

A Tutorial on Text-Independent Speaker Verification

F. Bimbot et al., 2004

Overview

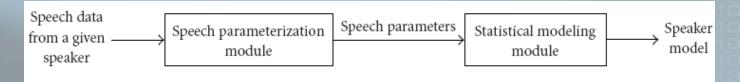
- Introduction
- Methods for Parameterization and Modeling
- Normalization of Scores
- Evaluation
- Extensions
- Applications

Introduction

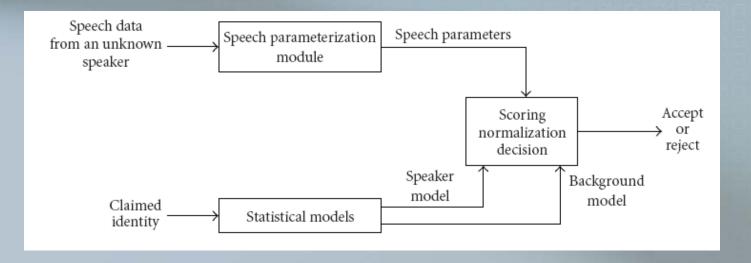
- Speaker Verification?
 - * Is this person who he/she claims to be? An example of biometrics
 - * Natural source of data: considered to be less intrusive than other methods
- Text-Independence?
 - * Most applications based on digit recognition or fixed vocabulary
 - * Text-independence implies operation independent of user cooperation and spoken utterance

Introduction

Training Phase

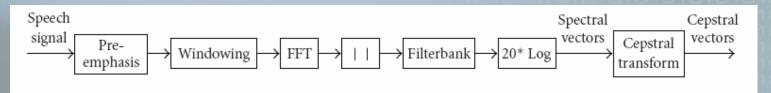


Test Phase

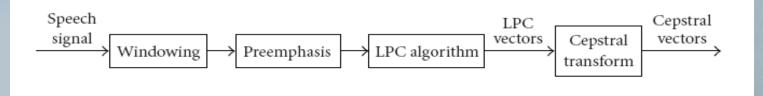


Parameterization

Filterbank Cepstral Parameters



LPC Cepstral Parameters



- Dynamic Information and log-Energy
- Discard Useless Information

Modeling

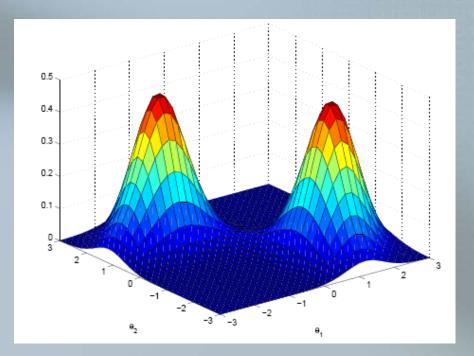
Likelihood Ratio Detection Given segment of speech Y and speaker S, determine if Y was spoken by S. For this, we need two hypotheses; Y was spoken by S and Y was not spoken by S. To implement this, we train two models; if X represents parameterization of Y, p(X|alpha) = speaker model p(X|alpha_) = non-speaker model

The ratio is the likelihood; if greater than threshold, accept, else reject

- Options for Non-speaker Model
 - * Train multiple models
 - * Train single model

Gaussian Mixture Models

- GMMs used to represent likelihood function p(X|alpha)
- Basically a weighted combination of M unimodal Gaussian densities of dimension D
- An example where D = 3 and M = 2



Interpretation: each unimodal component represents a broad acoustic class

Gaussian Mixture Models

- Given a set of training vectors of dimension D, we select M and then train the model using the EM algorithm
- $\mathbf{M} = 512$ on constrained, and 2048 on unconstrained speech
- The GMM is both parametric (has structure and parameters that can be tuned) and non-parametric (arbitrary density modeling)
- Advantages: computationally inexpensive, well-understood and insensitive to temporal aspects of speech
- The latter is actually a disadvantage in some cases; throws away information

Adapted GMM System

- Different methods of GMM training; one approach is to train background model first (using large M) and then train the speaker mode independently; often performs poorly
- Adaptation approach; train single GMM for background, and using training vectors for the speaker, create a new model for the speaker from the background GMM
- Method:
 - Step 1) compute statistics from the new data such as weights, mean and variance
 - Step 2) combine new information with the old s.t. mixtures with high counts of data from the speaker rely more on the new statistics and vice versa

Why Adaptation?

