

SPOKEN LANGUAGE IDENTIFICATION

Guide
Dr. K. G. Srinivasa

Adarsh Patil
Akshay V Joshi
Harsh K C
Pramod N

What is a Spoken Language Identification ?

- *Spoken Language Identification is the process of recognition of the language spoken in an utterance.*

Problem Statement

- ⦿ Build an optimum corpora
- ⦿ Identifying features for an acoustic model
- ⦿ Employ learning techniques for multi-class classification
- ⦿ Identify language

What are the characteristics of an ideal Language Identification System ?

Characteristics

- ⦿ Accurate language identification
- ⦿ Unaffected by noise
- ⦿ Speaker independent:
 - Gender
 - Accent
 - Pronunciations

Approaches to Spoken Language Identification system

Approaches

◎ Prosodic

- Rhythm, stress, and intonation of speech
- Syllable length, loudness, pitch, and the formant frequencies of speech sounds.

◎ Phonotactic

- Analysis at the phoneme or syllable level
- Rules that govern permissible sequence of phonemes in speech signals.

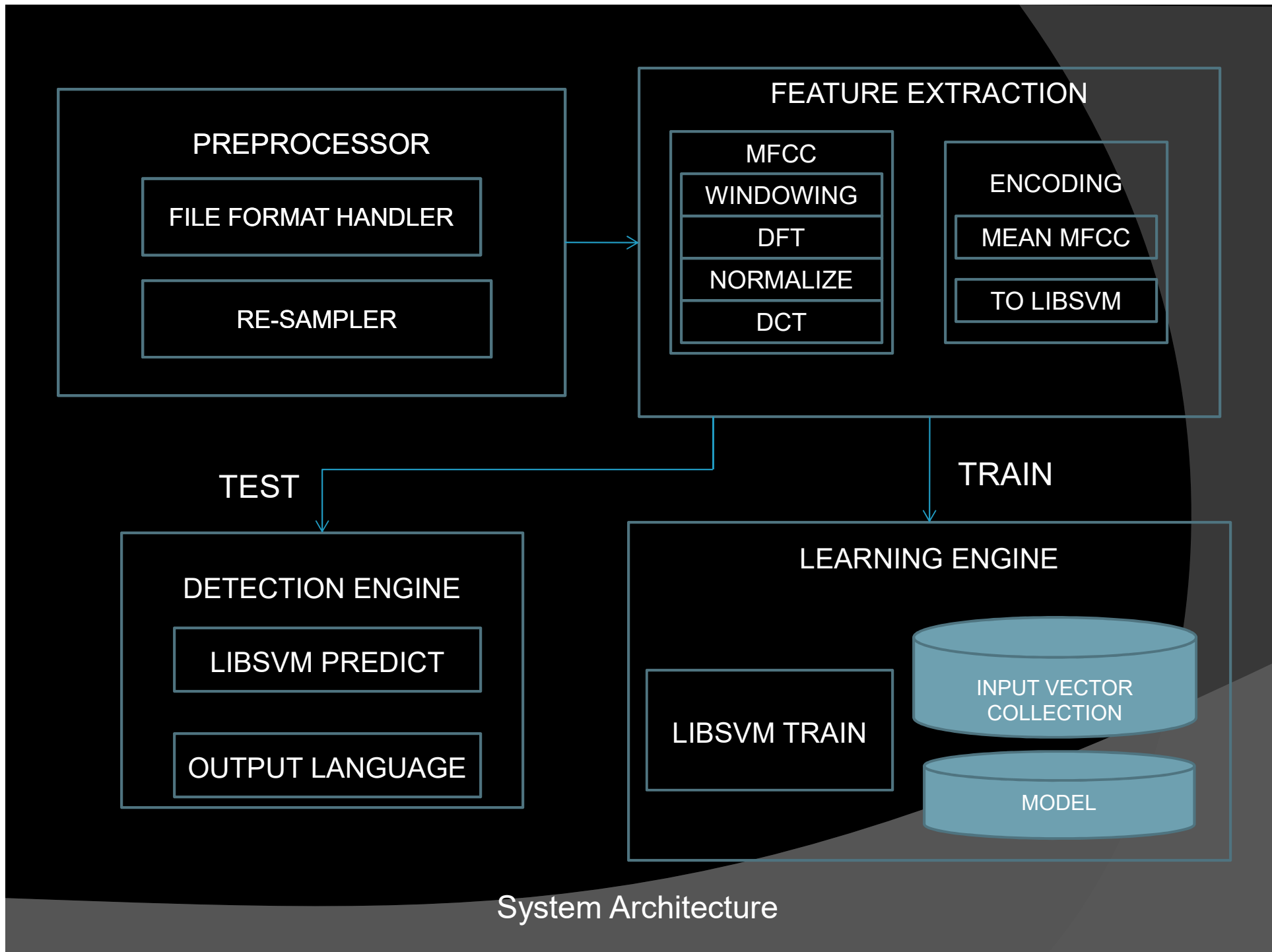
◎ Acoustic

- The power spectrum of a signal is indicative of acoustic information in speech.
- Speech signal characteristics

The Corpora

- ⦿ Unavailability of standard dataset
- ⦿ Custom Corpora : Derived from podcasts and online audio books
- ⦿ Uniform duration of 10 seconds
- ⦿ Semi spontaneous and casual human adapted interaction

System Architecture



Pre-processing

- File format handling

Converting all the input samples to WAV format

- Re-sampling

re-sampling all the input sample to 44.1KHz

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graph TD; subgraph PREPROCESSOR; direction TB; F[FILE FORMAT HANDLER]; R[RE-SAMPLER]; end
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PREPROCESSOR

