MULTIMODAL EMOTION RECOGNITION SYSTEM

A Major-Project Report Submitted to the JAWAHARLAL NEHRU TECHNOLOGICAL UNIVERSITY HYDERABAD in partial fulfillment of the requirements for the award of the degree of

BACHELOR OF TECHNOLOGY

IN

COMPUTER SCIENCE AND ENGINEERING

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This is to certify that the project work entitled "Multimodal Emotion Recognition Systems" submitted in VNR Vignana Jyothi Institute of Engineering & Technology in partial fulfilment of requirement for the award of Bachelor of Technology in Computer Science and Engineering is a bona fide report of the work carried out by us under the guidance and supervision of Bhanusree Yalamanchili (Asst Professor), Department of CSE, VNRVJIET. To the best of our knowledge, this report has not been submitted in any form to any university or institution for the award of any degree or diploma.

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We express our sincere thanks to our faculty of the department of Computer Science and Engineering and the remaining members of our college VNR VIGNANA JYOTHI INSTITUTE OF ENGINEERING

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ABSTRACT

The need for machines to react towards human emotions is required and is the need of the hour in the current digital era. The human-machine interaction (HCI) with responses to emotions will make them more useful and user friendly. So we developed a system which takes either audio or video as input and gives the status of emotion. We classified emotions into 7 categories like happy, sad, fearful, disgust, calm, angry and surprise. We used Ravdess dataset for speech using time distributed CNN which extracts MFCC, Mel-Scale, Spectrogram features. For video, Fer2013 dataset using CNN model with synthesized features like haar features, HOG sliding windows, HOG features and Facial landmarks are used. By applying these algorithms we have achieved an accuracy of about 96% on training and about 75% on validation when speech emotion model is used and about 89% accuracy on training and 76% on validation when face emotion model is used. Later, these modules can be integrated in interactive devices such as smartphones, smart wearables, voice assistants and other IoT devices.

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CHAPTER 1

INTRODUCTION

In today's world, the rapid growth of artificial intelligence has escalated the need for better and natural interaction between humans and machines. Emotion recognition systems can be deployed into diverse applications. The information obtained from these systems is being used in fields like health, education, tourism and commerce, etc. Unimodal systems cannot provide more information about the user and to various exterior factors. This has led to the development of multimodal systems rather than unimodal systems. The ways to recognize the emotions of users are asking from a user, Voice recognition, Tracking implicit parameters, Facial expression recognition, Gesture recognition, Vital signals, and Hybrid methods. Physiological features such as respiratory volume, skin temperature, heart rate, respiration pattern, EDA, PPG, and EMG can also be used to determine emotion. For multimodal systems to give accurate results using different types of data, fusion techniques are used. Various fusion techniques which can be used are feature-level, hybrid multimodal, decision- level, rule-based, classificationbased, model-level, and estimation-based fusions. The databases accumulated by researchers for these systems can contain only text, only audio, only image, only video, audio and video, physiological data or audio, text, and video data.

CHAPTER 2

EXISTING SYSTEM AND PROPOSED SYSTEM

2.1 EXISTING SYSTEM

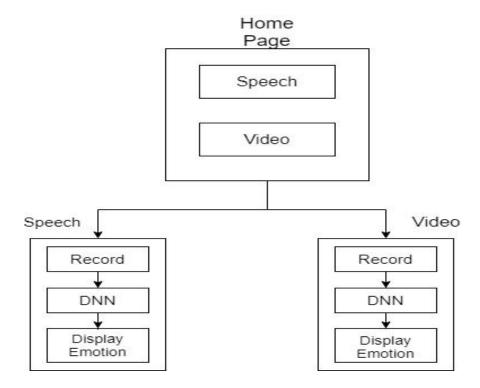
The emotion recognition systems developed till date try to percept information using data from facial expression, speech, text, gestures and other physiological features. The techniques such as signal processing, computer vision, machine learning, and speech processing are used. These techniques are classified into three categories- knowledge-based, statistical, and hybrid approaches. Knowledge based techniques use semantics and syntax of languages to determine the emotion. They can be dictionary based or corpus based. Statistical methods commonly use supervised machine learning algorithms to predict emotions and give more accurate results. Deep learning algorithms such as CNN and LSTM are also used in these methods. Hybrid approaches are a combination of knowledge based and statistical techniques. The computation in hybrid approaches is more complex which is the reason for its scarce usage.

The unimodal systems use only one type of data such as audio, video or text. These methods do not yield accurate results when there are external disturbances. The multimodal systems being used generally use audio, video and text data to extract features. The features extracted are then fused to predict emotions. While these methods give more accurate results, they need to access all types of data in order to deduce emotions. This may raise privacy concerns. If even one of the types of data is missing, the fusion is not possible and therefore results cannot be given.

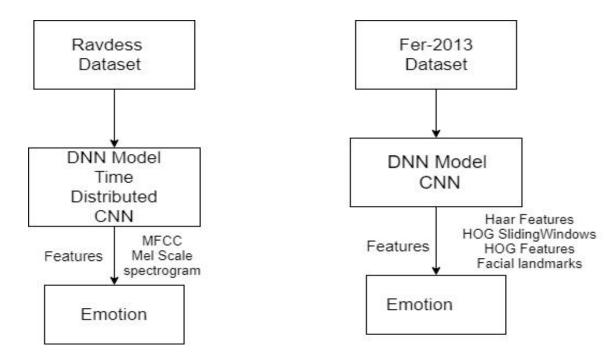
2.2 PROPOSED SOLUTION:

The proposed system tries to solve these issues in emotion recognition systems. The users are given a choice for what type of data they want to share in order to predict emotions. This resolves the privacy concerns and even if one type of data is not available, other types of data can be used. This ensures the working of the system under all conditions.

Proposed System:



Proposed Neural Network Architecture:



CHAPTER 3

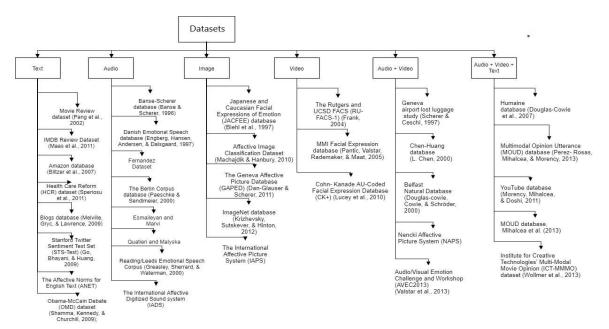
FEASIBILITY STUDY

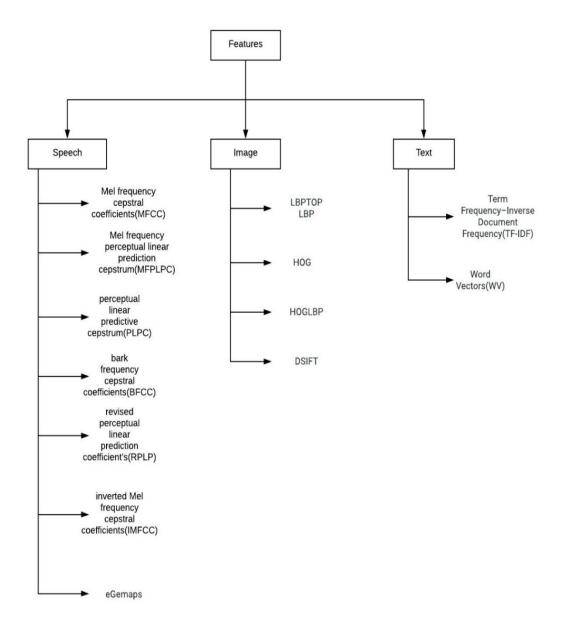
3.1 Description:

This project aims to recognize emotion with different forms of input like speech and facial expressions. By introducing this system in various applications we can find out the emotional status of the person using those applications which will be helpful in many ways.

3.2 Possible Solutions:

Few of them worked on individual inputs and few of them on multiple modes of inputs using different algorithms like RNN (Recurrent neural network), CNN (Convolutional neural network), HMM (Hidden Markov Models), SVMs (Support Vector Machines) etc. They used different features like MFCC, spectral etc. They even used different datasets like Fermandez, Berlin for audio and Fer2013, MMI Facial Expression database etc. for video.





3.3 Most Feasible Solution:

After reviewing all the possible solutions and the efficiencies achieved from different solutions, we came up with a solution of using Ravdess dataset for speech using time distributed CNN which extracts MFCC, Mel-Scale, Spectrogram features. For video Fer2013 dataset using CNN model with synthesized features like "HOG Features, Hybrid Features, Inception, Xception, DeXpression".

3.4 Conclusion:

So by using the above solution, we will develop a model with better accuracy than the previous models. After development, this model can be integrated in different applications where emotion is recognition is required.

CHAPTER 4

SYSTEM ANALYSIS

4.1. System Requirements

4.1.1. Software Requirements

4.1.1.1. Data sets

Either for synthesis or recognition of emotion by speech or face, we need emotional database for both speech and face. A single actor is not sufficient for recognizing emotions, we need multiple actors to identity emotions and also various styles of expressing the emotions.

While considering database, these are the other factors to be considered. They are:

- size of database
- Speaker/Actor
- gender
- Language (in case of speech system)

A) Speech Emotion Recognition:

In case of Speech Emotion Recognition there are three different methods to collect the data. They are:

- Simulated Databases
- Elicited Databases
- Naturalistic Databases

"Simulated databases are one of major resource of database where actors are asked to portray the given emotion".

"Elicited database are recorded by involving people in particular situation for extracting particular emotion".

"Naturalistic databases are recorded from our daily life activities".

There are various data sets for speech emotion recognition. They are: "Ravdess Emotional Speech data set, Emo-DB or Berlin dataset, IEMOCAP" etc., In this project we used "Ravdess data set for Speech Emotion Recognition".

RAVDESS Emotional Speech Audio:

"Ryerson Audio-Visual Database of Emotional Speech and Song" consists of 1440 audio files in .wav form. It contains 24 professional actors i.e., 12 male and 12 female actors. So, "60 trials per each actor x 24 actors constitutes 1440 audio files". We can derive seven emotions like happy, sad, fearful, disgust, calm, angry and surprise expressions from this Ravdess data set. Among 1440 files, each file has its unique name which represents various identifiers of the audio file. They are: "Modality, Vocal Channel, Emotion, Emotional Intensity, Statement, Repetition, Actor".

B) Face Emotion Recognition:

For Face Emotion Recognition there are many ways to collect the face data. They are:

- Take photographs manually
- Using a program to generate the pictures automatically
- Searching on the internet to collect similar kind of images
- Various data augmentation techniques like flipping the image, rotating the image, cropping the image etc.,

There are many data sets for "emotion recognition using facial expressions". They are:

"Fer-2013, IEMOCAP, FERET dataset" etc.,

FER- 2013 ("Facial Expression Recognition"):

FER data set contains gray scale images of face data with pixel value of 48x48 pixels. There are two .csv files (train and test) in this data set. The train.csv file consists of two

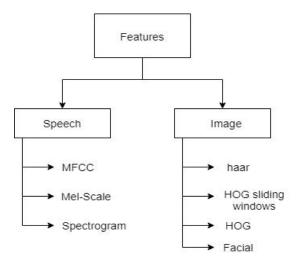
columns, "emotion and pixels". The test.csv file consists of only only one column i.e., "pixles column". The face data used for training comprises of "28,709 examples". The face data for testing contains around "3,589 examples". We can derive seven emotions like happy, sad, fearful, disgust, calm, angry and surprise expressions from this dataset.

