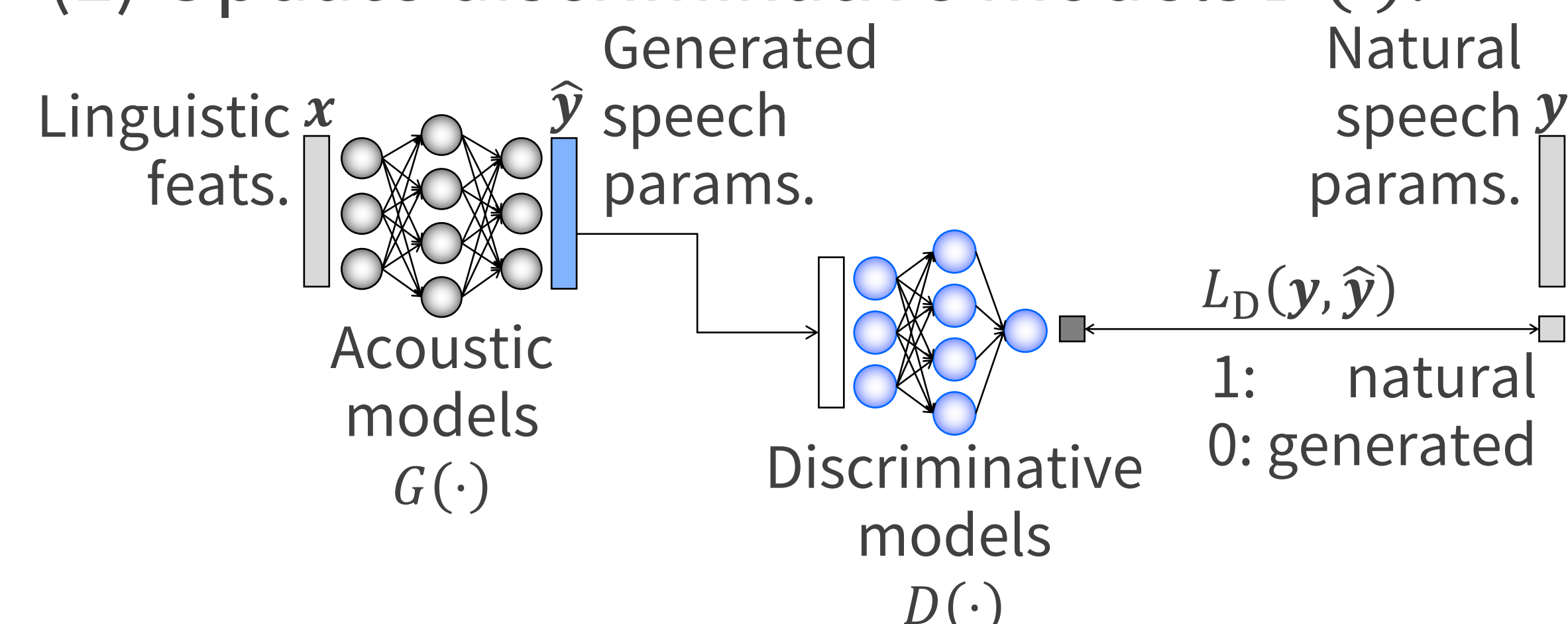
- (1) Generate raw STFT amplitude spectra.
- (2) Reconstruct phase spectra by using Griffin and Lim's method^[3].

Pros. **avoiding a vocoding process**

Cons. **over-smoothing of amplitude spectra**

GAN-based TTS w/ vocoders^[4]

