

NIST 2005 Speaker Recognition Evaluation

University of New Brunswick
System Submission

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Outline

- Background
- Detection systems
 - Core system description
 - Development data
 - Pitch / energy system
 - Fusion
- Performance
- Current research and future directions

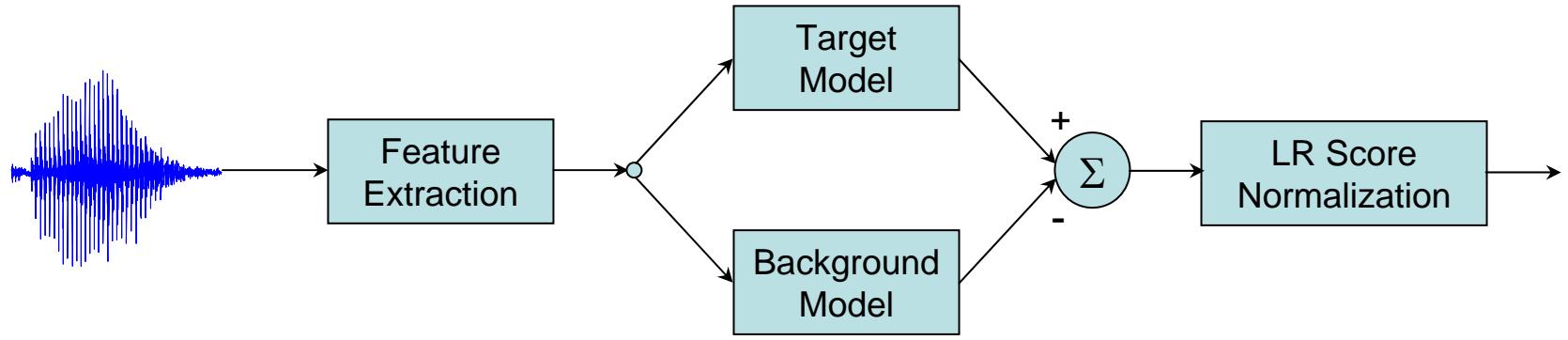
Background

- UNB R&D efforts began in October 2004
- Objectives: robust speaker identification and verification, with emphasis on short utterance length
- Team decided to participate in NIST SRE05
 - technical challenges are very close
 - motivated to become part of research community

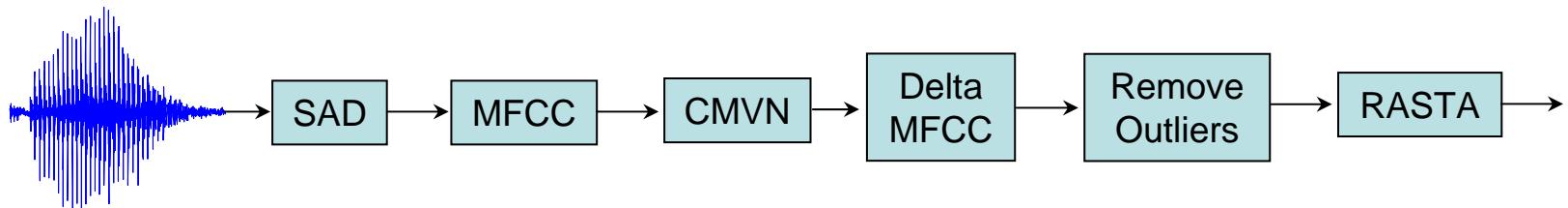
Timeline

- Core team assembled in January 2005
- MFCC-GMM system prototyped in February 2005
- Development data
 - Jan-March 2005: NIST SRE 1998 (landline only)
 - April 2005: NIST SRE 2004 (landline & cellular)

Scoring: Log-Likelihood Ratio



Front-End Processing



- Speech activity detection
 - Energy and frequency (zero crossings)
- Feature extraction
 - 20 ms window, 10 ms increment
 - 19 cepstral coefficients extracted from 300-3300Hz band
 - 19 delta cepstral coefficients
 - MFCC outliers removed (bottom, top 10%)
- Channel compensation
 - RASTA filtering
 - Cepstral mean and variance normalization (speech frames only)
 - Attempted Feature Mapping*

* Reynolds, D.A., "Channel robust speaker verification via feature mapping," ICASSP 2003

