

A Time Regularization Technique for Discrete Spectral Envelopes Through Frequency Derivative

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ABSTRACT

Issues in amplitude spectral envelope estimation:

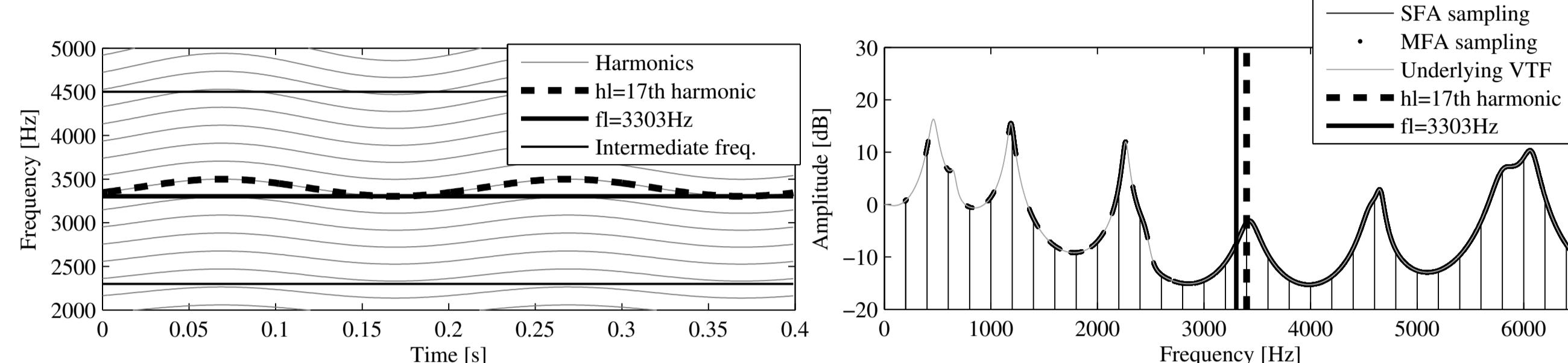
- Undersampled Vocal Tract Filter (VTF)
- Shape reconstruction
- Time regularity

Traditional approach: Single Frame Analysis (SFA)

Why not using Multiple Frame Analysis (MFA) [4] ?!

Evaluation shown for singing voice [1,2]

MFA size: 2 periods of vibrato $\Rightarrow 400\text{ms}$



MFA METHODS

The methods below use spectral peaks extracted on each frame (each 5ms).

