Assignment

Spoken Digit Recognition



In this notebook, You will do Spoken Digit Recognition.

Input - speech signal, output - digit number

It contains

- 1. Reading the dataset. and Preprocess the data set. Detailed instrctions are gi ven below. You have to write the code in the same cell which contains the instrction.
- 2. Training the LSTM with RAW data
- 3. Converting to spectrogram and Training the LSTM network
- 4. Creating the augmented data and doing step 2 and 3 again.

instructions:

- 1. Don't change any Grader Functions. Don't manipulate any Grader functions. If you manipulate any, it will be considered as plagiarised.
- 2. Please read the instructions on the code cells and markdown cells. We wil l explain what to write.
- 3. please return outputs in the same format what we asked. Eg. Don't return List of we are asking for a numpy array.
- 4. Please read the external links that we are given so that you will learn the concept behind the code that you are writing.
- 5. We are giving instructions at each section if necessary, please follow th em.

Every Grader function has to return True.

```
In [ ]:
```

In [1]:

```
import numpy as np
import pandas as pd
import librosa
import os
import seaborn as sns
import matplotlib.pyplot as plt
from sklearn.model_selection import train_test_split
from sklearn.metrics import accuracy score, classification report, f1 score, recall sco
re, precision_score
import tensorflow as tf
import tensorflow.keras as keras
from keras.preprocessing.sequence import pad sequences
from tensorflow.keras.layers import Input, LSTM, Dense, Flatten
from tensorflow.keras.models import Model
from keras.utils.vis_utils import plot_model
from keras import backend as K
from tqdm import tqdm
# import tensorflow addons as tfa
```

In [40]:

```
%load_ext tensorboard
import datetime
!rm -rf ./logs/

def warn(*args, **kwargs):
   pass
import warnings
warnings.warn = warn
```

We shared recordings.zip, please unzip those.

In []:

!wget --header="Host: doc-10-7k-docs.googleusercontent.com" --header="User-Agent: Moz
illa/5.0 (Windows NT 10.0; Win64; x64) AppleWebKit/537.36 (KHTML, like Gecko) Chrome/8
9.0.4389.114 Safari/537.36" --header="Accept: text/html,application/xhtml+xml,applicati
on/xml;q=0.9,image/avif,image/webp,image/apng,*/*;q=0.8,application/signed-exchange;v=b
3;q=0.9" --header="Accept-Language: en-GB,en-US;q=0.9,en;q=0.8,la;q=0.7" --header="Cook
ie: AUTH_e6r7g654r08n4c7rvpdb3hggohuuimpl=14748004131253493613|1618294350000|vnbtageljg
c7ne0ea7eqvugi1tqi080f" --header="Connection: keep-alive" "https://doc-10-7k-docs.googl
eusercontent.com/docs/securesc/ir88vhfas868utof5nd7gfa59g7jlumj/7hjeq0fs1u613rjir11ijjb
c21pvc7e1/1618294575000/00484516897554883881/14748004131253493613/17YGQheavMbM6aeHYjUcG
ssXfb7eQHo1z?e=download&authuser=0" -c -0 'recordings.zip'

In [94]:

```
!unzip recordings.zip
```

In [3]:

```
#read the all file names in the recordings folder given by us
#(if you get entire path, it is very useful in future)
#save those files names as list in "all_files"
all_files = []
all_file_paths = []

for fileName in os.listdir("recordings"):
    all_file_paths.append(os.path.join("recordings",fileName))
    all_files.append(fileName)
```

Grader function 1

In [4]:

```
def grader_files():
    temp = len(all_files)==2000
    temp1 = all([x[-3:]=="wav" for x in all_files])
    temp = temp and temp1
    return temp
grader_files()
```

Out[4]:

True

Create a dataframe(name=df_audio) with two columns(path, label).

You can get the label from the first letter of name.