Target and Background Models

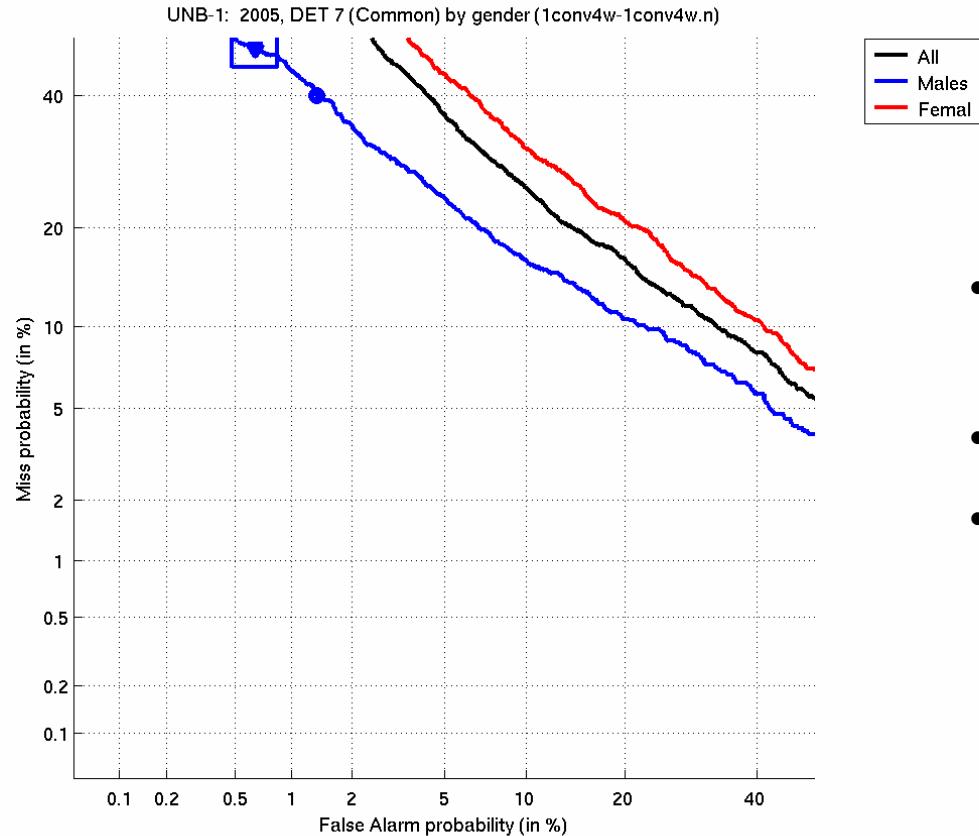
- Background models are 64 mixture GMMs
 - Trained using NIST 2004 data
 - Approximately 4 hours of data
 - 248 speakers, one minute per speaker (to avoid speaker bias)
 - **Unbalanced** (8:1 landline vs. cellular)
 - Performance degraded with more UBM mixtures
- Target models derived from UBM using MAP adaptation
 - One iteration
 - Variance limiting
 - Only mean vectors adapted

Prosodic Features

- Pitch + energy (1 minute of data)
- GMM classifier (64 mixtures)
- Fusion using a 2-layer multilayer perceptron network with 12 hidden layer nodes
- The training performance measure was adapted to minimize the DCF