Table 4.1: Datasets

Dataset	Audio	Video	Text
IEMOCAP	√	✓	√
CMU-MOSI	√	/	/
AFEW	√	V	√
Fernandez	√	×	×
RECOLA	√	√	×
eNterface	√	√	×
Berlin	√	×	×
Ravdess	✓	×	×
Fer2013	×	√	×

4.1.1.2. Features:

Feature Extraction plays a vital role in emotion recognition models. In this project we extracted various features for both "Speech Emotion Recognition" and "Emotion Recognition". For SER, the "Time-distributed CNN" extracts the features like "MFCC, Mel-Scale, Spectrogram" etc., For FER, the CNN synthesized features like "HOG Features, Hybrid Features, Inception, Xception, DeXpression" etc.,



A) Mel Frequency Cepstral Coefficients (MFCC):

The initial phase in a synthesizing features of SER is to recognize the segments of sound signal that are useful distinguishing the language specialist content. The second step is to remove the background noise and cut the .wav file to particular milli seconds.

"Mel Frequency Cepstral Coefficient" (MFCC) are an element generally utilized in programmed discourse and speaker acknowledgment. They were presented by Davis and Mermelstein in the 1980's, and have been best in class from that point onwards. Preceding the presentation of MFCCS "Linear Expectation Coefficients" (LPCs) and Linear Prediction Cepstral Coefficients (LPCC) and were the fundamental element type for "Automatic Speech Emotion Recognition (ASR)".

Steps:

- Frame the signal into short frames.
- For each frame compute the periodogram estimate of the power spectrum.
- Apply the Mel filterbank to the power or force spectra, entirety the vitality in each channel.
- Take the logarithm of all filterbank energies.
- Take the DCT of the log filterbank energies.
- Keep DCT coefficients 2-13, dispose of the rest.

B) Mel Scale:

The Mel scale relates apparent recurrence, or pitch of an unadulterated tone to its real estimated recurrence. People are greatly improved at observing little changes in pitch low frequencies than they are at high frequencies. Consolidating this scale makes our features coordinate intimately with human's ear.

The formula for changing over from recurrence to Mel scale is:

$$M(f) = 1125 In(1 + f/700)$$
 (1)

To go from Mels back to frequency:

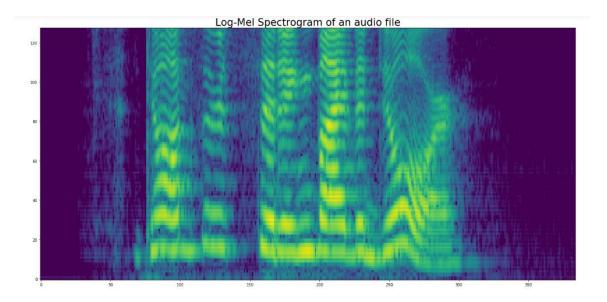
$$M-1(m) = 700(\exp(m/1126) - 1)$$
 (2)

C) Spectrogram:

A spectrogram of the discourse signal is a 2D portrayal of the frequencies as for time, that have more data than content translation words for perceiving the feelings of a speaker.

A typical arrangement is a chart with two geometric measurements: one dimension(x-axis) speaks to time, and the other dimension(y-axis) speaks to recurrence; a third measurement(z-axis) demonstrating the abundancy of a specific recurrence at a specific time is spoken to by the power or shade of each point in the picture.

Spectrograms are utilized broadly in the fields of music, sonar, radar, and audio processing, seismology, and others.



D) Haar features:

Therefore, a common Haar feature for face detection is a set of two adjacent rectangles that lie above the eye and the cheek region. The position of these rectangles is defined relative to a detection window that acts like a bounding box to the target object (the face in this case).

E) HOG:

Histogram of Oriented Gradients Feature changes over a picture of size width x tallness x 3 (channels) to an element vector/cluster of length n. On account of the HOG descriptor, the information picture is of size 64 x 128 x 3 and the yield include vector is of length 3780.

F) Facial:

Facial expressions are a universal language of emotion, instantly conveying happiness, sadness, anger, fear, and much more. Reading these expressions is essential to compassion and empathy. Some emotions appear more than once.

Other Features:

A smile on human face shows their bliss and it communicates eye with a bended shape. The sad articulation is the inclination of detachment which is ordinarily communicated as rising slanted eyebrows and frown. The anger on human face is identified with terrible and disturbing conditions. The outflow of annoyance is communicated with pressed eyebrows, thin and extended eyelids. The disgust expressions are communicated with pull down eyebrows and wrinkled nose. The suprise or stun articulation is communicated when some unpredicted occurs. This is communicated with eye-extending and mouth expanding and this expression is an effortlessly recognized one. The statement of fearfulness is connected with shock expression which is communicated as developing slanted eyebrows.

4.1.1.3. Deep Neural Network Models:

In general, there are two methodologies for "Multimodal Emotion Recognition Systems". In the first method, the extracted features are utilized to teach six AI classifiers namely "Support Vector Machines, Random Forest Classifier, Gradient Boosting Classifier, Logistic Regression" etc., while the subsequent methodology depends on deep learning. Emotion Recognition through audio signals using Machine learning classifiers has its own restrictions like, Language, accent dependent. Deep networks may conquer these restrictions. Deep neural networks collect the useful features from the Multimodal data by itself instead of passing the features explicitly to the model.

In the present research findings on Multimodal Emotion Recognition, several works are implemented using various deep neural networks like Recurrent Neural Networks (RNN), Multi-layer perceptron, Long-Short Term Memory Classifier, Convolutional Neural Networks, Time-distributed Neural Networks etc., However, there are few works identified to Hierarchical fusion by combination of one or more modalities like speech+video,text+audio, text+video, audio+video+text etc., In this project, we implemented Time-distributed Neural Networks for Speech Emotion Recognition and Convolutional Neural Networks for Face Emotion Recognition.

For Audio Emotion Recognition, the principle thought of a Time Dispersed Convolutional Neural System is to apply a moving window (fixed size and time-step) up and down the log-mel-spectrogram. Each one of these windows will be the section of a convolutional neural system, created by Four Local Feature Learning Blocks (LFLBs) and the yield of each of these convolutional systems will be taken care of into a repetitive neural system formed by 2 cells of "LSTM" to learn long-term contextual dependencies.

At last, a completely associated layer with softmax activation is utilized to predict the feeling recognized in the voice.

For Facial Emotion Recognition, the aim of using Convolutional Neural Networks is to diminish the pictures into a structure which is simpler to process, without losing features which are basic for getting a decent prediction. We implemented XCeption model to reduce overfitting as much as possible to build a robust model. The XCeption architecture depends on DepthWise Separable convolutions that permit to prepare many less parameters, and in this manner decrease training time on Colab's GPUs to under an hour and a half.

Table 4.2: Models

Models	Datasets	Data	Accuracy
RNN	IEMOCAP	Audio, Video	71.8%
CNN	CMU-MOSI	Audio, Video	49.17%
CNN+LSTM	AFEW	Audio, Video	60.34%
RNN	RECOLA	Audio, Video	76%
CNN+LSTM	eNterface	Audio, Video	90.85%
CNN+RNN	Berlin	Audio	88.9%
RNN	Ravdess	Audio	89.02%
CNN	Fer2013	Video	82.19%

4.1.1.4. Libraries:

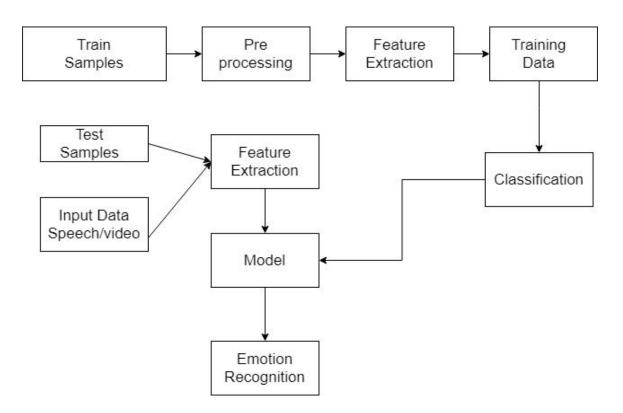
In this project we used several python and machine learning libraries like Librosa, Numpy, Pandas, Soundfile, Pyaudio, Scikit-learn, Tensorflow, Keras, Pytorch, matplotlib etc., for implementing deep neural networks.

4.1.2. Hardware Requirements:

• Processor – i5

- Hard Disk 5 GB
- Memory 8 GB RAM
- Nvidia (GPU)

4.1.3. System Architecture:



The proposed human emotion recognition system is of five components:

- 1. input data
- 2. pre-processing techniques
- 3. feature extraction and selection
- 4. classification
- 5. emotion recognition

We divide the dataset in the ration of 80:20 (80=training set & 20=testing set). On

training dataset pre-processing techniques are applied. From the pre-processed speech signal, we extract the useful features required for emotion recognition.

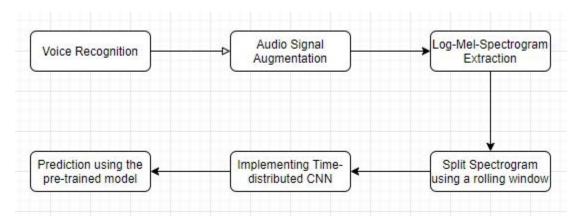
In the next step, Deep Neural Network algorithms like Time-distributed CNN and CNN are implemented on the data. Hence the Emotion Recognition Model is created in .hdf5 or .pickle format.

The input signal is recorded and we synthesize the features which are useful for emotion recognition. The recorded data is given as input for the emotion recognition model. In the final stage, one of the 7 emotions is identified and displayed on the webpage.

4.1.4. Methodology:

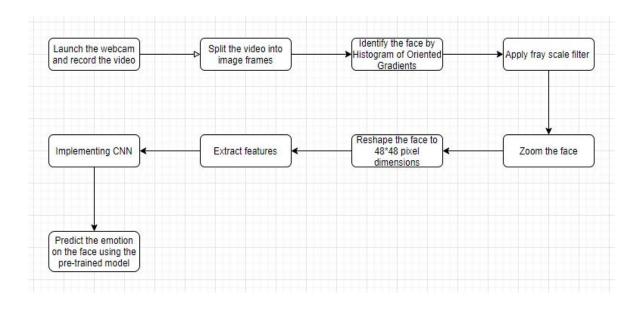
4.1.4.1. Speech Emotion Recognition:

The Speech Emotion Recognition methodology pipeline is built in the following way:



4.1.4.2. Face Emotion Recognition:

The Face Emotion Recognition methodology Pipeline is built in the following way:



CHAPTER 5

SOFTWARE DESIGN

5.1 UML Diagrams

Unified Modelling Language is a tool that assists a designer to present his thoughts about the task to his customer and his designer. Modelling theaters a vital role in scheming a software. A sick designed model can lead to a poorly settled software.

A UML system has using five diverse views that support in labelling systems from diverse outlooks. Each view has a set of diagrams that and modules that represent the real time objects.

a. User Model View:

- i. It models the operator behavior in a organization context.
- ii. All the diagrams are drawn keeping in mind the user's response and reaction towards a system.

b. Structural Model View

- This vision contains of class and object diagram which is used to model the static structures.
- ii. It uses objects, operations, attributes and relationships.

c. Behavioral Model View

- i. It largely contains the sequence diagram, collaboration diagram, state chart diagram and activity diagram. They mostly symbolize flow of movements among different objects tangled in the system.
- ii. They are used to envision numerous lively aspects of the system architecture.

d. Implementation Model View

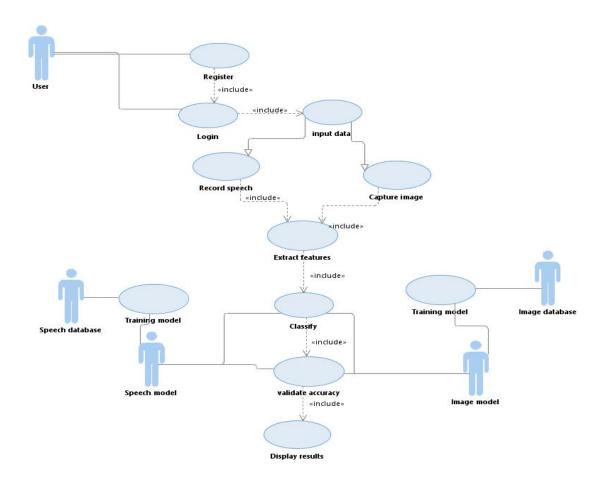
- a. This outlook consists of component diagrams and deployment diagrams. This view models the static software modules for an organization.
- b. This usually contains the data files, the executables, documentation and source code.
- c. These are the materially useable components of the system. They are modelled using component diagrams.

5.1.1. Use Case Diagram

The simple representation for the collaboration of the user with the system is symbolized by the use case diagram. It includes the connection between the user and many use cases with the actors being involved. There are diverse kind of relations that are involved among the use cases and the actors. They include:

- a. Association relationship
- b. Generalization
- c. Dependency
- d. Realizations
- e. Transitions

The following represents the use case diagram of the proposed system:



5.1.2. Class Diagram

They are stationary representation of an solicitation. Lone the class diagrams have the ability to be straight mapped with the OOP Languages since in OOPs the whole thing is model in the usage of classes and objects. As of this intention these diagrams are used extensively at the time of building. This is one of the most commonly used UML diagram in the exclusive community. A class diagram plays an crucial role in forward and reverse engineering.

- a. It deeds as a base for the component and deployment diagrams.
- b. It mostly defines and expresses the simple tasks of a system's application.
- c. It gears the scrutiny and design view for a static application.

In a class diagram, every object is modelled as a class. Each class comprises of

section or compartments.

- 1. Class name
- 2. Attributes of a class or operations
- 3. Methods or functions
- 4. Documentation (optional section)

The subsequent points have to to be summon up while drawing a class diagram:

- a. The labels of the class diagram need to be significant to depict the feature of the framework.
- b. Every component and their relations must be illustrious gaining of time.
- c. Every class has a accountability (attributes and methods) that need to be recognized openly.
- d. Total properties for all classes need to be minimum. Subsequently useless properties will create the diagram intricate.
- e. At all opinion required to show some part of the diagram use notes Since to the completion of the diagram it should be reasonable to the designer/coder.
- f. Afore concluding the previous version, the diagram must be drawn on normal paper and review whatever figure conditions as would be sensible to make it amends.

1. Scopes:

The UML diagrams have two types of possibilities for class members:

- i. instance members scope and
- ii. classifier members scope
- **2. Classifier members** are "static" members of a class in various programming languages. The scope is the class itself.

- i. Static attributes are mutual to all additional objects that raise the class.
- ii. Static methods are not instantiated.
- **3. Instance members** are the members that are local to an object.
 - i. The key purpose of instance members is to allow the objects to store their states.
 - ii. Statements exterior the methods are generally famous as instance members.

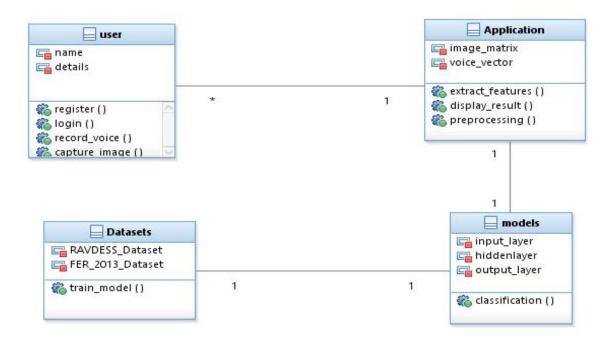


Fig 5.2: Class Diagram for Developed Model

5.1.3. Sequence Diagram

The Sequence Diagram portrays the time series amid many objects in an application. It shows the series of communications with which objects connect with each other so that they carry out the requisite functionality.

It contains of the salvations which are generally parallel vertical lines. It comprises of horizontal arrows which specifies the route of the communications that are swapped in a right order which makes the user cool to understand.

The lifeline for a given object represents a role. The synchronous calls are represented with the help of a solid arrow head whereas the asynchronous messages are represented with the help of open arrow heads.

All objects are represented according to their time ordering. Timing of messages plays a major role in sequence diagrams. An object is killed immediately after its use in sequence diagrams.

I). Common Properties:

An arrangement graph is much the same as unique sort of diagram and offers some indistinguishable properties from other diagrams. In any case, it varies from every single other diagram in its content.

II). Contents

Objects are normally named or unknown instances of class, however may likewise speak to occurrences of different things, for example components, collaboration and nodes. Graphically, object is represented as a rectangle by underlying its name.

III). Links

A link is a semantic association among objects i.e., an object of an affiliation is called as a connection. It is represented as a line.

IV). Messages

A message is a determination of a correspondence between objects that passes on the data with the desire that the action will follow.

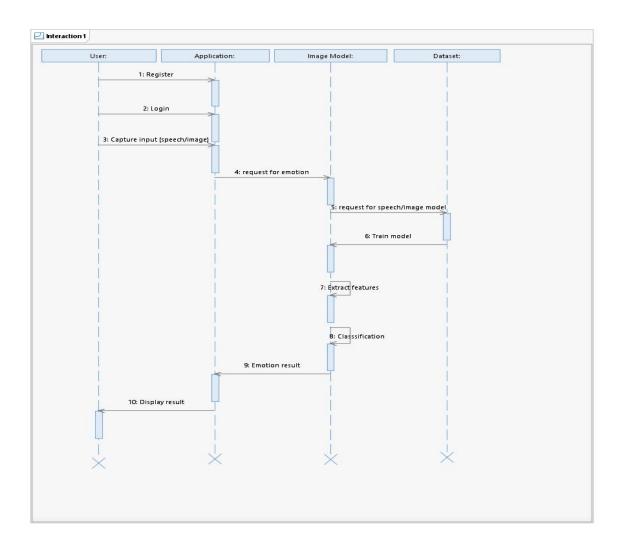


Fig 5.3 Sequence Diagram for Developed Model

5.1.4. Activity Diagram

The flow from one activity to another activity can be represented in the form of a flow chart which is usually an activity diagram. It forms a backbone for the UML diagrams. It depicts the dynamic aspects for all the objects within the system.

The control flow from one object to another object is drawn which shows the basic operations that are to be performed.

Activity diagrams are constructed using the following:

1. Actions are represented using rounded rectangles;

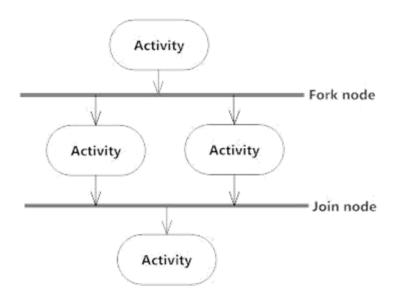


2. decisions are represented using diamonds;



3. concurrent activities bars are represented using the start (split) or end (join);

Synchronization



4. Time event is represented as



5. final state is represented using encircled black circle.



The basic purpose of an activity diagram is same as that of other UML

diagrams. The dynamic behavior of the system is viewed by the activity diagram. They are used to construct a system using the backward and forward engineering mechanisms.

The purpose of an activity diagram is as follows:

- 1) For drawing the flow (i.e. activity) in a system.
- 2) For showing the flow of sequence from one activity to another activity.
 - 1) For showing the concurrent and parallel flow of actions in the system.

 The elements that are used in an activity diagram are as follows:
 - i. Association Relationship
 - ii. Activities
 - iii. Conditions and Constraints.

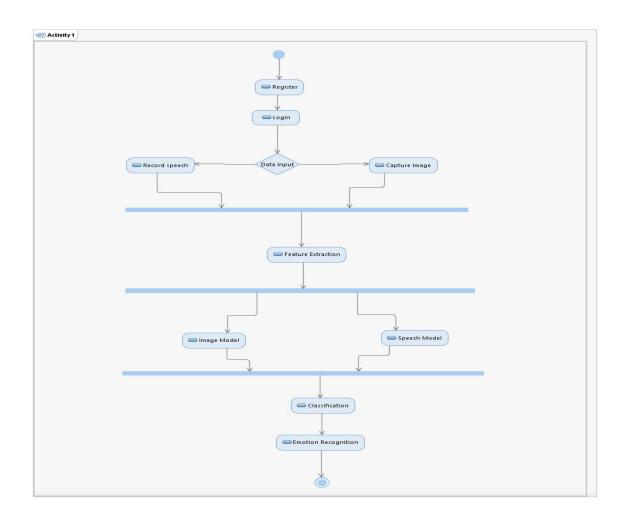


Fig 5.4 Activity Diagram

CHAPTER 6

IMPLEMENTATION

6.1. Source Code and Outputs

6.1.1. Speech Emotion Recognition

6.1.1.1. Signal Processing

The Signal Processing includes:

- Signal Discretization
- Audio Data Augmentation
- Log-mel-spectrogram extraction
- Time distributes framing
- Build train and test data set

General imports

import os

from glob import glob

import pickle

import itertools

import numpy as np

```
from scipy.stats import zscore
from sklearn.model_selection import train_test_split
# Graph imports
import matplotlib.pyplot as plt
from PIL import Image
# Audio imports
import librosa
import IPython
from IPython.display import Audio
#Set Labels
# RAVDESS Database
label dict ravdess = {'02': 'NEU', '03':'HAP', '04':'SAD', '05':'ANG', '06':'FEA', '07':'DIS',
'08':'SUR'}
# Set audio files labels
def set_label_ravdess(audio_file, gender_differentiation):
  label = label dict ravdess.get(audio file[6:-16])
  if gender differentiation == True:
    if int(audio file[18:-4])%2 == 0: # Female
       label = 'f' + label
```

```
if int(audio_file[18:-4])%2 == 1: # Male
     label = 'm' + label
return label
#Import Audio Files
# Start feature extraction
print("Import Data: START")
# Audio file path and names
#file path = '../Datas/Ravdess/'
#file names = os.listdir(file path)
#print(file_path)
#print(file names)
audio filepath="
audio fileslist=[]
#for i in range(1,25):
  #audio_filepath='../Datas/Ravdess/Actor_'+str(i)+'/'
  #print(audio filepath)
  #audio fileslist=os.listdir(audio filepath)
  #for i in audio_fileslist:
     #print(i)
#print(audio fileslist)
# Initialize features and labels list
```

```
signal = []
labels = []
# Sample rate (16.0 kHz)
sample rate = 16000
# Max pad lenghth (3.0 sec)
max pad len = 49100
# Compute spectogram for all audio file
for i in range(1,25):
  audio_filepath='../Datas/Ravdess/Actor_'+str(i)+'/'
  audio fileslist=os.listdir(audio filepath)
  for audio_index, audio_file in enumerate(audio_fileslist):
  #print(audio_file)
     #print(audio index)
     if audio file[6:-16] in list(label dict ravdess.keys()):
     #print(audio file[6:-16])
     # Read audio file
       y, sr = librosa.core.load(audio_filepath + audio_file, sr=sample_rate, offset=0.5)
     # Z-normalization
       y = zscore(y)
     # Padding or truncated signal
       if len(y) < max_pad_len:
          y padded = np.zeros(max pad len)
          y \text{ padded}[:len(y)] = y
```

```
y = y_padded
       elif len(y) > max_pad_len:
         y = np.asarray(y[:max pad len])
     # Add to signal list
       signal.append(y)
     # Set label
       labels.append(set label ravdess(audio file,False))
       #print(audio index)
     # Print running...
       if (audio index \% 100 == 0):
         print("Import Data: RUNNING ... {} files".format(audio_index))
# Cast labels to array
labels = np.asarray(labels).ravel()
#print(labels)
#print(signal)
# Stop feature extraction
print("Import Data: END \n")
print("Number of audio files imported: {}".format(labels.shape[0]))
# Select one random audio file
X=[]
print(len(labels))
```

```
random idx = np.random.randint(len(labels))
  print(random idx)
  random label = labels[random idx]
  print(random label)
  random signal = signal[random idx]
  print(random signal)
  for i in range(1,25):
    audio filepath='../Datas/Ravdess/Actor '+str(i)+'/'
    audio fileslist=os.listdir(audio filepath)
    #print(audio_fileslist)
    x=x+audio fileslist
    #print(x)
  random filename = x[random idx]
  print(random filename)
  # Plot signal wave
  plt.figure(figsize=(10,5))
  plt.plot(np.arange(len(random signal))/float(sample rate), random signal)
  plt.xlim((np.arange(len(random signal))/float(sample rate))[0], (np.arange(len(rando
m signal))/float(sample rate))[-1])
  plt.xlabel('Time (s)', fontsize=16)
  plt.ylabel('Amplitude (dB)', fontsize=16)
  plt.title("Signal wave of file '{}' with label {}".format(random_filename, random_label)
```

```
, fontsize=18)
  plt.show()
  # Play audio file
  print("Audio file '{}':".format(random filename))
  Audio(random signal, rate=sample rate)
  #Audio Data Augmentation
  # Number of augmented data
  nb augmented = 2
  # Function to add noise to a signals with a desired Signal Noise ratio (SNR)
  def noisy signal(signal, snr low=15, snr high=30, nb augmented=2):
    # Signal length
    signal len = len(signal)
    # Generate White noise
    noise = np.random.normal(size=(nb_augmented, signal_len))
    # Compute signal and noise power
    s_power = np.sum((signal / (2.0 ** 15)) ** 2) / signal_len
    n_power = np.sum((noise / (2.0 ** 15)) ** 2, axis=1) / signal_len
    # Random SNR: Uniform [15, 30]
    snr = np.random.randint(snr low, snr high)
```

```
# Compute K coeff for each noise
    K = np.sqrt((s power / n power) * 10 ** (- snr / 10))
    K = np.ones((signal len, nb augmented)) * K
    # Generate noisy signal
    return signal + K.T * noise
  # Generate noisy signals from signal list
  print("Data Augmentation: START")
  augmented signal = list(map(noisy signal, signal))
  print("Data Augmentation: END!")
  # Plot signal wave
  plt.figure(figsize=(20,5))
  plt.subplot(1,2,1)
  plt.plot(np.arange(len(random signal))/float(sample rate), random signal)
  plt.xlim((np.arange(len(random signal))/float(sample rate))[0], (np.arange(len(rando
m signal))/float(sample rate))[-1])
  plt.xlabel('Time (s)', fontsize=16)
  plt.ylabel('Amplitude (dB)', fontsize=16)
  plt.title("Signal wave of file '{}' ".format(random filename), fontsize=18)
  # Plot signal wave with noise
  plt.subplot(1,2,2)
  plt.plot(np.arange(len(random signal))/float(sample rate), augmented signal[random
```

```
idx][0]
  plt.xlim((np.arange(len(random signal))/float(sample rate))[0], (np.arange(len(rando
m signal))/float(sample rate))[-1])
  plt.xlabel('Time (s)', fontsize=16)
  plt.ylabel('Amplitude (dB)', fontsize=16)
  plt.title("Signal wave of file '{}' with Noise".format(random filename), fontsize=18)
  plt.show()
  # Play audio file
  print("Audio file '{}':".format(random filename))
  IPython.display.display(Audio(random signal, rate=sample rate))
  # Play same audio file with noise
  print("Audio file '{}' with noise:".format(random filename))
  IPython.display.display(Audio(augmented signal[random idx][0], rate=sample rate))
  #Feature Extraction
  def mel spectrogram(y, sr=16000, n fft=512, win length=256, hop length=128, wind
ow='hamming', n mels=128, fmax=4000):
    # Compute spectogram
    mel_spect = np.abs(librosa.stft(y, n_fft=n_fft, window=window, win_length=win_le
ngth, hop length=hop length)) ** 2
```

```
# Compute mel spectrogram
    mel spect = librosa.feature.melspectrogram(S=mel spect, sr=sr, n mels=n mels, f
max=fmax)
    # Compute log-mel spectrogram
    mel spect = librosa.power to db(mel spect, ref=np.max)
    return mel spect
  # Start feature extraction
  print("Feature extraction: START")
  # Compute spectogram for all audio file
  mel spect = np.asarray(list(map(mel spectrogram, signal)))
  augmented mel spect = [np.asarray(list(map(mel spectrogram, augmented signal[i])))
for i in range(len(augmented signal))]
  # Stop feature extraction
  print("Feature extraction: END!")
  # Plot one random Spectogram
  plt.figure(figsize=(20, 10))
  plt.imshow(mel spect[np.random.randint(len(mel spect))], origin='lower', aspect='aut
o', cmap='viridis')
  plt.title('Log-Mel Spectrogram of an audio file', fontsize=26)
  plt.tight layout()
  plt.show()
```

```
#Train and Test set
  # Build Train and test dataset
  MEL SPECT train, MEL SPECT test, AUG MEL SPECT train, AUG MEL SPE
CT_test, label_train, label_test = train_test_split(mel_spect, augmented_mel_spect, labels,
test size=0.2)
  # Build augmented labels and train
  aug label train = np.asarray(list(itertools.chain.from iterable([[label] * nb augmented
for label in label train])))
  AUG MEL SPECT train = np.asarray(list(itertools.chain.from iterable(AUG MEL
SPECT_train)))
  # Concatenate original and augmented
  X_train = np.concatenate((MEL_SPECT_train, AUG_MEL_SPECT_train))
  y_train = np.concatenate((label_train, aug_label_train))
  # Build test set
  X_{test} = MEL_{SPECT_{test}}
  y test = label test
  # Delete
  del MEL SPECT train, AUG MEL SPECT train, label train, aug label train, AUG
MEL SPECT test, MEL SPECT test, label test
```

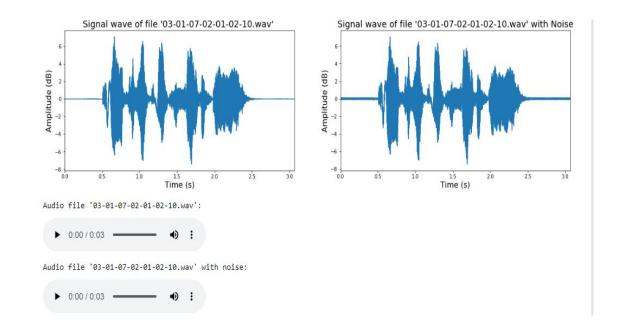
```
del mel spect, augmented mel spect, labels
  #Time distributed Framing
  # Time distributed parameters
  win ts = 128
  hop ts = 64
  # Split spectrogram into frames
  def frame(x, win step=128, win size=64):
    nb frames = 1 + int((x.shape[2] - win size) / win step)
    frames = np.zeros((x.shape[0], nb frames, x.shape[1], win size)).astype(np.float32)
    for t in range(nb frames):
       frames[:,t,:,:] = np.copy(x[:,:,(t * win step):(t * win step + win size)]).astype(np.
float32)
    return frames
  # Frame for TimeDistributed model
  X train = frame(X train, hop ts, win ts)
  X test = frame(X_test, hop_ts, win_ts)
  #Saving
  # Save Train and test set
  pickle.dump(X train.astype(np.float16), open('../Datas/Pickle/RAVDESS/[RAVDESS]
[MEL SPECT][X train].p', 'wb'))
  pickle.dump(y train, open('../Datas/Pickle/RAVDESS/[RAVDESS][MEL SPECT][y
```

train].p', 'wb'))

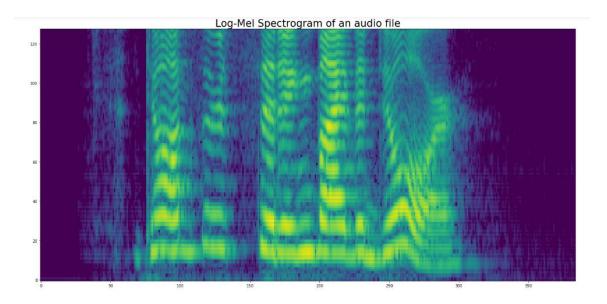
pickle.dump(X_test.astype(np.float16), open('../Datas/Pickle/RAVDESS/[RAVDESS][MEL_SPECT][X_test].p', 'wb'))

pickle.dump(y_test, open('../Datas/Pickle/RAVDESS/[RAVDESS][MEL_SPECT][y_t est].p', 'wb'))

Output of Data Augmentation:



Output of Spectrogram:



6.1.1.2. Time Distributed Neural Network

#General Imports

import os

from glob import glob

import pickle

import numpy as np

Plot imports

from IPython.display import Image

import matplotlib.pyplot as plt

Time Distributed ConvNet imports

import tensorflow as tf

from tensorflow.keras.models import Sequential, Model

from tensorflow.keras.layers import Input, Dense, Dropout, Activation, TimeDistributed, concatenate

from tensorflow.keras.layers import Conv2D, MaxPooling2D, AveragePooling2D, Batch

Normalization, LeakyReLU, Flatten

from tensorflow.keras.layers import LSTM

from tensorflow.keras.optimizers import Adam, SGD

from tensorflow.keras.callbacks import EarlyStopping, ModelCheckpoint, ReduceLROn Plateau

from tensorflow.keras import backend as K

from keras.utils import np utils

from keras.utils import plot model

from sklearn.preprocessing import LabelEncoder

from tensorflow.keras.utils import plot_model

#Warning imports

import warnings

warnings.filterwarnings('ignore')

Connect Colab to google drive

from google.colab import drive

drive.mount('/content/drive')

#Import datas

RAVDESS mel-Spectrogram

X_train = pickle.load(open('drive/My Drive/SpeechEmotionRecognition/[RAVDESS][M
EL_SPECT][X_train].p', 'rb'))

y_train = pickle.load(open('drive/My Drive/SpeechEmotionRecognition/[RAVDESS][M
EL_SPECT][y_train].p', 'rb'))

```
y test = pickle.load(open('drive/My Drive/SpeechEmotionRecognition/[RAVDESS][ME
L SPECT][y test].p', 'rb'))
X test = pickle.load(open('drive/My Drive/SpeechEmotionRecognition/[RAVDESS][ME
L SPECT][X test].p', 'rb'))
#Encode Label
# Encode Label from categorical to numerical
lb = LabelEncoder()
y train = np utils.to categorical(lb.fit transform(np.ravel(y train)))
y test = np utils.to categorical(lb.transform(np.ravel(y test)))
#Reshape train and test set
# Reshape for convolution
X train = X train.reshape(X train.shape[0], X train.shape[1], X train.shape[2], X train.
shape[3], 1)
X \text{ test} = X \text{ test.reshape}(X \text{ test.shape}[0], X \text{ test.shape}[1], X \text{ test.shape}[2], X \text{ test.shape}[2]
[3], 1)
#Time-distributed Convolutional Neural Network Model
K.clear session()
# Define two sets of inputs: MFCC and FBANK
input y = Input(shape=X train.shape[1:], name='Input MELSPECT')
## First LFLB (local feature learning block)
y = TimeDistributed(Conv2D(64, kernel size=(3, 3), strides=(1, 1), padding='same'), na
me='Conv_1_MELSPECT')(input y)
y = TimeDistributed(BatchNormalization(), name='BatchNorm 1 MELSPECT')(y)
```

```
y = TimeDistributed(Activation('elu'), name='Activ 1 MELSPECT')(y)
y = TimeDistributed(MaxPooling2D(pool size=(2, 2), strides=(2, 2), padding='same'), na
me='MaxPool 1 MELSPECT')(y)
y = TimeDistributed(Dropout(0.2), name='Drop 1 MELSPECT')(y)
## Second LFLB (local feature learning block)
y = TimeDistributed(Conv2D(64, kernel size=(3, 3), strides=(1, 1), padding='same'), na
me='Conv_2_MELSPECT')(y)
y = TimeDistributed(BatchNormalization(), name='BatchNorm 2 MELSPECT')(y)
y = TimeDistributed(Activation('elu'), name='Activ 2 MELSPECT')(y)
y = TimeDistributed(MaxPooling2D(pool size=(4, 4), strides=(4, 4), padding='same'), na
me='MaxPool 2 MELSPECT')(y)
y = TimeDistributed(Dropout(0.2), name='Drop 2 MELSPECT')(y)
## Second LFLB (local feature learning block)
y = TimeDistributed(Conv2D(128, kernel size=(3, 3), strides=(1, 1), padding='same'), na
me='Conv 3 MELSPECT')(y)
y = TimeDistributed(BatchNormalization(), name='BatchNorm_3_MELSPECT')(y)
y = TimeDistributed(Activation('elu'), name='Activ 3 MELSPECT')(y)
y = TimeDistributed(MaxPooling2D(pool size=(4, 4), strides=(4, 4), padding='same'), na
me='MaxPool 3 MELSPECT')(y)
y = TimeDistributed(Dropout(0.2), name='Drop 3 MELSPECT')(y)
## Second LFLB (local feature learning block)
```

```
y = TimeDistributed(Conv2D(128, kernel size=(3, 3), strides=(1, 1), padding='same'), na
me='Conv_4_MELSPECT')(y)
y = TimeDistributed(BatchNormalization(), name='BatchNorm 4 MELSPECT')(y)
y = TimeDistributed(Activation('elu'), name='Activ 4 MELSPECT')(y)
y = TimeDistributed(MaxPooling2D(pool size=(4, 4), strides=(4, 4), padding='same'), na
me='MaxPool 4 MELSPECT')(y)
y = TimeDistributed(Dropout(0.2), name='Drop 4 MELSPECT')(y)
## Flat
y = TimeDistributed(Flatten(), name='Flat MELSPECT')(y)
# Apply 2 LSTM layer and one FC
y = LSTM(256, return sequences=False, dropout=0.2, name='LSTM 1')(y)
y = Dense(y train.shape[1], activation='softmax', name='FC')(y)
# Build final model
model = Model(inputs=input y, outputs=y)
# Plot model graph
plot model(model, show shapes=True, show layer names=True, to file='model.png')
Image(retina=True, filename='model.png')
# Compile model
model.compile(optimizer=SGD(lr=0.01, decay=1e-
6, momentum=0.8), loss='categorical crossentropy', metrics=['accuracy'])
# Save best model
best model save = ModelCheckpoint('drive/My Drive/SpeechEmotionRecognition/[CN
N-LSTM]Model.hdf5', save best only=True, monitor='val acc', mode='max')
# Early stopping
```

```
early stopping = EarlyStopping(monitor='val acc', patience=30, verbose=1, mode='max')
# Fit model
history = model.fit(X train, y train, batch size=64, epochs=100, validation data=(X test,
y test), callbacks=[early stopping, best model save])
# Loss Curves
plt.figure(figsize=(25, 10))
plt.subplot(1, 2, 1)
plt.plot(history.history['loss'],'-g',linewidth=1.0)
plt.plot(history.history['val loss'],'r',linewidth=1.0)
plt.legend(['Training loss', 'Validation Loss'],fontsize=14)
plt.xlabel('Epochs',fontsize=16)
plt.ylabel('Loss',fontsize=16)
plt.title('Loss Curves',fontsize=22)
# Accuracy Curves
plt.subplot(1, 2, 2)
plt.plot(history.history['acc'],'-g',linewidth=1.0)
plt.plot(history.history['val acc'],'r',linewidth=1.0)
plt.legend(['Training Accuracy', 'Validation Accuracy'],fontsize=14)
plt.xlabel('Epochs',fontsize=16)
plt.ylabel('Accuracy',fontsize=16)
plt.title('Accuracy Curves',fontsize=22)
plt.show()
```

#Save the model

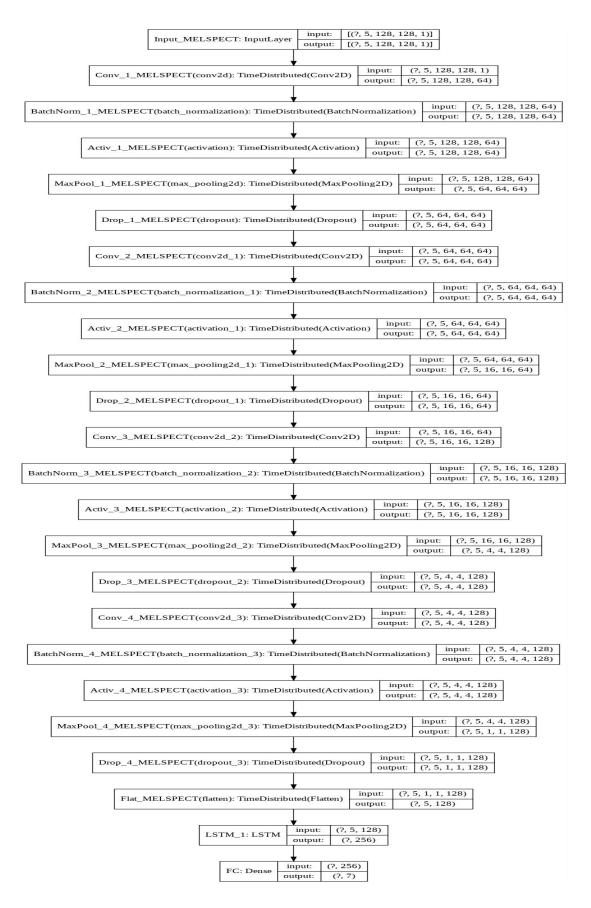
model.save('drive/My Drive/SpeechEmotionRecognition/[CNN-LSTM]M.h5')
model.save weights('drive/My Drive/SpeechEmotionRecognition/[CNN-LSTM]W.h5')

Output of Accuracy:

```
EDOCU 83/100
3225/3225 [===============] - 285 9ms/sample - loss: 0.3919 - acc: 0.8549 - val loss: 1.6001 - val acc: 0.6059
Epoch 84/100
Epoch 85/100
3225/3225 [=========================== ] - 28s 9ms/sample - loss: 0.3944 - acc: 0.8589 - val_loss: 0.8618 - val_acc: 0.6840
Epoch 86/100
3225/3225 [============== ] - 28s 9ms/sample - loss: 0.4044 - acc: 0.8481 - val_loss: 0.9878 - val_acc: 0.6691
Epoch 87/100
Epoch 88/100
Epoch 89/100
Epoch 90/100
3225/3225 [================= ] - 28s 9ms/sample - loss: 0.3525 - acc: 0.8750 - val_loss: 0.9331 - val_acc: 0.6803
Epoch 91/100
3225/3225 [============================ ] - 285 9ms/sample - loss: 0.3497 - acc: 0.8744 - val_loss: 0.8997 - val_acc: 0.6952
Epoch 92/100
3225/3225 [================ ] - 28s 9ms/sample - loss: 0.3405 - acc: 0.8809 - val_loss: 0.8597 - val_acc: 0.7323
Epoch 93/100
Enoch 94/100
Epoch 95/100
3225/3225 [===============] - 285 9ms/sample - loss: 0.3014 - acc: 0.8890 - val loss: 1.1308 - val acc: 0.6877
Epoch 96/100
3225/3225 [================ ] - 28s 9ms/sample - loss: 0.3167 - acc: 0.8878 - val_loss: 1.0205 - val_acc: 0.7249
Epoch 97/100
3225/3225 [=========================== ] - 28s 9ms/sample - loss: 0.2862 - acc: 0.9029 - val_loss: 0.9757 - val_acc: 0.6989
Epoch 98/100
3225/3225 [================= ] - 28s 9ms/sample - loss: 0.2941 - acc: 0.8936 - val_loss: 1.3295 - val_acc: 0.6431
Epoch 99/100
Enoch 188/188
3225/3225 [============] - 28s 9ms/sample - loss: 0.2774 - acc: 0.9029 - val_loss: 1.2179 - val_acc: 0.6840
```

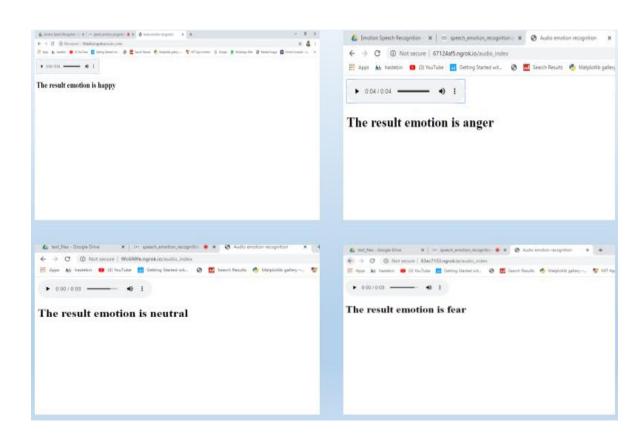
Output of

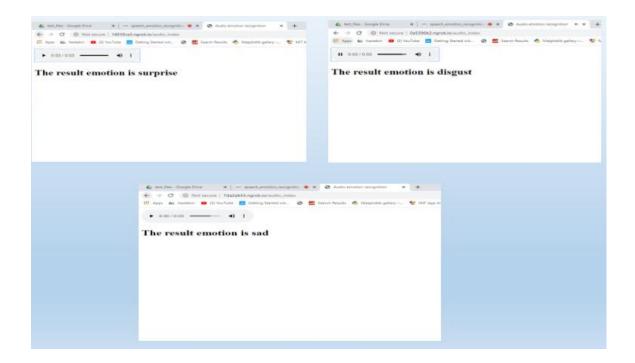
Architecture:



Output of Speech Emotion Recognition in Web page







6.1.2. Face Emotion Recognition

6.1.2.1. Pre-Processing

The models explored include:

- Manual filters
- Deep Learning Architectures
- DenseNet Inspired Architectures

General imports

import numpy as np

import pandas as pd

import matplotlib.pyplot as plt

from time import time

from time import sleep import re import os import argparse from collections import OrderedDict import matplotlib.animation as animation #Image processing from scipy.ndimage import zoom from scipy.spatial import distance import imutils from scipy import ndimage import cv2 import dlib from future import division from imutils import face utils **#CNN** models %tensorflow_version 1.x import keras from keras.preprocessing.image import ImageDataGenerator, array to img, img to arra y, load img from keras.callbacks import TensorBoard from keras.models import Sequential from keras.layers.core import Dense, Dropout, Activation, Flatten

from keras.layers.convolutional import Conv2D, MaxPooling2D, SeparableConv2D

from keras.utils import np_utils

from keras.regularizers import 12#, activity 12

from keras.optimizers import SGD, RMSprop

from keras.utils import to categorical

from keras.layers.normalization import BatchNormalization

from keras import models

from keras.utils.vis utils import plot model

from keras.layers import Input, GlobalAveragePooling2D

from keras.models import Model

from tensorflow.keras import layers

Build SVM models

from sklearn.preprocessing import OneHotEncoder

from sklearn.model selection import train test split

from sklearn.metrics import accuracy score

from sklearn import svm

#Same trained models

import h5py

from keras.models import model from json

import pickle

#Import Datas from Google Drive

from google.colab import drive

drive.mount('/content/drive')

```
local_path = '/content/drive/My Drive/Colab Notebooks/'

pd.options.mode.chained_assignment = None # default='warn' #to suppress SettingWith
CopyWarning

#Reading the dataset

dataset = pd.read_csv(local_path + 'fer2013.csv')
```

#Obtaining train data where usage is "Training"

path = '/content/drive/My Drive/Colab Notebooks/'

train = dataset[dataset["Usage"] == "Training"]

#Obtaining test data where usage is "PublicTest"

test = dataset[dataset["Usage"] == "PublicTest"]

#Converting " " separated pixel values to list

 $train['pixels'] = train['pixels'].apply(lambda \ image_px : np.fromstring(image_px, sep = ' \ '))$

 $test['pixels'] = test['pixels'].apply(lambda\ image_px: np.fromstring(image_px,\ sep = '\ '))$

dataset.head()

	emotion	pixels	Usage
0	0	70 80 82 72 58 58 60 63 54 58 60 48 89 115 121	Training
1	0	151 150 147 155 148 133 111 140 170 174 182 15	Training
2	2	231 212 156 164 174 138 161 173 182 200 106 38	Training
3	4	24 32 36 30 32 23 19 20 30 41 21 22 32 34 21 1	Training
4	6	4 0 0 0 0 0 0 0 0 0 0 3 15 23 28 48 50 58 84	Training

dataset[dataset['emotion'] == 1].head()

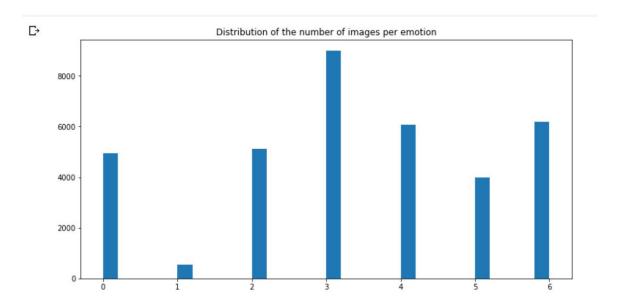
₽		emotion	pixels	Usage
	299	1	126 126 129 120 110 168 174 172 173 174 170 15	Training
	388	1	89 55 24 40 43 48 53 55 59 41 33 31 22 32 42 4	Training
	416	1	204 195 181 131 50 50 57 56 66 98 138 161 173	Training
	473	1	14 11 13 12 41 95 113 112 111 122 132 137 142	Training
	533	1	18 25 49 75 89 97 100 100 101 103 105 107 107	Training

plt.figure(figsize=(12,6))

plt.hist(dataset['emotion'], bins=30)

plt.title("Distribution of the number of images per emotion")

plt.show()



train.shape

test.shape

#Create the dataset

$$shape_x = 48$$

$$shape_y = 48$$

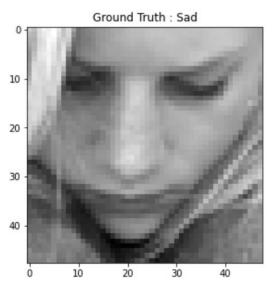
```
X train = train.iloc[:, 1].values
y train = train.iloc[:, 0].values
X \text{ test} = \text{test.iloc}[:, 1].values
y test = test.iloc[:, 0].values
#np.vstack stack arrays in sequence vertically (picking element row wise)
X train = np.vstack(X train)
X \text{ test} = \text{np.vstack}(X \text{ test})
#Reshape X train, y train, X test, y test in desired formats
X train = np.reshape(X train, (X train.shape[0],48,48,1))
y train = np.reshape(y train, (y train.shape[0],1))
X \text{ test} = \text{np.reshape}(X \text{ test}, (X \text{ test.shape}[0],48,48,1))
y test = np.reshape(y test, (y test.shape[0], 1))
print("Shape of X train and y train is " + str(X train.shape) +" and " + str(y train.shape)
+" respectively.")
print("Shape of X test and y test is " + str(X test.shape) +" and " + str(y test.shape) +" r
espectively.")
# Change to float datatype
train data = X train.astype('float32')
test data = X test.astype('float32')
# Scale the data to lie between 0 to 1
train data /= 255
test data /= 255
# Change the labels from integer to categorical data
```

```
train labels one hot = to categorical(y train)
test labels one hot = to categorical(y test)
#Define the number of classes
# Find the unique numbers from the train labels
classes = np.unique(y train)
nClasses = len(classes)
print('Total number of outputs : ', nClasses)
print('Output classes : ', classes)
# Find the shape of input images and create the variable input shape
nRows,nCols,nDims = X train.shape[1:]
input shape = (nRows, nCols, nDims)
#Defining labels
def get label(argument):
  labels = {0:'Angry', 1:'Disgust', 2:'Fear', 3:'Happy', 4:'Sad', 5:'Surprise', 6:'Neutral'}
  return(labels.get(argument, "Invalid emotion"))
plt.figure(figsize=[10,5])
# Display the first image in training data
plt.subplot(121)
plt.imshow(np.squeeze(X train[25,:,:], axis = 2), cmap='gray')
plt.title("Ground Truth : {}".format(get label(int(y train[25]))))
# Display the first image in testing data
plt.subplot(122)
```

 $plt.imshow(np.squeeze(X_test[26,:,:], axis = 2), cmap='gray')$

plt.title("Ground Truth : {}".format(get_label(int(y_test[26]))))





Detect Faces:

!pip install git+git://github.com/PnS2019/pnslib.git

from pnslib import utils

def detect_face(frame):

#Cascade classifier pre-trained model

#cascPath=files.upload()

#cascPath = '/usr/local/lib/python3.7/site-

packages/cv2/data/haarcascade frontalface default.xml'

#faceCascade = cv2.CascadeClassifier(cascPath)

faceCascade = cv2.CascadeClassifier(

utils.get haarcascade path('/content/drive/My Drive/Colab

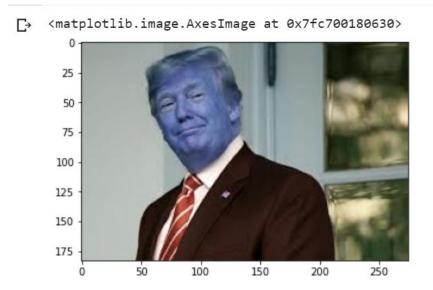
Notebooks/haarcascade_frontalface_default.xml'))

#eye_cascade = cv2.CascadeClassifier(

```
# utils.get haarcascade path('haarcascade eye.xml'))
  #BGR -> Gray conversion
  gray = cv2.cvtColor(frame, cv2.COLOR BGR2GRAY)
  #Cascade MultiScale classifier
 # detected faces =
faceCascade.detectMultiScale(gray,scaleFactor=1.1,minNeighbors=6,
  #
                             minSize=(shape x, shape y),
                             flags=cv2.CASCADE SCALE IMAGE)
  #
  detected faces = faceCascade.detectMultiScale(gray,scaleFactor=1.1,minNeighbors=6,
                            minSize=(shape x,shape y),
                            flags=cv2.CASCADE SCALE IMAGE)
  coord = []
  for x, y, w, h in detected faces:
    if w > 100:
      sub_img=frame[y:y+h,x:x+w]
      #cv2.rectangle(frame,(x,y),(x+w,y+h),(0, 255,255),1)
      coord.append([x,y,w,h])
  return gray, detected_faces, coord
```

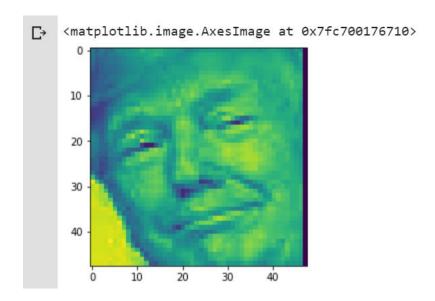
```
#Extract facial features
def extract face features(faces, offset coefficients=(0.075, 0.05)):
  gray = faces[0]
  detected face = faces[1]
  new_face = []
  for det in detected face:
    #Region in which the face is detected
    x, y, w, h = det
    #X and y correspond to the conversion to gray by gray, and w, h correspond to the
height / width
    #Offset coefficient, np.floor takes the lowest integer (delete border of the image)
    horizontal_offset = np.int(np.floor(offset coefficients[0] * w))
    vertical offset = np.int(np.floor(offset coefficients[1] * h))
    #gray = cv2.cvtColor(frame, cv2.COLOR_BGR2GRAY)
    #gray transforms the image
    extracted face = gray[y+vertical offset:y+h, x+horizontal offset:x-
horizontal_offset+w]
    #Zoom on the extracted face
```

```
new_extracted_face = zoom(extracted_face, (shape x /
extracted face.shape[0],shape y / extracted face.shape[1]))
    #cast type float
    new_extracted_face = new_extracted_face.astype(np.float32)
    #scale
    new extracted face /= float(new extracted face.max())
    #print(new extracted face)
    new_face.append(new_extracted_face)
  return new_face
from io import BytesIO
from PIL import Image
pic="/content/drive/My Drive/Colab Notebooks/test.jpg"
trump = Image.open(pic)
trump_face = cv2.imread(pic, cv2.COLOR_BGR2RGB)
plt.imshow(trump face)
```



Extracted Face:

face = extract_face_features(detect_face(trump_face))[0]
plt.imshow(face)



6.1.2.2. Deep Learning Model Architectures (CNN)

A Simple Model:

def createModel():

```
#Model Initialization
  model = Sequential()
  #Adding Input Layer
  model.add(Conv2D(32, (3, 3), padding='same', activation='relu', input shape=input
shape))
  #Adding more layers
  model.add(Conv2D(32, (3, 3), activation='relu'))
  model.add(MaxPooling2D(pool size=(2, 2)))
  model.add(Dropout(0.25))
  model.add(Conv2D(64, (3, 3), padding='same', activation='relu'))
  model.add(Conv2D(64, (3, 3), activation='relu'))
  model.add(MaxPooling2D(pool size=(2, 2)))
  model.add(Dropout(0.25))
  model.add(Conv2D(64, (3, 3), padding='same', activation='relu'))
  model.add(Conv2D(64, (3, 3), activation='relu'))
  model.add(MaxPooling2D(pool size=(2, 2)))
  model.add(Dropout(0.25))
```

```
#Flattening
  model.add(Flatten())
  #Adding fully connected layer
  model.add(Dense(512, activation='relu'))
  model.add(Dropout(0.6))
  #Adding Output Layer
  model.add(Dense(nClasses, activation='softmax'))
  return model
Prevent Overfitting:
def createModel2():
  #Model Initialization
  model = Sequential()
  model.add(Conv2D(32, (3, 3), padding='same', activation='relu', input shape=input
shape))
  model.add(MaxPooling2D(pool size=(2, 2)))
  model.add(BatchNormalization())
```

```
model.add(Conv2D(32, (3, 3), activation='relu'))
  model.add(MaxPooling2D(pool_size=(2, 2)))
  model.add(BatchNormalization())
  model.add(Conv2D(32, (3, 3), padding='same', activation='relu'))
  model.add(MaxPooling2D(pool size=(2, 2)))
  model.add(Conv2D(32, (3, 3), padding='same', activation='relu'))
  #Flattening
  model.add(Flatten())
  #Adding fully connected layer
  model.add(Dense(512, activation='relu'))
  #Adding Output Layer
  model.add(Dense(nClasses, activation='softmax'))
  return model
Deeper networks:
def createModel3():
  #Model Initialization
  model = Sequential()
```

```
model.add(Conv2D(20, (3, 3), padding='same', activation='relu',
input shape=input shape))
  model.add(Conv2D(30, (3, 3), padding='same', activation='relu'))
  model.add(MaxPooling2D(pool size=(2, 2)))
  model.add(BatchNormalization())
  model.add(Dropout(0.2))
  model.add(Conv2D(40, (3, 3), padding='same', activation='relu'))
  model.add(Conv2D(50, (3, 3), padding='same', activation='relu'))
  model.add(MaxPooling2D(pool size=(2, 2)))
  model.add(BatchNormalization())
  model.add(Dropout(0.2))
  model.add(Conv2D(60, (3, 3), padding='same', activation='relu'))
  model.add(Conv2D(70, (3, 3), padding='same', activation='relu'))
  model.add(MaxPooling2D(pool size=(2, 2)))
  model.add(Dropout(0.2))
  model.add(Conv2D(80, (3, 3), padding='same', activation='relu'))
  model.add(Conv2D(90, (3, 3), padding='same', activation='relu'))
  #Flattening
  model.add(Flatten())
```

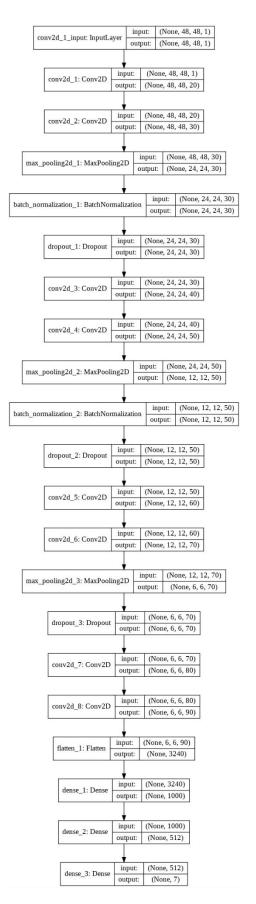
```
#Adding fully connected layer
model.add(Dense(1000, activation='relu'))
model.add(Dense(512, activation='relu'))

#Adding Output Layer
model.add(Dense(nClasses, activation='softmax'))
Build Model:
model = createModel3()
model.summary()
```

Layer (type)	Output	Shape	Param :
conv2d_1 (Conv2D)	(None,	48, 48, 20)	200
conv2d_2 (Conv2D)	(None,	48, 48, 30)	5430
max_pooling2d_1 (MaxPooling2	(None,	24, 24, 30)	0
batch_normalization_1 (Batch	(None,	24, 24, 30)	120
dropout_1 (Dropout)	(None,	24, 24, 30)	0
conv2d_3 (Conv2D)	(None,	24, 24, 40)	10840
conv2d_4 (Conv2D)	(None,	24, 24, 50)	18050
max_pooling2d_2 (MaxPooling2	(None,	12, 12, 50)	0
batch_normalization_2 (Batch	(None,	12, 12, 50)	200
dropout_2 (Dropout)	(None,	12, 12, 50)	0
conv2d_5 (Conv2D)	(None,	12, 12, 60)	27060
conv2d_6 (Conv2D)	(None,	12, 12, 70)	37870
max_pooling2d_3 (MaxPooling2	(None,	6, 6, 70)	0
dropout_3 (Dropout)	(None,	6, 6, 70)	0
conv2d_7 (Conv2D)	(None,	6, 6, 80)	50480
conv2d_8 (Conv2D)	(None,	6, 6, 90)	64890
flatten_1 (Flatten)	(None,	3240)	0
dense_1 (Dense)	(None,	1000)	324100
dense_2 (Dense)	(None,	512)	512512
dense_3 (Dense)	(None,	7)	3591
Total params: 3,972,243 Trainable params: 3,972,083 Non-trainable params: 160			

And visualize the model architecture:

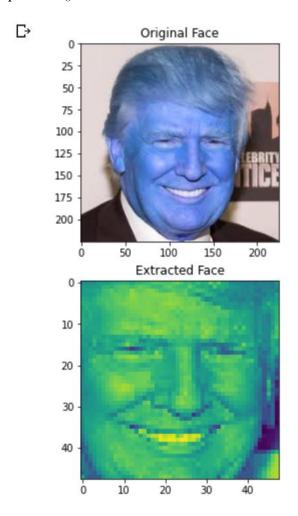
plot_model(model, to_file='model_plot.png', show_shapes=True, show_layer_names=
True)



Visualize layers and output

```
layer outputs = [layer.output for layer in model.layers[:12]]
# Extracts the outputs of the top 12 layers
activation model = models.Model(inputs=model.input, outputs=layer outputs)
layer names = []
for layer in model.layers[:12]:
  layer names.append(layer.name)
# Names of the layers
images per row = 16
pic='/content/drive/My Drive/Colab Notebooks/trump.png'
trump = Image.open(pic)
trump face = cv2.imread("/content/drive/My Drive/Colab Notebooks/trump.png")
face = extract face features(detect face(trump face))[0]
to predict = np.reshape(face.flatten(), (1,48,48,1))
res = model.predict(to predict)
activations = activation model.predict(to predict)
plt.figure(figsize=(12,8))
plt.subplot(211)
plt.title("Original Face")
plt.imshow(trump face)
plt.subplot(212)
plt.title("Extracted Face")
plt.imshow(face)
```

plt.show()



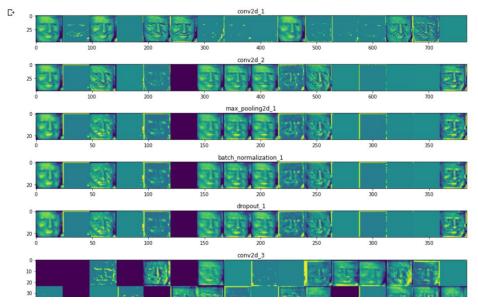
for layer_name, layer_activation in zip(layer_names, activations): # Displays the featur e maps

n_features = layer_activation.shape[-1] # Number of features in the feature map
size = layer_activation.shape[1] #The feature map has shape (1, size, size, n_feature
s).

n_cols = n_features // images_per_row # Tiles the activation channels in this matrix
display_grid = np.zeros((size * n_cols, images_per_row * size))

for col in range(n_cols): # Tiles each filter into a big horizontal grid for row in range(images per row):

```
channel_image = layer_activation[0,:,:,col * images_per_row + row]
       channel image -= channel image.mean() # Post-
processes the feature to make it visually palatable
       channel_image /= channel_image.std()
       channel image *= 64
       channel image += 128
       channel image = np.clip(channel image, 0, 255).astype('uint8')
       display grid[col * size : (col + 1) * size, # Displays the grid
               row * size : (row + 1) * size] = channel image
  scale = 1. / size
  plt.figure(figsize=(scale * display_grid.shape[1],
              scale * display grid.shape[0]))
  plt.title(layer name)
  plt.grid(False)
  plt.imshow(display grid, aspect='auto', cmap='viridis')
```



Create and train the model:

```
datagen = ImageDataGenerator(
    zoom range=0.2,
                            # randomly zoom into images
    rotation range=10,
                           # randomly rotate images in the range (degrees, 0 to 180)
    width shift range=0.1, #randomly shift images horizontally (fraction of total wi
dth)
    height shift range=0.1, # randomly shift images vertically (fraction of total heig
ht)
    horizontal flip=True, # randomly flip images
     vertical flip=False) # randomly flip images
#Creating 2nd model and training(fitting)
model.compile(optimizer='adam', loss='categorical crossentropy', metrics=['accuracy'])
batch size = 256
epochs = 100
# Fit the model on the batches generated by datagen.flow().
history = model.fit generator(
  datagen.flow(train data, train labels one hot, batch size=batch size),
  steps_per_epoch=int(np.ceil(train_data.shape[0] / float(batch_size))),
  epochs = epochs,
  validation data=(test data, test labels one hot)
```

```
Epoch 90/100
113/113 [====
Epoch 91/100
        =================] - 17s 149ms/step - loss: 0.8281 - accuracy: 0.6879 - val_loss: 1.0447 - val_accuracy: 0.6406
                      - 17s 149ms/step - loss: 0.8258 - accuracy: 0.6872 - val_loss: 1.0355 - val_accuracy: 0.6481
Epoch 92/100
                      - 17s 149ms/step - loss: 0.8203 - accuracy: 0.6908 - val_loss: 1.0535 - val_accuracy: 0.6411
Epoch 93/100
                      - 17s 151ms/step - loss: 0.8261 - accuracy: 0.6859 - val_loss: 1.0429 - val_accuracy: 0.6397
Fnoch 94/199
         Epoch 95/100
113/113 [====
         Epoch 96/100
=========================] - 17s 150ms/step - loss: 0.7973 - accuracy: 0.6948 - val_loss: 1.0347 - val_accuracy: 0.6478
113/113 [=====
113/113 [============] - 17s 150ms/step - loss: 0.7962 - accuracy: 0.7010 - val_loss: 1.0804 - val_accuracy: 0.6434
```

Evaluate the model:

#Plotting accuracy and loss curves for 2nd model

```
# Loss Curves

plt.figure(figsize=[8,6])

plt.plot(history.history['loss'],'r',linewidth=2.0)

plt.plot(history.history['val_loss'],'b',linewidth=2.0)

plt.legend(['Training loss', 'Validation Loss'],fontsize=18)

plt.xlabel('Epochs ',fontsize=16)

plt.ylabel('Loss',fontsize=16)

plt.title('Loss Curves',fontsize=16)

# Accuracy Curves

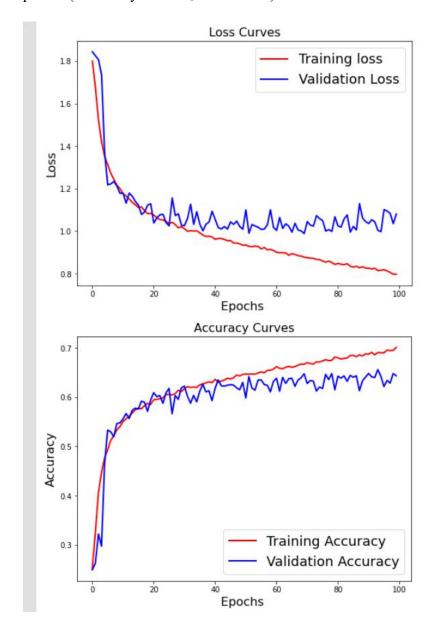
plt.figure(figsize=[8,6])

plt.plot(history.history['acc'],'r',linewidth=2.0)

plt.plot(history.history['val_acc'],'b',linewidth=2.0)

plt.legend(['Training Accuracy', 'Validation Accuracy'],fontsize=18)
```

plt.xlabel('Epochs ',fontsize=16)
plt.ylabel('Accuracy',fontsize=16)
plt.title('Accuracy Curves',fontsize=16)



Save and reopen the model:

#save the model weights

json_string = model.to_json()

model.save weights('/content/drive/My Drive/Colab Notebooks/model 3.h5')

open('/content/drive/My Drive/Colab Notebooks/model 3.json', 'w').write(json string) model.save weights(local path + 'savedmodels/Emotion Face Detection Model.h5') with open('/content/drive/My Drive/Colab Notebooks/model 3.json','r') as f: json = f.read()

model = model from json(json)

model.load weights('/content/drive/My Drive/Colab Notebooks/model 3.h5') model.save('/content/drive/My Drive/Colab Notebooks/model fer.h5') print("Loaded model from disk")

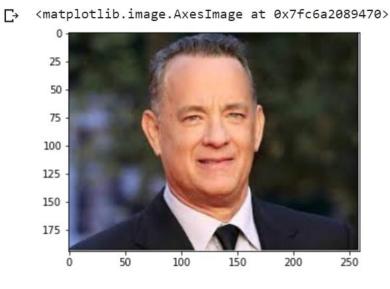
Making prediction on an image:

pic='/content/drive/My Drive/Colab Notebooks/hanks.png'

hanks = Image.open(pic)

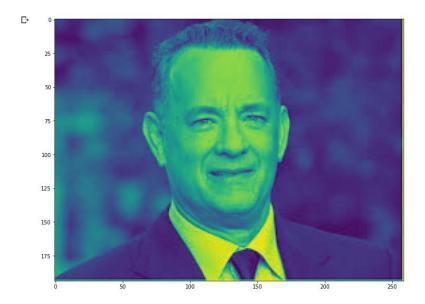
hanks face = cv2.imread(pic)

plt.imshow(cv2.cvtColor(hanks face, cv2.COLOR BGR2RGB))



plt.figure(figsize=(12,12)) plt.imshow(detect face(hanks face)[0])

plt.show()

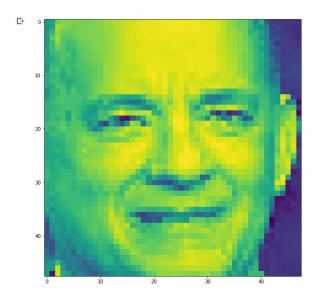


 $for \ face \ in \ extract_face_features(detect_face(hanks_face)):$

plt.figure(figsize=(10,10))

plt.imshow(face)

plt.show()



for face in extract_face_features(detect_face(hanks_face)) :

to_predict = np.reshape(face.flatten(), (1,48,48,1))

```
res = model.predict(to_predict)
result_num = np.argmax(res)
print(result_num)

$\textstyle 3 \quad 3 \quad \text{3}$
```

This corresponds to the Happy Labels which is a good prediction.

Enhanced visualization:

Frequency of eye bink:

```
def eye_aspect_ratio(eye):

A = distance.euclidean(eye[1], eye[5])

B = distance.euclidean(eye[2], eye[4])

C = distance.euclidean(eye[0], eye[3])

ear = (A + B) / (2.0 * C)

return ear

thresh = 0.25

frame_check = 20

face_detect = dlib.get_frontal_face_detector()

predictor_landmarks = dlib.shape_predictor(local_path+"shape_predictor_68_face_landmarks.dat")

(lStart, lEnd) = face_utils.FACIAL_LANDMARKS_IDXS["left_eye"]

(rStart, rEnd) = face_utils.FACIAL_LANDMARKS_IDXS["right_eye"]

Detect Keypoints to plot them:
```

(nStart, nEnd) = face utils.FACIAL LANDMARKS IDXS["nose"]

```
(mStart, mEnd) = face utils.FACIAL LANDMARKS IDXS["mouth"]
(jStart, jEnd) = face utils.FACIAL LANDMARKS IDXS["jaw"]
(eblStart, eblEnd) = face utils.FACIAL LANDMARKS IDXS["left eyebrow"]
(ebrStart, ebrEnd) = face utils.FACIAL LANDMARKS IDXS["right eyebrow"]
Face Alignment:
desiredLeftEye=(0.35, 0.35)
def align(gray, rect):
  # convert the landmark (x, y)-coordinates to a NumPy array
  shape = predictor(gray, rect)
  shape = shape to np(shape)
  \# extract the left and right eye (x, y)-coordinates
  (lStart, lEnd) = FACIAL LANDMARKS IDXS["left eye"]
  (rStart, rEnd) = FACIAL LANDMARKS IDXS["right eye"]
  leftEyePts = shape[1Start:1End]
  rightEyePts = shape[rStart:rEnd]
  # compute the center of mass for each eye
  leftEyeCenter = leftEyePts.mean(axis=0).astype("int")
  rightEyeCenter = rightEyePts.mean(axis=0).astype("int")
  # compute the angle between the eye centroids
```

```
dY = rightEyeCenter[1] - leftEyeCenter[1]
dX = rightEyeCenter[0] - leftEyeCenter[0]
angle = np.degrees(np.arctan2(dY, dX)) - 180
# compute the desired right eye x-coordinate based on the
# desired x-coordinate of the left eye
desiredRightEyeX = 1.0 - desiredLeftEye[0]
# determine the scale of the new resulting image by taking
# the ratio of the distance between eyes in the *current*
# image to the ratio of distance between eyes in the
# *desired* image
dist = np.sqrt((dX ** 2) + (dY ** 2))
desiredDist = (desiredRightEyeX - desiredLeftEye[0])
desiredDist *= self.desiredFaceWidth
scale = desiredDist / dist
\# compute center (x, y)-coordinates (i.e., the median point)
# between the two eyes in the input image
eyesCenter = ((leftEyeCenter[0] + rightEyeCenter[0]) // 2,
     (leftEyeCenter[1] + rightEyeCenter[1]) // 2)
```

grab the rotation matrix for rotating and scaling the face

```
M = cv2.getRotationMatrix2D(eyesCenter, angle, scale)
```

update the translation component of the matrix

tX = self.desiredFaceWidth * 0.5

tY = self.desiredFaceHeight * self.desiredLeftEye[1]

M[0, 2] += (tX - eyesCenter[0])

M[1, 2] += (tY - eyesCenter[1])

apply the affine transformation

(w, h) = (self.desiredFaceWidth, self.desiredFaceHeight)

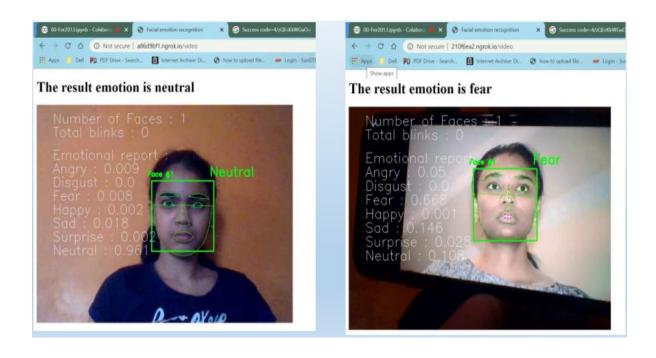
#output = cv2.warpAffine(image, M, (w, h), flags=cv2.INTER_CUBIC)

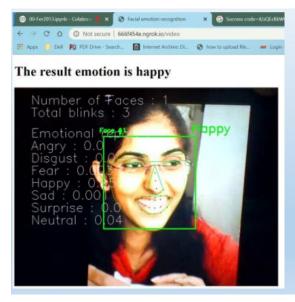
output = cv2.warpAffine(gray, M, (w, h), flags=cv2.INTER CUBIC)

return the aligned face

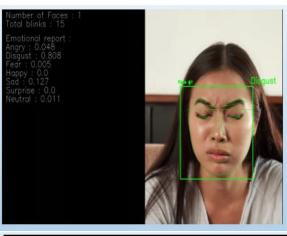
return output





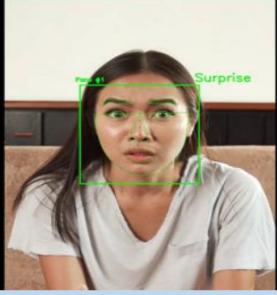












CHAPTER 7 TESTING

7.1. Speech Emotion Recognition:

Testing is basically, evaluating the accuracy of a model, when speech emotion recognition model is tested, we have achieved an accuracy of about 96% on training and about 68% on validation.

```
EDOCU 83/100
Epoch 84/100
3225/3225 [============] - 285 9ms/sample - loss: 0.4080 - acc: 0.8493 - val_loss: 1.0905 - val_acc: 0.6543
Epoch 85/100
Epoch 86/100
Epoch 88/100
Epoch 89/100
Epoch 90/100
3225/3225 [=============] - 28s 9ms/sample - loss: 0.3525 - acc: 0.8750 - val_loss: 0.9331 - val_acc: 0.6803
Epoch 91/100
Epoch 92/100
3225/3225 [===========] - 285 9ms/sample - loss: 0.3405 - acc: 0.8809 - val loss: 0.8597 - val acc: 0.7323
Epoch 93/100
Epoch 94/100
Epoch 95/100
3225/3225 [===========] - 285 9ms/sample - loss: 0.3014 - acc: 0.8890 - val_loss: 1.1308 - val_acc: 0.6877
Epoch 96/100
Epoch 97/100
Epoch 98/100
Epoch 99/100
Epoch 100/100
```

```
1344

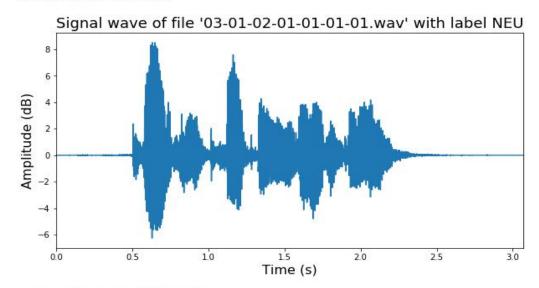
4

NEU

[-0.00020196 -0.00020196 -0.00020196 ... -0.00020215 -0.00020169

-0.00020233]

03-01-02-01-01-01-01.wav
```

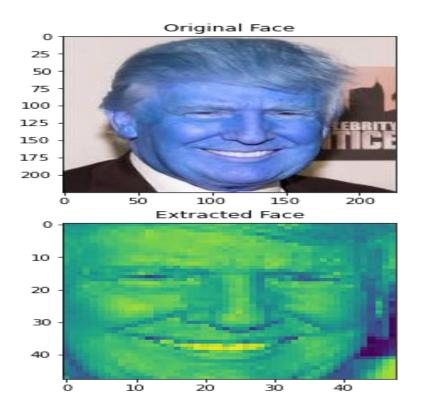


Audio file '03-01-02-01-01-01-01.wav':

7.2. Face Emotion Recognition:

For face emotion recognition model, we have achieved an accuracy of about 89% on training and about 72% on validation.

```
Epoch 90/100
Epoch 91/100
Epoch 92/100
Epoch 93/100
Epoch 94/100
      113/113 [====
Fnoch 95/100
113/113 [====
       Epoch 96/100
     113/113 [=====
Epoch 97/100
113/113 [==========] - 17s 151ms/step - loss: 0.8140 - accuracy: 0.6954 - val_loss: 1.0943 - val_accuracy: 0.6339
Epoch 98/100
113/113 [===========] - 17s 150ms/step - loss: 0.8044 - accuracy: 0.6945 - val loss: 1.0828 - val accuracy: 0.6283
Epoch 99/100
113/113 [==========] - 17s 150ms/step - loss: 0.7973 - accuracy: 0.6948 - val_loss: 1.0347 - val_accuracy: 0.6478
Epoch 100/100
113/113 [==========] - 17s 150ms/step - loss: 0.7962 - accuracy: 0.7010 - val_loss: 1.0804 - val_accuracy: 0.6434
```



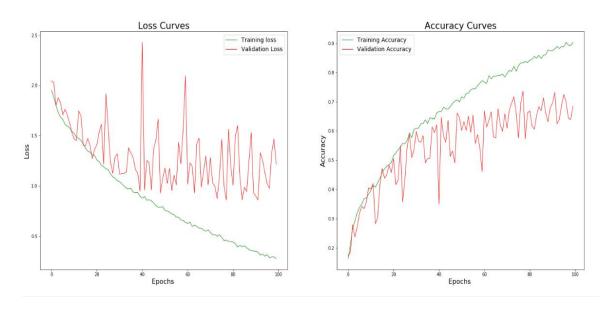
CHAPTER 8

CONCLUSION AND FUTURE WORK

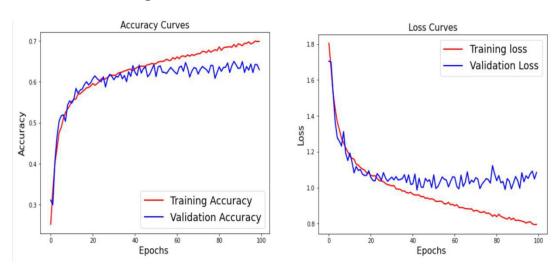
8.1 Conclusion:

Most of the current research concentrate on investigating different features and their correlation with emotional state in audio, video and text. In this fact some researcher develop their own feature like MLS to achieve high performance in recognition rate. In most of them, it is hard even for human to specify different emotion of certain collected utterances. In conclusion, there are only limited studies that considered applying multiple classifier to speech emotion recognition. We reviewed and discussed various emotional recognition systems based approaches. We also compare its performance in terms of classifier, features, recognition rate, and datasets. Well-design classifiers have obtain high classification accuracies between different types of emotions. In this current approach, we presented an automatic multimodal emotion recognition (MER) system using CNN as the machine learning algorithm which classify 7 emotions. Thus, We used Ravdess dataset for speech using time distributed CNN which extracts MFCC, Mel-Scale, Spectrogram features. For video, Fer2013 dataset using CNN model with synthesized features like haar features, HOG sliding windows, HOG features and Facial landmarks are used. Infact, we study how classifiers and features impact recognition accuracy of emotions in audio and video. A subset of highly discriminant features is selected. Feature selection techniques show that more information is not always good in machine learning applications. The machine learning models were trained and evaluated to recognize emotional states from these features. When we implemented deep networks on ravdess dataset, we acheived an accuracy of 96% on training set and 78% of accuracy on testing set and about 89% accuracy on training and 79% on validation when face emotion is recognized using Fer2013 dataset and features like haar features, HOG sliding windows, HOG features and Facial landmarks are used. From this result, we can see that CNN often perform better with more data and it suffers from the problem of very long training times.

Speech Emotion Recognition:



Face Emotion Recognition:



8.2 Future Scope:

- 1. In future, we can do fusion of audio, video and text emotion recognition models using different models.
- **2.** We can integrate this system in several applications where emotion plays an important role.
- **3.** The suicidal rate of people can also be decreased when these applications can find the emotional status (like depressed state) of the person well before.

CHAPTER 9

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