

# Performance Comparison of Single-Microphone Speech Enhancement Using Speech-Distortion Weighted Inter-Frame Wiener Filters



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### Introduction

- Filters for single-microphone speech enhancement can be applied to suppress background noise while preventing speech distortion [1, 2, 3, 4]
- Single- & multi-frame trade-off filters (also known as speech-distortion weighted interframe Wiener filters) can increase noise reduction while limiting speech distortion
- Single-frame filters disregard correlation across time frames and/or frequency bins
- Multi-frame filters can exploit temporal correlation across present and past frames

#### IN THIS POSTER

Real- and complex-valued **trade-off filters** are analysed using **low-delay** STFT filterbanks for **speech enhancement** with a single-microphone.

## Framework

Filterbank	Analysis window lengt	n Synthesis window length
Low resolution STFT	64 (4ms) 🔨	64 (4ms)
High resolution STFT	256 (16ms) —	64 (4ms)

Sampling frequency: 16 kHz

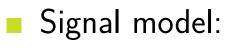
Both filterbanks use a frame-shift of 16 (1 ms) and have a synthesis delay of 3 ms

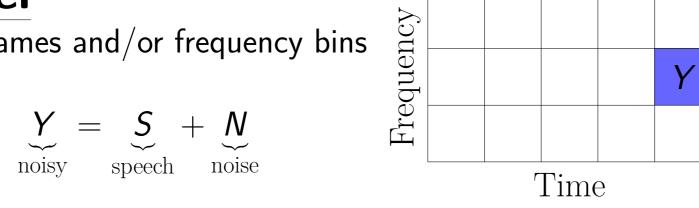
### **Evaluation**

Speech with crossroad noise at SNR: 0 dB. Trade-off parameter  $\mu = 0.8$ 

# **Signal Models**

Single-Frame Signal Model
Assumption: no correlation over time frames and/or frequency bins





• Estimate speech S by applying gain G to to each noisy speech STFT coefficient Y independently:

 $\hat{S} = GY$ 

### Multi-Frame Signal Model

- Assumption: speech correlated over consecutive time frames
- Speech correlation vector:

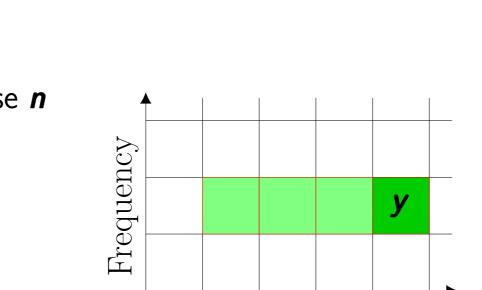
$$oldsymbol{\gamma_s} = rac{\mathbb{E}[oldsymbol{s}S^*]}{\mathbb{E}[|oldsymbol{S}|^2]} = rac{oldsymbol{R_s}}{\phi_S}$$

Signal model of noisy speech y in terms of speech s and noise n or correlated speech  $\gamma_s S$  and undesired u:

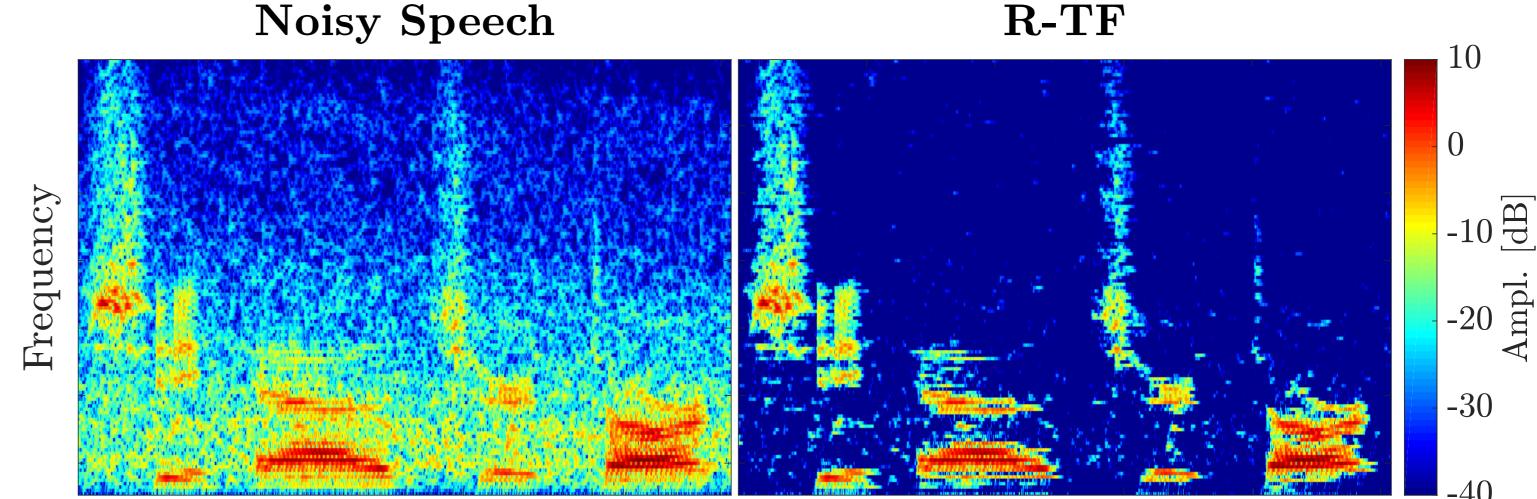
$$oldsymbol{y} = oldsymbol{s} + oldsymbol{n} \ = oldsymbol{\gamma_s} S + oldsymbol{u}$$

