

Loading data

```
import librosa

from glob import glob

#directories of normal audios

data_dir = "D:\\signal processing\\Language Detection Dataset\\Marathi"

audio = librosa.util.find_files(data_dir , ext=['wav'])

audio
```

Extracting features from audio data

```
import librosa

from librosa import feature

import numpy as np

list_i = [

    feature.chroma_stft,

    feature.spectral_centroid,

    feature.spectral_bandwidth,

    feature.spectral_rolloff

]

list_ii = [

    feature.zero_crossing_rate

]

def feature_vector(y,sr):

    feat_vect_i = [ np.mean(func(y,sr)) for func in list_i]

    feat_vect_ii = [ np.mean(func(y)) for func in list_ii]

    feature_vector = feat_vect_i + feat_vect_ii

    return feature_vector
```

Converting features into list

```
list = []

for file in audio:

    y , sr = librosa.load(file,sr=None,offset = 0.0,duration = 5.0)

    feature_vector = get_feature_vector(y, sr)
```

```
list.append(feature_vector)
```

Creating csv file of features

```
import csv
output = "data.csv"
header =[
    "chroma_stft",
    "spectral_centroid",
    "spectral_bandwidth",
    "spectral_rolloff",
    "zero_crossing_rate"
]
with open(output,"+w") as f:
    csv_writer = csv.writer(f, delimiter = ",")
    csv_writer.writerow(header)
    csv_writer.writerows(norm_audios_feat)
output
```

Loading dataset

```
import pandas as pd
data= pd.read_csv('data.csv')
data
```

```
# Importing the libraries
```

```
import numpy as np
```

```
import pandas as pd
```

```
import pandas as pd
```

Loading data

```
Data1 = "C:\\Users\\Rahul\\Desktop\\language recognition\\Final Data\\English.csv"
```

```
Data2 = "C:\\Users\\Rahul\\Desktop\\language recognition\\Final Data\\hindi.csv"
```

```
Data3 = "C:\\Users\\Rahul\\Desktop\\language recognition\\Final Data\\marathi.csv"
```

```
files = [Data1,Data2,Data3]
```

```
# Combine all three CSV files using the concat method
```

```
data = pd.concat([pd.read_csv(f) for f in files])

# Export to csv
data.to_csv( "DATA.csv", index=False, encoding='utf-8-sig')

data = pd.read_csv("DATA.csv")

data
```

Modelling-

Encoding language column

```
from sklearn.preprocessing import LabelEncoder

le = LabelEncoder()

y_data = le.fit_transform(data['Lang'])

print(y_data)

X = data.iloc[:, 0:5].values

y = y_data
```

Train test split

```
from sklearn.model_selection import train_test_split

X_train, X_test, y_train, y_test = train_test_split(X, y, test_size = 0.3, random_state = 1,
shuffle=True)

print(y_train)
```

Random Forest algorithm

```
# Fitting Random Forest Regression to the dataset

# import the regressor

from sklearn.ensemble import RandomForestClassifier


# create regressor object

classifier= RandomForestClassifier(n_estimators = 100, bootstrap=True,random_state = 1)


# fit the regressor with x and y data

classifier.fit(X_train, y_train)


y_pred = classifier.predict(X_train)
```

```
y_pred
classifier.score(X_test,y_test)
```

Recording Audio

```
import sounddevice as sd
import soundfile as sf
from tkinter import *

def Voice_rec():
    fs = 48000

    # seconds
    duration = 5
    myrecording = sd.rec(int(duration * fs),
        samplerate=fs, channels=2)
    sd.wait()

    # Save as FLAC file at correct sampling rate
    return sf.write('Audio1.wav', myrecording, fs)

master = Tk()

Label(master, text=" Voice Recoder : "
).grid(row=0, sticky=W, rowspan=5)

b = Button(master, text="Start", command=Voice_rec)
b.grid(row=0, column=2, columnspan=2, rowspan=2,
    padx=5, pady=5)

mainloop()
```

Loading Audio data

```
import librosa

list = []

file="C:\\Users\\Rahul\\Desktop\\language recognition\\Audio1.wav"

y , sr = librosa.load(file,sr=None,offset = 0.0,duration = 5.0)

feature_vector = get_feature_vector(y, sr)

list.append(feature_vector)
```

Creating Csv from features of audio

```
import csv

output = "Input_audio.csv"

header =[

    "chroma_stft",

    "spectral_centroid",

    "spectral_bandwidth",

    "spectral_rolloff",

    "zero_crossing_rate"

]

with open(norm_output,"+w") as f:

    csv_writer = csv.writer(f, delimiter = ",")

    csv_writer.writerow(header)

    csv_writer.writerows(list)
```

```
import pandas as pd

Input= pd.read_csv('Input_audio.csv')

Input

Input=data.loc[0].values.tolist()
```

Predicting Output

```
print(classifier.predict([[Input[0:5]]]))
```