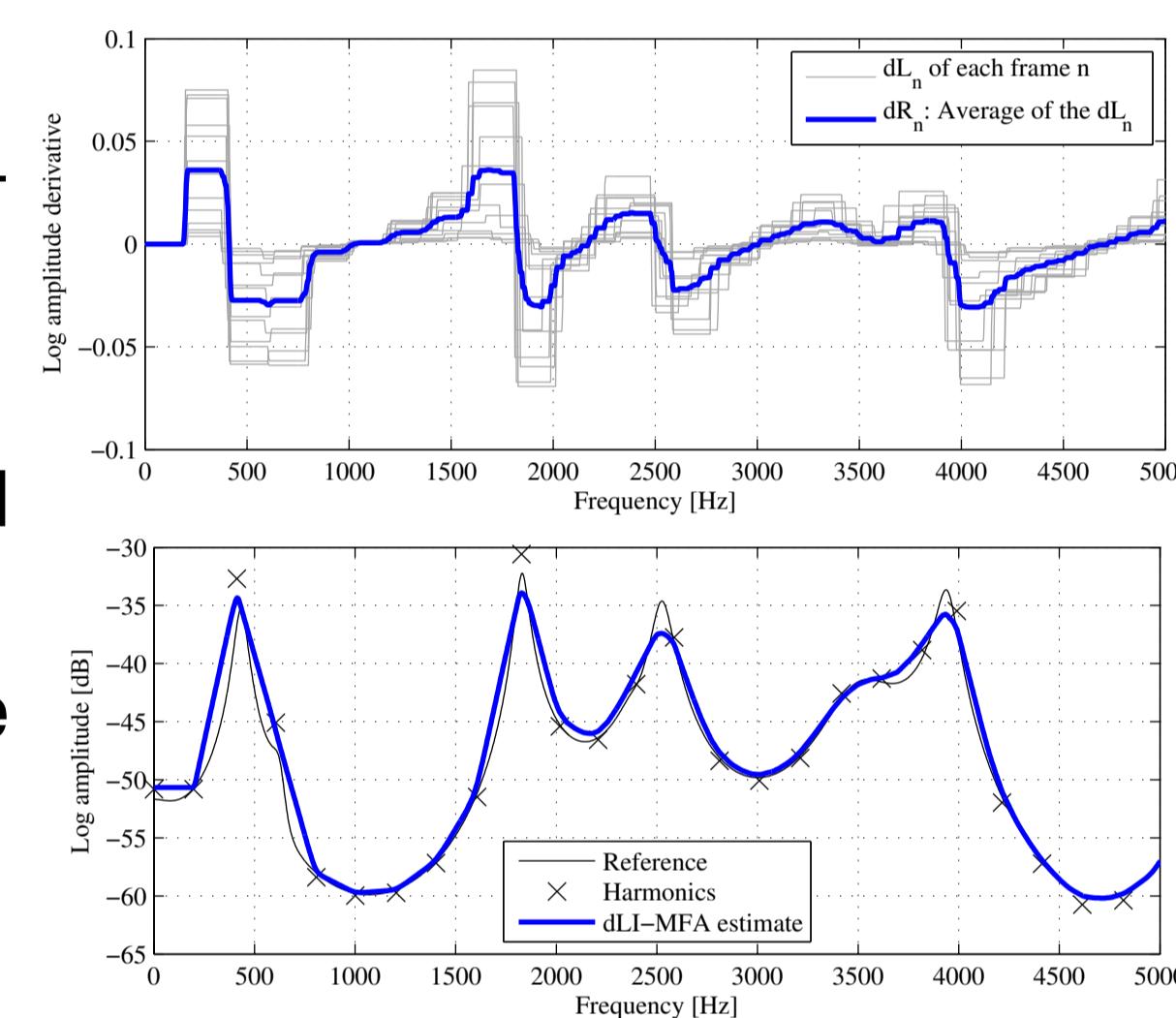
Extra issue: The frames' energy follows the amplitude modulation (AM) of the signal (e.g. tremolo in singing voice) \Rightarrow **How to align the frames ?**

DLINEAR-MFA[THIS LETTER]

Steps:

1. Compute derivative of the linear envelope (on log scale).
2. Averaging of K frames across time.
3. Compute integral of the averaged derived envelope.
4. Final alignment on the central frame $(K - 1)/2$.

(No need of frame pre-alignment!)



SDCE-MFA[1,2]

Model for the Discrete Cepstral Envelope (DCE):

$$E(f) = c_0 + 2 \sum_{n=1}^P c_n \cos(n2\pi f/f_s) \quad (1)$$

c_n : the cepstral coefficients, P : the cepstral order.

For MFA, Shiga et al. [4] suggested to minimize:

$$\epsilon = \sum_{k=1}^K \| \mathbf{a}_k - d_k \mathbf{u}_k - \mathbf{B}_k \mathbf{c} \| \quad (2)$$

k : the frame, a_k : the log amplitudes, d_k : correction term for the AM, $\mathbf{u}_k = [1, \dots, 1]^T$, \mathbf{c} : the cepstral coefficients, \mathbf{B} : the Fourier basis.

Steps:

1. Compute:

$$\mathbf{c} = \sum_{k=1}^K \left(\sum_{l=1}^K \mathbf{B}_l^T \mathbf{B}_l \right)^{-1} \cdot \left(\mathbf{B}_k^T \mathbf{a}_k \right) \quad (3)$$

(No need of frame pre-alignment! Shown in [1,2])

2. Final alignment on the central frame.

We can increase the order ! (e.g. x1.4 in the following experiments)

LINEAR-MFA-LIFT[1,2]

MFA version of the basic linear interpolation, which has already been used for comparison [5]. We only add a low-pass filtering.

Steps:

1. Pre-alignment of K successive frames using energy in [0-4]kHz.
2. Linear interpolation of all peaks of the K frames [5] \Rightarrow Linear-MFA.
3. Low-pass filter of the Linear-MFA to alleviate erratic shapes.
4. Final alignment on the central frame.