In [5]:

```
#Create a dataframe(name=df_audio) with two columns(path, label).
#You can get the label from the first letter of name.
#Eg: 0_jackson_0 --> 0
#0_jackson_43 --> 0
df_list = [(path,int(file.split('_')[0])) for file,path in zip(all_files,all_file_paths) ]
df_audio = pd.DataFrame(df_list,columns=["path","label"])
```

In [6]:

```
df_audio.head()
```

Out[6]:

	path	label
0	recordings/3_jackson_16.wav	3
1	recordings/2_nicolas_3.wav	2
2	recordings/0_nicolas_34.wav	0
3	recordings/5_yweweler_25.wav	5
4	recordings/5_theo_3.wav	5

In [7]:

```
#info
df_audio.info()
```

Grader function 2

In [8]:

```
def grader_df():
    flag_shape = df_audio.shape==(2000,2)
    flag_columns = all(df_audio.columns==['path', 'label'])
    list_values = list(df_audio.label.value_counts())
    flag_label = len(list_values)==10
    flag_label2 = all([i==200 for i in list_values])
    final_flag = flag_shape and flag_columns and flag_label and flag_label2
    return final_flag
grader_df()
```

Out[8]:

True

In [9]:

```
from sklearn.utils import shuffle
df_audio = shuffle(df_audio, random_state=33)#don't change the random state
```

Train and Validation split

-

In [10]:

```
#split the data into train and validation and save in X_train, X_test, y_train, y_test
#use stratify sampling
#use random state of 45
#use test size of 30%
X = df_audio["path"]
y = df_audio["label"]
X_train, X_test, y_train, y_test = train_test_split(X,y,test_size=0.3,stratify=y,random_state = 45)
```

Grader function 3

In [11]:

```
def grader_split():
    flag_len = (len(X_train)==1400) and (len(X_test)==600) and (len(y_train)==1400) and
(len(y_test)==600)
    values_ytrain = list(y_train.value_counts())
    flag_ytrain = (len(values_ytrain)==10) and (all([i==140 for i in values_ytrain]))
    values_ytest = list(y_test.value_counts())
    flag_ytest = (len(values_ytest)==10) and (all([i==60 for i in values_ytest]))
    final_flag = flag_len and flag_ytrain and flag_ytest
    return final_flag
grader_split()
```

Out[11]:

True

Preprocessing



All files are in the "WAV" format. We will read those raw data files using the librosa

In [12]:

```
sample_rate = 22050
def load_wav(x, get_duration=True):
    '''This return the array values of audio with sampling rate of 22050 and Duratio
n'''
    #loading the wav file with sampling rate of 22050
    samples, sample_rate = librosa.load(x, sr=22050)
    if get_duration:
        duration = librosa.get_duration(samples, sample_rate)
        return [samples, duration]
    else:
        return samples
```

In [13]:

```
#use load_wav function that was written above to get every wave.
#save it in X_train_processed and X_test_processed
# X_train_processed/X_test_processed should be dataframes with two columns(raw_data, du ration) with same index of X_train/y_train

# data : pd.Series()
def process_audio(data):
    samples_list, duration_list, index_list = [], [], []
    for index,filePath in data.iteritems():
        res = load_wav(filePath,True)
        samples_list.append(res[0])
        duration_list.append(res[1])
        index_list.append(index)

res_df = pd.DataFrame(data = list(zip(samples_list,duration_list)), columns=["raw_data","duration"], index=index_list)
        return res_df
```

In [14]:

```
X_train_processed = process_audio(X_train)
X_test_processed = process_audio(X_test)
```

In [15]:

```
X_train_processed.head()
```

Out[15]:

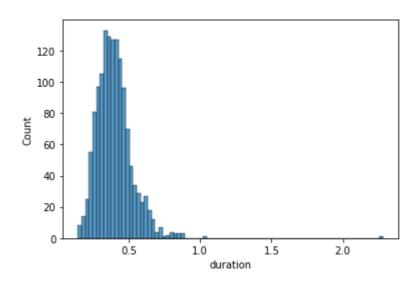
	raw_data	duration
1266	[6.2426276e-05, -8.5246946e-05, -0.00021928802	0.460000
1630	[0.005607843, 0.003237236, 0.001920043, 0.0027	0.290522
1585	[-0.0010017226, -0.0041279504, -0.0062265065,	0.390658
1317	[0.010659263, 0.011418013, 0.009967709, 0.0078	0.405896
1377	[0.00064825395, 0.00068861473, 0.0006723904, 0	0.567891

In [16]:

```
#plot the histogram of the duration for trian
sns.histplot(data=X_train_processed, x="duration")
```

Out[16]:

<matplotlib.axes._subplots.AxesSubplot at 0x7fe038a2db50>



In [17]:

```
#print 0 to 100 percentile values with step size of 10 for train data duration.
for i in range(0,101,10):
   percentile = np.percentile(X_train_processed["duration"].values,i)
   print("{0} th Percentile is {1}".format(i,percentile))
```

```
0 th Percentile is 0.1435374149659864
10 th Percentile is 0.2599909297052154
20 th Percentile is 0.2984308390022676
30 th Percentile is 0.3310204081632653
40 th Percentile is 0.3575238095238095
50 th Percentile is 0.38834467120181404
60 th Percentile is 0.4160362811791383
70 th Percentile is 0.44450793650793646
80 th Percentile is 0.4799818594104308
90 th Percentile is 0.5540317460317462
100 th Percentile is 2.282766439909297
```

In [18]:

```
##print 90 to 100 percentile values with step size of 1.
for i in range(90,101,1):
   percentile = np.percentile(X_train_processed["duration"].values,i)
   print("{0} th Percentile is {1}".format(i,percentile))
```

```
90 th Percentile is 0.5540317460317462
91 th Percentile is 0.569807709750567
92 th Percentile is 0.5811446712018141
93 th Percentile is 0.5915165532879821
94 th Percentile is 0.6080870748299319
95 th Percentile is 0.622421768707483
96 th Percentile is 0.6366947845804989
97 th Percentile is 0.6574408163265306
98 th Percentile is 0.6884154195011338
99 th Percentile is 0.790681179138322
100 th Percentile is 2.282766439909297
```

Grader function 4

In [19]:

```
def grader_processed():
    flag_columns = (all(X_train_processed.columns==['raw_data', 'duration'])) and (all(
X_test_processed.columns==['raw_data', 'duration']))
    flag_shape = (X_train_processed.shape ==(1400, 2)) and (X_test_processed.shape==(60
0,2))
    return flag_columns and flag_shape
grader_processed()
```

Out[19]:

True

Based on our analysis 99 percentile values are less than 0.8sec so we will limit maximum length of X_train_processed and X_test_processed to 0.8 sec. It is simil ar to pad_sequence for a text dataset.

While loading the audio files, we are using sampling rate of 22050 so one sec will give array of length 22050. so, our maximum length is 0.8*22050 = 17640

Pad with Zero if length of sequence is less than 17640 else Truncate the number.

Also create a masking vector for train and test.

masking vector value = 1 if it is real value, 0 if it is pad value. Masking vect or data type must be bool.

```
In [20]:
```

```
max_length = 17640
```

In [21]:

```
## as discussed above, Pad with Zero if length of sequence is less than 17640 else Trun
cate the number.
## save in the X_train_pad_seq, X_test_pad_seq
## also Create masking vector X_train_mask, X_test_mask

## all the X_train_pad_seq, X_test_pad_seq, X_train_mask, X_test_mask will be numpy arr
ays mask vector dtype must be bool.

def pad_data(data):
    max_length = 17640
    padded_seq = pad_sequences(data,maxlen=max_length, padding='post',truncating='post',d
type='float32')

mask_arr = []
for arr in padded_seq:
    mask = [True if val == 0 else False for val in arr]
    mask_arr.append(np.array(mask))

return padded_seq, np.array(mask_arr)
```

In [22]:

```
X_train_pad_seq, X_train_mask = pad_data(X_train_processed["raw_data"].values)
X_test_pad_seq, X_test_mask = pad_data(X_test_processed["raw_data"].values)
```

Grader function 5

In [23]:

```
def grader_padoutput():
    flag_padshape = (X_train_pad_seq.shape==(1400, 17640)) and (X_test_pad_seq.shape==(
600, 17640)) and (y_train.shape==(1400,))
    flag_maskshape = (X_train_mask.shape==(1400, 17640)) and (X_test_mask.shape==(600,
17640)) and (y_test.shape==(600,))
    flag_dtype = (X_train_mask.dtype==bool) and (X_test_mask.dtype==bool)
    return flag_padshape and flag_maskshape and flag_dtype
grader_padoutput()
```

Out[23]:

True

1. Giving Raw data directly.

Now we have

Train data: X_train_pad_seq, X_train_mask and y_train
Test data: X_test_pad_seq, X_test_mask and y_test

We will create a LSTM model which takes this input.