- Results indicate better performance
- The background models an acoustic space; tuning the existing one for speaker model leads to less surprises; likelihood ratio unaffected by "unseen" acoustic events
- Fast-Scoring:
 Step 1) For each feature vector, determine C top-scoring mixtures in background model; compute likelihood using only these
 Step 2) Score the vector against the top C mixtures in the speaker model
- Alternative Methods:
 - * ANNs
 - * SVMs

Normalization

- The problem: once the likelihood is calculated, compare with a threshold to make decision. How do we calculate the threshold?
- Score variability a major issue; speaker may be tired, in poor health etc, or there might be environmental issues (like background noise). Composition of training set for background also affects scores.
- Normalization of score variability makes decision threshold tuning easier: $\tilde{L_{\lambda}}(X) = \frac{L_{\lambda} \mu_{\lambda}}{\sigma_{\lambda}}$

Normalization Methods

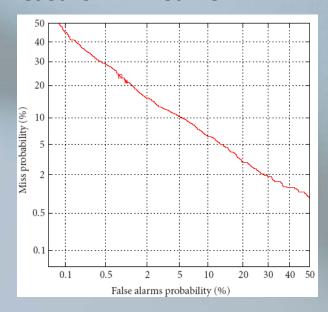
World-model and cohort-based normalizations

$$\tilde{L_{\lambda}}(X) = \frac{L_{\lambda}(X)}{L_{\bar{\lambda}}(X)}$$

- Centered/Reduced Imposter Distribution
 Most commonly used (derived from equation on previous slide)
- The Norms
 Znorm, Hnorm, Tnorm, Htnorm, Cnorm, Dnorm
- WMAP
 World-model Maximum A Posteriori normalization. Produces a meaningful score in probability space

Evaluation

- 2 forms of errors; false negatives and false positives. Depends on application as to which is more serious.
- Performance measure: DET Curve



 Factors affecting performance: environmental issues, speaker "performance", "goats and lambs", training set size and diversity

Extensions

- Multiple Speaker Detection
- Speaker Tracking
- Segmentation

Applications: General

- On-Site
 In a given facility, voice recognition required for access to certain features or places
- Remote Applications
 Secure access to remote databases or services
- Information Structuring
 Automatic annotation of audio archives, speaker indexing, speaker change in subtitles etc.
- Games
 Personalized toys (seemingly humanity's most pressing need)

Applications: Forensic

- Forensics
 Refers to criminal investigation, i.e. voice identification of a suspect.
- Difficulties
 Situation for recognition more difficult; more noise, variability etc.
- Controversy A "voice print" is not the same thing as a finger-print; not physiological because of the psychological factors involved. Because of possible errors, the concept of nonzero errors creates difficulties in judicial process.
- Systems
 Semiautomatic systems require expert input; "supervised selection of acoustic phonetic events". Automatic systems exist, and are based on the preceding discussion.

Future Work

- Robustness Issues
 Channel variability and mismatched conditions, especially in microphones, play havoc with acoustic feature extraction. These need to be addressed, especially in a real-world, and not a laboratory setting.
- Exploitation of Higher Levels of Information
 Word usage, prosodic (manner of speech) measures etc.
- Emphasis on Unconstrained Tasks
 No prior assumptions on the state of the environment, for a given value of "no."

References

- Bimbot et al. "A Tutorial on Text-Independent Speaker Verification."
- Frank Dellart, "The Expectation Maximization Algorithm" (for GMM picture.)