

THALES



THALES COMMUNICATIONS-LIA NIST SRE 06 Submission

Bertrand Ravera, Francois Capman



n NIST SRE 2006 Evaluation : System description

n Classical GMM-UBM system including:

- ¹ UMB Universal background Model:
 - ¹ 2048 Gaussians
 - ¹ Data: NIST 03 and NIST 04, 600 speakers, 300 male and 300 females,
 - ¹ Learning by classical EM Algorithm
- ¹ Client learning:
 - ¹ Adaptation from MAP algorithm adaptation from Universal Background by MAP algorithm
- ¹ Scoring phase
 - ¹ The 10 top Gaussians form UBM and target model are used
- ¹ Score Normalisation:
 - ¹ T-NORM: 380 impostors
 - ¹ Data NIST 03 et NIST 04, 160 males and 160 females
- ¹ Acoustic parameters normalisation:
 - ¹ Classical cepstral mean suppression and variance normalisation



n Acoustic Parameters

n ESTI Standard AURORA-2 including:

- 1 MFCC parameters extraction,
- 1 Noise reduction stage,
- 1 Blind channel equalisation stage (acoustic parameters normalisation)
- L Compression stage turned off

n Acoustic vector used:

- 1 20 MFCC, 20 Δ MFCC, 20 $\Delta\Delta$ MFCC, $\Delta \log E$, $\Delta\Delta \log E$: 62 coefficients every 10 ms

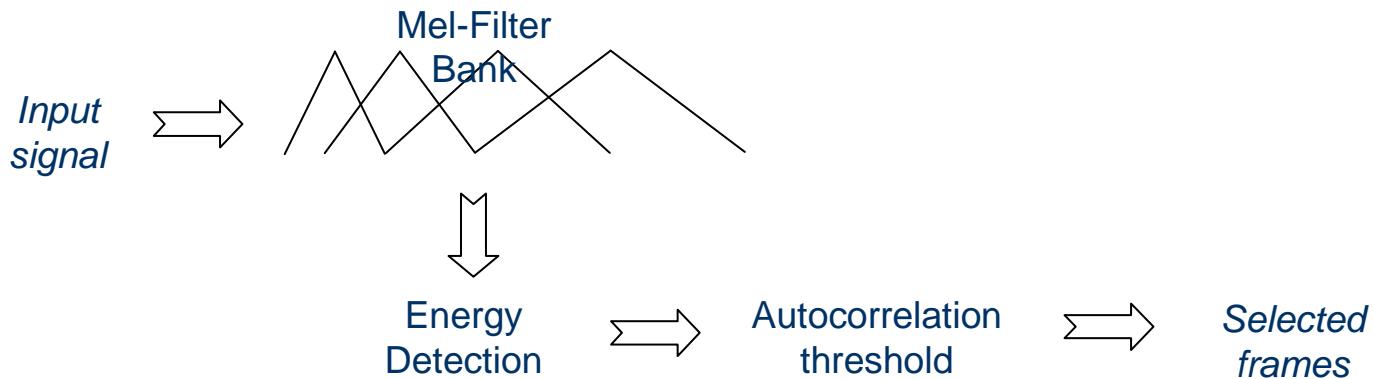
n Statistic Tools

n Collaboration with LIA: Toolkit ALIZE and LiaSpeakerDet 05 tools

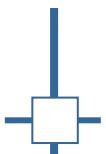


n Description of the frame selection process

- n Sub-band energy detection for pre-selection
 - l GMM modelling of sub-band energy distributions
 - l 4 Mel-spaced sub-bands
 - l 3 Gaussians per sub-band
- n Autocorrelation coefficient
 - l Post-processing of pre-selected frames focusing on **voiced** frames
 - l Autocorrelation coefficient is maximised on candidate pitch lags in the range of 60 Hz to 400 Hz
- n Percentage of frames
 - l On NIST evaluation data less than 30 % of frames are selected

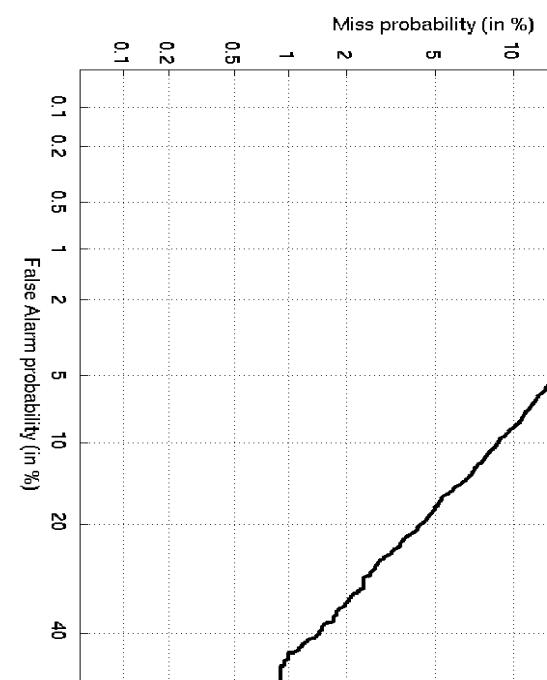
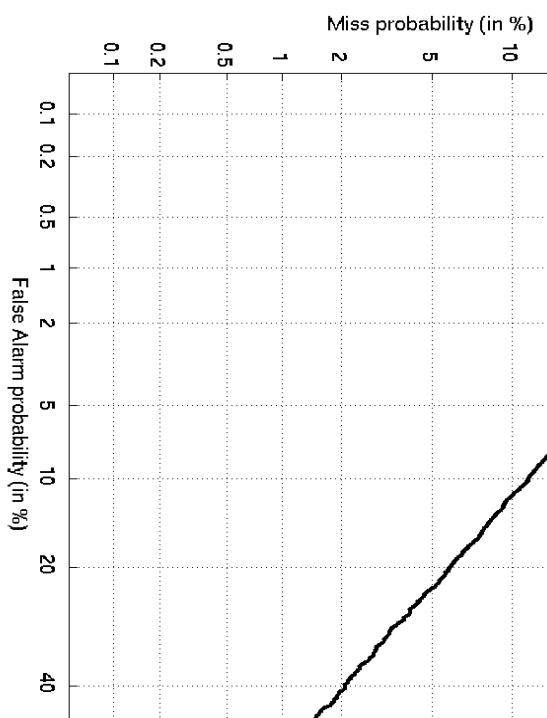


THALES-LIA NIST SRE06 submission



THALES: (THL-1, 1conv4w-1conv4w,n) DET 1 All Trials SRE06

THALES: (THL-1, 1conv4w-1conv4w,n) DET 3 English Trials (Common Test) SRE06



n DCF min DET1: 0,54

n DCF min DET3: 0,45

06/06/2006

NIST SRE06

THALES



4



Some future experiments:

- n Robust extraction of acoustic features
- n Distributed architecture for Speaker Verification (DSV)
 - n Evaluation of AURORA 2 compression module
 - n Evaluation of new compression schemes
- n Targeted applications
 - n Professional network monitoring (specific allocated bit-rate)
 - n Secured-access to server-based applications