Speech_detection

August 9, 2020

```
[3]: import numpy as np
import pandas as pd
import librosa
import os

from pathlib import Path
```

We shared recordings.zip, please unzip those.

```
[5]: RECORDINGS_DIR = Path('/content/recordings/')
all_files = os.listdir(RECORDINGS_DIR)
```

Grader function 1

```
[]: 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()
```

[]: True

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 = pd.DataFrame({
           'path': path,
           'label': label
       })
       return df
     df_audio = create_dataframe(all_files)
[]: #info
     df_audio.info()
    <class 'pandas.core.frame.DataFrame'>
    RangeIndex: 2000 entries, 0 to 1999
    Data columns (total 2 columns):
         Column Non-Null Count Dtype
    --- ----- -----
                 2000 non-null object
     0
         path
     1
         label
                 2000 non-null object
    dtypes: object(2)
    memory usage: 31.4+ KB
    Grader function 2
[]: 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()
[]: True
[]: from sklearn.utils import shuffle
     df_audio = shuffle(df_audio, random_state=33)#don't change the random state
[7]: #split the data into train and validation and save in X_train, X_test, y_train, \( \sqrt{2} \)
     \rightarrow y_t test
     #use stratify sampling
     #use random state of 45
     #use test size of 30%
     from sklearn.model_selection import train_test_split
     X, y = df_audio['path'], df_audio['label']
```

Grader function 3

[]: True

```
[8]: sample_rate = 22050

def load_wav(x, get_duration=True):
    '''This return the array values of audio with sampling rate of 22050 and
    →Duration'''
    #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
```

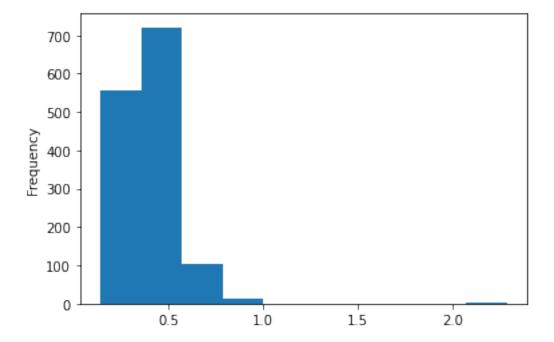
```
'duration': duration,
})

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

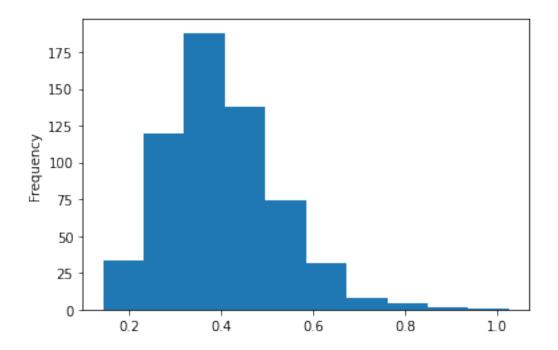
CPU times: user 32.8 s, sys: 39.2 s, total: 1min 12s

Wall time: 3min 54s

[]: #plot the histogram of the duration for trian X_train_processed['duration'].plot.hist();



```
[]: #plot the histogram of the duration for trian
X_test_processed['duration'].plot.hist();
```



```
def cal_percentiles(values, start, stop, step):
       for i in range(start, stop, step):
         print(f'{i} th percentile is {np.percentile(values, i)}')
    cal_percentiles(X_train_processed['duration'].values, 0, 101, 10)
    0 th percentile is 0.1435374149659864
    10 th percentile is 0.25839909297052155
    20 th percentile is 0.30187755102040814
    30 th percentile is 0.3346485260770975
    40 th percentile is 0.36226757369614515
    50 th percentile is 0.3925396825396825
    60 th percentile is 0.41944671201814054
    70 th percentile is 0.4493378684807256
    80 th percentile is 0.48478911564625854
    90 th percentile is 0.5557097505668935
    100 th percentile is 2.282766439909297
[]: ##print 90 to 100 percentile values with step size of 1.
    cal_percentiles(X_train_processed['duration'].values, 90, 101, 1)
    90 th percentile is 0.5549160997732426
    91 th percentile is 0.5659854875283448
    92 th percentile is 0.581082993197279
    93 th percentile is 0.601049433106576
```

