

# Speaker Recognition

Abraham W. Zewoudie, Post-Doctoral Researcher

Department of Signal Processing and Acoustics

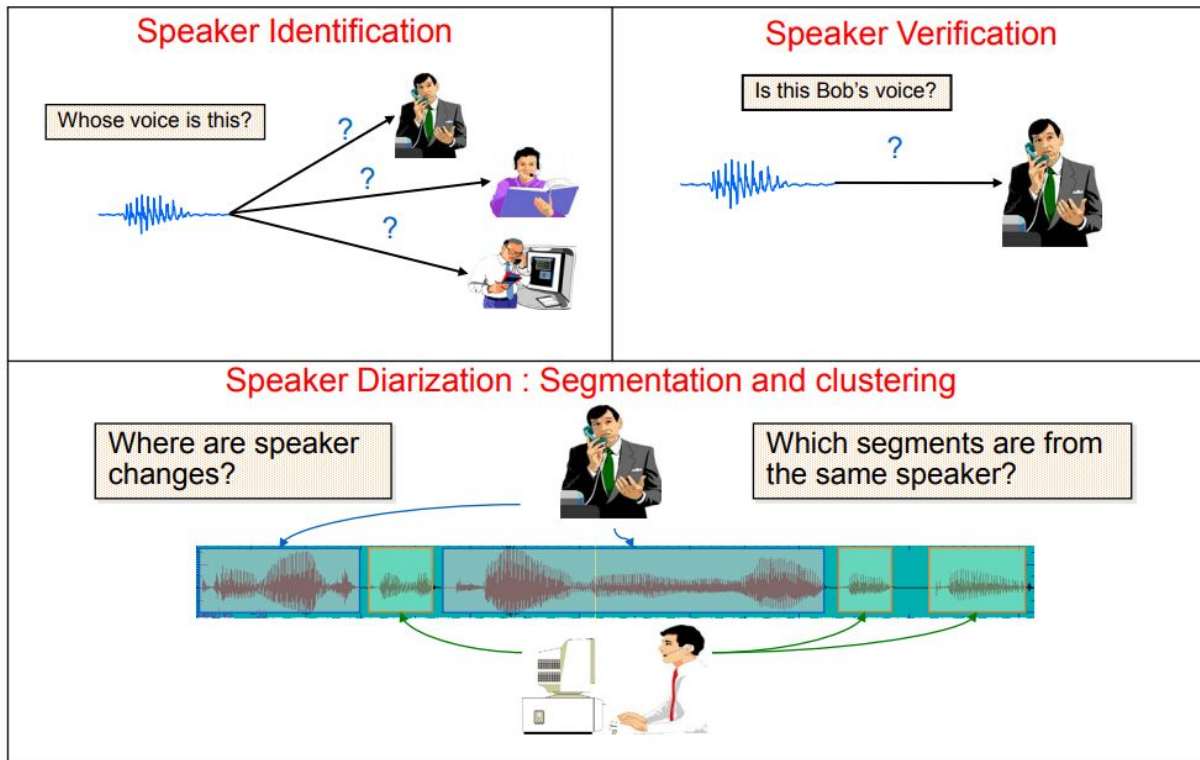
Aalto University

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## Outline

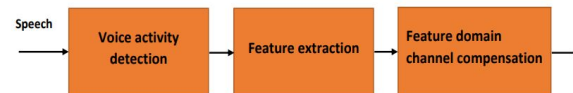
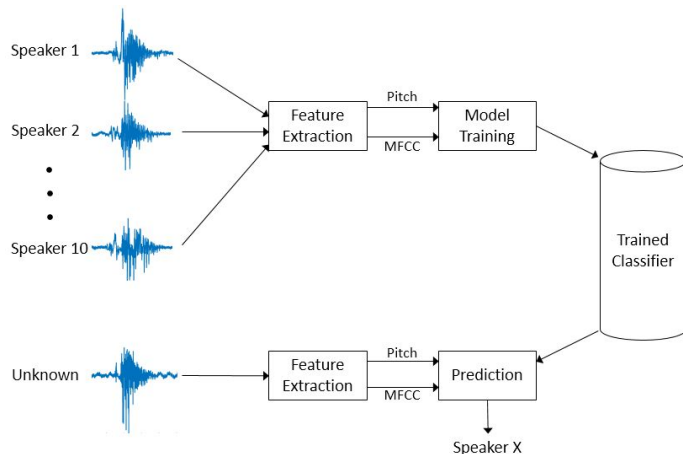
- Overview of Speaker Recognition
- State-of-the-art in Speaker Recognition
  - GMM-UBM
  - i-Vector
  - DNN
- Application Areas
- Performance Evaluations

