# Increased MFCC Filter Bandwidth For Noise-Robust Phoneme Recognition

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#### Introduction

- Many speech recognition systems use mel-frequency coefficient(mfcc) feature extraction as a front-end. In the algorithm, a speech spectrum passes through a filter bank of mel-spaced triangular filters, and the filter output energies are log-compressed and transformed to the cepstral domain by the DCT.
- With complex cochlear models of human auditory systems, the filters in the model's filter bandwidth are much wider and overlap with neighboring filters more so than mfcc filters.

◆ The first scheme for widening the mfcc filters increase this overlap while maintaining the bandwidth of entire filter bank. Thus the triangle base length L for each filter is:

$$L = \frac{\hat{f}_{max} - \hat{f}_{min}}{N(1-m)+m}$$

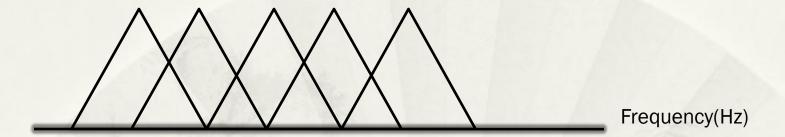
 $\hat{f}_{max}$ : the maximum frequency of the filter bank

 $\hat{f}_{min}$ : the minimum frequency of the filter bank

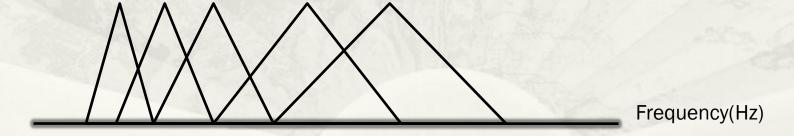
N: the number of filters in the filter bank

m: the percent overlap between adjacent filters bases  $(0 \le m \le 1)$ 

Mel-Frequency:



Linear-Frequency:



• Since  $\hat{f}_{max}$  and  $\hat{f}_{min}$  are constant in this scheme, filter bank center  $\hat{f}_{n}$  is a function of m:

$$\hat{f}_{n} = \frac{L}{2} + (n-1) \frac{\hat{f}_{max} - \hat{f}_{min} - L}{N-1} + \hat{f}_{min}$$

 $\hat{f}_n$ : the center frequency of the  $n^{th}$  filter  $(1 \le n \le N)$ 

 We refer to this algorithm as mfccVW since the filers are variable width according to the free parameter m.

**Example:** N = 3, m=75%,  $\hat{f}_{max} = 1500$ ,  $\hat{f}_{min} = 0$ 

$$\hat{f}_1$$
  $\hat{f}_2$   $\hat{f}_3$ 

$$L = \frac{\hat{f}_{\text{max}} - \hat{f}_{\text{min}}}{N(1-m)+m} = \frac{1500-0}{3(1-0.75)+0.75} = 1000$$

$$\frac{L}{2}$$
=500,  $\frac{\hat{f}_{max}-\hat{f}_{min}-L}{N-1} = \frac{1500-0-1000}{3-1} = 250$ 

$$\hat{f}_1 = 500 + 250(1-1) = 500$$

$$\hat{f}_2 = 500 + 250(2-1) = 750$$

$$\hat{f}_3 = 500 + 250(3-1) = 1000$$

• Equivalent rectangular bandwidth(ERB, in Hz) is the bandwidth of a rectangular filter centered at the center frequency of a critical band whose magnitude is maximum magnitude of the critical band and whose energy is the same as that of the critical band:

$$ERB = \frac{\int |H(f)|^2 df}{|H(f_0)|^2}$$

• Using psychoacoustical measurements of ERB of human auditory filter with a center frequency of  $\hat{f}_0$  (in kHz)