Task:

- 1. Create an LSTM network which takes "X_train_pad_seq" as input, "X_train_mask" as mask input. You can use any number of LSTM cells. Please read LSTM documentat ion(https://www.tensorflow.org/api_docs/python/tf/keras/layers/LSTM) in tensorfl ow to know more about mask and also https://www.tensorflow.org/guide/keras/masking_and_padding
- 2. Get the final output of the LSTM and give it to Dense layer of any size and then give it to Dense layer of size 10(because we have 10 outputs) and then compile with the sparse categorical cross entropy(because we are not converting it to one hot vectors).
- 3. Use tensorboard to plot the graphs of loss and metric(use micro F1 score as m etric) and histograms of gradients.
- 4. make sure that it won't overfit.
- 5. You are free to include any regularization

In [41]:

```
X_test_pad_seq.shape
Out[41]:
```

(600, 17640)

In [42]:

```
input = Input(shape=(X_train_pad_seq.shape[1],1),name="Input_Data")
input_mask = Input(shape=(X_train_mask.shape[1]), name='Input_Mask', dtype='bool')
lstm = LSTM(64)(input,mask=input_mask)
dense = Dense(32)(lstm)
output = Dense(10, activation="softmax")(dense)
```

In [43]:

```
## as discussed above, please write the LSTM
model1 = Model(inputs=[input,input_mask], outputs =[output])
```

In [44]:

```
model1.summary()
Model: "model_4"
Layer (type)
                                 Output Shape
                                                        Param #
                                                                    Connected
Input_Data (InputLayer)
                                 [(None, 17640, 1)]
Input_Mask (InputLayer)
                                 [(None, 17640)]
                                                        0
1stm_5 (LSTM)
                                 (None, 64)
                                                        16896
                                                                    Input_Dat
a[0][0]
                                                                    Input Mas
k[0][0]
dense_12 (Dense)
                                 (None, 32)
                                                        2080
                                                                    1stm_5[0]
[0]
dense_13 (Dense)
                                  (None, 10)
                                                        330
                                                                    dense_12
[0][0]
Total params: 19,306
Trainable params: 19,306
Non-trainable params: 0
```

In [45]:

```
class Metrics Callback(tf.keras.callbacks.Callback):
 def __init__(self,x_val,y_val):
    self.x_val = x_val
    self.y_val = y_val
 def on_train_begin(self, logs={}):
    self.history = {"micro f1":[]}
 def on_epoch_end(self, epoch, logs={}):
    y_pred = self.model.predict(self.x_val)
    y_pred = [np.argmax(arr) for arr in y_pred]
    f1_s = f1_score(self.y_val, y_pred, average='micro')
    self.history["micro_f1"].append(f1_s)
    print('val_f1_score: ',f1_s)
micro_f1_callback = Metrics_Callback([X_test_pad_seq,X_test_mask],y_test)
log_dir="logs/fit/" + datetime.datetime.now().strftime("%Y%m%d-%H%M%S")
tensorboard_callback = tf.keras.callbacks.TensorBoard(log_dir=log_dir, histogram_freq=1
 write graph=True)
```

```
In [ ]:
```

```
set(y_test)
Out[ ]:
\{0, 1, 2, 3, 4, 5, 6, 7, 8, 9\}
In [48]:
model1.compile(optimizer='adam',
              loss=keras.losses.SparseCategoricalCrossentropy(),
              metrics=["accuracy"])
```

In [49]:

```
model1.fit([X_train_pad_seq,X_train_mask],y_train,
          epochs=5,
          batch_size=256,
          callbacks = [micro f1 callback, tensorboard callback],
          validation_data = ([X_test_pad_seq,X_test_mask],y_test)
```

```
Epoch 1/5
racy: 0.1001 - val_loss: 2.3026 - val_accuracy: 0.1000
val_f1_score: 0.100000000000000002
Epoch 2/5
racy: 0.0871 - val_loss: 2.3025 - val_accuracy: 0.1000
val_f1_score: 0.10000000000000002
Epoch 3/5
racy: 0.0888 - val loss: 2.3025 - val accuracy: 0.1000
val f1 score: 0.100000000000000002
Epoch 4/5
racy: 0.0957 - val_loss: 2.3025 - val_accuracy: 0.1000
val f1 score: 0.10000000000000002
Epoch 5/5
6/6 [=========== ] - 262s 44s/step - loss: 2.3025 - accu
racy: 0.1012 - val loss: 2.3025 - val accuracy: 0.1000
val_f1_score: 0.100000000000000002
Out[49]:
```

<tensorflow.python.keras.callbacks.History at 0x7fe036049a90>

In [50]:

```
%tensorboard --logdir logs/fit
```

2. Converting into spectrogram and giving spectrogram data as input

We can use librosa to convert raw data into spectrogram. A spectrogram shows the features in a two-dimensional representation with the intensity of a frequency at a point in time i.e we are converting Time domain to frequency domain. you can read more about this in https://pnsn.org/spectrograms/what-is-a-spectrogram