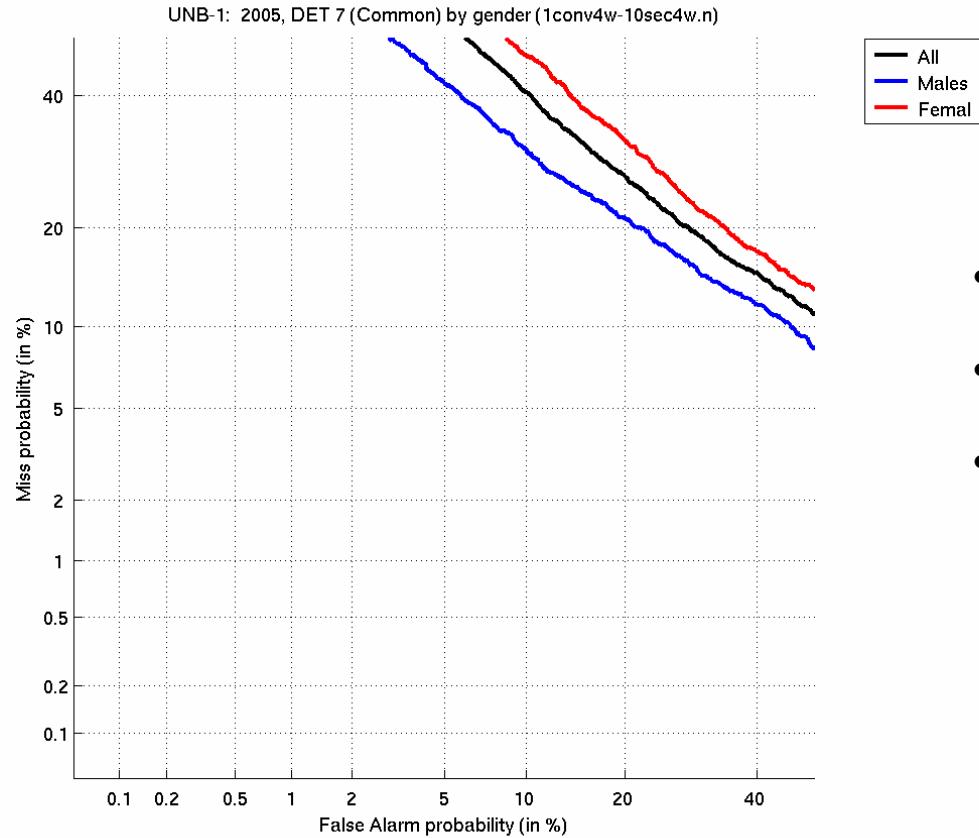
NIST SRE'05 Results

1conv4w-1conv4w



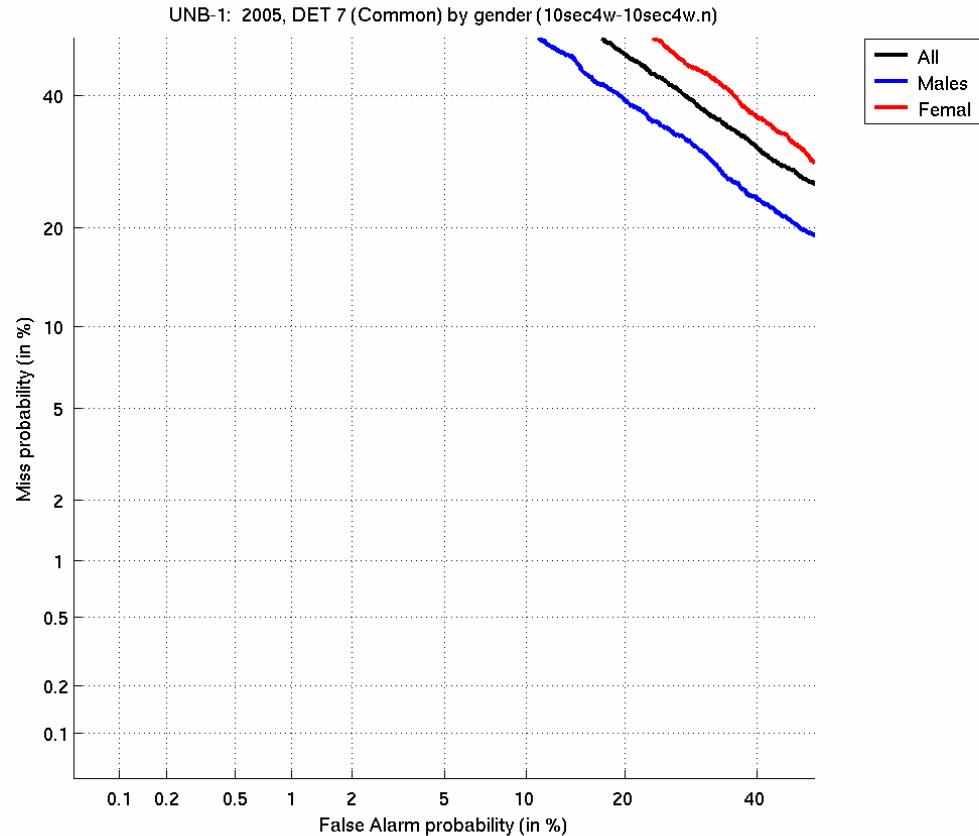
- 2 minutes for training and 1 minute for verification
- No score normalization
- No handset compensation

