

Temporal Modulation Processing of Speech signals for Noise Robust ASR

Author: Hong You , Abeer Alwan

Professor: 陳嘉平

Reporter :許峰閣

Outline

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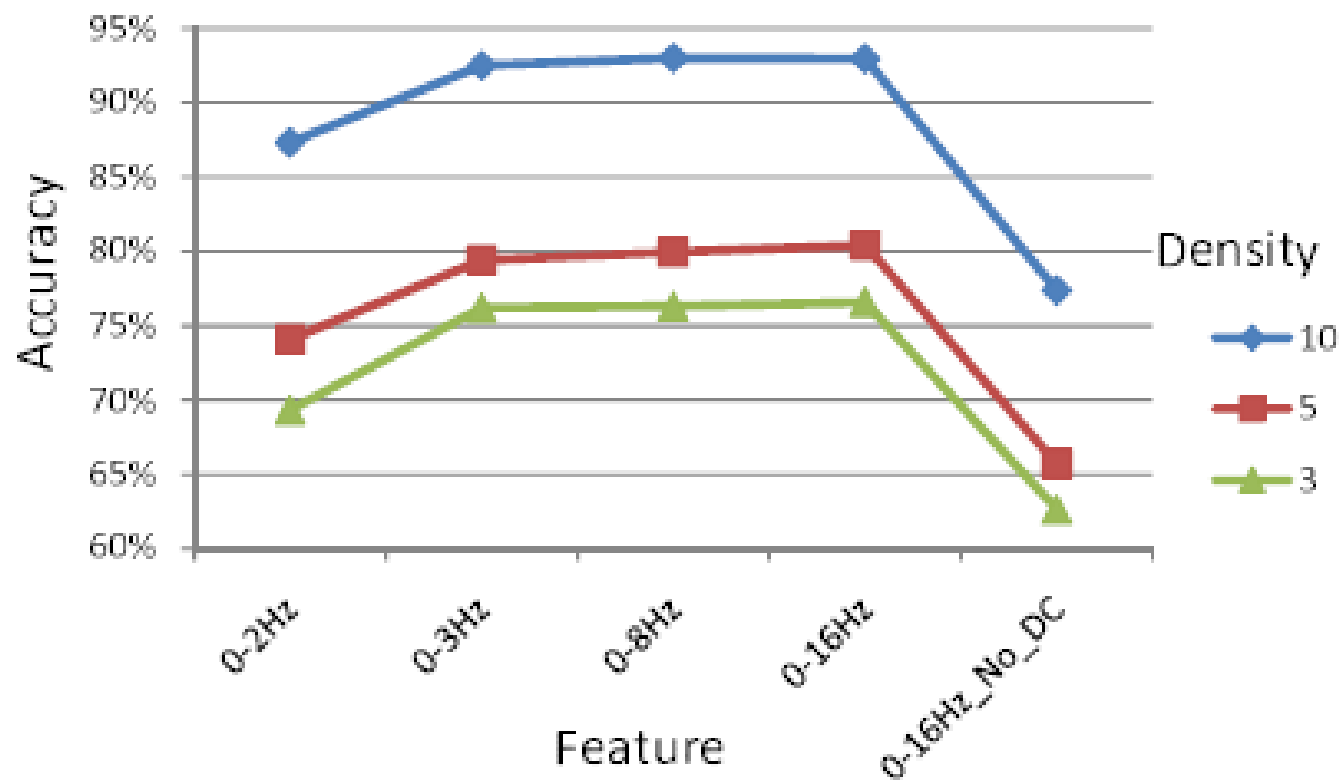
Introduction

- Speech and noise is more accurately classified by low-passed modulation frequency than band-passed.
- The frequency adaptive modulation processing algorithm can attenuate noise sensitive MFs based on estimation from noise robust MFs.

Analysis

- Speech and noise signal have distinct modulation frequency(MF) characteristics.
- Using SVM trained to classify speech/noise modulation features.