Estimate speech S by applying filter **h** to noisy speech vector:  $\hat{S} = \mathbf{h}^H \mathbf{y}$ 

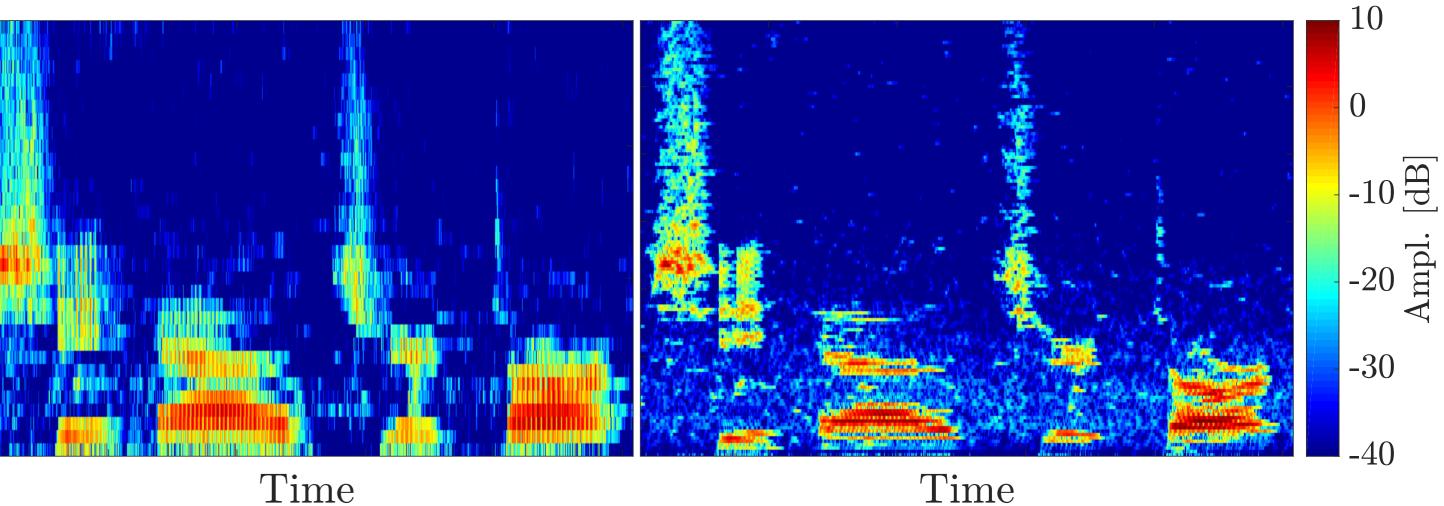
# **Trade-off Filters**



Time



C-MFTF



**R-MFTF** 

### Results

Frequency

Objective measures: Perceptual speech quality (PESQ), segmental speech distortion (SD),



Single-Frame Trade-off Filter:

$$G = \underset{G}{\operatorname{argmin}} \left\{ \underbrace{\mathbb{E}\left[|GS - S|^2\right]}_{\text{Filtered speech}} + \mu \underbrace{\mathbb{E}\left[|GN|^2\right]}_{\text{Filtered noise}} \right\}$$