- [1] G. Degottex, L. Ardaillon, A. Roebel, "Simple Multi Frame Analysis methods for estimation of Amplitude Spectral Envelope estimation in Singing Voice", in ICASSP, 2016.
[2] G. Degottex, L. Ardaillon, A. Roebel, "Multi-Frame Amplitude Envelope Estimation for Modification of Singing Voice", IEEE Transactions on Audio, Speech, and Language Processing, Accepted 2016.
[3] M. Campedel-Oudot, O. Cappe, E. Moulines "Estimation of the spectral envelope of voiced sounds using a penalized likelihood approach", IEEE Transactions on Speech and Audio Processing, 2001.

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with follow-up works with

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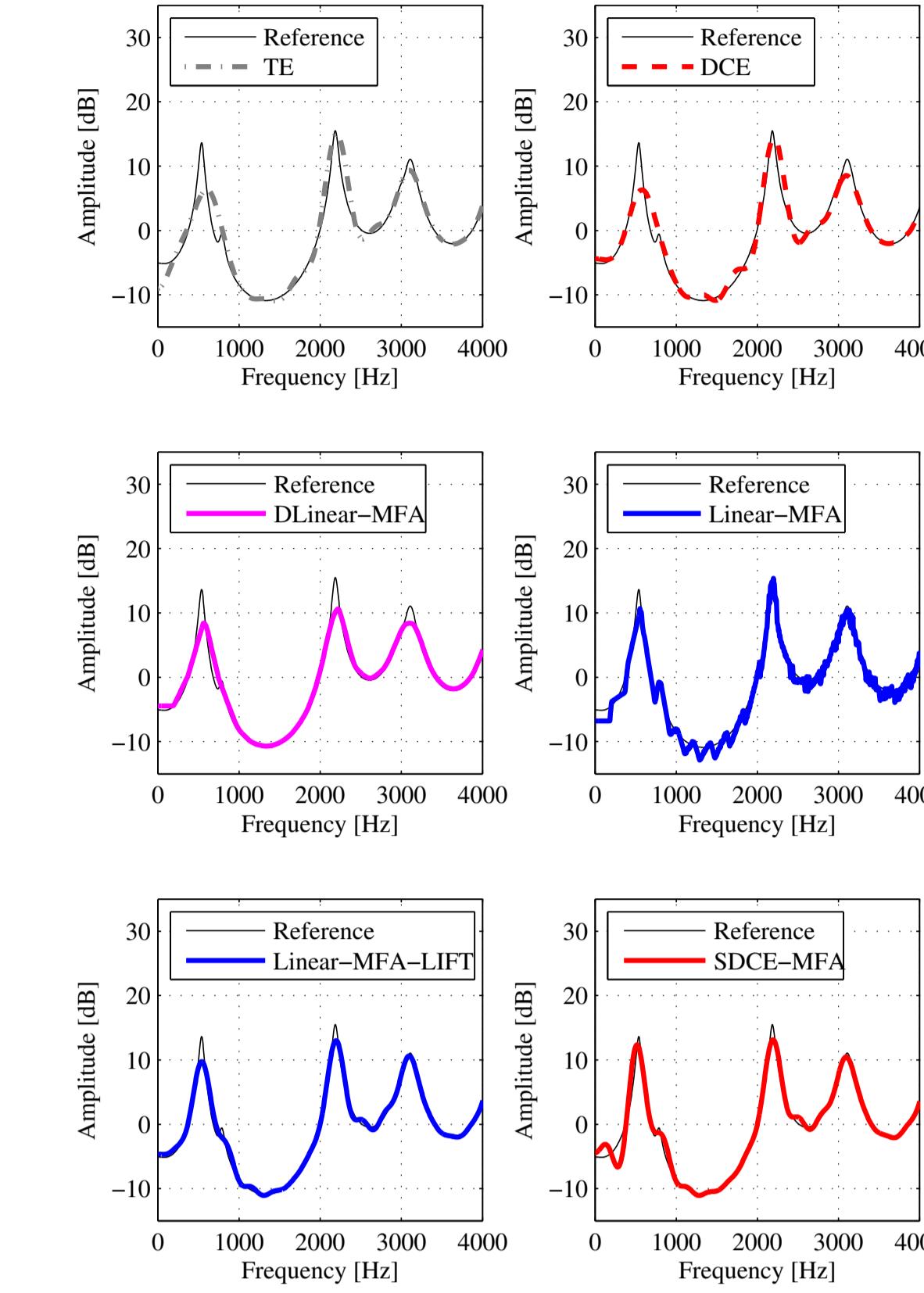
ircam
Centre Pompidou