Analysis



Analysis

- Band-passed MF performs consistently worse than low-passed ones.
- The result show the important of combining DC modulation along with low MF for the task of speech/non-speech feature classification.

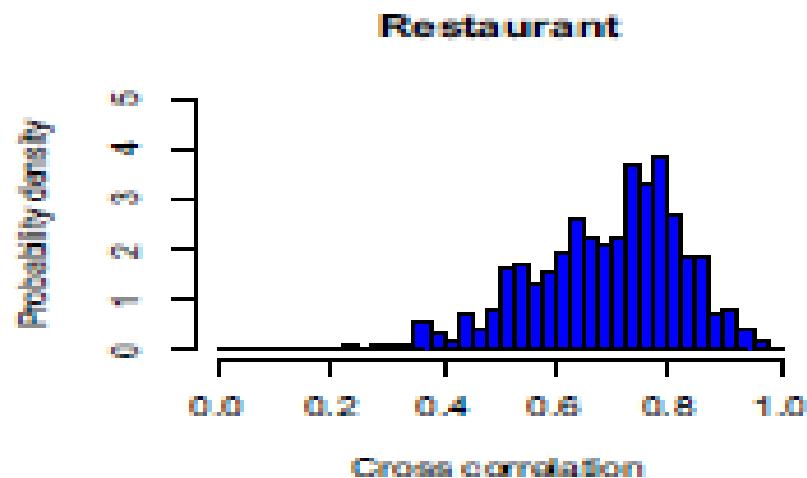
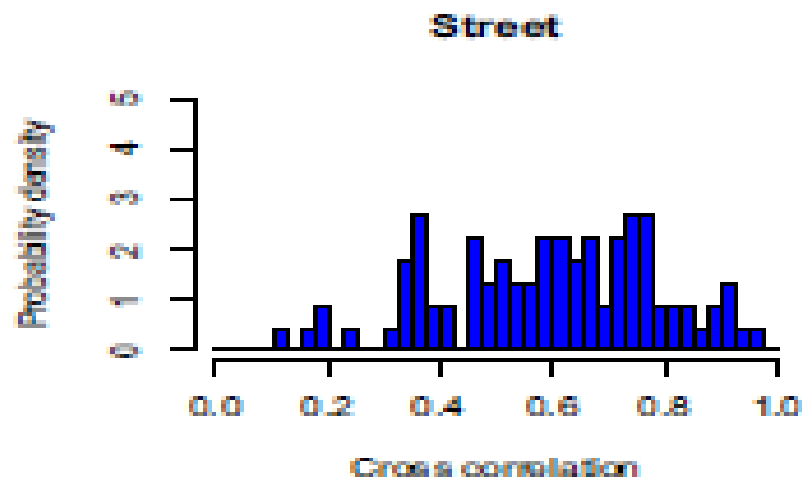
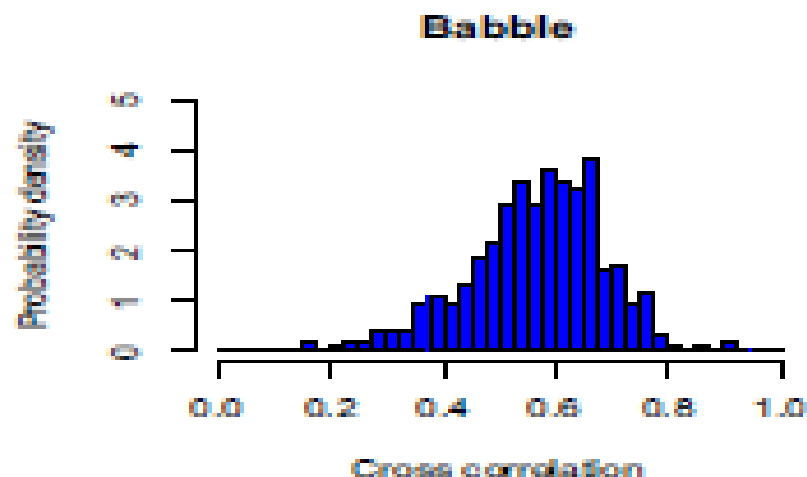
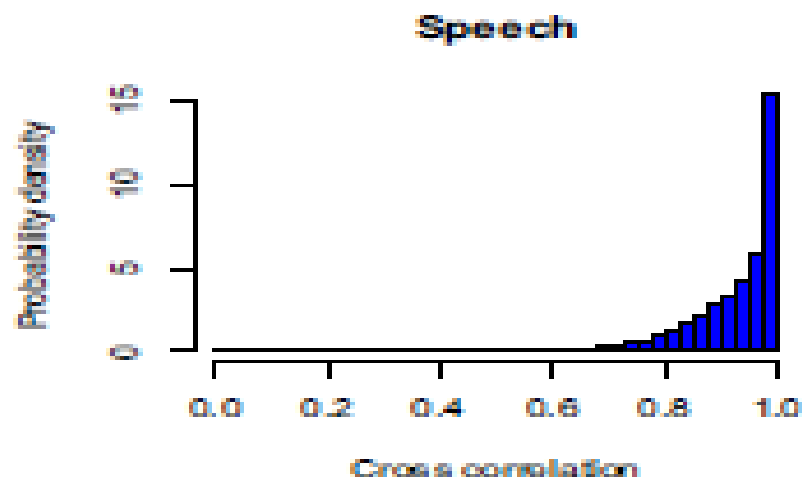
Speech and Noise Modulation characteristics