# Speaker Recognition



# Steps of Speaker Recognition

1. **Feature Extraction**: Used features include **MFCC**, **Spectrogram**.....
2. **Speaker Modeling** : The extracted features are used to generate models corresponding to each speaker and stored for comparison during testing. Speaker modeling techniques: **GMM**, **i-Vector** and **Deep Neural Network (DNN)**.
3. **Classification**: Relative scores are computed for each of the speaker models and the one with the **highest score** is identified to be the test speaker. Scoring methods include **Log Likelihood Ratio**, **Cosine Distance Scoring** and **Probabilistic Linear Discriminant Analysis (PLDA)**.



# Speech Modalities

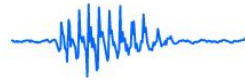
- **Text-dependent Speaker Recognition**
  - Recognition system knows text spoken by person
  - Examples: fixed phrase, prompted phrase
  - Used for applications with strong control over user input
  
- **Text-independent Speaker Recognition**
  - Recognition system does not know text spoken by person
  - Examples: User selected phrase, conversational speech
  - Used for applications with less control over user input
  - More flexible system but also more difficult problem.

# Two Phases of Speaker Verification System

## Enrollment Phase



Enrollment speech for each speaker



Bob

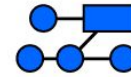


Sally

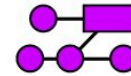
Feature extraction

Model training

Voiceprints (models) for each speaker

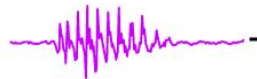


Bob



Sally

## Verification Phase



Feature extraction

Verification decision

Accepted!

Claimed identity: Sally

# Speaker Recognition Challenges

- Speaker verification performance is often degraded in the presence of **channel/session variability** between enrolment and verification speech signals. Various factors affect channel/session variability:
  - **Channel mismatch** between **enrolment** and **verification speech signals**.
  - Environmental noise and reverberation conditions.
  - The differences in speaker voice (e.g. ageing, health, speaking style and emotional state)
  - Transmission channel (e.g. landline, mobile phone, microphone and voice over Internet protocol (VoIP)).
- Various channel compensation techniques:
  - cepstral mean subtraction (CMS)
  - feature warping
  - cepstral mean variance normalization.

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# GMM-UBM Approach

- A GMM is a weighted sum of M Gaussian densities as given by:

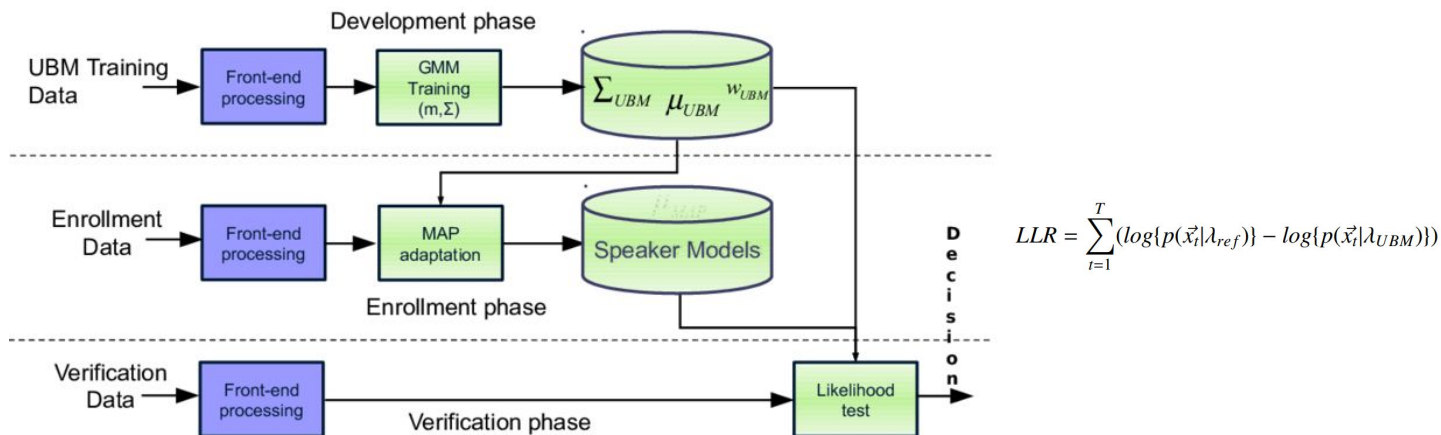
$$p(\mathbf{x}|\lambda) = \sum_{i=1}^M w_i g(\mathbf{x}|\boldsymbol{\mu}_i, \boldsymbol{\Sigma}_i)$$