[]: #print 0 to 100 percentile values with step size of 10 for train data duration.

```
94 th percentile is 0.6133478458049886

95 th percentile is 0.6230385487528345

96 th percentile is 0.6400816326530612

97 th percentile is 0.6635741496598639

98 th percentile is 0.6897750566893422

99 th percentile is 0.7963859410430838

100 th percentile is 2.282766439909297
```

Grader function 4

```
[]: 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==(600,2))
    return flag_columns and flag_shape
    grader_processed()
```

[]: True

Grader function 5

```
return flag_padshape and flag_maskshape and flag_dtype
grader_padoutput()
```

[]: True

0.0.1 1. Giving Raw data directly.

```
[12]: from tensorflow.keras.layers import Input, LSTM, Dense
    from tensorflow.keras.models import Model
    from tensorflow.keras.utils import to_categorical

[]: y_train = to_categorical(y_train, 10)
    y_test = to_categorical(y_test, 10)

[]: X_train_lstm_seq_inp = np.expand_dims(X_train_pad_seq, 2)
    X_test_lstm_seq_inp = np.expand_dims(X_test_pad_seq, 2)

    X_train_lstm_inp = [X_train_lstm_seq_inp, X_train_mask]
    X_test_lstm_inp = [X_test_lstm_seq_inp, X_test_mask]
```

1 Model 1

```
[13]: ## as discussed above, please write the LSTM

class SpeechDetectionModel(Model):

    def __init__(self, units):
        super(SpeechDetectionModel, self).__init__()
        self.lstm = LSTM(units)
        self.dense = Dense(1, activation='relu')
        self.output_layer = Dense(10, activation='softmax')

    def call(self, inputs):
        sequences, mask = inputs
        x = self.lstm(sequences, mask=mask)
        x = self.dense(x)
        x = self.output_layer(x)

    return x
```

```
[14]: from sklearn.metrics import f1_score
from tensorflow.keras.callbacks import EarlyStopping, TensorBoard,

→ReduceLROnPlateau, Callback

class F1Callback(Callback):
```

```
def __init__(self, validation_data):
        super(F1Callback, self).__init__()
        self.validation_data = validation_data
      def on_epoch_end(self, epoch, logs={}):
        probs = self.model.predict(self.validation_data[0])
        preds = np.argmax(probs, axis=1)
        y_true = np.argmax(self.validation_data[1], axis=1)
        score = f1_score(y_true, preds, average='micro')
        logs['micro_f1_score'] = score
[]: f1_callback = F1Callback(validation_data=(X_test_lstm_inp, y_test))
    tensorboard = TensorBoard(write_images=True)
    callbacks = [
                 f1_callback,
                 tensorboard,
    ]
    model = SpeechDetectionModel(1)
    model.compile(optimizer='adam', loss='categorical_crossentropy')
[]: | #train your model
    !rm -rf ./logs/*
    model.fit(X_train_lstm_inp, y_train, epochs=3,
              validation_data=(X_test_lstm_inp, y_test),
              callbacks=callbacks)
    Epoch 1/3
    val_loss: 2.3026 - micro_f1_score: 0.1000
    Epoch 2/3
    44/44 [============ ] - 1686s 38s/step - loss: 2.3029 -
    val_loss: 2.3026 - micro_f1_score: 0.1133
    Epoch 3/3
    44/44 [=============== ] - 1677s 38s/step - loss: 2.3029 -
    val_loss: 2.3026 - micro_f1_score: 0.1000
    1.0.1 2. Converting into spectrogram and giving spectrogram data as input
[]: def convert_to_spectrogram(raw_data):
        '''converting to spectrogram'''
        spectrum = librosa.feature.melspectrogram(y=raw_data, sr=sample_rate,__
     \rightarrown_mels=64)
```

logmel_spectrum = librosa.power_to_db(S=spectrum, ref=np.max)

```
return logmel_spectrum

f make_spectrograms(data):
```

```
[]: def make_spectrograms(data):
    spectograms = []
    for i in data:
        spectograms.append(convert_to_spectrogram(i))
    return np.array(spectograms)
```