- We observe that speech MFs have smoother energy transition from 0 Hz to low MFs.
- For noise, sharp energy decrease occur from 0 Hz to low MFs.

Speech and Noise Modulation characteristics

- We analyze the cross-correlation between different MFs across acoustic frequencies.
- The hypothesis that the modulation pattern of speech signals over long segments is more consistent than that of noise.

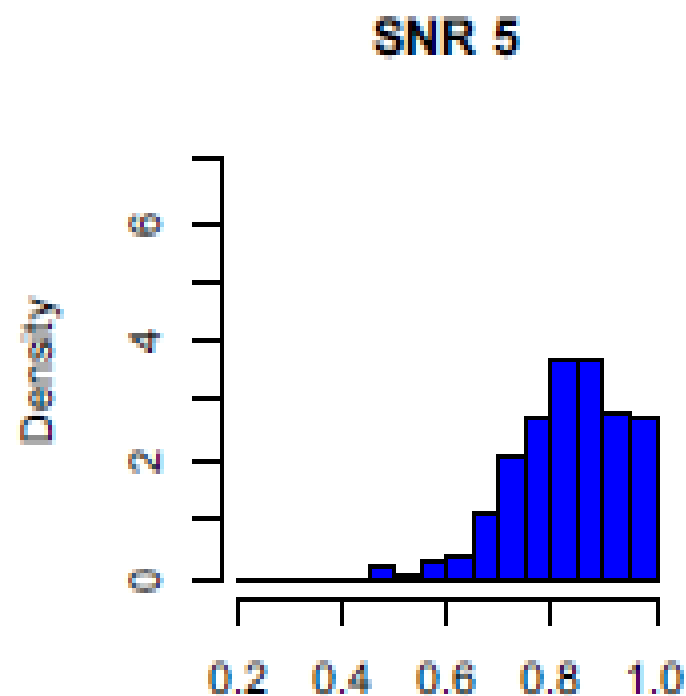
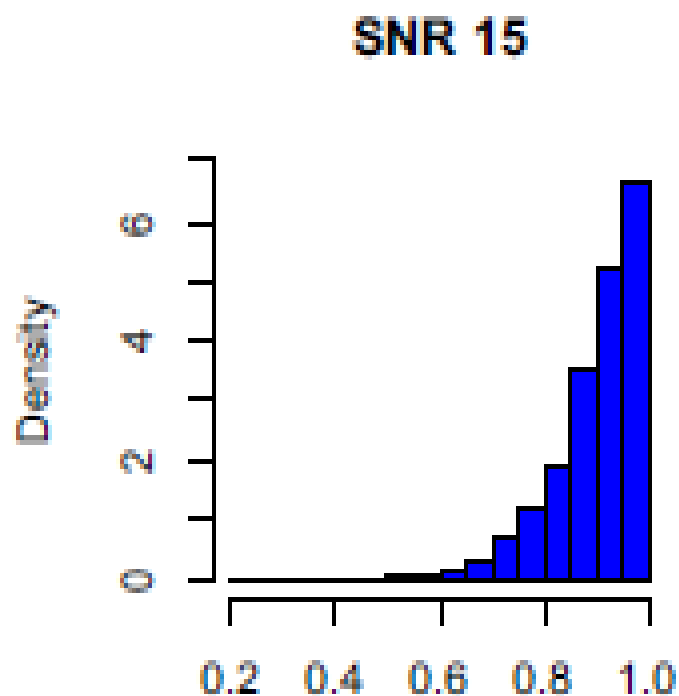
Speech and Noise Modulation characteristics