#### Real-Valued Trade-off filter (R-TF)

Implemented in STFT filterbank with high frequency resolution & high time resolution

Parameters are estimated using [5], [6]  $G = \frac{\phi_S}{\mu \phi_N + \phi_S}$ 

For  $\mu = 0$  applies **no noise reduction** 

For  $\mu = 1$  is a Wiener gain

Multi-Frame Trade-off Filter:

$$\boldsymbol{h}^{\mathrm{MFTF}} = \underset{\boldsymbol{h}}{\operatorname{argmin}} \left\{ \underbrace{\mathbb{E}\left[|\boldsymbol{h}^{H}\boldsymbol{\gamma_{s}}\boldsymbol{S} - \boldsymbol{S}|^{2}\right]}_{\boldsymbol{h}} + \mu \underbrace{\mathbb{E}\left[|\boldsymbol{h}^{H}\boldsymbol{u}|^{2}\right]}_{\boldsymbol{h}} \right\}$$

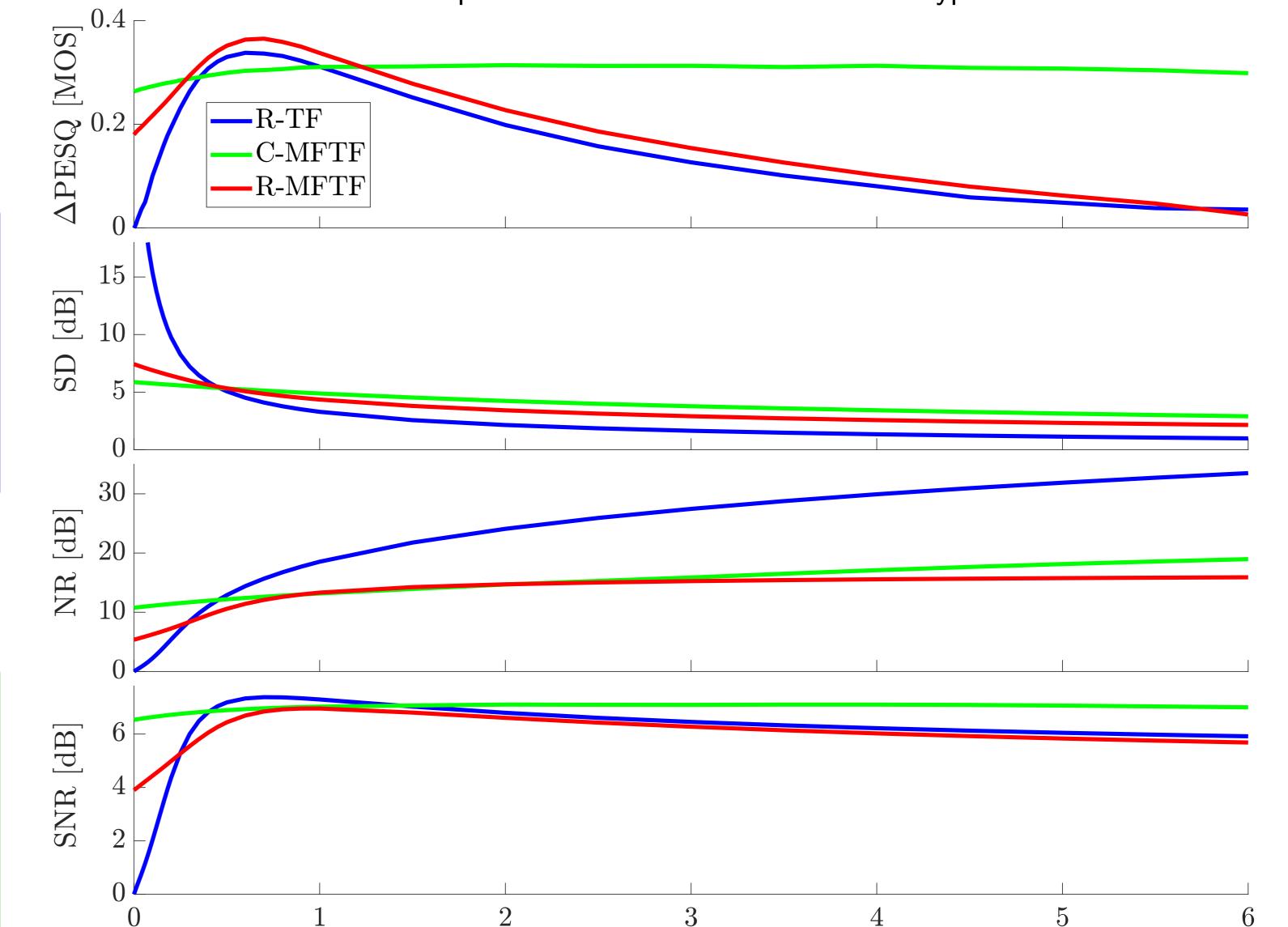
Filtered correlated speech Filtered undesired

