

# The USTC Systems for The NIST-2006 Speaker Recognition Evaluation

Beiqian Dai, Yanlu Xie, Xi zhou,  
Zhiqiang Yao, Jixu Chen, Minghui Liu

## *Introduction*

Participant Task:

		Test Segment Condition			
		10 sec 2-chan	1 conv 2-chan	1 conv summed- chan	1 conv aux mic
Training Condition	10 seconds 2-channel	<input type="radio"/>			
	1 conversation 2-channel	<input type="radio"/>	<input type="radio"/>		
	3 conversation 2-channel	<input type="radio"/>	<input type="radio"/>		
	8 conversation 2-channel	<input type="radio"/>	<input type="radio"/>		
	3 conversation summed- channel		<input type="radio"/>	<input type="radio"/>	



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A slide titled "Main Modules" with a list of seven processing steps. The slide includes the USTC logo and a globe icon.

**Main Modules**

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- FrontEnd Processing
- Universal Background Model Training
- Speaker Model Adaptation
- LLR Score Computation
- Fusion
- Making Decision



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## FrontEnd Processing

- FrontEnd Processing for MFCC
- FrontEnd Processing for LPCC
- FrontEnd Processing for Pitch
- FrontEnd Processing with Wavelet



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## FrontEnd Processing for MFCC & LPCC

- Band-limited (300Hz – 3400Hz)
- MFCC+Delta(16+16) with the 0th removed
- RASTA
- CMS
- Remove Silence
- Feature Warping



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## Silence Removal

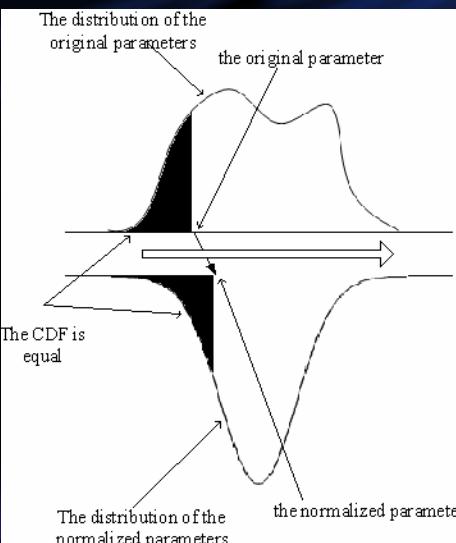
- Energy based threshold to remove long period silence
- Predictive Segment
  - $H_0$  : current frame is a new segment first frame
  - $H_1$ : current frame is belong to previous segment
  - $|X_t - Seedt-1| < |X_t - O|$  , choose  $H_0$ ,
  - Else, choose  $H_1$
- Energy & Duration based threshold to remove silence segment



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## Feature Warping or Short-time Gaussianization



$$T(x) = \Phi^{-1}(F_X(x))$$

$$F_X(x) = p_X(X \leq x) = \int_{-\infty}^x p_X(t) dt$$

$$\Phi(x) = \int_{-\infty}^x \phi(t) dt = \int_{-\infty}^x \frac{1}{\sqrt{2\pi}} e^{-\frac{t^2}{2}} dt$$



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## FrontEnd Processing for pitch

We firstly split pitch and energy contours into segment with 7 frames length. 4 parameters related to pitch were extracted:

- log (mean\_F0) averaged over a segment
- log (max\_F0) of a segment
- log (min\_F0) of a segment
- F0\_slop of a segment



Another 4 parameters related to energy are extracted as above. Total 8 parameters of a segment comprise an 8-dimension vector.

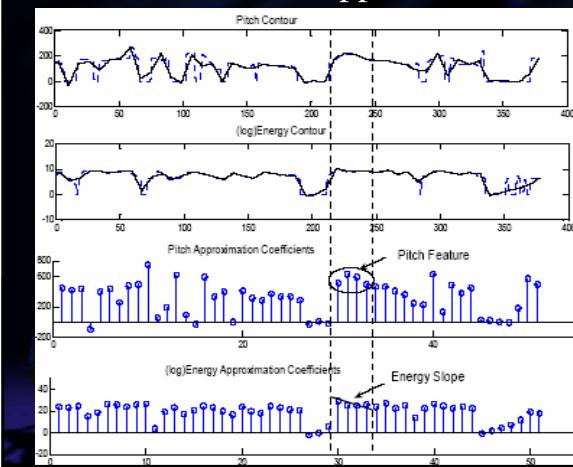


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## FrontEnd Processing with wavelet

We made wavelet analysis of the f0 and energy contour. Subsequently, the prosodic features were extracted only from the 3rd level approximation coefficients



Prosodic Feature:

[cA1 cA2 cA3 cA4 ESlope]



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## Universal Background Model

- Model Type
  - GMM consist of 2048 mixtures (1conv)
  - GMM consist of 512 mixtures (10seconds)
  - UBM\_F for female and UBM\_M for male
- Training data
  - Selected from NIST'04&05 training and test data
- Training Algorithm
  - EM Algorithm



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## Speaker Model Adaptation

- Model Type
  - Same as UBM
- Training data
  - Training data in NIST'06
- Training algorithm
  - MAP from UBM\_M or UBM\_F



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## LLR Score Computation

- Log Likelihood Ratio

$$\Lambda(\mathbf{O}) = \frac{1}{T} \sum_{t=1}^T (\log p(\mathbf{O}_t | \lambda_{tar}) - \log p(\mathbf{O}_t | \lambda_{UBM}))$$

- TNORM

- A speaker-specific T-norm selection
- The closest set of P cohort models are used to Tnorm during run time where P is chosen to be 50.