$$ERB=6.23f_0^2+93.39f_0+28.52$$

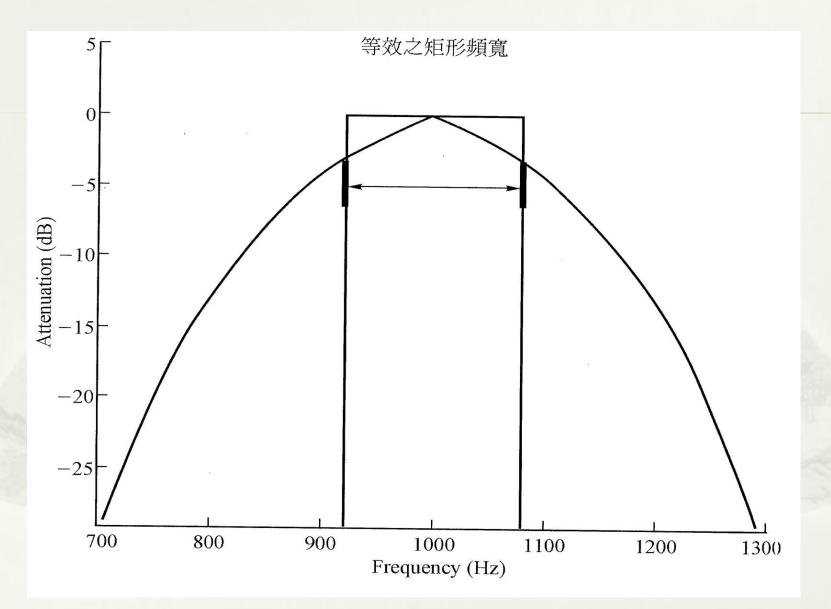


Fig. ERB

- The center frequencies used by the traditional mfcc are kept constant, and the ERB for each center frequency is calculated.
- The mel-frequency warping function between linear frequency  $\hat{\mathbf{f}}$  and mel-frequency  $\hat{\mathbf{f}}$  is :

$$\hat{f}=2595\log_{10}(1+\frac{f}{700})$$

and

$$\hat{f}_0 = \frac{1}{2}(\hat{f}_H + \hat{f}_L)$$
; ERB= $\frac{1}{3}(f_H - f_L)$ 

are solved for  $f_{\rm H}$  and  $f_{\rm L}$  when  $|f_0\>$  is given and ERB is calculated.

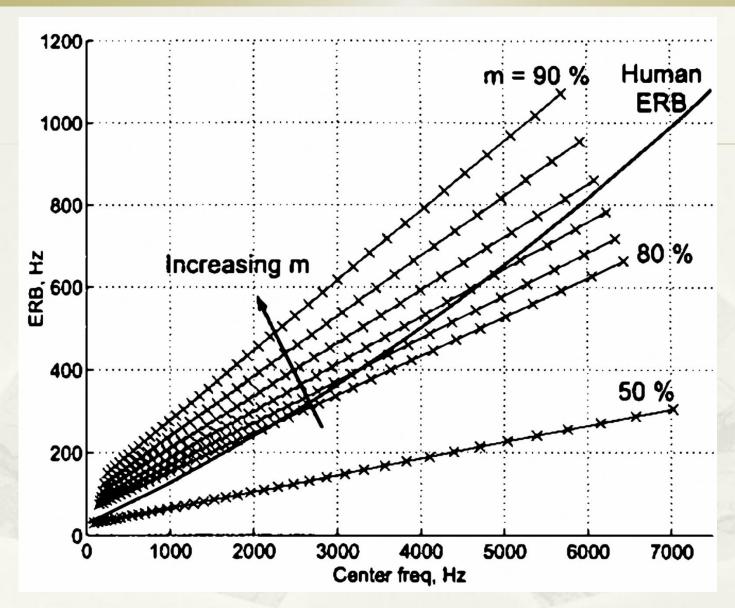


Fig1. ERB vs Frequency for mfcc (m=50%) and mfccVW (m=80%,82%,...,90%) as well as for the human auditory system. Each x corresponds to a filter center frequency.