#### Complex-Valued Multi-Frame Trade-off Filter (C-MFTF)

- Implemented in STFT filterbank with low frequency resolution & high time resolution
- Parameters are estimated as in [1]

$$\boldsymbol{h}^{\text{C-MFTF}} = \frac{\boldsymbol{R}_{\boldsymbol{y}}^{-1}\boldsymbol{\gamma}_{\boldsymbol{s}}}{\boldsymbol{\gamma}_{\boldsymbol{s}}^{H}\boldsymbol{R}_{\boldsymbol{v}}^{-1}\boldsymbol{\gamma}_{\boldsymbol{s}}}\frac{\phi_{\boldsymbol{S}}}{\mu\phi_{U}^{\text{out}}+\phi_{\boldsymbol{S}}}$$

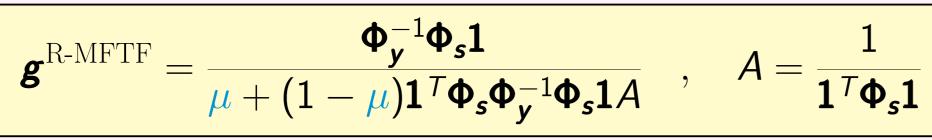
segmental noise reduction (NR), and segmental signal to noise ratio (SNR). Evaluated over 105 s speech material for 5 different noise types at SNR: 0 dB



#### For $\mu = 0$ is a C-MFMPDR filter For $\mu = 1$ is a C-MFWF

#### Real-Valued Multi-Frame Trade-off Filter (R-MFTF)

- Implemented in STFT filterbank with high frequency resolution & high time resolution
- Parameters are estimated as in [2]
- Solution is obtained by applying DFT to cost function: g = Dh(D: DFT matrix)



Real-valued filter vector  $\mathbf{g}^{\text{R-MFTF}}$  can be overlapped into high-resolution scalar gain:  $\mathbf{G}^{\text{R-MFTF}}$ 

For  $\mu = 0$  is a **R-MFMPDR filter** For  $\mu = 1$  is a **R-MFWF** 

#### • At $\mu = 0$ , the C-MFTF (i.e. C-MFMPDR filter) is the most effective filter

- R-MFTF applies best overall PESQ improvement at  $\mu = 0.7$
- R-TF can apply highest noise reduction but with high speech distortion

### Conclusions

- Complex- and real-valued trade-off filters are effective for speech enhancement
- Increasing trade-off parameter  $\mu$  increases both noise reduction and speech distortion

• C-MFMPDR filter performs better than R-MFMPDR filter (MFTFs for  $\mu = 0$ )

#### References

] D. Fischer, K. Brümann, and S. Doclo, "Comparison of parameter estimation methods for Single-Microphone Multi-Frame wiener filtering," in <u>27th European Signal Processing</u> <u>Conference (EUSIPCO) (Submitted)</u> .	[4] A. Schasse and R. Martin, "Estimation of subband speech correlations for noise reduction via MVDR processing," IEEE Trans. Audio, Speech, Language Process., vol. 22, no. 9, pp. 1355–1365, Sep. 2014.
] K. T. Andersen and M. Moonen, "Robust speech-distortion weighted interframe Wiener filters for single-channel noise reduction," IEEE Trans. Audio, Speech, Language Process., vol. 26, no. 1, pp. 97–107, Jan. 2018.	[5] Y. Ephraim and D. Malah, "Speech enhancement using a minimum mean-square error log-spectral amplitude estimator," IEEE Trans. Acoust., Speech, Signal Process., vol. 33, no. 2, pp. 443–445, Apr. 1985.
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