EVALUATION

Compared methods

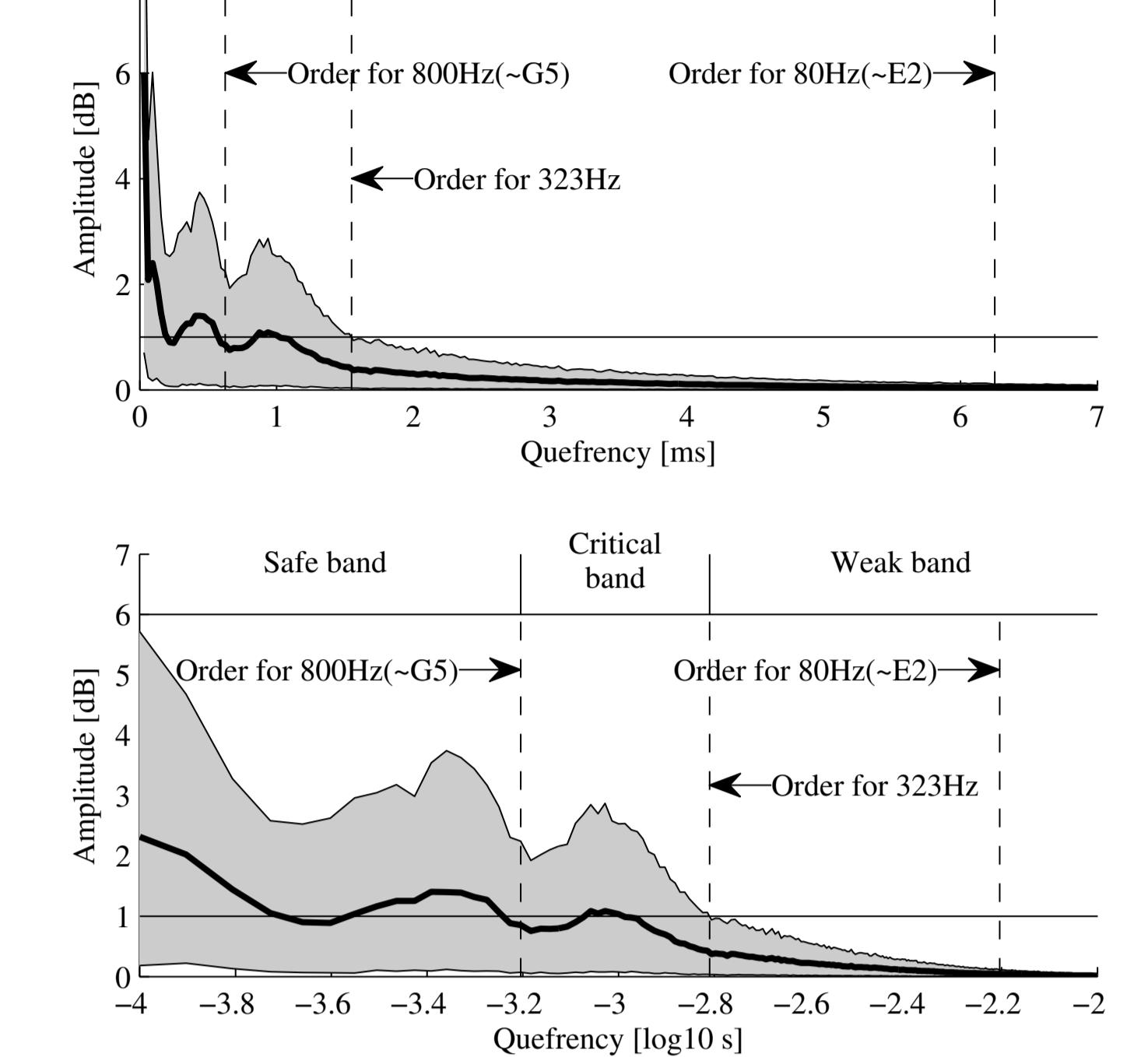
TE=True Envelope (SFA)

DCE=Discrete Cepstral Envelope (SFA)



Material

1000 synthetic signals of 2s; f_0 : vibrato with central freq. in [80, 800]Hz; AM component: low-passed filtered Gaussian noise; VTF from digital acoustic synthesizer.