FILE FORMAT
HANDLER

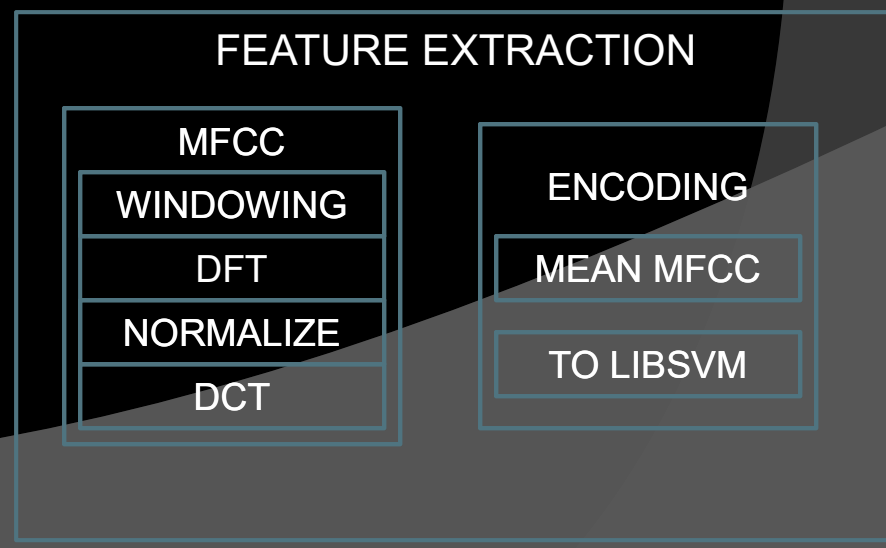
RE-SAMPLER

Acoustic model - MFCC

Mel Frequency Cepstral Coefficient

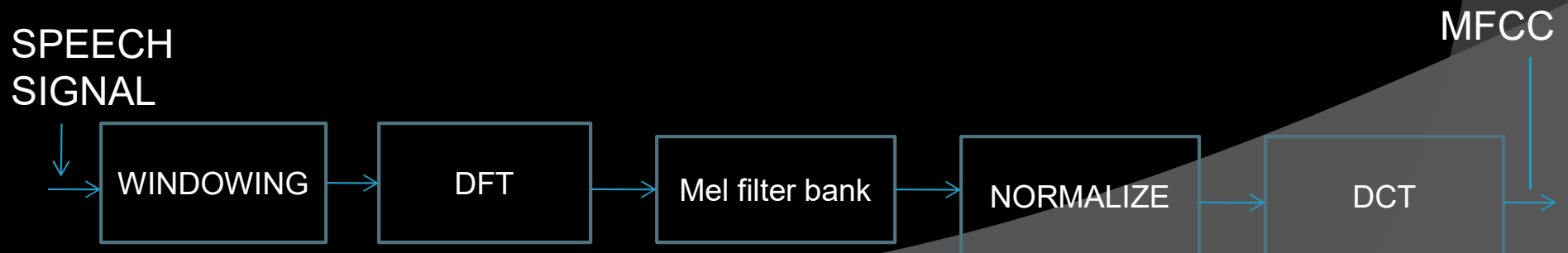
- A representation of the short term power spectrum of a sound

[-12.4518 , 2.30226 , 0.862566 , 0.585914 , 0.387642 , 0.591962 ,