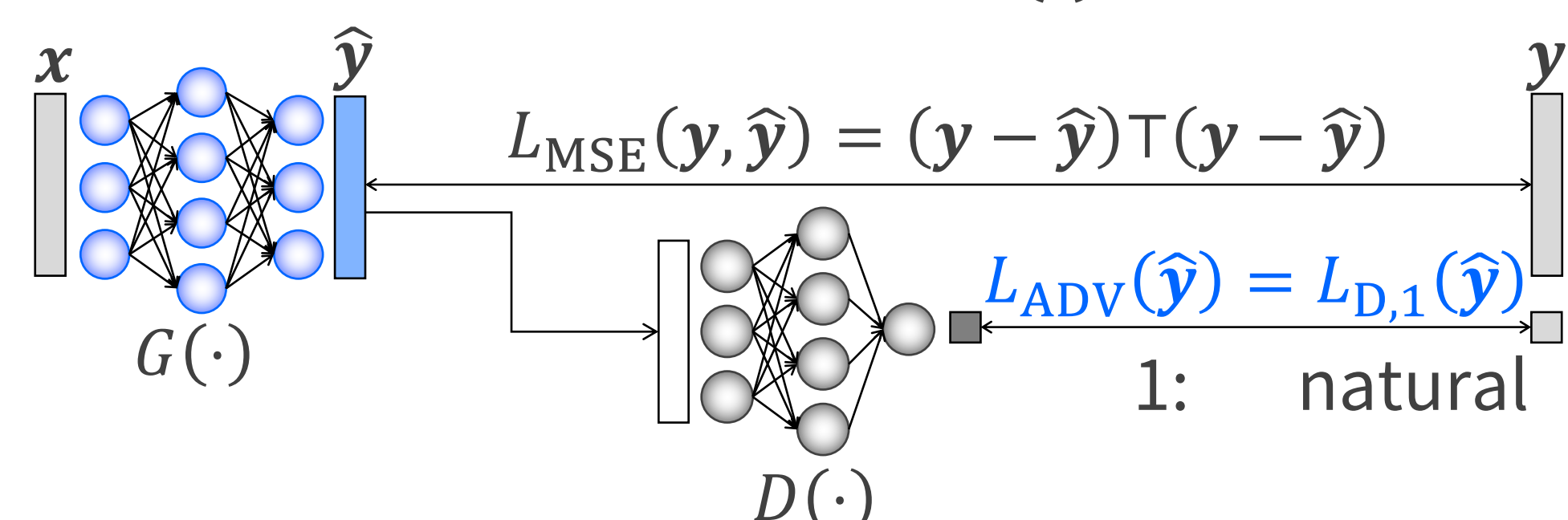
- (1) Update discriminative models $D(\cdot)$.



$$L_D(\mathbf{y}, \hat{\mathbf{y}}) = -\sum_t \log D(\mathbf{y}_t) - \sum_t \log(1 - D(\hat{\mathbf{y}}_t))$$

$L_{D,1}(\mathbf{y})$ (loss for natural params.) $L_{D,0}(\hat{\mathbf{y}})$ (loss for generated params.)

- (2) Update acoustic models $G(\cdot)$.



$$L_G(\mathbf{y}, \hat{\mathbf{y}}) = L_{MSE}(\mathbf{y}, \hat{\mathbf{y}}) + \omega_D \frac{\mathbb{E}_{\hat{\mathbf{y}}} [L_{MSE}]}{\mathbb{E}_{\hat{\mathbf{y}}^{(L)}} [L_{ADV}]} L_{ADV}(\hat{\mathbf{y}})$$