MEASUREMENTS

Absolute cepstral error (AC Error)

$$\epsilon_n = \frac{1}{M} \sum_{m=1}^M |c_n^* - c_{m,n}|$$

M : the number of frames

c_n^* : The known VTF.

Relative cepstral error (RC Error)

$$\rho_n = \frac{1}{M} \sum_{m=1}^M \left| \frac{c_n^* - c_{m,n}}{c_n^*} \right|$$

Relative Cepstral Excess

(RC Excess): Risk of degeneration or incoherent resonances.

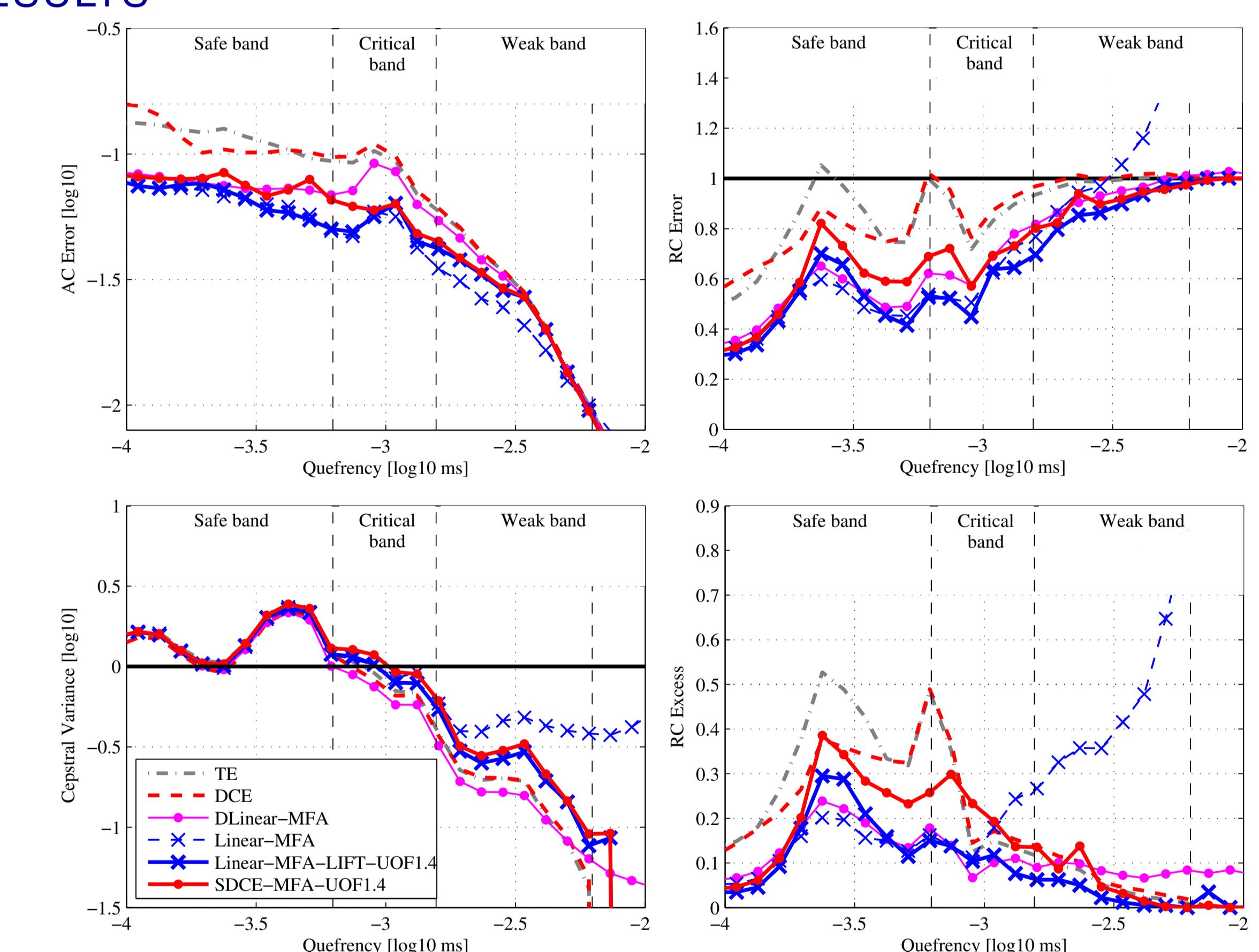
$$\bar{\sigma}_n = \frac{\text{std}_i(\bar{c}_{n,i})}{\text{std}_i(\bar{c}_n^*)} \quad \bar{c}_{n,i} = \frac{1}{M} \sum_{m=1}^M c_{m,n,i}$$