Speech and Noise Modulation characteristics

- And we will show the cross correlation of speech signal corrupted by subway noise at SNR 15db and 5db.
- It shows that additive noise considerably reduced cross-correlation between MFs.

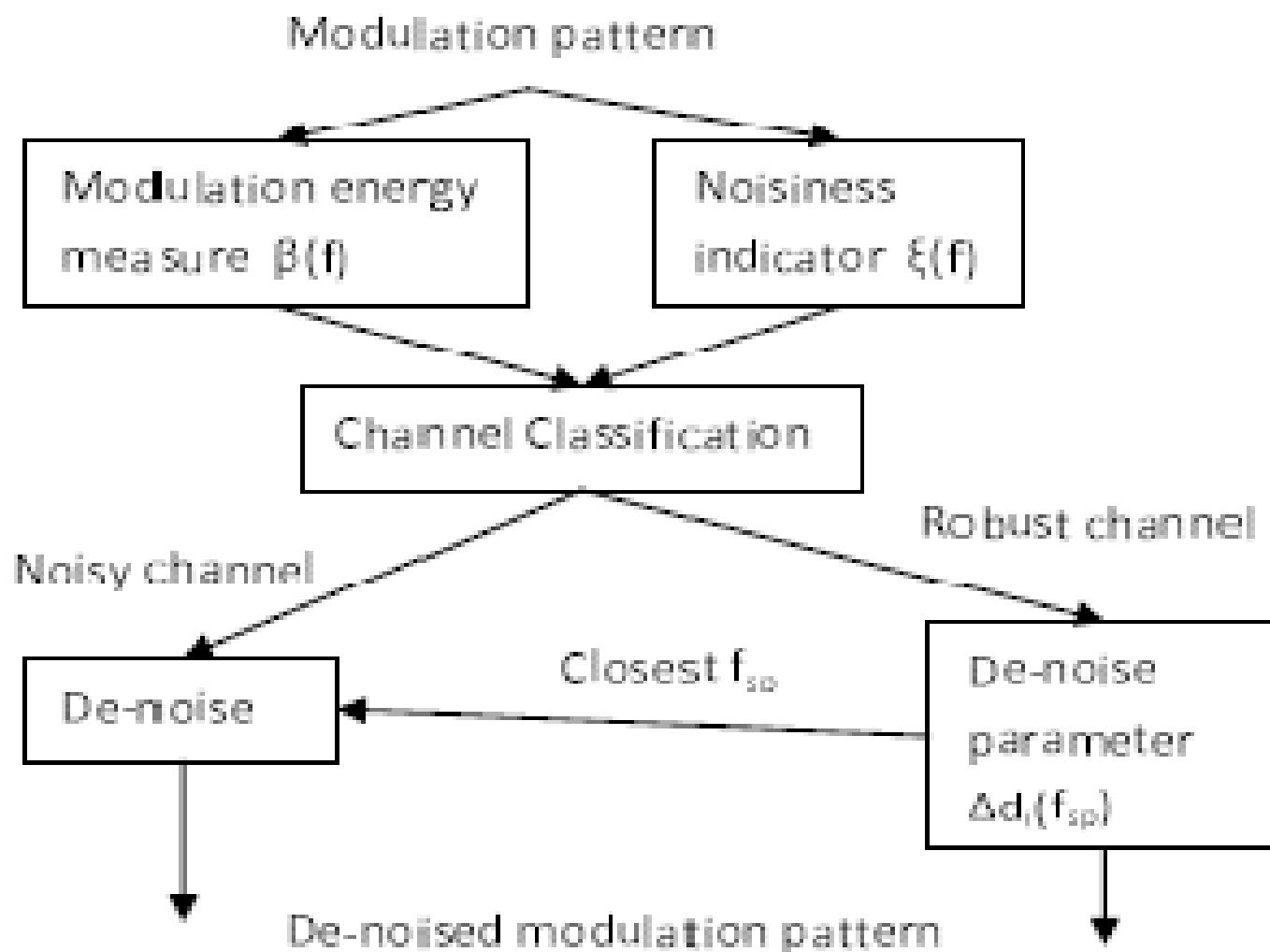
Speech and Noise Modulation characteristics