Grader function 6

[]: True

2 Model 2

```
[]: from tensorflow.keras import backend

class SpectrogramModel(Model):

    def __init__(self, units):
        super(SpectrogramModel, self).__init__()
        self.lstm_1 = LSTM(units, return_sequences=True)
        self.lstm_2 = LSTM(units, return_sequences=True)
        self.lstm_3 = LSTM(units, return_sequences=True)

        self.dense = Dense(64, activation='relu')
        self.output_layer = Dense(10, activation='softmax')

        def feature_average(self, sequences):
            return backend.mean(sequences, axis=2)

        def call(self, inputs):
            x = self.lstm_1(inputs)
```

```
x = self.lstm_2(x)
       x = self.lstm_3(x)
       x = self.feature_average(x)
       x = self.dense(x)
       x = self.output_layer(x)
       return x
[]: f1_callback = F1Callback(validation_data=(X_test_spectrogram, y_test))
    early_stopping = EarlyStopping(patience=5)
    tensorboard = TensorBoard(write_images=True)
    reduce_lr = ReduceLROnPlateau(patience=3)
    callbacks = [
               early_stopping,
               f1_callback,
               tensorboard,
               reduce_lr,
    ]
    model = SpectrogramModel(32)
    model.compile(optimizer='adam', loss='categorical_crossentropy')
[]: #train your model
    !rm -rf ./logs/*
    model.fit(X_train_spectrogram, y_train, epochs=100,
            validation_data=(X_test_spectrogram, y_test),
            callbacks=callbacks)
   Epoch 1/100
    2/44 [>...] - ETA: 10s - loss:
   2.2998WARNING:tensorflow:Method (on_train_batch_end) is slow compared to the
   batch update (0.211291). Check your callbacks.
   val_loss: 2.1142 - micro_f1_score: 0.2100 - lr: 0.0010
   Epoch 2/100
   1.8749 - micro_f1_score: 0.2983 - lr: 0.0010
   Epoch 3/100
   44/44 [==============] - 4s 86ms/step - loss: 1.7555 - val_loss:
   1.7060 - micro_f1_score: 0.3517 - lr: 0.0010
   Epoch 4/100
   44/44 [=============] - 4s 86ms/step - loss: 1.6187 - val_loss:
   1.5659 - micro_f1_score: 0.4283 - lr: 0.0010
   Epoch 5/100
```

```
1.4748 - micro_f1_score: 0.4317 - lr: 0.0010
Epoch 6/100
1.4971 - micro_f1_score: 0.4383 - lr: 0.0010
Epoch 7/100
1.2320 - micro_f1_score: 0.5567 - lr: 0.0010
Epoch 8/100
1.1170 - micro_f1_score: 0.5817 - lr: 0.0010
Epoch 9/100
1.1674 - micro_f1_score: 0.5467 - lr: 0.0010
Epoch 10/100
44/44 [=============] - 4s 88ms/step - loss: 0.9983 - val_loss:
1.0755 - micro_f1_score: 0.6300 - lr: 0.0010
