

iVOCALISE: An i-vector-based automatic speaker recognition system using spectral and phonetic features

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In this article we present a novel framework based on ‘i-vectors’ for extracting phonetic or spectral features of the voice and representing them as individual points in a compact lower dimensional space. This represents a generational change from the last version of VOCALISE, an automatic speaker recognition system, originally based on Gaussian mixture modelling (GMM) and Mel-frequency cepstral coefficients. Within the GMM-based system we have previously developed capabilities such as modeling long-term distributions of automatically extracted phonetic features, including formants and user-provided features, as well as selective processing of annotated regions.

In the latest version of VOCALISE (called iVOCALISE) we have incorporated the dominant approach in high-performing algorithms, namely an i-vector PLDA (Probabilistic Linear Discriminant Analysis) framework (described in Dehak et al, 2011), and have applied this to both phonetic and spectral features. The i-vector-based approach represents a significant performance improvement over GMMs, particularly when there is significant mismatch in the recording conditions of the samples under comparison. The conversion from speech sample to i-vector attempts to preserve speaker-specific information and discard information not related to the identity of the speaker. The i-vector approach has been extensively explored in literature, particularly with regard to spectral features such as MFCCs. Phonetic features such as formants are not commonly analysed within this framework however. We have sought to provide exactly the same i-vector-based framework to phonetic features.

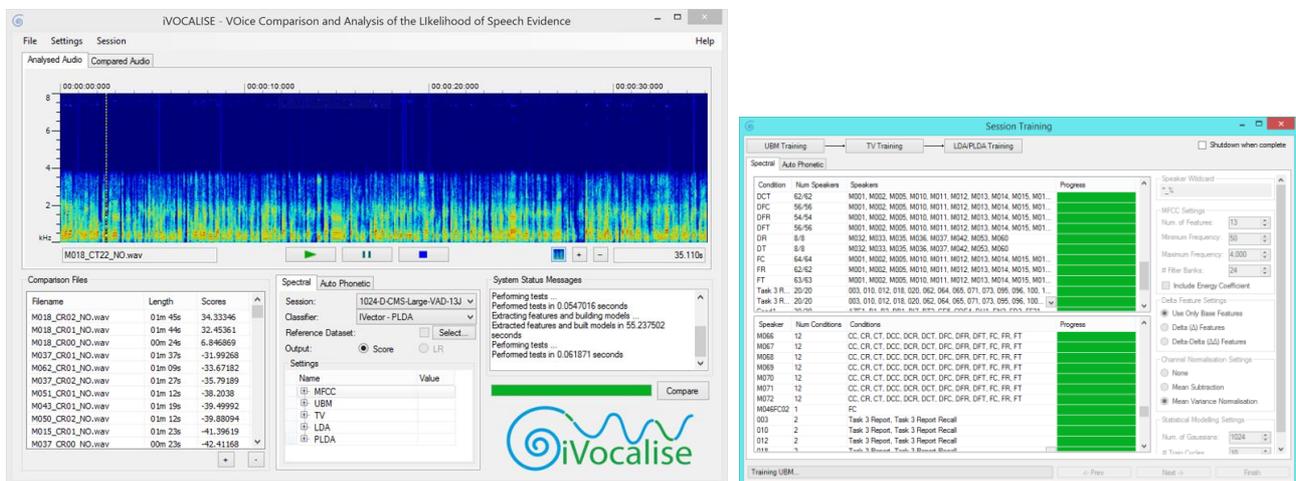


Figure 1. The iVOCALISE spectral comparison user interface (left), with user-provided data being used to train the universal background model, total variability matrix, and linear discriminant analysis (right).

This system has been developed with an ‘open-box’ architecture in which the user may adjust most of the feature and modelling parameters, and introduce new data at every step of the speaker recognition process. The user is, by design, not limited to manufacturer-provided models or configurations. They can either train the i-vector system and corresponding background models from scratch, or adapt the provided models with their own data.

The iVOCALISE system has been used with real forensic data, including the NFI-FRITS corpus that contains real telephone intercept data (van der Vloed et al, 2014), and the German real-case corpus from the BKA (Solewicz et al, 2012). While it is possible to train and adapt the system to context-specific data, simply using a pre-trained session file (containing UBM, TV matrix and PLDA information) that was trained with spectral MFCC data, (with delta coefficients, 1024 Gaussians, and a 400 element i-vector), iVOCALISE has obtained promising results when compared to those previously reported. In the German real-case data, the iVOCALISE system obtained an equal error rate (EER) of 6.89% using a pre-trained session file. This result is better than the highest-performing system with this dataset as described in Solewicz et al, 2012. iVOCALISE demonstrates a successful application of the i-vector framework to both phonetic and spectral features.

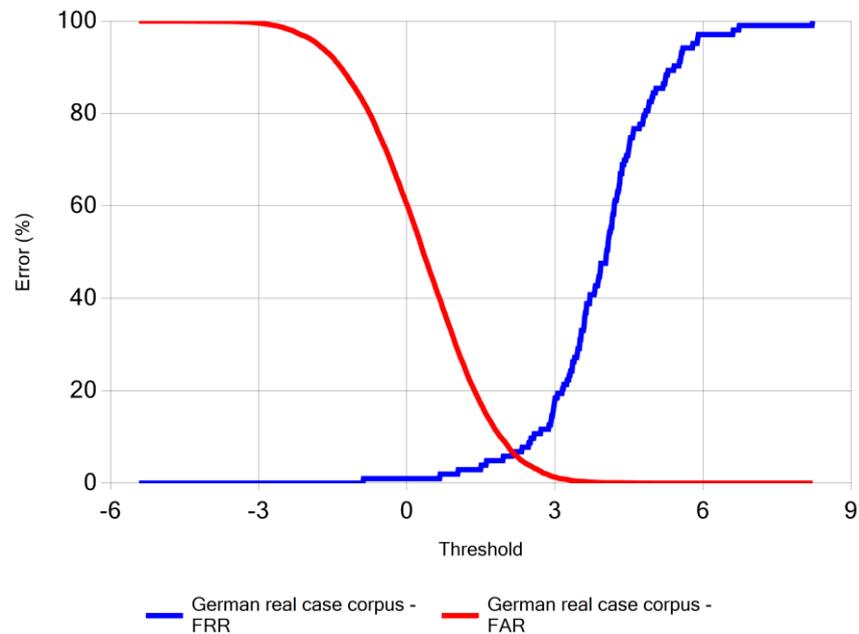


Figure 2. An example equal error graph with raw i-vector comparison scores obtained using the German real case dataset

References

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