Frequency Adaptive Modulation processing Algorithm

- The basic idea is to denoised noise-sensitive MFs for noisy channels based on a linear combination of noise-sensitive and noise-robust MFs.
- We assume that noise-sensitive MFs range from 0 Hz to 2 Hz and noise-robust MFs range from 3 Hz to 8 Hz.

Frequency Adaptive Modulation processing Algorithm



Frequency Adaptive Modulation processing Algorithm

- Define:

$d_i(f)$ is the log magnitude at MF i and frequency f .

$$\beta(f) = \sum_{i=3}^8 d_i(f) \quad \text{Modulation energy measure}$$

$$\varepsilon(f) = d_0(f) - d_{1.5}(f) \quad \text{Noisiness indicator}$$

Frequency Adaptive Modulation processing Algorithm

- To classify frequency channels, a recursive min-max scheme is used.

- There are 2 thresholds:

T_{β} is an adaptive threshold adjusted according to local maximum of $\beta(f)$

$$T_{\varepsilon} = 5.5 + 0.33 \times \text{mean of } \varepsilon(f) \text{ over } 0 < f < 50$$

Frequency Adaptive Modulation processing Algorithm

- Speech modulation parameters, $\Delta d_i(f_{sp})$
 f_{sp} is the classified noise robust channel.
For $i = 0, 1, 2$ Hz(MF)

$$\Delta d_i(f_{sp}) = d_i(f_{sp}) - \frac{1}{3} \sum_{j=3}^5 d_j(f_{sp})$$

Frequency Adaptive Modulation processing Algorithm

- For noisy channel f , the closet f_{sp} is used to denoise noisy MFs. For $i=0 \sim 2$ Hz

$$d'_i(f) =_{\Delta} d_i(f_{sp}) + \frac{1}{3} \sum_{j=3}^5 d_j(f)$$

- A noisy MF is denoised if $|d'_i(f) - d_i(f)| > 1.5$ db.

Experiments

- Using Aurora2 database.
- Using HTK tools.
- The MFCC features are used as our baseline system.

Experiments

	MFCC	RASTA-MFCC	ETSI-AFE	Adaptive
SNR20	97.05	95.00	98.4	95.60
SNR15	93.49	90.94	96.32	94.23
SNR10	78.72	80.2	91.53	86.34
SNR5	52.2	63.83	77.92	70.34
SNR0	26.01	37.98	50.91	43.75
SNR-5	11.18	16.79	20.4	18.67
Average	59.77	64.13	72.58	68.16