Epoch 11/100
0.9762 - micro_f1_score: 0.6400 - lr: 0.0010
Epoch 12/100
1.0116 - micro_f1_score: 0.6117 - lr: 0.0010
Epoch 13/100
0.9672 - micro_f1_score: 0.6433 - lr: 0.0010
Epoch 14/100
0.9108 - micro_f1_score: 0.6667 - lr: 0.0010
0.8710 - micro_f1_score: 0.6683 - lr: 0.0010
Epoch 16/100
0.8163 - micro_f1_score: 0.7217 - lr: 0.0010
Epoch 17/100
0.8225 - micro_f1_score: 0.7017 - lr: 0.0010
Epoch 18/100
0.8036 - micro_f1_score: 0.7133 - lr: 0.0010
Epoch 19/100
0.8068 - micro_f1_score: 0.6967 - lr: 0.0010
Epoch 20/100
44/44 [=============] - 4s 88ms/step - loss: 0.6590 - val_loss:
0.7383 - micro_f1_score: 0.7367 - lr: 0.0010
Epoch 21/100
```

```
0.7602 - micro_f1_score: 0.7400 - lr: 0.0010
Epoch 22/100
0.7003 - micro_f1_score: 0.7517 - lr: 0.0010
Epoch 23/100
0.7776 - micro_f1_score: 0.7150 - lr: 0.0010
Epoch 24/100
0.7194 - micro_f1_score: 0.7367 - lr: 0.0010
Epoch 25/100
0.6845 - micro_f1_score: 0.7733 - lr: 0.0010
Epoch 26/100
0.7880 - micro_f1_score: 0.7283 - lr: 0.0010
Epoch 27/100
0.7566 - micro_f1_score: 0.7483 - lr: 0.0010
Epoch 28/100
0.6883 - micro_f1_score: 0.7733 - lr: 0.0010
Epoch 29/100
0.6182 - micro_f1_score: 0.7950 - lr: 1.0000e-04
Epoch 30/100
0.5989 - micro_f1_score: 0.7900 - lr: 1.0000e-04
0.5927 - micro_f1_score: 0.7983 - lr: 1.0000e-04
Epoch 32/100
0.5918 - micro_f1_score: 0.8000 - lr: 1.0000e-04
Epoch 33/100
0.5904 - micro_f1_score: 0.8033 - lr: 1.0000e-04
Epoch 34/100
0.5933 - micro_f1_score: 0.8033 - lr: 1.0000e-04
Epoch 35/100
44/44 [=============] - 4s 88ms/step - loss: 0.3716 - val_loss:
0.5934 - micro_f1_score: 0.8050 - lr: 1.0000e-04
Epoch 36/100
44/44 [=============] - 4s 90ms/step - loss: 0.3662 - val_loss:
0.5885 - micro_f1_score: 0.8133 - lr: 1.0000e-04
Epoch 37/100
```

```
0.5986 - micro_f1_score: 0.8000 - lr: 1.0000e-04
   Epoch 38/100
   0.5894 - micro_f1_score: 0.8100 - lr: 1.0000e-04
   Epoch 39/100
   0.5860 - micro_f1_score: 0.8083 - lr: 1.0000e-04
   Epoch 40/100
   0.5866 - micro_f1_score: 0.8067 - lr: 1.0000e-04
   Epoch 41/100
   0.5864 - micro_f1_score: 0.8117 - lr: 1.0000e-04
   Epoch 42/100
   0.5928 - micro_f1_score: 0.8117 - lr: 1.0000e-04
   Epoch 43/100
   0.5899 - micro_f1_score: 0.8117 - lr: 1.0000e-05
   Epoch 44/100
   0.5873 - micro_f1_score: 0.8100 - lr: 1.0000e-05
[]: <tensorflow.python.keras.callbacks.History at 0x7f0c90c6dcf8>
   2.0.1 3. data augmentation
[15]: ## 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)
```