In [51]:

```
def convert_to_spectrogram(raw_data):
    '''converting to spectrogram'''
    spectrum = librosa.feature.melspectrogram(y=raw_data, sr=sample_rate, n_mels=64)
    logmel_spectrum = librosa.power_to_db(S=spectrum, ref=np.max)
    return logmel_spectrum
```

In [52]:

```
##use convert_to_spectrogram and convert every raw sequence in X_train_pad_seq and X_te
st_pad-seq.
## save those all in the X_train_spectrogram and X_test_spectrogram ( These two arrays
must be numpy arrays)
X_train_spectrogram = np.array([convert_to_spectrogram(x) for x in X_train_pad_seq])
X_test_spectrogram = np.array([convert_to_spectrogram(x) for x in X_test_pad_seq])
```

Grader function 6

In [53]:

```
def grader_spectrogram():
    flag_shape = (X_train_spectrogram.shape==(1400,64, 35)) and (X_test_spectrogram.sha
pe == (600, 64, 35))
    return flag_shape
grader_spectrogram()
```

Out[53]:

True

Now we have

```
Train data: X_train_spectrogram and y_train
Test data: X_test_spectrogram and y_test
```

We will create a LSTM model which takes this input.

Task:

- 1. Create an LSTM network which takes "X_train_spectrogram" as input and has to return output at every time step.
- 2. Average the output of every time step and give this to the Dense layer of any size.

(ex: Output from LSTM will be (#., time_steps, features) average the output of
every time step i.e, you should get (#.,time_steps)
and then pass to dense layer)

- 3. give the above output to Dense layer of size 10(output layer) and train the network with sparse categorical cross entropy.
- 4. Use tensorboard to plot the graphs of loss and metric(use micro F1 score as m etric) and histograms of gradients.
- 5. make sure that it won't overfit.
- 6. You are free to include any regularization

In [54]:

```
X_train_spectrogram.shape
```

Out[54]:

(1400, 64, 35)

In [55]:

```
input = Input(shape=(X_train_spectrogram.shape[1],X_train_spectrogram.shape[2]),name="I
nput_Data")
lstm = LSTM(120,return_sequences=True)(input)
X = tf.reduce_mean(lstm, axis=-1)
dense = Dense(128)(X)
dense2 = Dense(64)(dense)
flatten = Flatten()(dense2)
output = Dense(10, activation="softmax")(flatten)
```

In [56]:

```
model2 = Model(input, output)
```

In [57]:

```
model2.summary()
```

Model: "model_5"

Layer (type)	Output Shape	Param #
Input_Data (InputLayer)	[(None, 64, 35)]	0
lstm_6 (LSTM)	(None, 64, 120)	74880
tf.math.reduce_mean_4 (TFOpL	(None, 64)	0
dense_14 (Dense)	(None, 128)	8320
dense_15 (Dense)	(None, 64)	8256
flatten_4 (Flatten)	(None, 64)	0
dense_16 (Dense)	(None, 10)	650
Total params: 92,106 Trainable params: 92,106		

Trainable params: 92,106 Non-trainable params: 0

In [59]:

```
class Metrics_Callback_2(tf.keras.callbacks.Callback):
 def __init__(self,x_val,y_val):
    self.x_val = x_val
    self.y_val = y_val
 def on_train_begin(self, logs={}):
    self.history = {"micro_f1":[]}
 def on_epoch_end(self, epoch, logs={}):
    y pred = self.model.predict(self.x val)
    y_pred = [np.argmax(arr) for arr in y_pred]
    f1_s = f1_score(self.y_val, y_pred, average='micro')
    self.history["micro_f1"].append(f1_s)
    print(' val_f1_score: ',f1_s)
micro f1 callback = Metrics Callback 2(X test spectrogram, y test)
log dir="logs/fit/" + datetime.datetime.now().strftime("%Y%m%d-%H%M%S")
tensorboard_callback = tf.keras.callbacks.TensorBoard(log_dir=log_dir, histogram_freq=1
, write_graph=True)
```

In [60]:

In [61]:

```
model2.fit(X train spectrogram, y train,
          epochs=10,
          batch size=32,
          callbacks = [micro f1 callback, tensorboard callback],
          validation_data = (X_test_spectrogram, y_test)
          )
```

```
Epoch 1/10
uracy: 0.1506 - val_loss: 1.9301 - val_accuracy: 0.3067
val f1 score: 0.3066666666666664
Epoch 2/10
uracy: 0.3625 - val_loss: 1.4271 - val_accuracy: 0.5267
val_f1_score: 0.5266666666666666
Epoch 3/10
44/44 [================ ] - 4s 81ms/step - loss: 1.3490 - acc
uracy: 0.5407 - val_loss: 1.0407 - val_accuracy: 0.6267
val f1 score: 0.6266666666666667
Epoch 4/10
uracy: 0.6576 - val_loss: 0.8490 - val_accuracy: 0.7000
val f1 score: 0.7
Epoch 5/10
uracy: 0.7195 - val_loss: 0.7192 - val_accuracy: 0.7617
val_f1_score: 0.7616666666666667
Epoch 6/10
uracy: 0.7322 - val loss: 0.6881 - val accuracy: 0.7783
Epoch 7/10
uracy: 0.7796 - val_loss: 0.6117 - val_accuracy: 0.7883
Epoch 8/10
uracy: 0.7979 - val_loss: 0.5270 - val_accuracy: 0.8250
val f1 score: 0.825
Epoch 9/10
44/44 [============= ] - 3s 77ms/step - loss: 0.5352 - acc
uracy: 0.8040 - val loss: 0.5159 - val accuracy: 0.8233
val f1 score: 0.8233333333333334
Epoch 10/10
uracy: 0.8404 - val_loss: 0.4573 - val_accuracy: 0.8517
Out[61]:
```

<tensorflow.python.keras.callbacks.History at 0x7fe024d9f390>

In [65]:

```
%tensorboard --logdir logs/fit
```

3. data augmentation

Till now we have done with 2000 samples only. It is very less data. We are givin g the process of generating augmented data below.

There are two types of augmentation:

- 1. time stretching Time stretching either increases or decreases the length of the file. For time stretching we move the file 30% faster or slower
- 2. pitch shifting pitch shifting moves the frequencies higher or lower. For pitch shifting we shift up or down one half-step.

In [66]:

```
## generating augmented data.
def generate_augmented_data(file_path):
    augmented_data = []
    samples = load_wav(file_path,get_duration=False)
    for time_value in [0.7, 1, 1.3]:
        for pitch_value in [-1, 0, 1]:
            time_stretch_data = librosa.effects.time_stretch(samples, rate=time_value)
            final_data = librosa.effects.pitch_shift(time_stretch_data, sr=sample_rate,
n_steps=pitch_value)
            augmented_data.append(final_data)
    return augmented_data
```

In [67]:

```
temp_path = df_audio.iloc[0].path
aug_temp = generate_augmented_data(temp_path)
```

In [68]:

```
len(aug_temp)
```

Out[68]:

9

As discussed above, for one data point, we will get 9 augmented data points.

Split data into train and test (80-20 split)

We have 2000 data points(1600 train points, 400 test points)

Do augmentation only on train data, after augmentation we will get 14400 train points.

do the above steps i.e training with raw data and spectrogram data with augmentation.

In [69]:

```
X = df_audio["path"]
y = df_audio["label"]
X_train, X_test, y_train, y_test = train_test_split(X,y,test_size=0.2,stratify=y,random
_state = 45)
```

```
In [70]:
```

```
X_train_aug = []
y_train_aug = []
for fileName, y_val in zip(X_train,y_train):
    aug_data = generate_augmented_data(fileName)
    for data in aug_data:
        X_train_aug.append(data)
# X_train_aug +=
    y_train_aug += [y_val] * 9
```

In [71]:

```
print(len(X_train_aug))
print(len(y_train_aug))

X_train_processed = pd.DataFrame(data={"raw_data":X_train_aug})
y_train = pd.DataFrame(data={"label":y_train_aug})

X_test_processed = X_test.apply(lambda x: load_wav(x,get_duration=False))
```

14400 14400

In [72]:

```
X_train_aug_pad_seq, X_train_aug_mask = pad_data(X_train_processed["raw_data"].values)
X_test_aug_pad_seq, X_test_aug_mask = pad_data(X_test_processed.values)
```

In []:

Raw data Model.

