Improved feature extraction based on spectral noise reduction and nonlinear feature normalization

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Introduction

- Results for Noisy TI-Digits at ICASSP'02
 - ★ Histogram Equalization (HE) can reduce the mismatch of noisy speech better than CMS and CMVN
 - ★ Its performance is increased when applied over partially compensated speech features
- ❖ Results for AURORA 2 and 3 at ICSLP'2002
 - ★ Feature extraction combining spectral noise reduction and cepstral histogram equalization for robust ASR
- In this work we explore CDF-matching performance in combination with Wiener filtering





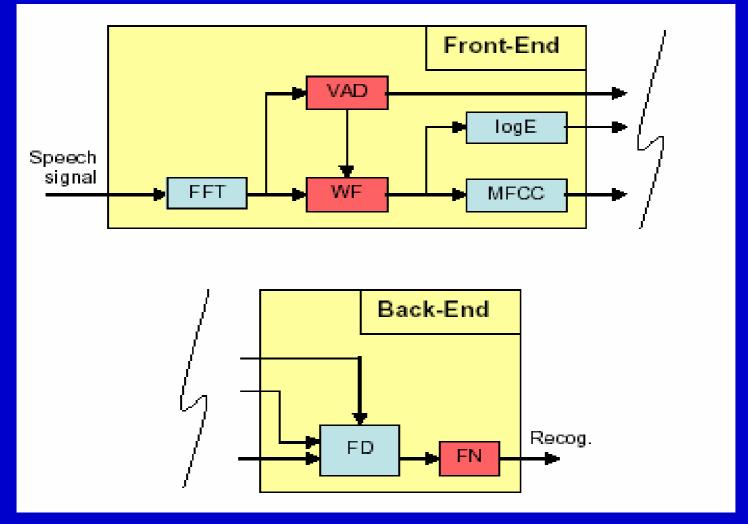
Outline

- System description
- Front-End Spectral Noise Reduction
 - **★** Speech/Non-Speech Detection
 - **★** Spectral noise reduction
- Back-End Processing
 - **★** Frame-Dropping
 - **★** Feature Normalization
- Experimental set-up
- Results and discussion





System Description







Speech/Non-Speech Detection (I)

- Long Term Spectral Estimation VAD algorithm
- LTSE estimation using a sliding window of 3 frames

$$LTSE(k) = \max\{X(k, n+l)\}_{l=-N}^{l=+N}$$

Decision rule

$$LTSD = 10\log_{10}\left(\frac{1}{NFFT}\sum_{k=0}^{NFFT-1}\frac{LTSE^{2}(k)}{Ne^{2}(k)}\right)$$

* LTSD is compared with an adaptive threshold γ





Speech/Non-Speech Detection (II)

Threshold γ function of the noise energy

$$\gamma = \begin{cases} \gamma_0 & E \le E_0 \\ \frac{\gamma_0 - \gamma_1}{E_0 - E_1} (E - E_0) + \gamma_0 & E_0 < E < E_1 \\ \gamma_1 & E \ge E_1 \end{cases}$$

VAD parameters

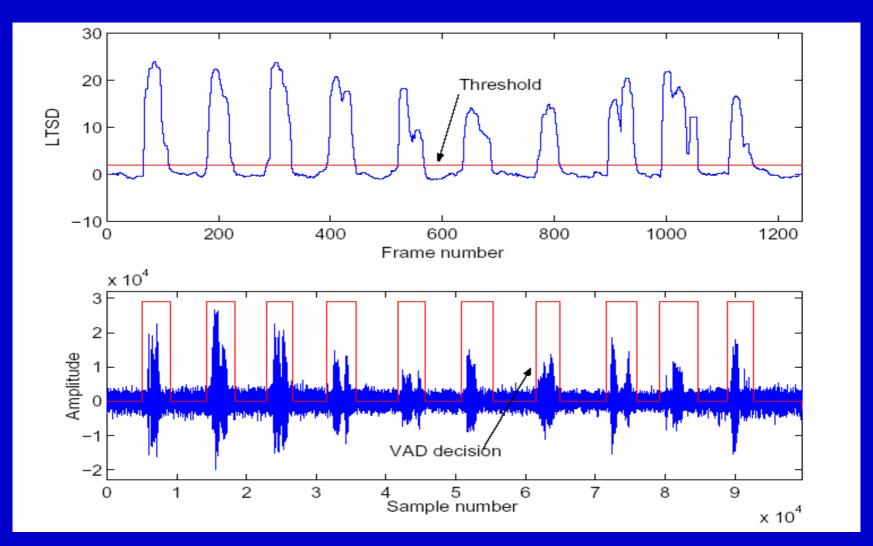
$$N = 3$$
 $NFFT = 256$
 $\gamma_0 = 5dB$ $E_0 = 30dB$ (low noise energy)
 $\gamma_1 = 1.5dB$ $E_1 = 50dB$ (high noise energy)

- Adaptive VAD to time varying noise environments
- Details of the algorithm
 - ★ A New Adaptive Long-Term Spectral Estimation Voice Activity Detector (EUROSPEECH'03)