Minimizing approx. JS-divergence

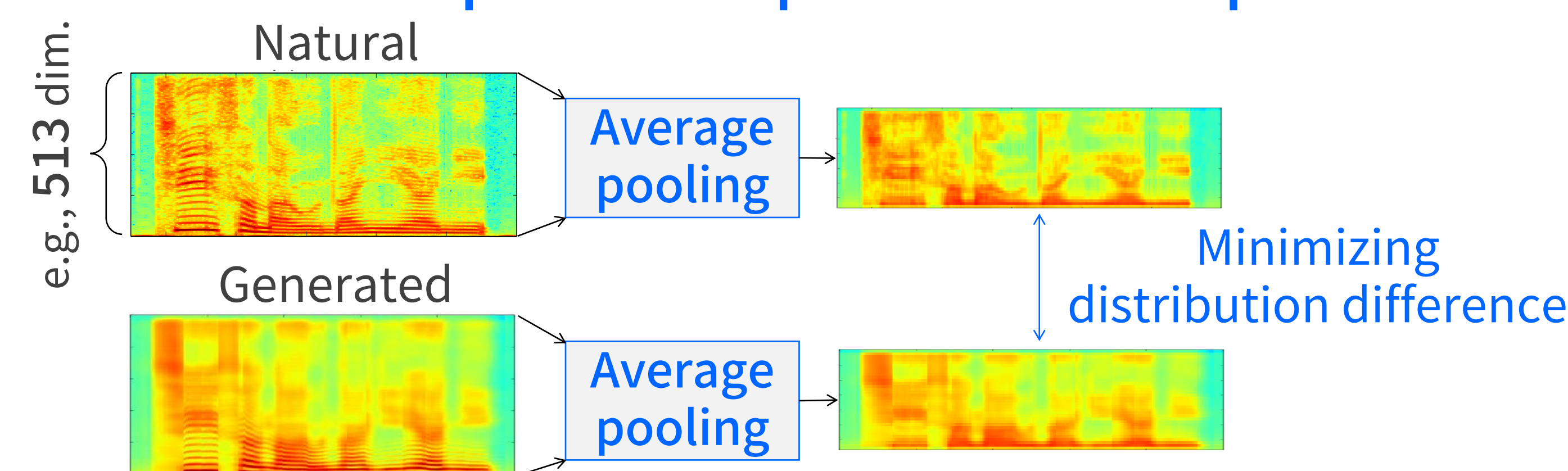
3. PROPOSED ALGORITHMS

Motivation

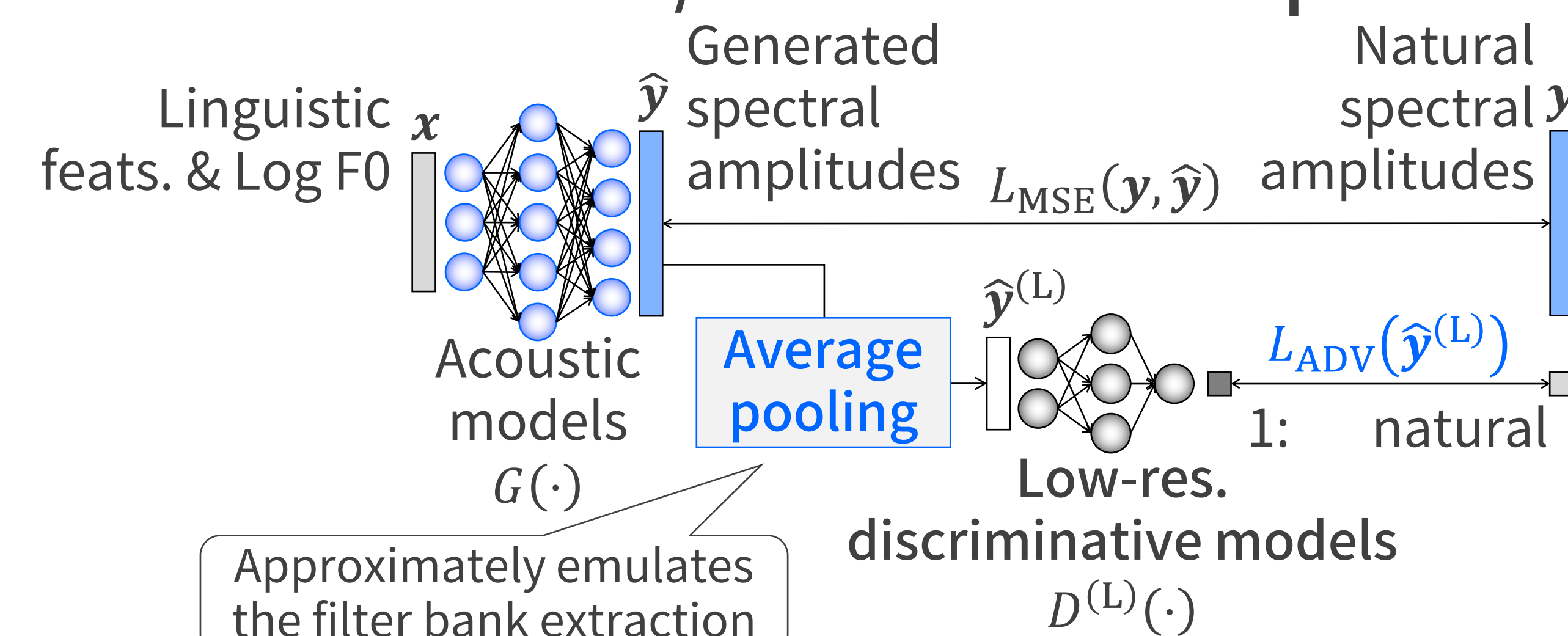
Amplitude spectra: high-dimensional features including spectral & excitation characteristics.
GAN-based TTS^[3]: effective for spectral envelopes (i.e., mel-cepstral coefficients).

Assumption:

low-res. spectra \approx spectral envelopes



GAN-based TTS w/ low-res. STFT spectra



$$L_G^{(Low)}(\mathbf{y}, \hat{\mathbf{y}}) = L_{MSE}(\mathbf{y}, \hat{\mathbf{y}}) + \omega_D^{(L)} \frac{\mathbb{E}_{\hat{\mathbf{y}}} [L_{MSE}]}{\mathbb{E}_{\hat{\mathbf{y}}^{(L)}} [L_{ADV}]} L_{ADV}(\hat{\mathbf{y}}^{(L)})$$

GAN-based TTS w/ multi-res. STFT spectra

$$L_G^{(Multi)}(\mathbf{y}, \hat{\mathbf{y}}) = L_{MSE}(\mathbf{y}, \hat{\mathbf{y}}) + \omega_D^{(L)} \frac{\mathbb{E}_{\hat{\mathbf{y}}} [L_{MSE}]}{\mathbb{E}_{\hat{\mathbf{y}}^{(L)}} [L_{ADV}]} L_{ADV}(\hat{\mathbf{y}}^{(L)}) + \omega_D \frac{\mathbb{E}_{\hat{\mathbf{y}}} [L_{MSE}]}{\mathbb{E}_{\hat{\mathbf{y}}} [L_{ADV}]} L_{ADV}(\hat{\mathbf{y}})$$