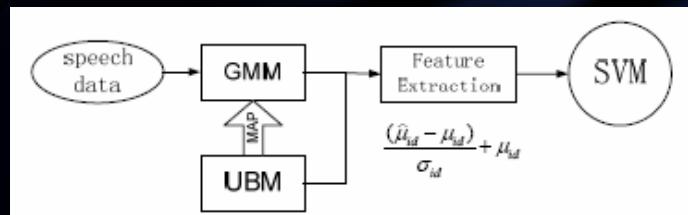


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## SVM system

- Feature: extracted by adapted GMM.
- RBF kernel



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## Fusion

- The scores from the sub-systems are fused with a perceptron classifier. The number of input nodes of the perceptron is the same as the number of sub-systems applied. There is no hidden layers and only one output node.



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## Fusion Step 1

- Clustering and training
  - Clustering the models in NIST'05 for each gender
  - Using the Kullback-Leibler distance and hierarchical agglomerative clustering
  - Each gender contain 4 clusters
  - A perceptron classifier is trained for each cluster and the threshold in each cluster is got, respectively



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## Fusion Step 2

- Classifying and fusion
  - Classify each model in NIST'06 to 1 of the former clusters for each gender
  - Fusion the score of the sub-systems of each 06's model with the corresponding perceptron classifier and threshold.



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## Making Decision

- Threshold is tested with NIST'05 test utterances when the minimal DCF is reached.



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This slide has a dark blue background with a globe in the lower-left corner. At the top left is the logo of the Institute of Acoustics and Technologies (IAT). The title "Main Modules" is centered in a large, white, bold, sans-serif font. Below the title is a horizontal line. To the right of the line is a bulleted list of seven modules: "FrontEnd Processing", "Universal Background Model Training", "Segmentation", "Speaker Model Adaptation", "LLR Score Computation", and "Making Decision".

**Main Modules**

- FrontEnd Processing
- Universal Background Model Training
- Segmentation
- Speaker Model Adaptation
- LLR Score Computation
- Making Decision



## FrontEnd Processing

- Feature for 2-sp Segmentation
  - Band-limited(0Hz - 4000Hz)
  - MFCC(23) (without delta)



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## FrontEnd Processing

- Feature for Speaker Verification
  - Band-limited(300Hz - 3400Hz)
  - MFCC + Delta(16 + 16)
  - RASTA
  - CMS
  - Remove Silence
  - Kurtosis Normalization



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## Universal Background Model

- UBM-F training
- UBM-M training
- Gender Independent UBM training



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## Gender Dependent UBM training (UBM-F and UBM-M)

- Setting
  - 2048 x 1
- Training Data:
  - NIST'04&05 Dev Training Data (IDs are selected)
- Training Algorithm:
  - EM algorithm



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## Gender Independent UBM training

- Setting
  - 4096 x 1
- Training Algorithm
  - Merge from UBM-F and UBM-M



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## Unsupervised Speaker Segmentation

- Hierarchical agglomerative clustering
  - Divide the speech into 1sec segments as initial clusters.
  - Merge two clusters which have minimum pair distance.
  - Until obtain two clusters ( speaker 1, speaker 2)
  - Refine clustering (rescore each 1sec segment by new speaker model and discard some segments with low score)



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## Pair-wise Distance Computing

- Likelihood Ratio Score for Segment

$$L(x:\theta_x) = \prod_{j=1}^r \sum_{k=1}^K g_k(x) N_k(v_j)$$

- Likelihood Ratio

$$\lambda_L = \frac{L(z:\theta_z)}{L(x:\theta_x)L(y:\theta_y)}$$



## Pair-wise Distance Computing

- Transition Probability

$$f(n) \equiv \Pr[S_{i+n} = S_i] = \frac{1 + (2p - 1)^n}{2}$$

- Duration time bias

$$\lambda_D = \frac{\prod_i^C f(n_i)}{\prod_i^C (1 - f(n_i))}$$





## Pair-wise Distance Computing

$$d(x, y) = -\log(\lambda_L) - \alpha \log(\lambda_D)$$

$\alpha = 4$



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## Speaker Model Adaptation

- Setting
  - Same as UBM
- Training data
  - 3 of the 9 Clusters are selected
    - Select most similar 3 clusters from 9 clusters.
- Training algorithm
  - MAP from UBM



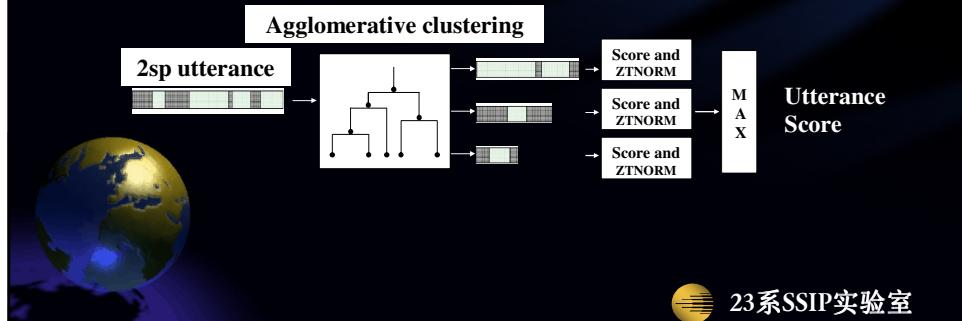
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# LR Score Computation

- Likelihood Ratio Score

$$\Lambda(\mathbf{O}) = \frac{1}{T} \sum_{t=1}^T (\log p(\mathbf{O}_t | \lambda_{tar}) - \log p(\mathbf{O}_t | \lambda_{UBM}))$$



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# Making Decision

- Threshold Selecting
  - NIST05 2-spk Evaluation Test Segments
  - Minimal DCF



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