# A Simple and Effective Framework for A Priori SNR Estimation

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#### Abstract

FШF

Der Wissenschaftsfonds.

- The a priori SNR is key-parameter in DFT-based speech enhancement schemes
- **Decision-directed** (DD) *a priori* SNR estimation: linear combination of estimates along fixed DFT bin k.
- Can speech enhancement performance be improved by combining estimates along harmonic trajectories instead of fixed DFT bins?

#### **Speech Enhancement**

## **Proof-of-Concept**

## Noisy signal: Speech and white noise mixed at 0 dB SNR.



**DFT** based speech enhancement: multiplicative gain function  $G(\cdot)$ Speech Estimate is obtained by

 $\hat{X}(k,\ell) = G(k,\ell,\xi(k,\ell),\zeta(k,\ell)) \cdot Y(k,\ell)$  $\boldsymbol{\zeta}(k,\ell) = \frac{|Y(k,\ell)|^2}{\sigma_d^2(k,\ell)} \dots \text{ a posteriori SNR}$  $\boldsymbol{\xi}(k,\ell) = \frac{\sigma_x^2(k,\ell)}{\sigma_d^2(k,\ell)} \dots \text{ a priori SNR}$ 

## The Decision-Directed A Priori SNR Estimator

## **DD** a priori SNR estimate:

 $\hat{\xi}_{\mathsf{DD}}(k,\ell) = (1 - \alpha_{\mathsf{DD}}) \max[\hat{\xi}_{\mathsf{ML}}(k,\ell),0] + \alpha_{\mathsf{DD}}\hat{\xi}_{\ell-1}(k,\ell)$ 

with

$$\hat{\xi}_{\mathsf{ML}}(k,\ell) = \hat{\zeta}(k,\ell) - 1$$
$$\hat{\xi}_{\ell-1}(k,\ell) = \frac{|\hat{X}(k,\ell-1)|^2}{\hat{\gamma}^2(1-\ell-1)}$$

#### **DD**:

• Spurious spectral peaks  $\rightarrow$  musical noise Harmonics are smeared along time

## **PADDi**:

Less isolated spectral peaks Harmonic fine structure is preserved

# Results (1)

**Characteristics** of speech estimator strongly depend on  $G(\cdot)$ • We compared **DD** and **PADDi** for various  $G(\cdot)$ s Evaluation: Segmental Speech to Speech Distortion Ratio (SSDR<sub>seg</sub>) vs. Segmental Noise Attenuation (NA<sub>seg</sub>)

**A A** 

 $\times$  Wiener filter

## $\sigma_d^2(\kappa, \ell-1)$

Is there a better choice for  $\xi_{\ell-1}(k, \ell)$ ?

## **PADDi - The Proposed Method**

Speech exhibits **harmonic structure** Fundamental frequency is time-varying • Main idea of this work: ensure that k is dominated similarly by the same harmonic at frames  $\ell'$  and  $\ell' - 1$ Pitch-adaptive discrete STFT (PADSTFT):

$$N_{\mathsf{DFT}}(\ell) = \operatorname{round} \left[ K \frac{f_s}{f_0(\ell)} \right]$$
$$k_h(\ell) = \operatorname{argmin}_k \left| k - N_{\mathsf{DFT}}(\ell) \frac{h f_0(\ell)}{f_s} \right| = Kh$$
$$\mathsf{independent of } \ell!$$

■ PADSTFT:

path coincide



**Red**: harmonic trajectory Green: smoothing path of a priori SNR estimator

Fixed mapping from h to k



• LSA ∆ jMAP □ MMSE-STSA

■ PADDi increases NA<sub>seg</sub> while preserving SSDR<sub>seg</sub> compared to DD

**Results (2)** 

 $\Delta$ -improvement in terms of **PESQ** and **SNR**<sub>seg</sub> over noisy speech



 $\square \Delta SNR_{seg}$ : PADDi brings improved or similar performance compared to benchmarks

All methods perform

similarly