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

→sr=sample_rate, n_steps=pitch_value)

return augmented_data

augmented_data.append(final_data)

final_data = librosa.effects.pitch_shift(time_stretch_data,__

```
[17]: len(aug_temp)
```

[17]: 9

As discussed above, for one data point, we will get 9 augmented data points. We have 2000 data points(train plus test) so, after augmentation we will get 18000 (train - 12600, test - 5400).

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

```
[18]: def augment_data(data):
        augmented_data, labels = [], []
        for i in range(len(data)):
          row = data.iloc[i]
          ret = generate_augmented_data(row['path'])
          augmented_data += [i for i in ret]
          labels += [row['label'] for _ in ret]
        return pd.DataFrame({
            'augmented_data': augmented_data,
            'labels': labels,
        })
[19]: %%time
      augmented_df = augment_data(df_audio)
     CPU times: user 6min 14s, sys: 1min 24s, total: 7min 38s
     Wall time: 11min 9s
[20]: X, y = augmented_df['augmented_data'], augmented_df['labels']
      X_train, X_test, y_train, y_test = train_test_split(X, y, stratify=y,_
       →random_state=45, test_size=0.3)
 []: cal_percentiles(X_train.map(len), 90, 101, 1)
     90 th percentile is 14336.0
     91 th percentile is 15360.0
     92 th percentile is 15360.0
     93 th percentile is 15872.0
     94 th percentile is 15872.0
     95 th percentile is 16384.0
     96 th percentile is 17408.0
     97 th percentile is 17920.0
     98 th percentile is 19456.0
     99 th percentile is 21504.0
     100 th percentile is 72192.0
     Over 98 percentile of data has len 20000 or below
[21]: MAX_SEQUENCE_LENGTH = 20000
```

```
X_train_pad_seq = pad_sequences(X_train, MAX_SEQUENCE_LENGTH, 'float32')
     X_test_pad_seq = pad_sequences(X_test, MAX_SEQUENCE_LENGTH, 'float32')
     X_train_mask = (X_train_pad_seq != 0.0)
     X_test_mask = (X_test_pad_seq != 0.0)
[22]: y_train = to_categorical(y_train, 10)
     y_test = to_categorical(y_test, 10)
[23]: X_train_lstm_seq_inp = np.expand_dims(X_train_pad_seq, 2)
     X_test_lstm_seq_inp = np.expand_dims(X_test_pad_seq, 2)
     X_train_lstm_inp = [X_train_lstm_seq_inp, X_train_mask]
     X_test_lstm_inp = [X_test_lstm_seq_inp, X_test_mask]
       Model 3
     3
[24]: f1_callback = F1Callback(validation_data=(X_test_lstm_inp, y_test))
     early_stopping = EarlyStopping(patience=5)
     tensorboard = TensorBoard(write_images=True)
     reduce_lr = ReduceLROnPlateau(patience=3)
     callbacks = [
                  f1_callback,
                  tensorboard,
     ]
     model = SpeechDetectionModel(1)
     model.compile(optimizer='adam', loss='categorical_crossentropy')
[25]: !rm -rf ./logs/*
     model.fit(X_train_lstm_inp, y_train, epochs=2,
               validation_data=(X_test_lstm_inp, y_test),
               callbacks=callbacks)
     Epoch 1/2
       1/394 [...] - ETA: Os - loss:
     2.3026WARNING:tensorflow:From /usr/local/lib/python3.6/dist-
     packages/tensorflow/python/ops/summary_ops_v2.py:1277: stop (from
     tensorflow.python.eager.profiler) is deprecated and will be removed after
     2020-07-01.
     Instructions for updating:
     use `tf.profiler.experimental.stop` instead.
```

val_loss: 2.3026

4 Model 4

```
val_loss: 0.6187 - micro_f1_score: 0.7748 - lr: 0.0010
Epoch 5/100
val_loss: 0.4959 - micro_f1_score: 0.8248 - lr: 0.0010
Epoch 6/100
394/394 [============== ] - 36s 92ms/step - loss: 0.4756 -
val_loss: 0.5461 - micro_f1_score: 0.8039 - lr: 0.0010
Epoch 7/100
394/394 [============= ] - 36s 92ms/step - loss: 0.4161 -
val_loss: 0.3773 - micro_f1_score: 0.8631 - lr: 0.0010
Epoch 8/100
394/394 [============= ] - 37s 93ms/step - loss: 0.3940 -
val_loss: 0.3569 - micro_f1_score: 0.8750 - lr: 0.0010
Epoch 9/100
val_loss: 0.3422 - micro_f1_score: 0.8761 - lr: 0.0010
Epoch 10/100
val_loss: 0.3032 - micro_f1_score: 0.8963 - lr: 0.0010
Epoch 11/100
val_loss: 0.2934 - micro_f1_score: 0.8933 - lr: 0.0010
Epoch 12/100
394/394 [============= ] - 37s 93ms/step - loss: 0.2960 -
val_loss: 0.3084 - micro_f1_score: 0.8881 - lr: 0.0010
Epoch 13/100
val_loss: 0.3160 - micro_f1_score: 0.8833 - lr: 0.0010
Epoch 14/100
val_loss: 0.3575 - micro_f1_score: 0.8711 - lr: 0.0010
```

[]: <tensorflow.python.keras.callbacks.History at 0x7f0c87e56128>