In [73]:

```
input = Input(shape=(X_train_aug_pad_seq.shape[1],1),name="Input_Data")
input_mask = Input(shape=(X_train_aug_mask.shape[1]), name='Input_Mask', dtype='bool')
lstm = LSTM(64)(input,mask=input_mask)
dense = Dense(32)(lstm)
output = Dense(10, activation="softmax")(dense)
```

In [74]:

```
## as discussed above, please write the LSTM
model3 = Model(inputs=[input,input_mask], outputs =[output])
```

In [75]:

```
model3.summary()
Model: "model_6"
Layer (type)
                                                        Param #
                                  Output Shape
                                                                     Connected
Input_Data (InputLayer)
                                  [(None, 17640, 1)]
Input_Mask (InputLayer)
                                  [(None, 17640)]
                                                        0
lstm_7 (LSTM)
                                  (None, 64)
                                                        16896
                                                                     Input_Dat
a[0][0]
                                                                     Input Mas
k[0][0]
dense_17 (Dense)
                                  (None, 32)
                                                        2080
                                                                     1stm_7[0]
[0]
dense_18 (Dense)
                                  (None, 10)
                                                        330
                                                                     dense_17
[0][0]
Total params: 19,306
Trainable params: 19,306
Non-trainable params: 0
```

In [76]:

```
class Metrics Callback(tf.keras.callbacks.Callback):
 def __init__(self,x_val,y_val):
    self.x_val = x_val
    self.y_val = y_val
 def on_train_begin(self, logs={}):
    self.history = {"micro f1":[]}
 def on_epoch_end(self, epoch, logs={}):
    y_pred = self.model.predict(self.x_val)
    y_pred = [np.argmax(arr) for arr in y_pred]
    f1_s = f1_score(self.y_val, y_pred, average='micro')
    self.history["micro_f1"].append(f1_s)
    print(' val_f1: ',f1_s)
micro_f1_callback = Metrics_Callback([X_test_aug_pad_seq,X_test_aug_mask],y_test)
log_dir="logs/fit/" + datetime.datetime.now().strftime("%Y%m%d-%H%M%S")
tensorboard callback = tf.keras.callbacks.TensorBoard(log dir=log dir, histogram freq=1
, write_graph=True)
```

```
In [ ]:
```

In [78]:

```
Epoch 1/5
curacy: 0.1114 - val_loss: 2.3040 - val_accuracy: 0.1000
val f1: 0.100000000000000002
Epoch 2/5
curacy: 0.1222 - val_loss: 2.3110 - val_accuracy: 0.1000
val_f1: 0.100000000000000002
Epoch 3/5
curacy: 0.1353 - val loss: 2.3120 - val accuracy: 0.1000
val_f1: 0.100000000000000002
Epoch 4/5
curacy: 0.1218 - val_loss: 2.3106 - val_accuracy: 0.1000
val f1: 0.100000000000000002
Epoch 5/5
curacy: 0.1263 - val_loss: 2.3099 - val_accuracy: 0.1000
val f1: 0.100000000000000002
Out[78]:
```

<tensorflow.python.keras.callbacks.History at 0x7fdfe632cf90>

In [85]:

```
%tensorboard --logdir logs/fit
```

2. Converting into spectrogram and giving spectrogram data as input

We can use librosa to convert raw data into spectrogram. A spectrogram shows the features in a two-dimensional representation with the intensity of a frequency at a point in time i.e we are converting Time domain to frequency domain. you can read more about this in https://pnsn.org/spectrograms/what-is-a-spectrogram

In [79]:

```
def convert_to_spectrogram(raw_data):
    '''converting to spectrogram'''
    spectrum = librosa.feature.melspectrogram(y=raw_data, sr=sample_rate, n_mels=64)
    logmel_spectrum = librosa.power_to_db(S=spectrum, ref=np.max)
    return logmel_spectrum
```

In [80]:

```
##use convert_to_spectrogram and convert every raw sequence in X_train_pad_seq and X_te
st_pad-seq.
## save those all in the X_train_spectrogram and X_test_spectrogram ( These two arrays
    must be numpy arrays)
X_train_spectrogram = np.array([convert_to_spectrogram(x) for x in X_train_aug_pad_seq
])
X_test_spectrogram = np.array([convert_to_spectrogram(x) for x in X_test_aug_pad_seq])
```

In [81]:

```
input = Input(shape=(X_train_spectrogram.shape[1],X_train_spectrogram.shape[2]),name="I
nput_Data")
lstm = LSTM(120,return_sequences=True)(input)
X = tf.reduce_mean(lstm, axis=-1)
dense = Dense(128)(X)
dense2 = Dense(64)(dense)
flatten = Flatten()(dense2)
output = Dense(10, activation="softmax")(flatten)
```