#### **Experiment And Conclusion**

- ◆ To characterize our modified filter banks, we perform two experiments on vowel extracted from the TIMIT database. The vocabulary consists of 10 vowels (/IY/,/IH/,/EH/,/AE/,/AA/,/UH/,/UW/,/AH/,/ER/) extracted from read sentences according to the phonetic labels provided by the corpus(~50,000 phonemes in all).
- The first experiment using the Fisher discriminant (J-measure)
- The second experiment using a Bayes classifier.

## Experiment (1)

- The first experiment measures the Fisher discriminant (J-measure) for the 10-class problem. This measure compares variance between classes to variance within each class.
- Larger J-measures denote greater separation between classes.

$$J=trace(S_W^{-1}S_B)$$

$$S_{W} = \sum_{k=1}^{c} S_{k} = \sum_{k=1}^{c} N_{k} \sum k$$
  $S_{B} = \sum_{k=1}^{c} N_{k} (m_{k} - m_{0}) (m_{k} - m_{0})$ 

S<sub>w</sub>: the within-class scatter

 $S_{\rm B}$ : the between-class scatter

m<sub>o</sub>: the mean vector of the entire data set

 $N_k$ : the number of samples in the  $k^{th}$  class(c classes total)

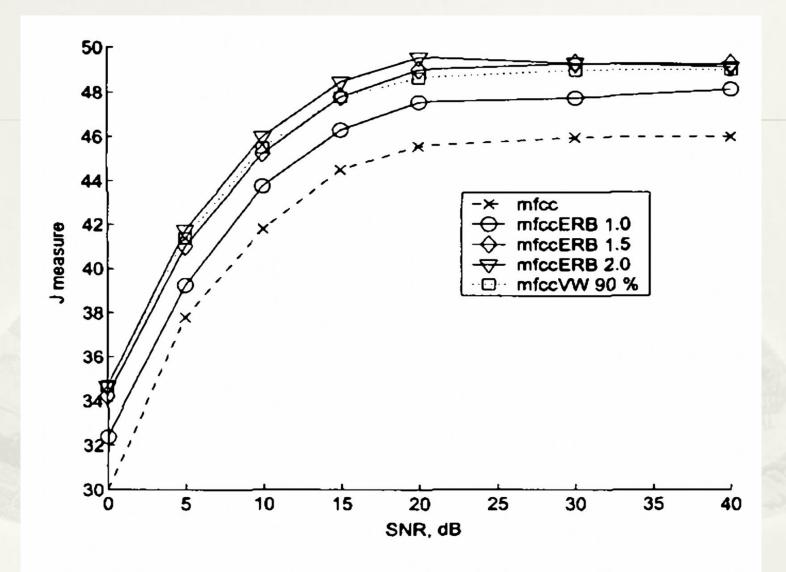


Fig. 2. J-measure vs SNR for mfcc, mfccVW, and mfccERB (inflation factors 1.0, 1.5, and 2.0).

#### Experiment (2)

- The second experiment using a Bayes classifier, each class is divided into test and train data(80% train).
- Classification is determined by maximizing the discriminant function:

$$g_k = x^T W_k x + w_k^T x + \omega_{k0}$$

where

$$W_k = -\frac{1}{2} \sum_{k}^{-1}$$

$$w = \sum_{k=0}^{-1} m_k$$

$$\omega_{ko} = -\frac{1}{2} m_k^T \sum_{k=1}^{-1} m_k - \frac{1}{2} \log \left| \sum_{k=1}^{-1} k \right| + \log P_k$$

for covariance  $\sum k$  and mean  $m_k$  of the  $k^{th}$  class with a priori probability  $P_k$ . x is the vector of cepstral coefficients for the test data and is classified as  $arg({}^{max}_kg_k)$ .

#### The mfccVW results

(1) Performance at all SNRs is nearly maximized for m near 90%.

