

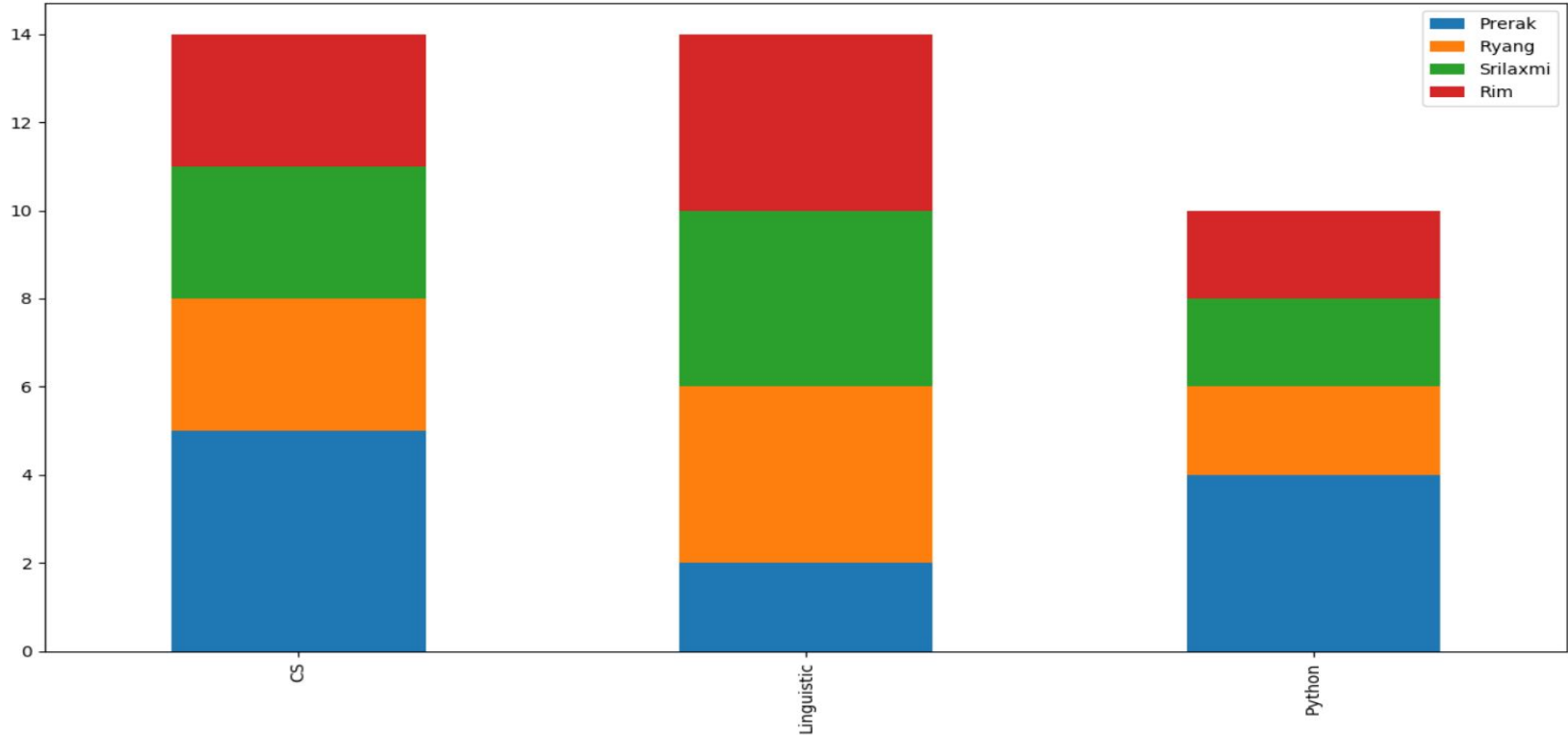
Speaker Recognition

Software project progress report

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Background of Group Members



Outline

- Abstract
 - Application
 - Important Terms
 - Baseline Models
 - Objectives
 - Our Approach
 - Timelines
 - Further Work
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Abstract

- *Popular technique*
- **Sub areas:**
 - Speaker identification
 - Speaker verification
- **Motivation**
- **Data:**
 - *Voxforge*
- **Purpose of our work:**
 - Train our model in French corpus for now
- **The method:**
 - Deep Neural Network Embeddings (i-vectors)
 - Transferring learning (kaldi tool)

Applications

- Secure access control by voice
 - Voice dialing, banking transaction over telephone, database access, multi-speaker tracking, etc.
- Verification of identity in e-commerce
- Make daily life more convenient