Speech/Non-Speech Detection (III)







Spectral noise reduction

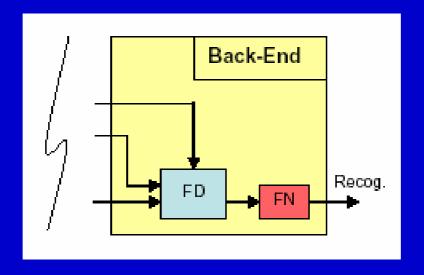
- ❖ Noise reduction implementation as in the first stage of the ETSI ES 202 050 without mel-scale warping.
- * Temporal and frequency smoothing of the magnitude spectrum of the noisy frames is applied.
- Maximum attenuation is fixed at 22dB.
- FIR filter with 17 taps is obtained.
- ❖ Noise of spectrum estimation as (with λ = 0.99)

$$\left| \hat{N}_{t}(w) \right| = \begin{cases} \lambda \left| \hat{N}_{t-1}(w) \right| + (1-\lambda) \left| Y_{t}(w) \right| & \text{Non-Speech} \\ \left| \hat{N}_{t-1}(w) \right| & \text{Speech} \end{cases}$$





Back-end processing



- Frame Dropping
 - ★ Remove all the frames labeled as non-speech
- Feature Normalization
 - **★** ECDF-matching





Feature Normalization (I)

CDF-matching for non-linear distortion compensation

★ Given a zero-memory one-to-one general transformation y=T[x]

$$x \to p_X(x)$$

$$y = T[x] \rightarrow p_Y(T[x]) = p_Y(y)$$

$$C_X(x) = \int_{-\infty}^x p_X(u) \, du$$

$$C_Y(y) = \int_{-\infty}^{y} p_Y(u) \, du$$

$$C_X(x) = C_Y(y)$$

$$\Rightarrow x = T^{-1}[y] = C_X^{-1}(C_Y(y))$$





Feature Normalization (II)

- CDF-matching for feature normalization
 - \star A predefined $C_X(x)$ is selected (usually Gaussian)
 - ***** For both training and test, features are transformed to match the reference distribution using an estimate of $C_Y(y)$
 - ★ Can be viewed as an extension of CMVN
- Implementation details
 - ★ CDF-matching is applied in the cepstrum domain in a feature transformation scheme
 - ★ Each cepstral coefficient is transformed independently to match a Gaussian reference distribution





Feature Normalization (III)

Ecdf Algorithm:

★ Temporal buffer for a given distorted features

$$Y_t = \{y_{-T}, \dots, y_t, \dots, y_T\}$$

★ Order statistics of data

$$y_{(1)} \le y_{(2)} \le \dots \le y_{(r)} \dots \le y_{(2T+1)}$$

★ Asymptotically unbiased point estimation of the CDF

$$\hat{x}_{t} = C_{x}^{-1} \left(\hat{C}(y_{t}) \right) = C_{x}^{-1} \left(\frac{r(y_{t}) - 0.5}{2T + 1} \right)$$

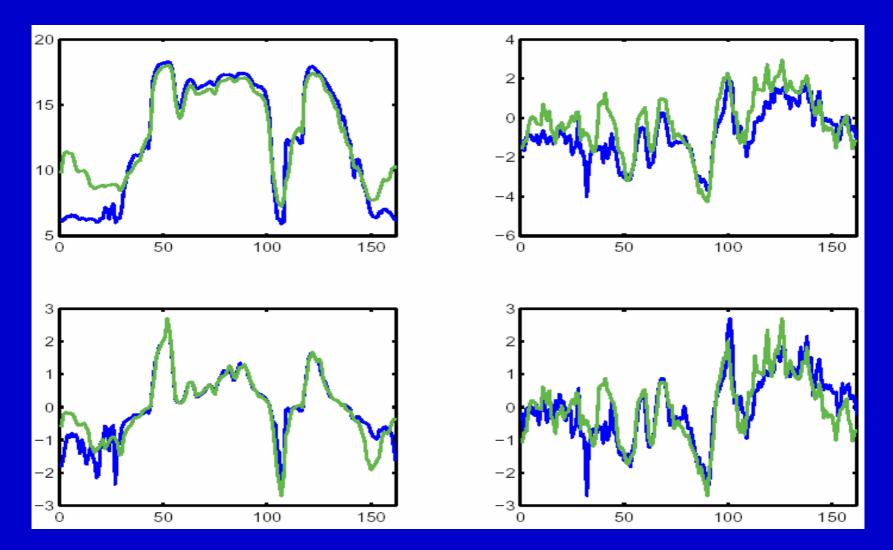
★ Estimation of the transformed value of the distorted feature

$$\hat{C}(y_{(r)}) = \frac{r - 0.5}{2T + 1}$$
 $r = 1, \dots, 2T + 1$





Feature Normalization (IV)







Experimental set-up (I)

Database end-pointing

- ★ Noisy TI-digits and SpeechDat Car databases have been automatically end-pointed
- ★ SND algorithm is used on clean speech (channel 0) utterances
- ★ 200ms of silence have been added at the end-points