(2) The dramatic change in recognition results for m > 90%.

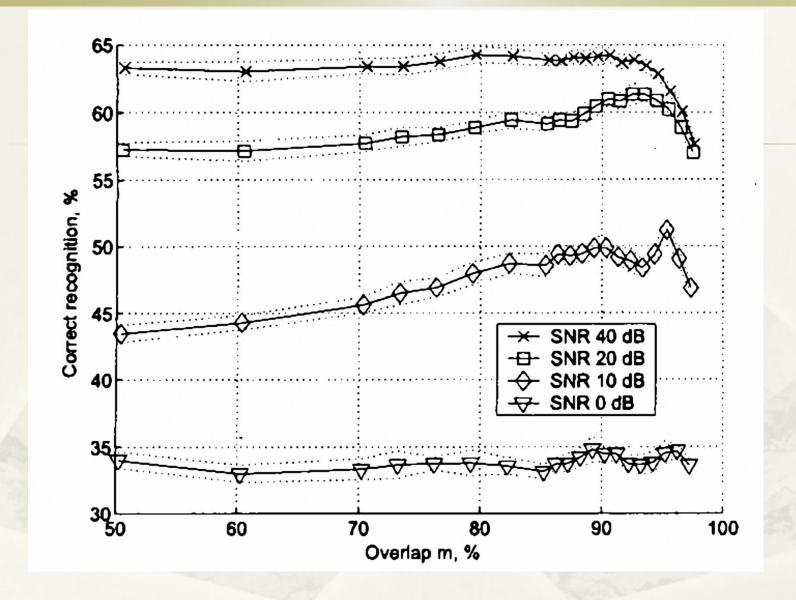


Fig.3 (a)mfcc and mfccVW for various SNRs vs Overlap percentage m

#### The mfccERB results

- (1) Recognition is nearly the same between 30-40 dB SNR while the other filter banks increased in performance over the same range.
- (2) The results using the largest mfccERB scale factor are highest for moderate noise (5-30 dB).
- (3) Below 5 dB SNR

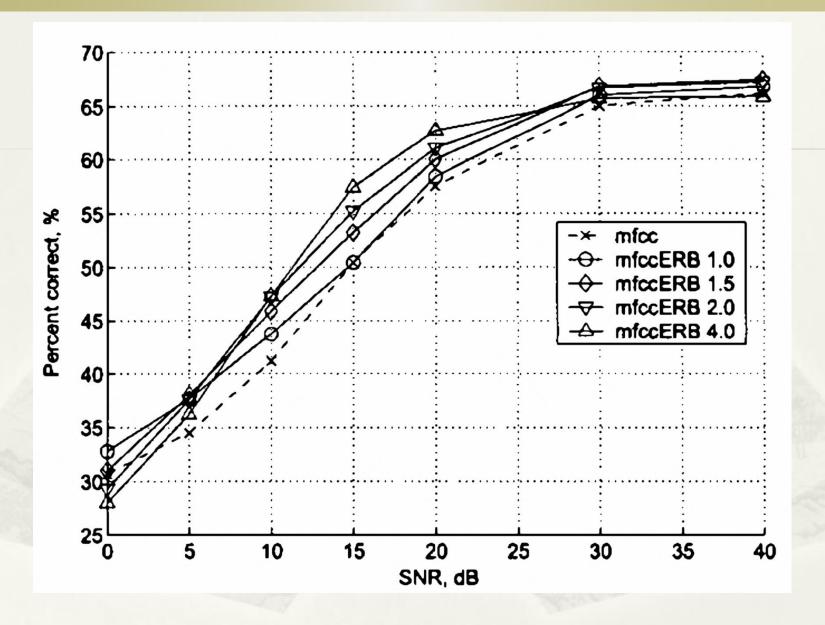


Fig.3 (b) mfcc and mfccERB with ERB scale factors 1.0, 1.5, 2.0 and 4.0