4. EXPERIMENTAL EVALUATION

Experimental conditions

Dataset	4,007 utterances taken from Japanese female speaker (subset of JSUT ^[5] corpus, 3,808/199 for training/evaluation)
STFT analysis	Frame length: 400, shift length: 80, FFT length: 1024, analysis window: Hamming
Average pooling	Zero padding: 6, pooling width w : {14, 30, 70}, stride: $w/2$
ω_D and $\omega_D^{(L)}$	1.0
Dims. of $\mathbf{x}/\mathbf{y}/\mathbf{y}^{(L)}$	444 (linguistic feats, durations, Log F0, UV)/513/{74, 34, 14} (dim. of $\mathbf{y}^{(L)}$ was changed in accordance with w)
DNN architectures	Feed-Forward (see our paper)

Subjective evaluation (preference AB tests)

MSE: minimizing $L_{MSE}(\mathbf{y}, \hat{\mathbf{y}})$ ^[2]

ADV: minimizing $L_G(\mathbf{y}, \hat{\mathbf{y}})$ ^[4]

ADV-Low: minimizing $L_G^{(Low)}(\mathbf{y}, \hat{\mathbf{y}})$ (proposed)

ADV-Multi: minimizing $L_G^{(Multi)}(\mathbf{y}, \hat{\mathbf{y}})$ (proposed)

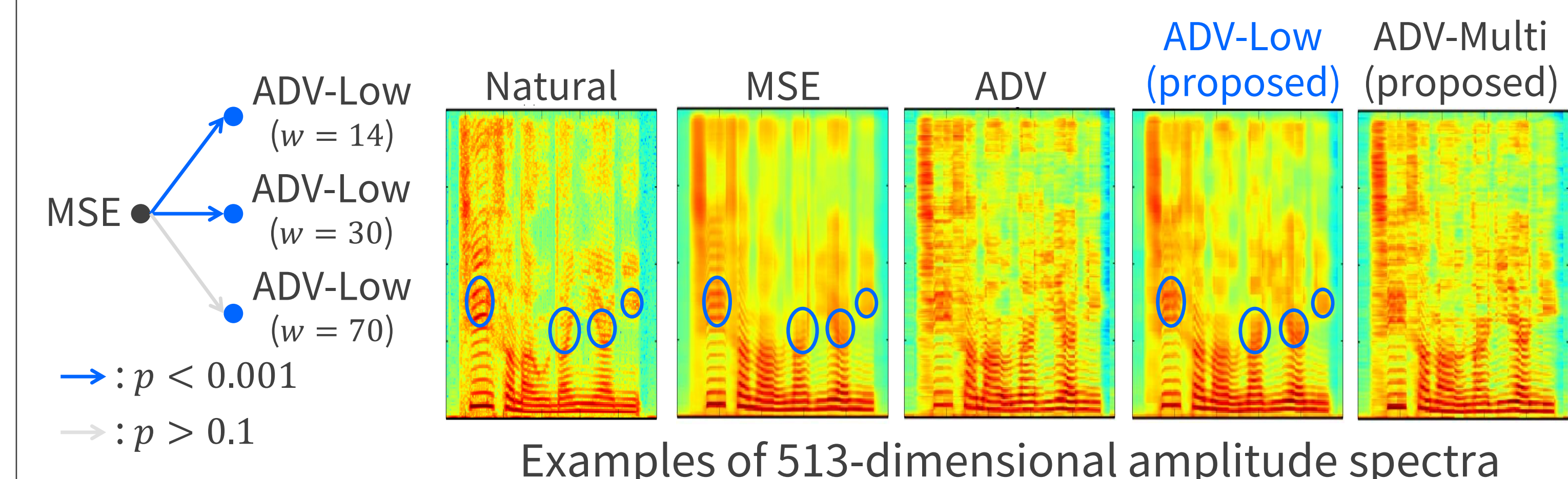
Results:

(1) ADV & ADV-Multi didn't improve speech quality.

⇒ Because of difficulty in minimizing distribution differences in original resolution

(2) ADV-Low improved speech quality.

⇒ Effect of the spectral envelopes compensation



[References]

[1] Goodfellow et al., *Proc. NIPS*, 2014. [2] Takaki et al., *Proc. INTERSPEECH*, 2017. [3] Griffin et al., *IEEE Trans. on ASLP*, 1984. [4] Saito et al., *IEEE/ACM Trans. on ASLP*, 2018. [5] Sonobe et al., *arXiv:1711.00354*, 2017.