1conv4w-10sec4w



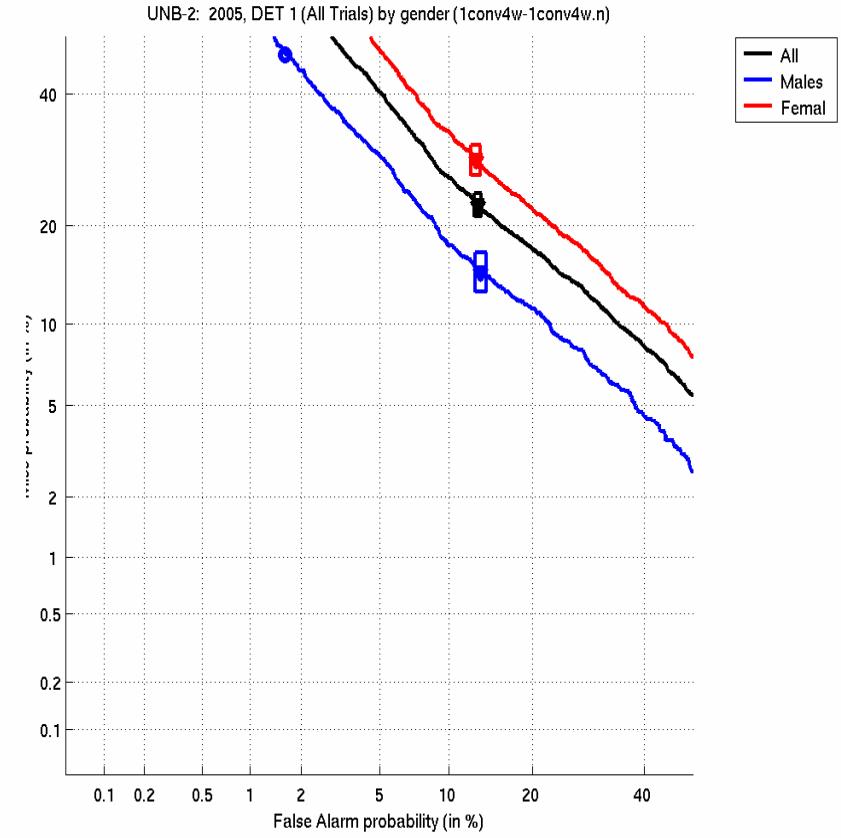
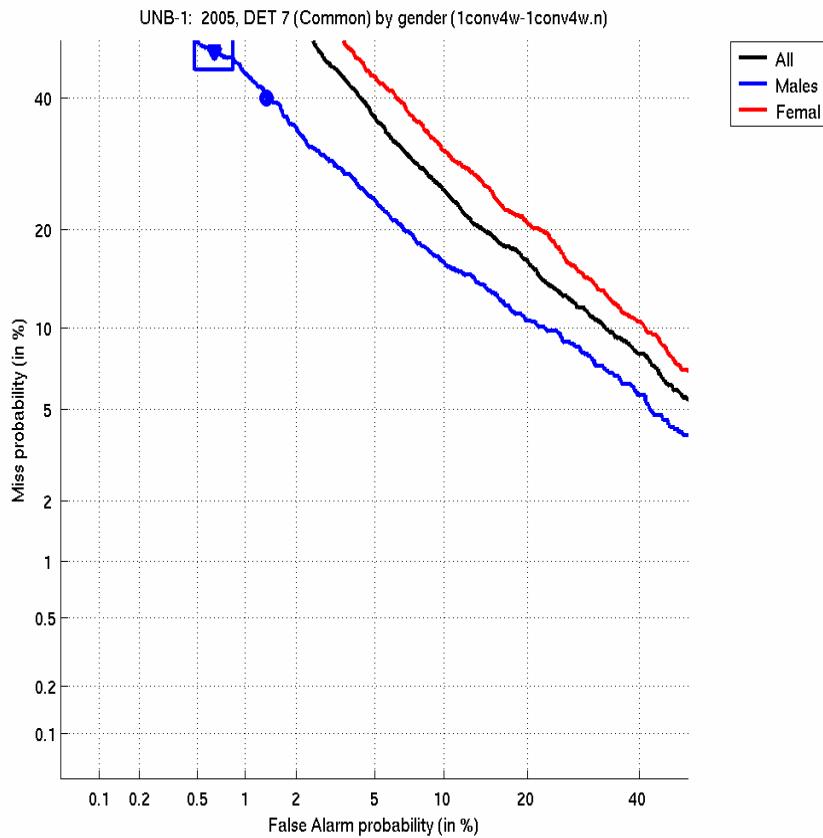
- 2 minutes for training
- No score normalization
- No handset compensation

10sec4w-10sec4w



- No score normalization
- No handset compensation

Fusion



Current Areas of Research

- Score normalization
 - T-norm with cohort selection
- Feature mapping
- Handset and phone type normalization
 - Cell (analog & digital)
 - Landline (regular & cordless)

Future Directions

- Exploit more development data
 - Balanced UBM creation
 - Handset dependent UBMs
- Speaker dependent threshold selection