In [82]:

```
model4 = Model(input, output)
model4.summary()
```

Model: "model_7"

	======
[(None, 64, 35)]	0
(None, 64, 120)	74880
(None, 64)	0
(None, 128)	8320
(None, 64)	8256
(None, 64)	0
(None, 10)	650
	(None, 64) (None, 64) (None, 64)

Trainable params: 92,106 Non-trainable params: 0

In [88]:

```
class Metrics_Callback_2(tf.keras.callbacks.Callback):
 def __init__(self,x_val,y_val):
    self.x_val = x_val
    self.y_val = y_val
 def on_train_begin(self, logs={}):
    self.history = {"micro_f1":[]}
 def on_epoch_end(self, epoch, logs={}):
    y pred = self.model.predict(self.x val)
    y_pred = [np.argmax(arr) for arr in y_pred]
    f1_s = f1_score(self.y_val, y_pred, average='micro')
    self.history["micro f1"].append(f1 s)
    print(' val_f1_score: ',f1_s)
micro_f1_callback = Metrics_Callback_2(X_test_spectrogram, y_test)
log_dir="logs/fit/" + datetime.datetime.now().strftime("%Y%m%d-%H%M%S")
tensorboard callback = tf.keras.callbacks.TensorBoard(log dir=log dir, histogram freq=1
, write_graph=True)
```

In [89]:

```
In [90]:
model4 = model4.fit(X_train_spectrogram,y_train,
         epochs=10,
         batch_size=64,
         callbacks = [micro_f1_callback, tensorboard_callback],
         validation data = (X test spectrogram, y test)
         )
Epoch 1/10
225/225 [============= ] - 23s 94ms/step - loss: 1.9063 -
accuracy: 0.2948 - val_loss: 0.8202 - val_accuracy: 0.6725
val_f1_score: 0.6725
Epoch 2/10
225/225 [============= ] - 21s 93ms/step - loss: 0.7923 -
accuracy: 0.7046 - val loss: 0.5856 - val accuracy: 0.8100
val f1 score: 0.81
Epoch 3/10
225/225 [================ ] - 21s 93ms/step - loss: 0.5787 -
accuracy: 0.7887 - val_loss: 0.4752 - val_accuracy: 0.8100
val f1 score: 0.81
Epoch 4/10
225/225 [============= ] - 21s 95ms/step - loss: 0.4836 -
accuracy: 0.8251 - val loss: 0.3597 - val accuracy: 0.8875
val_f1_score: 0.8875
Epoch 5/10
225/225 [============= ] - 22s 99ms/step - loss: 0.4501 -
accuracy: 0.8365 - val loss: 0.3472 - val accuracy: 0.8675
val f1 score: 0.8675
Epoch 6/10
225/225 [================ ] - 22s 97ms/step - loss: 0.4164 -
accuracy: 0.8474 - val_loss: 0.3616 - val_accuracy: 0.8850
val_f1_score: 0.885
Epoch 7/10
225/225 [=============== ] - 21s 94ms/step - loss: 0.3965 -
accuracy: 0.8617 - val_loss: 0.3250 - val_accuracy: 0.8900
val_f1_score: 0.89
Epoch 8/10
225/225 [============ ] - 22s 99ms/step - loss: 0.3719 -
accuracy: 0.8677 - val loss: 0.2860 - val accuracy: 0.9000
val f1 score: 0.9
Epoch 9/10
225/225 [================ ] - 21s 93ms/step - loss: 0.3820 -
```

In [91]:

Epoch 10/10

```
%tensorboard --logdir logs/fit
```

val f1 score: 0.91

val_f1_score: 0.895

accuracy: 0.8630 - val_loss: 0.2741 - val_accuracy: 0.9100

accuracy: 0.8785 - val_loss: 0.3072 - val_accuracy: 0.8950

225/225 [============] - 21s 94ms/step - loss: 0.3465 -

```
In [92]:
```

```
micro_f1_callback.history
```

Out[92]:

```
{'micro_f1': [0.6725,
0.81,
0.81,
0.8875,
0.8675,
0.885,
0.89,
0.9,
0.91,
0.895]}
```

In [93]:

```
!jupyter nbconvert --to html "/content/Speech_detection_Assignment.ipynb"
```

[NbConvertApp] Converting notebook /content/Speech_detection_Assignment.ip ynb to html

[NbConvertApp] Writing 498976 bytes to /content/Speech_detection_Assignmen t.html

In []: