



I3A NIST SRE2010 System Description

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Introduction

I3A has submitted several systems to NIST SRE2010:

⇒ core-core/coreext-coreext:

- ① Fusion of *JFA-SVM* and *JFA-LLR*, both Gender Dependent (GD)
- ② *JFA-SVM* Gender Dependent
- ③ *JFA-LLR* Gender Dependent

⇒ core-10sec/10sec-10sec:

- ① Fusion of *JFA-LLR* and *iVectors*, both Gender Dependent
- ② *JFA-LLR* Gender Dependent

⇒ core-summed

- ① *JFA-LLR* Gender Independent (GI)

Feature Extraction

⇒ 19 MFCC + C0 + Δ + $\Delta\Delta$

⇒ VAD comparing the Long-Term Spectral Divergence (LTSD) against a threshold.

- ❑ Crosstalk removal in phonecall using non target channel
- ❑ Interviewer removal using ASR labels

⇒ Short Time Gaussianization with a 3 seconds window.

Summed Channel Speaker Segmentation

⇒ First Viterbi segmentation using 20 speaker factors, calculated using 12 MFCC and 256 Gaussians, as features and modeling each speaker with a Gaussian.

⇒ Two Viterbi re-segmentations using 12 MFCC as features and modeling each speaker with a GMM.

⇒ Softclustering in the second re-segmentation.

⇒ Every output stream is short time Gaussianized separately.

Classification

⇒ 2048 GMM *UBM* trained on SRE04, SRE05, SRE06 telephone data.

⇒ *JFA* Hyperparameters:

- ❑ v (300 speaker factors), u_{phn} (100 channel factors) and d from tel data in SRE04, SRE05 and SRE06
- ❑ u_{mic} (100 channel factors) from mic data in SRE05, SRE06 and 50 speakers from SRE08.

⇒ *JFA-LLR*

- ❑ Speaker mean supervector given by the speaker factors (y,z) MAP estimates:

$$M_s = m_{UBM} + vy + dz \quad (1)$$

- ❑ Scoring is performed using first order Taylor approximation of the LLR:

$$LLR \approx (vy_{trn} + dz_{trn})^t \Sigma^{-1} (F_{tst} - N_{tst} u_{x_{tst}}) \quad (2)$$

- ❑ GD ZTNorm using telephone segments from SRE04,SRE05 and SRE06 (2363 male and 3318 female)

⇒ *JFA-SVM*

- ❑ Enrollment and scoring is done by an SVM with the following kernel:

$$k(x_1, x_2) = (vy_1 + dz_1)^t \Sigma^{-1} (F_2 - N_2 u_{x_2}) + (vy_2 + dz_2)^t \Sigma^{-1} (F_1 - N_1 u_{x_1}) \quad (3)$$

- ❑ GD SVM Background and ZTNorm using tel and mic segments from SRE04, SRE05, SRE06 and 50 speakers from SRE08 (4241 male and 5400 female)

⇒ *iVectors*

- ❑ T (400 total variability factors), LDA (200 dimensions) and WCCN trained from telephone data in SRE04,SRE05 and SRE06.
- ❑ Scoring is done by Cosine distance between enrollment and testing iVectors
- ❑ GD ZTNorm using telephone segments from SRE04,SRE05 and SRE06 (2363 male and 3318 female)

Calibration and Fusion

Core-core

⇒ Calibration

- ❑ Dev trial list including most of the common conditions included in SRE2010.
- ❑ All training vs. all testing short, long and follow-up SRE08 English data keeping out the 50 speakers used in JFA training (4M male trials and 10M female trials).
- ❑ Calibration has been done by linear logistic regression using FoCal package.
- ❑ We do multiple condition calibration iteratively until convergence:
 - ① Gender dependent (male, female)
 - ② Channel dependent (mic-mic same channel, mic-mic different channel, mic-phn, phn-phn)
 - ③ Length dependent (short-short, long-long, long-short, short-long)

⇒ Same/Different Microphone detection

- ❑ iVectors like system using mic channel factors as features.
- ❑ We train LDA (12 dimensions) and WCCN using SRE08 mic data from the 50 keep out speakers.
- ❑ The scores are normalized using SNorm and calibrated to provide a soft decision probability.

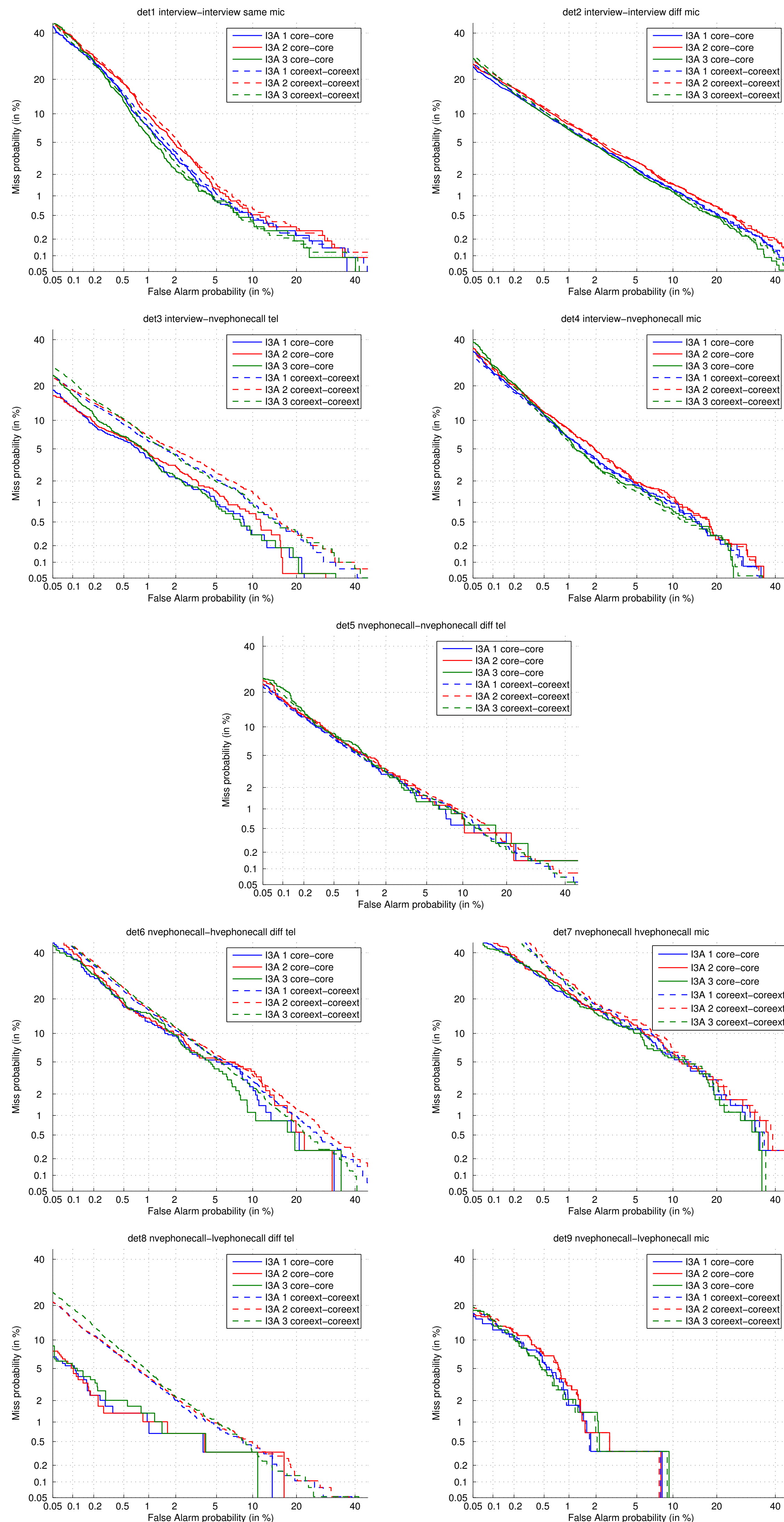
⇒ Fusion: The calibrated systems are channel dependent fused.

Non core-core

⇒ Gender dependent calibration and fusion using the det7 matching conditions of SRE08.

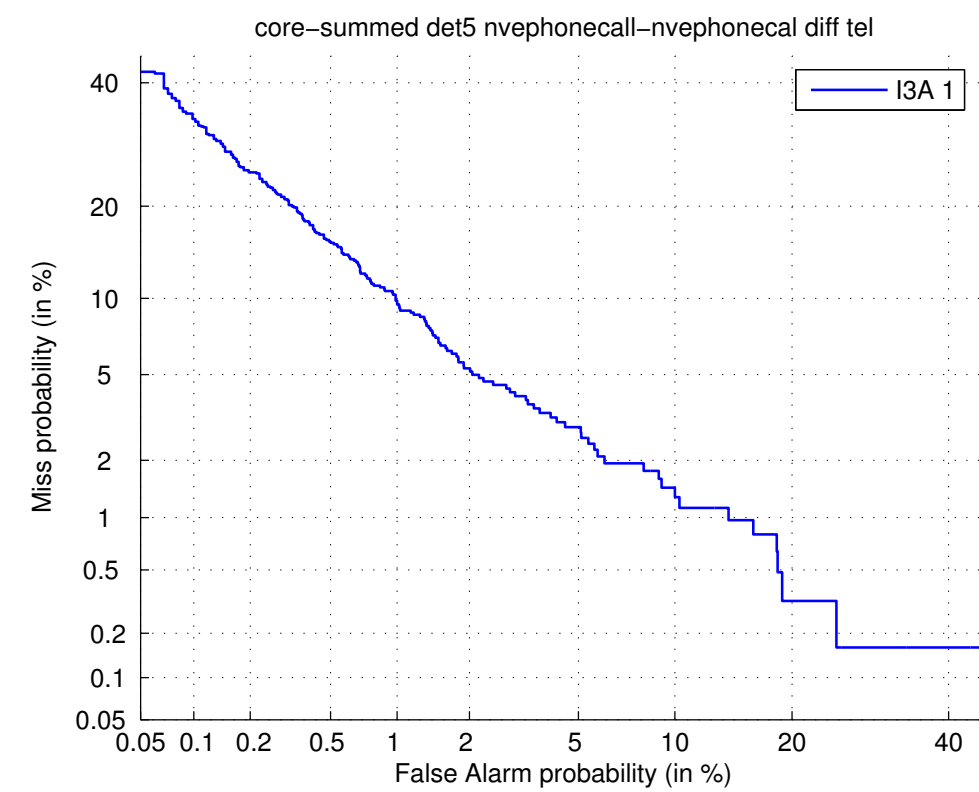
Results

core-core/coreext-coreext



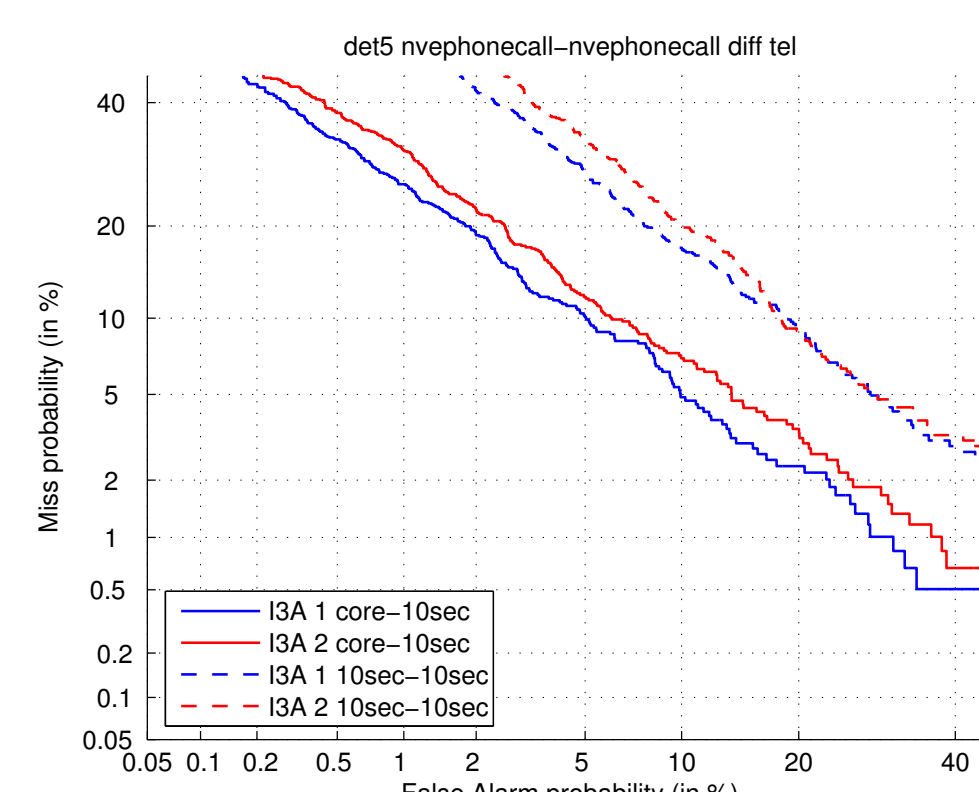
actCost/minCost	det1	det2	det3	det4	det5	det6	det7	det8	det9
ISA 1 core-core	2.21/ 0.70	0.49/0.48	0.61/0.41	1.21/0.62	0.55/0.40	0.84/0.77	0.99/ 0.69	0.19/0.17	1.06/0.40
ISA 2 core-core	3.12/0.72	0.51/0.50	0.97/ 0.35	1.79/ 0.61	0.81/0.38	0.85/0.80	1.52/0.70	0.19/0.18	1.24/0.39
ISA 3 core-core	1.18/0.75	0.58/0.57	0.58/0.49	0.82/0.67	0.54/0.50	0.81/0.73	0.81/0.75	0.31/0.22	0.73/0.44
ISA 2 core-core recal	1.22/0.73	0.54/0.50	0.50/0.40	0.99/0.64	0.73/ 0.35	0.87/0.81	0.75/0.71	0.24/ 0.17	0.60/0.38
ISA 3 core-core recal	0.74/0.71	0.69/0.58	0.59/0.58	0.70/0.63	0.53/0.49	0.81/0.72	0.75/0.70	0.33/0.24	0.43/0.39
ISA 1 coreext-coreext	2.06/ 0.69	0.51/0.51	0.66/0.48	1.16/ 0.60	0.54/0.48	0.93/0.90	2.87/ 0.99	0.48/0.46	1.22/ 0.40
ISA 2 coreext-coreext	3.22/0.72	0.53/ 0.51	0.97/ 0.45	1.75/0.61	0.70/0.49	0.97/0.91	3.73/0.99	0.47/0.45	1.43/0.43
ISA 3 coreext-coreext	1.08/0.73	0.58/0.57	0.63/0.56	0.74/0.66	0.56/0.52	0.94/0.91	1.88/1.00	0.60/0.53	0.82/0.46

core-summed



actCost/minCost	det5
ISA 1	0.20/0.19

core-10sec/10sec-10sec



actCost/minCost	det5 core-10sec	det5 10sec-10sec
ISA 1	0.36/0.35	0.64/0.61
ISA 2	0.40/0.39	0.69/0.66

Conclusions

- ⇒ We have built state of the art systems with good performance in most of the conditions
- ⇒ Especially good results for summed channel and 10sec conditions.
- ⇒ SVM system has better minDCF than LLR but it is more difficult to get a robust calibration.
- ⇒ Need of big number of non target trials without key errors for robust calibration in the new operating point.
 - ❑ Improvement in actCost recalibrating after correcting SRE08 key errors.
- ⇒ Score shift between same/diff mic conditions difficulties calibration ⇒ Need channel detector.
- ⇒ High vocal effort speech degrades performance considerably.

Acknowledgments

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