Acoustic features

- ★ Standard front-end: 12 MFCC + logE
- ★ Delta and acceleration coefficients are appended at the recognizer with regression lengths of 7 and 11 frames respectively

Acoustic modeling

- ★ One 16 emitting states left-to-right continuous HMM per digit
- ★ 3 Gaussian mixture per state for AURORA 3
- ★ 20 Gaussian mixture per state for AURORA 2





Experimental set-up (II)

Batch implementation

★ Using all the features of a given input utterance to perform the normalization

Segmental implementation

- **★** Non-stationary noise
- ★ Using a short temporal window around the frame to be normalized
- ★ 121 frames of temporal window





Experimental Results (I)

- Results with Batch implementation
 - **★** Comparative results over the previous system (ICSLP'02)

Aurora 2 Relative Improvement							
	Set A Set B Set C Overall						
Multi	16,47%	21,79%	20,70%	19,44%			
Clean	30,46%	30,59%	28,78%	30,18%			
Average	23,46%	26,19%	24,74%	24,81%			

Aurora 3 Relative Improvement							
	Finnish Spanish German Danish Average						
Well (x40%)	23,62%	6,57%	19,52%		16,57%		
Mid (x35%)	20,12%	-8,98%	15,34%		8,83%		
High (x25%)	52,81%	21,36%	19,19%		31,12%		
Overall	29,69%	4,82%	17,97%		17,50%		

- Spectral Subtraction ----- Wiener filtering
- Quantile based VAD
 LTSD VAD
- Histogram Equalization ---- ECDF





★ Comparative results over AFE

Aurora 2 Word Error Rate								
	Set A Set B Set C Overall							
Multi	6,16%	6,50%	7,27%	6,52%				
Clean	12,83%	12,07%	13,63%	12,69%				
Average	9,49%	9,28%	10,45%	9,60%				

Aurora 2 Relative Improvement							
	Set A Set B Set C Overall						
Multi	-8,59%	-4,21%	-3,86%	-5,89%			
Clean	-21,13%	-10,50%	-4,46%	-13,54%			
Average	-14,86%	-7,35%	-4,16%	-9,72%			

Aurora 3 Word Error Rate								
	Finnish	Finnish Spanish German Danish Averag						
Well (x40%)	4,14%	3,13%	4,37%	6,01%	4,41%			
Mid (x35%)	10,60%	6,43%	10,10%	14,31%	10,36%			
High (x25%)	12,69%	10,20%	8,93%	21,07%	13,22%			
Overall	8,54%	6,05%	7,52%	12,68%	8,70%			

Aurora 3 Relative Improvement							
	Finnish Spanish German Danish Average						
Well (x40%)	-5,88%	6,85%	10,63%	9,35%	5,24%		
Mid (x35%)	44,44%	-5,76%	-10,26%	22,69%	12,78%		
High (x25%)	5,23%	-20,71%	-2,06%	-3,23%	-5,19%		
Overall	14,51%	-4,45%	0,15%	10,87%	5,27%		





Segmental Implementation

Aurora 2 Word Error Rate									
Set A Set B Set C Overall									
Multi	6,33%	6,55%	7,51%	6,65%					
Clean	13,16%	12,04%	13,64%	12,81%					
Average									

Aurora 2 Relative Improvement							
	Set A Set B Set C Overall						
Multi	-12,68%	-7,21%	-8,87%	-9,73%			
Clean	-28,78%	-14,51%	-9,10%	-19,14%			
Average	-20,73%	-10,86%	-8,99%	-14,43%			

Aurora 3 Word Error Rate							
	Finnish Spanish German Danish Average						
Well (x40%)	4,14%	3,31%	5,09%	6,68%	4,80%		
Mid (x35%)	10,60%	6,61%	11,27%	16,86%	11,34%		
High (x25%)	13,25%	8,99%	10,78%	20,44%	13,37%		
Overall	8,68%	5,89%	8,68%	13,68%	9,23%		

Aurora 3 Relative Improvement							
	Finnish Spanish German Danish Average						
Well (x40%)	-5,88%	1,49%	-4,09%	-0,75%	-2,31%		
Mid (x35%)	44,44%	-8,72%	-23,03%	8,91%	5,40%		
High (x25%)	1,05%	-6,39%	-23,20%	-0,15%	-7,17%		
Overall	13,46%	-4,05%	-15,50%	2,78%	-0,83%		





Conclusions

- ❖ Feature extraction algorithm based on the combination of spectral noise reduction and nonlinear features normalization
- New VAD based on Long Term spectral envelope
 - ★ Improve the noise estimation
 - **★** Frame dropping
 - **★** Better discrimination speech/noise
- More computational efficiency of the feature normalization algorithm
- Segmental version of the feature normalization algorithm
 - ★ Performance is only slightly worse
- Results presented for AURORA 2 and AURORA 3







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These slides are available at http://sirio.ugr.es/segura/pdfdocs/eurospeech'03_sl.pdf