where  $x$  is a D-dimensional feature vector,  $i$  is the index of the  $i^{\text{th}}$  Gaussian mixture,  $g(x|\boldsymbol{\mu}_i, \boldsymbol{\Sigma}_i)$  are Gaussian mixtures.

- As the amount of the enrollment data for each speaker is usually few, it is not so efficient to train a GMM for each speaker from scratch.
- Therefore, a global GMM, which is referred to as Universal Background Model (UBM), is first trained using a large number of utterances, and then the UBM is adapted to each speaker.
- The adaptation is typically performed using the Maximum a Posteriori (MAP) estimation which includes two steps.
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  - Sufficient statistics, which are known as Baum-Welch statistics, are calculated given the new feature vectors.
  - The adapted parameters are obtained by the combination of the new statistics for a given speaker and the UBM parameters

## Steps of GMM-UBM Approach

- A **Universal Background Model (UBM)** is first generated using speech samples from all the different speakers.
- **MAP (Maximum A Posteriori)** estimation is used to obtain models for each of the individual speakers.
- For testing, the feature vectors are extracted from test signal and are compared against all the speaker models in the database
- The model with the highest log likelihood ratio (LLR) is chosen to be the test speaker.



# i-Vectors

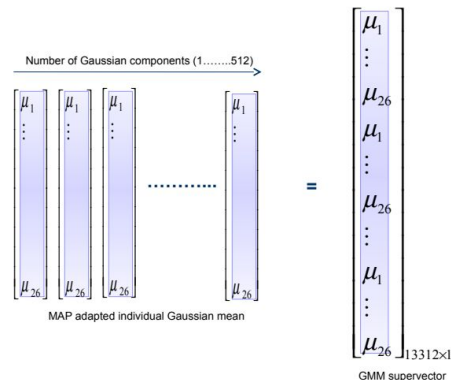
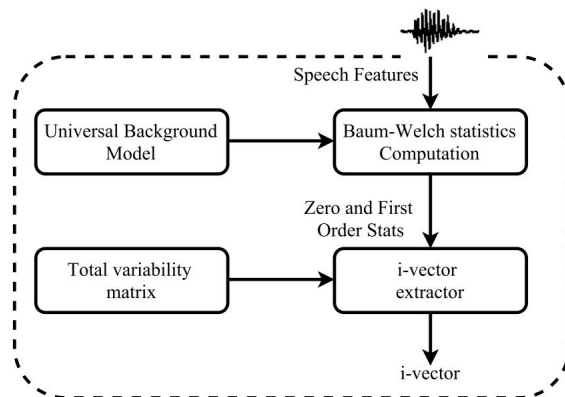
- A significant contributor to the performance degradation of traditional GMM-UBM speaker verification is the presence of **session variability** between the **training** and **testing** conditions.
- Different approaches have been developed recently to improve the performance of speaker recognition systems. The most popular ones were based on **GMM-UBM**.
- In i-Vector approach, a given speech recording is represented by a new vector, called total factors as it contains the **speaker** and **channel variabilities** simultaneously.
- Speaker recognition based on the *i-vector/x-vector framework* is currently the state-of-the-art in the field.
- Given an utterance, the speaker and channel dependent GMM supervector,  $M$ , is defined as follows:

$$M = m + T w$$

$m$  is the speaker and channel independent background UBM super-vector

$T$  is the total variability matrix

$w$  is the extracted i-vector



# i-Vector Scoring

- The i-vector based speaker recognition is implemented using two types of classifiers - [Cosine distance](#) and [Probabilistic Linear Discriminant Analysis \(PLDA\)](#).
- Both [Cosine distance](#) and [PLDA](#) make use two different ways for computing the likelihood scores and in either case, the speaker model with the [highest score](#) is identified to be the speaker.
- [Cosine distance](#) directly compares two inputs and gives out the degree of similarity between them.
- Given two i-vectors  $w_1$  and  $w_2$ , [PLDA](#) computes the likelihood ratio of the two i-vectors as follows:

$$Score(w_1, w_2) = \frac{p(w_1, w_2|H_1)}{p(w_1|H_2)p(w_2|H_2)}$$

where  $H_1$  indicates that both i-vectors belong to the same speaker and  $H_2$  indicates they belong to two different speakers.

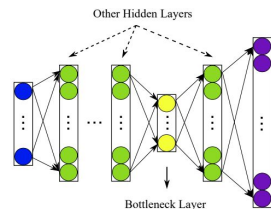
- In PLDA, assuming that the training data consists of J i-vectors where each of these i-vectors belong to speaker I, the j'th i-vector of the I'th speaker is denoted by:

$$w_{ij} = \mu + Fh_i + Gy_{ij} + \Sigma_{ij}$$

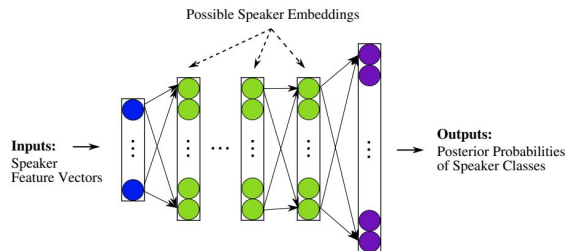
$\mu$  is the overall speaker and segment independent mean of the i-vectors in the training dataset  
F is the eigen voice matrix (speaker variability)  
G is the eigen channel matrix (within-speaker).  
 $\Sigma_{ij}$  represents any unexplained data variation.  
 $h_i$  are the speaker factors and  $y_{ij}$  are channel factors.

# DNN

- The traditional i-vector approach consists in three main stages: *Baum-Welch statistics computation*, *i-vector extraction*, and *PLDA backend*.
  - Using i-Vector as input to the network
  - Using bottleneck features as input to the network.



- Using *speaker embeddings* (i.e., the speaker characteristics of a speech signal with a single low dimensional vector).



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# Application Areas

## 1. Speaker Recognition for Authentication

- It allows users to identify person using their voices.
- *Voice sample vs PIN/credit card (lost/stolen) vs PIN or password (forgotten)*

## 2. Speaker Recognition for Surveillance

- *Security agencies have several means of collecting information.*
- *One of these is **electronic eavesdropping** of telephone and radio conversations.*
- *As these results in high quantities of data, filter mechanism must be applied to find the relevant information.*
- *One of these filters could be recognition of target speakers that are of interest for the service.*

## 3. Forensic Speaker Recognition

- *If there is a speech sample that was recorded **during a crime**, the suspect's voice can be compared with this to find the similarity of two voices.*

## 4. Security

- *It is the most obvious application of any biometric authentication applications.*
- *Examples: Access control, credit card transactions, banking access*

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# Performance evaluations

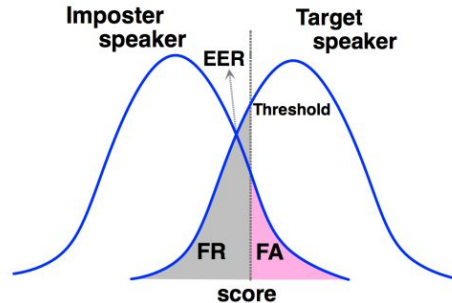
- **False Acceptance Rate (FAR):** It occurs when the speech segments from an impostor speaker are falsely accepted as a target speaker by the system.

$$FAR = \frac{\text{Total number of false acceptance errors}}{\text{Total number of imposter speaker attempts}}$$

- **False Rejection Rate (FRR):** A false rejection occurs when the target speaker is rejected by the verification systems.

$$FRR = \frac{\text{Total number of false rejection errors}}{\text{Total number of enrolled speaker attempts}}$$

```
id10270/x6uYqmc31kE/00001 id10270/8jEAjG6Segy/00006 1
id10270/x6uYqmc31kE/00001 id10380/ize_eiCFEg/00003 0
id10270/x6uYqmc31kE/00001 id10270/gW6xj]-xAW/00017 1
id10270/x6uYqmc31kE/00001 id10273/800CHkZyq/00001 0
id10270/x6uYqmc31kE/00001 id10270/8jEAjG6Segy/00022 1
id10270/x6uYqmc31kE/00001 id10284/0zcv7Am32z/00001 0
id10270/x6uYqmc31kE/00001 id10270/gW6xj]-xAW/00033 1
id10270/x6uYqmc31kE/00001 id10284/7y9d0yZLk/00029 0
id10270/x6uYqmc31kE/00002 id10270/5r0dWky17C8/00026 1
id10270/x6uYqmc31kE/00002 id10285/m-uLLTo09s/00009 0
id10270/x6uYqmc31kE/00002 id10270/gW6xj]-xAW/00035 1
id10270/x6uYqmc31kE/00002 id10386/uz136PBzTzW/00001 0
id10270/x6uYqmc31kE/00002 id10270/gW6xj]-xAW/00038 1
id10270/x6uYqmc31kE/00002 id10387/kp_GcJLq4A/00004 0
id10270/x6uYqmc31kE/00002 id10270/gW6xj]-xAW/00033 1
id10270/x6uYqmc31kE/00002 id10275/Mdk1SxywHck/00024 0
id10270/x6uYqmc31kE/00003 id10270/gW6xj]-xAW/00038 1
id10270/x6uYqmc31kE/00003 id10293/1vffth1tspLg/00004 0
id10270/x6uYqmc31kE/00003 id10270/5r0dWky17C8/00004 1
id10270/x6uYqmc31kE/00003 id10270/8e4fJE7PHp/00004 0
id10270/x6uYqmc31kE/00003 id10270/8jEAjG6Segy/00039 1
id10270/x6uYqmc31kE/00003 id10380/502wPHRqk/00012 0
id10270/x6uYqmc31kE/00003 id10270/5r0dWky17C8/00010 1
id10270/x6uYqmc31kE/00003 id10305/G50_1x7IVjU/00001 0
id10270/x6uYqmc31kE/00004 id10270/gW6xj]-xAW/00010 1
id10270/x6uYqmc31kE/00004 id10386/z5aEEN8yYz4/00011 0
id10270/x6uYqmc31kE/00004 id10270/gW6xj]-xAW/00045 1
```



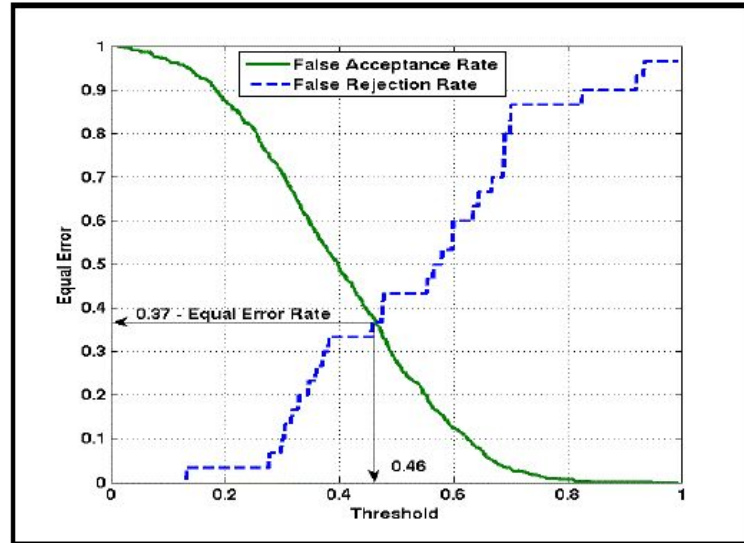
The main goal for speaker verification must be to minimize those errors.

The tradeoff between the errors depend on the application.

```
0.39320746
0.43270904
0.55977297
0.41281128
0.531512
0.35153052
0.5532894
0.43684848
0.46815294
0.39521956
0.5227014
0.45907044
0.45167223
0.45562238
0.47765756
0.44990146
0.3408629
0.46217704
0.5683596
0.46842796
0.39797342
0.5182556
0.5431458
0.5363163
0.5385692
0.49122933
0.44365984
0.38439894
```

## Performance evaluations

- The performance metrics of speaker verification systems can be measured using the **equal error rate (EER)**.
- The EER is obtained when the **false acceptance rate** and **false rejection rate** are equal.
- The performance of the system improves if the value of **EER is lower** because the sum of total error of the false acceptance and the false rejection at the point of EER decreases



# Implementation

## 1. Read training List

```
id10001/1zcIwhmdeo4/00001 id10001
id10001/1zcIwhmdeo4/00002 id10001
id10001/1zcIwhmdeo4/00003 id10001
id10001/7gWzIy6yIIk/00001 id10001
id10002/0_laIeN-Q44/00001 id10002
id10002/6w0410Q0euo/00001 id10002
id10002/6w0410Q0euo/00002 id10002
id10002/6w0410Q0euo/00003 id10002
.....
.....
.....
id11251/s4R4hvqrhFw/00006 id11251
id11251/s4R4hvqrhFw/00007 id11251
id11251/s4R4hvqrhFw/00008 id11251
id11251/s4R4hvqrhFw/00009 id11251
```

## 1. Encode target labels with value between 0 and find unique number of classes

[ 0 0 0 0 1 1 1 .....39 39 39 39]

Number of classes = 40



# Implementation

6. We are given a list of trials

```
id10270/x6uYqmx31kE/00001 id10270/8jEAjG6SegI/00008 1
id10270/x6uYqmx31kE/00001 id10300/1ze_4lFEF9p/00003 0
id10270/x6uYqmx31kE/00001 id10270/G00uj1->AVM/00037 1
id10270/x6uYqmx31kE/00001 id10273/00CK1Hh2yq/00001 0
id10270/x6uYqmx31kE/00001 id10270/8jEAjG6SegI/00022 1
id10270/x6uYqmx31kE/00001 id10284/Uzxy7Awh3Z8/00001 0
id10270/x6uYqmx31kE/00001 id10270/G00uj1->AVM/00033 1
id10270/x6uYqmx31kE/00001 id10284/7yx9A0yz1Yk/00029 0
id10270/x6uYqmx31kE/00002 id10270/Sr00kvy17C8/00026 1
id10270/x6uYqmx31kE/00002 id10385/rp-idLT0095s/00009 0
id10270/x6uYqmx31kE/00002 id10270/G00uj1->AVM/00035 1
id10270/x6uYqmx31kE/00002 id10306/uzt3zF9s-t2w/00001 0
id10270/x6uYqmx31kE/00002 id10270/G00uj1->AVM/00038 1
id10270/x6uYqmx31kE/00002 id10307/kp_GcJLq4qA/00004 0
id10270/x6uYqmx31kE/00002 id10270/G00uj1->AVM/00033 1
id10270/x6uYqmx31kE/00002 id10270/r0h15Vw+hKck/00024 0
id10270/x6uYqmx31kE/00003 id10270/G00uj1->AVM/00030 1
id10270/x6uYqmx31kE/00003 id10283/TyftHlTqplq/00004 0
id10270/x6uYqmx31kE/00003 id10270/Sr00kvy17C8/00004 1
id10270/x6uYqmx31kE/00003 id10273/BcFyJEV7hP8/00004 0
id10270/x6uYqmx31kE/00003 id10270/8jEAjG6SegI/00038 1
id10270/x6uYqmx31kE/00003 id10300/SQWYPrRqmk/00012 0
id10270/x6uYqmx31kE/00003 id10270/Sr00kvy17C8/00010 1
id10270/x6uYqmx31kE/00003 id10385/s50_1zT1VU/00001 0
id10270/x6uYqmx31kE/00004 id10270/G00uj1->AVM/00010 1
id10270/x6uYqmx31kE/00004 id10306/25aEB0hYz4/00011 0
id10270/x6uYqmx31kE/00004 id10270/G00uj1->AVM/00045 1
```

7. We predict using trained model

8. Compute EER using 7 and 8.

```
0.39320746
0.43270904
0.55977297
0.41281128
0.531512
0.35153052
0.5532894
0.43084848
0.46015294
0.39521956
0.5227014
0.45907044
0.45167223
0.45562238
0.47765756
0.44900146
0.3400629
0.46217704
0.5683596
0.46842796
0.39797342
0.5182556
0.5431458
0.5363163
0.5385692
0.49122933
0.44365984
0.38430894
```