$\bar{c}_{n,i}$: the average cepstrum of sample i .

$$\chi_n = \frac{1}{M} \sum_{m=1}^M \max(\{\rho_{m,n}, 1\}) - 1$$

$$\rho_{m,n} = \left| \frac{c_n^* - c_{m,n}}{c_n^*} \right|$$

RESULTS



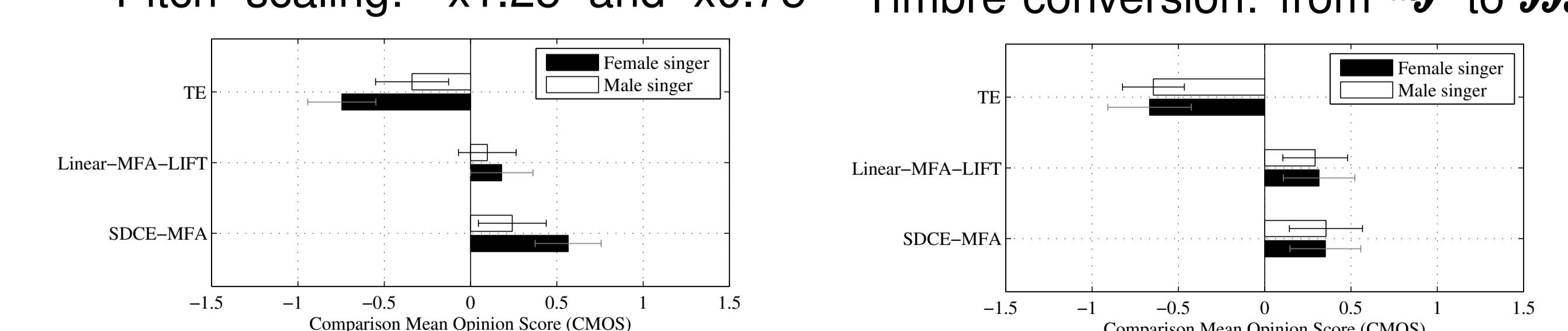
- RC Error, safe band: MFA divides the error by ~ 2 compared to SFA.
- Critical and weak bands, cepstral Variance $< 1 \Rightarrow$ averaging of the envelopes (without any statistical modeling!). The DL-MFA suffers the most. SDCE-MFA is the best.
- RC Excess: SFA produces substantial erratic shapes, even in the safe band. Much less problems for the MFA.

LISTENING TESTS

Material: 1-3s of sustained vibrato; 15 French vowels; 2 voices (!)

Method: Harmonic synthesis[6]

Pitch scaling: x1.25 and x0.75 Timbre conversion: from **mf** to **ffff**



• MFA clearly preferred to TE.

• Linear-MFA-LIFT: good results and efficient !

[4] Y. Shiga, S. King, "Estimating the spectral envelope of voiced speech using multi-frame analysis", EUROSPEECH, 2003.

[5] T. Wang, T. Quatieri, "High-pitch formant estimation by exploiting temporal change of pitch", IEEE Transactions on Audio, Speech, and Language Processing, 2010.

[6] G. Kafentzis, G. Degottex, O. Rosec, Y. Stylianou, "Pitch Modifications of speech based on an Adaptive Harmonic Model", in ICASSP, 2014.