-0.0487 , 0.206065 , 0.163443 , 0.254867 , 0.031689 , 0.042887 ,
0.115736 , 0.025846 , 0.136068 , -0.02038 , 0.04294 ,
0.082779 , 0.01792 , 0.025043]



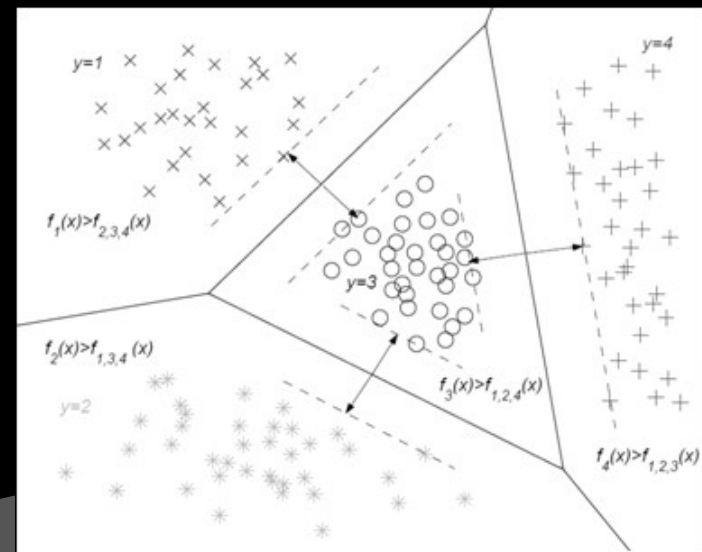
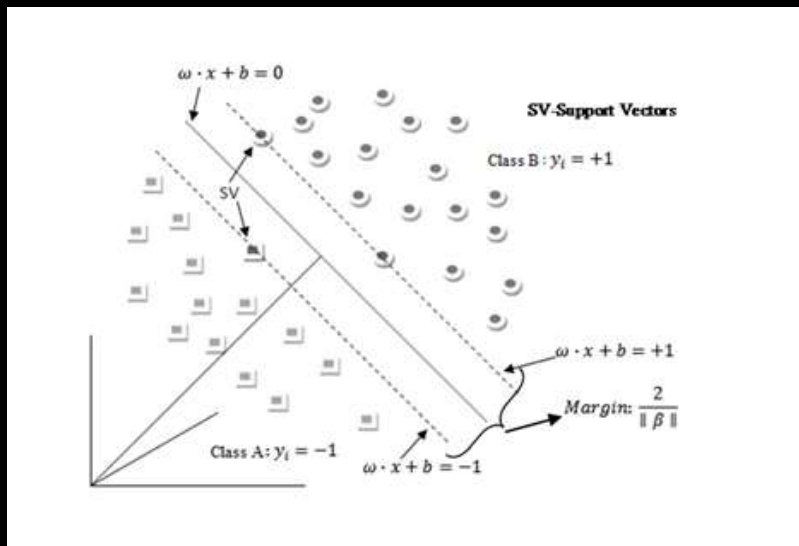
MFCC Computation

- Take the Fourier transform of a windowed excerpt of a signal
- Map the powers of the spectrum obtained above onto the Mel scale
- Normalize the powers at each of the Mel frequencies
- Take the discrete cosine transform of the list of Mel log powers
- The MFCCs are the amplitudes of the resulting spectrum



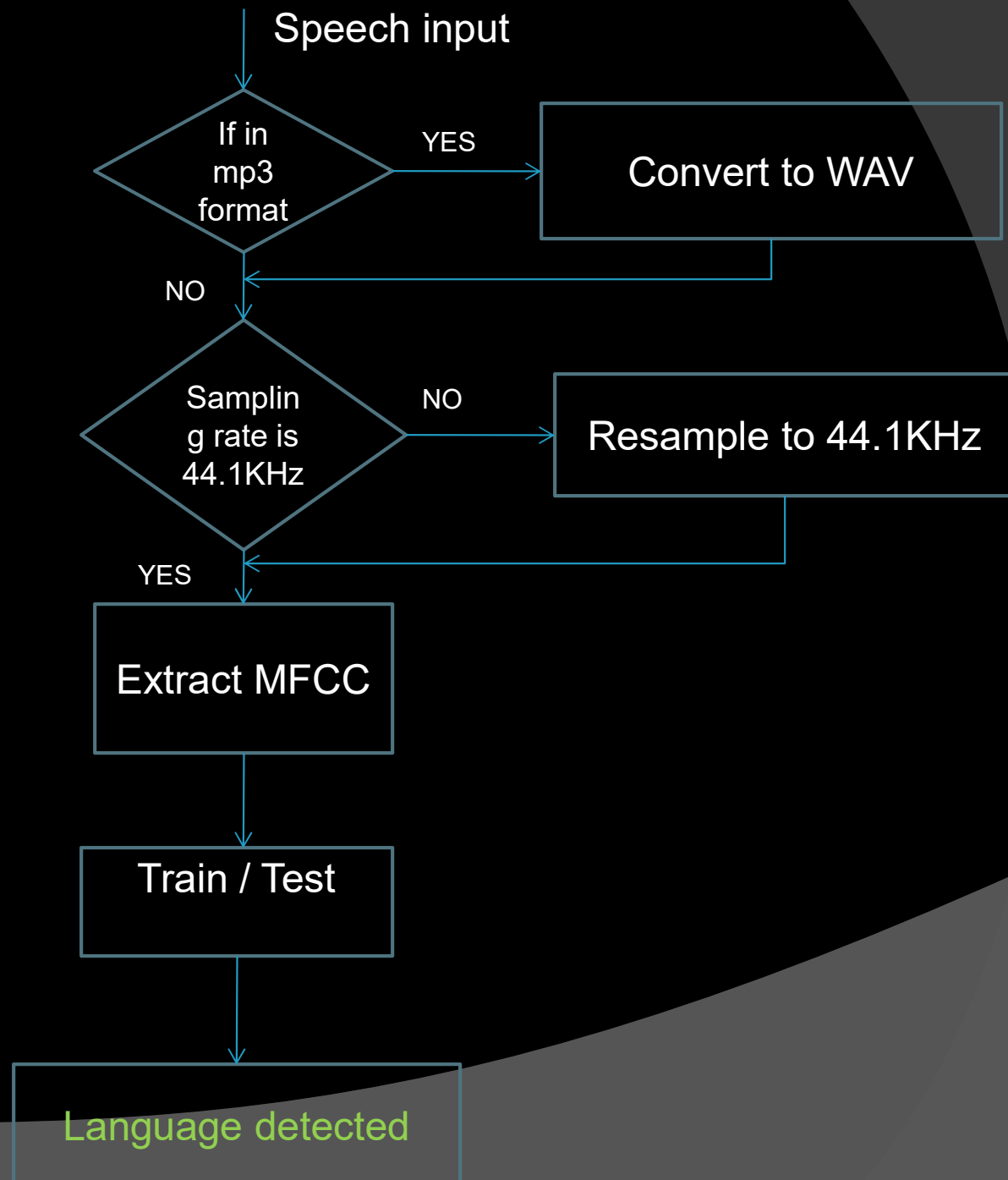
Learning Engine

- Support Vector Machines:
 - Uni-model multiclass classifier
 - Equation of the hyperplane: $\omega \cdot X + \beta = 0$



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Performance Analysis

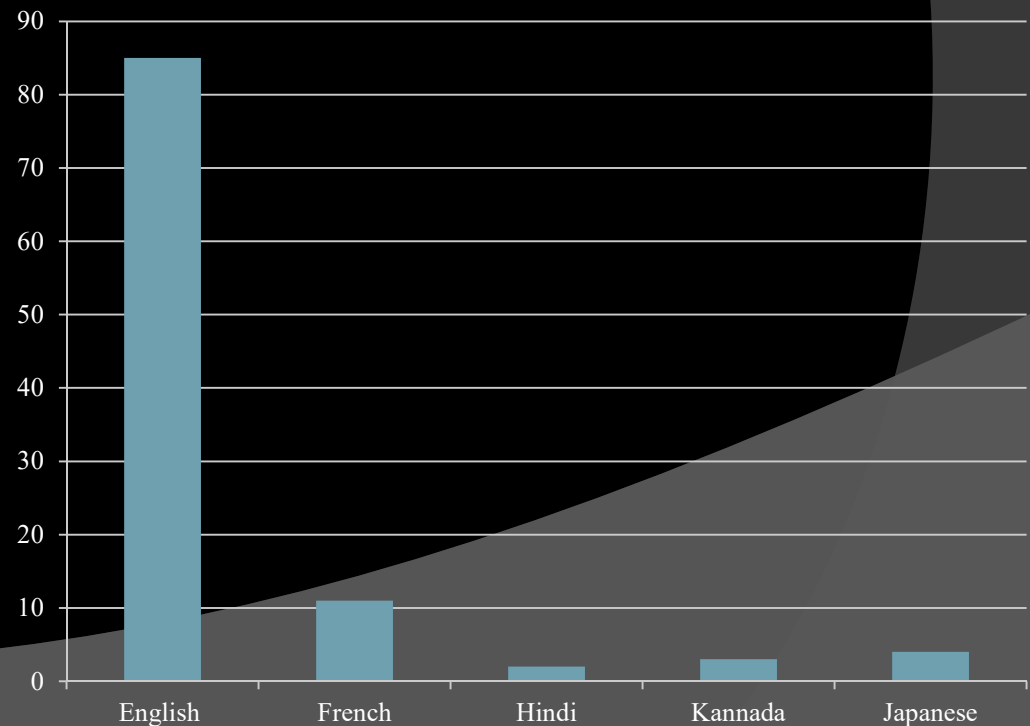
Trivial Experiment

- The entire training set is used for testing
- Number of samples:
 - English: 1093
 - French : 1069
 - Hindi : 853
 - Japanese : 539
 - Kannada : 868

	Eng	Fr	Hin	Kan	Jap
Eng	98.558	0.0108	0.0027	0	0
Fr	0.0935	97.0065	0	0.0935	2.8
Hin	7.735	0.351	91.79	0.1172	0
Kan	2.9935	0.4608	0.1152	96.42	0
Jap	0	1.29	0	0.371	98.3302

Performance Analysis

- ⦿ Test with open source multi lingual speech corpus – VoxForge
- ⦿ Number of English Speech samples : 105
- ⦿ Samples classified as :
 - English : 85
 - French : 11
 - Hindi : 2
 - Kannada : 3
 - Japanese : 4
- ⦿ **Accuracy : 80.95 %**



Challenges

- ⦿ Standard multi lingual speech corpus
- ⦿ Identifying features
- ⦿ Differentiating accents and dialects
- ⦿ Eliminating noise
- ⦿ Scaling system to high throughput classification

Applications

- In Spoken language translators
- In tourist information systems (Airports)
- In customer support switchboard (call routing)
- Military applications (spy applications)

Future Enhancement

- Introduce incremental learning (feedback)
- Increase the number of languages supported
- Make the corpora robust
- Enhance the feature space
- A real time mobile App

Thank you