Important Terms

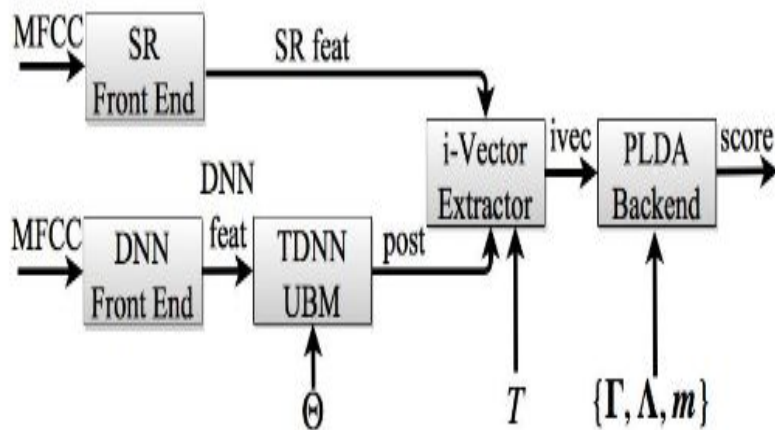
- Data Augmentation
- Variable-length Utterance
- Text Dependent and Text Independent
- In-Domain data and Out Of Domain Data
- Channel Variabilities
- Mel Frequency Cepstral Coefficient (MFCC)

Baseline (i-vector Model)

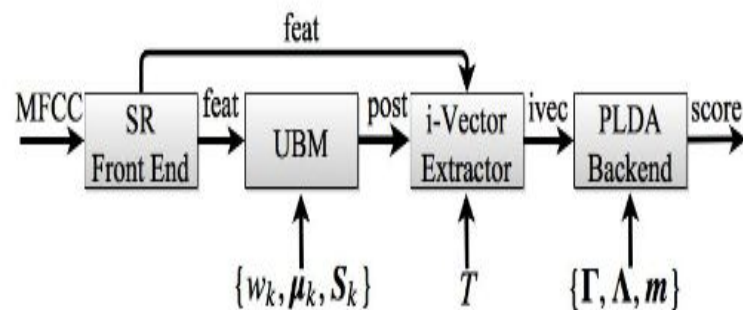
- It is based on Gaussian mixture model–universal background model (GMM-UBM)
- The I-vector is a compact representation that summarizes what is happening in a given speech recording
- it's a discrete spectrogram
- I-vector extraction is essentially a dimensionality reduction of the GMM supervector
- It factorizes the speech signal into the phonetic factor and the speaker factor

Comparison Between GMM-UBM v/s TDNN GMM

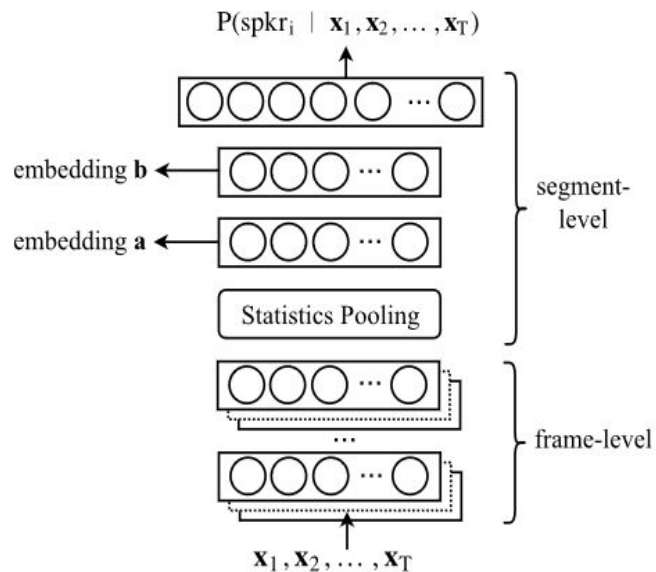
TDNN Based I-vector



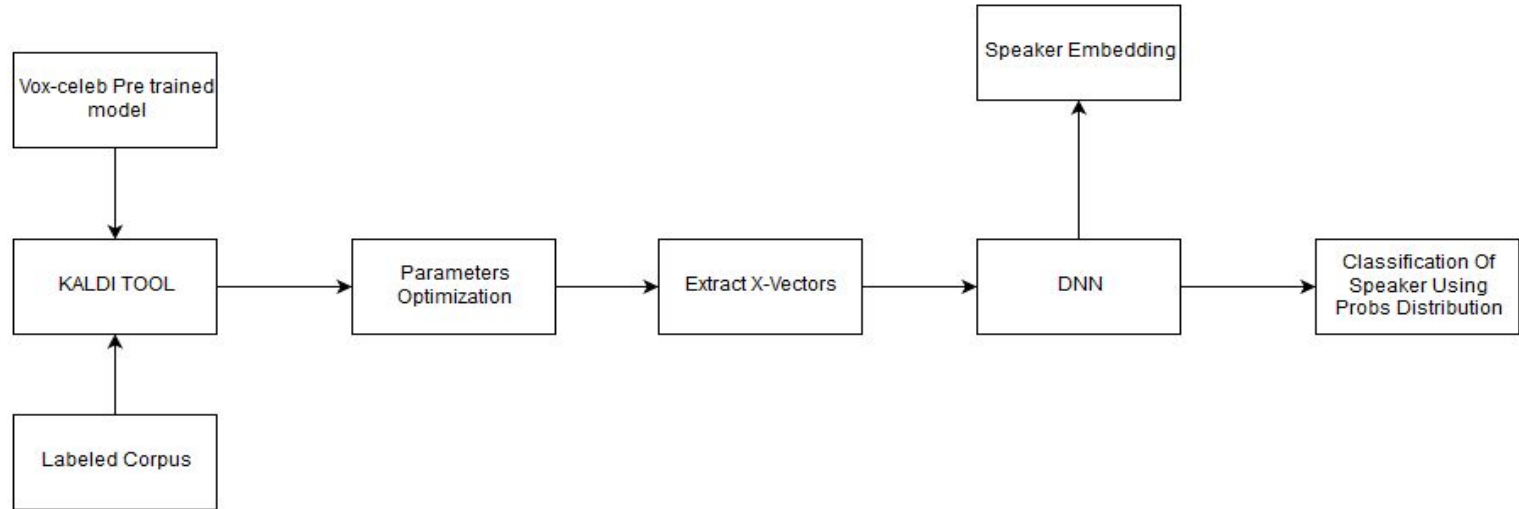
GMM-UBM



X-vectors



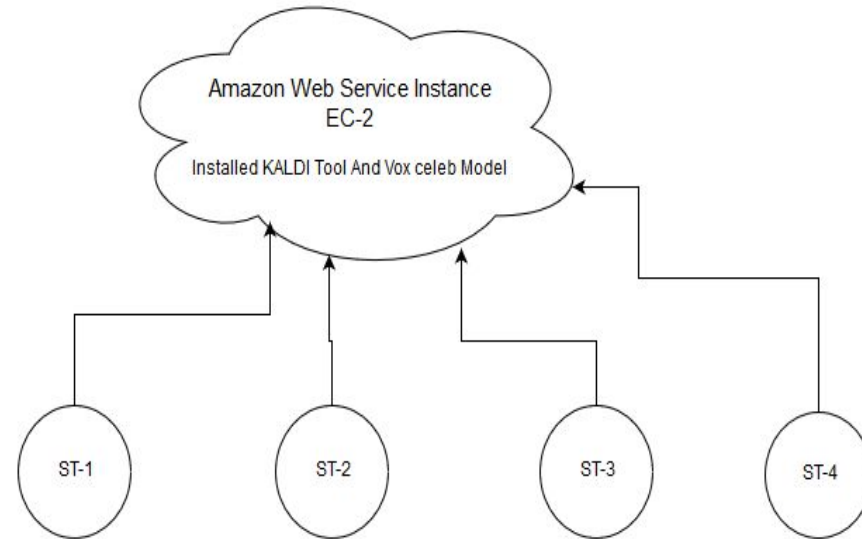
Our Approach



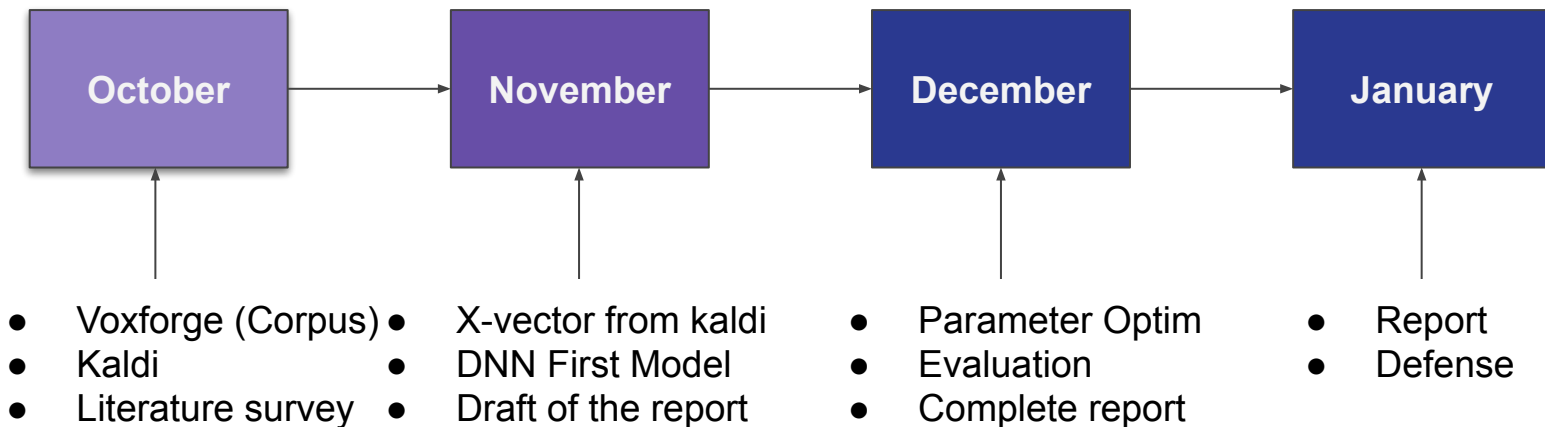
Evaluation

- **Precision:** how relevant are the positive detection
- **Recall:** probability of detection
- **Average accuracy**
- **F-score:** relation between data positive label and those given by classifier based on per-class average

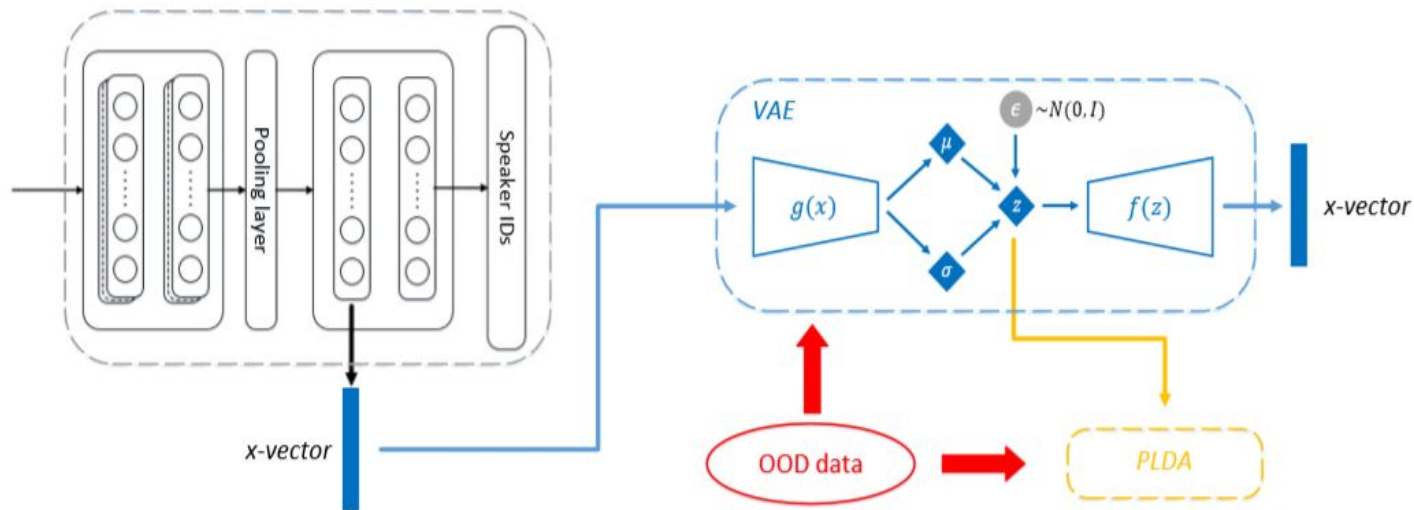
Kaldi Tool Robust Usage



Milestones



Future Work With Variational Autoencoder



GitLab Repo

<https://gitlab.com/prerakshrivastava/asr-sv>

The project structure of this repo as of now

- Data
- Models
- Utils
- Docs/Reports
- Kaldi
- Plots

Bibliography

- Snyder, David, Daniel Garcia-Romero, Gregory Sell, Daniel Povey and Sanjeev Khudanpur. “X-Vectors: Robust DNN Embeddings for Speaker Recognition.” *2018 IEEE International Conference on Acoustics, Speech and Signal Processing (ICASSP)* (2018): 5329-5333.
- Snyder, David, Daniel Garcia-Romero, Daniel Povey and Sanjeev Khudanpur. “Deep Neural Network Embeddings for Text-Independent Speaker Verification.” *INTERSPEECH* (2017).
- Wang, X., Li, L., & Wang, D. (2019). VAE-based Domain Adaptation for Speaker Verification. ArXiv:1908.10092 [Cs, Eess]. Retrieved from <http://arxiv.org/abs/1908.10092>
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Thank you!

Merci de votre attention!

Milestones

- **October:**
 - Bibliography
 - Kaldi
 - Corpora
- **November:**
 - Neural network
- **December:**
 - Evaluation, realization and preparing the report and defense
- **January:**
 - Report submission and defense