

# Configuration Manual

MSc Research Project  
MSC in Data Analytics

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Project Submission Sheet  
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# Configuration Manual

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## 1 Introduction

As per the requirements of MSc Research Project submission of National College of Ireland, the Configuration Manual is concluded in relation to implementation of work. This manual reports the software used and settings considered to analyse the Deception in the audio signal using Convolutional Neural Network..

## 2 System Specification

The hardware and software requirements for the implementation of the model is discussed in this section.

### 2.1 Hardware Requirements

The hardware requirement in the development of the model is covered here detailing processor, RAM, storage and OS. Processor: 11th Gen Intel(R) Core(TM) i5-1125G7 @2.40GHz 2.42GHz Random Access Memory: 8 Giga Bytes Storage: 256GB SSD/1TB HDD Operating System: 64-bit operating system, x64-based Operating System

### 2.2 Software Requirements

The programming tools and development environments used in the research work implementation is explained below.

- Jupyter Notebook
- Python Version 3
- Overleaf

## 3 Environment Setup

### 3.1 Anaconda Installation

Anaconda was installed from <https://www.anaconda.com/distribution/#download-section> and the Python 3.7 version of Anaconda was installed for the operating system. The default settings for the same must be retained.

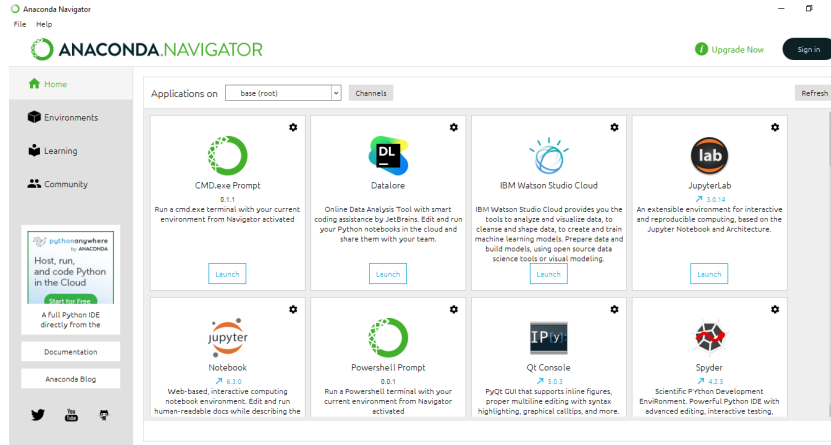


Figure 1: Anaconda Navigator

### 3.2 Jupyter Notebook

After starting Anaconda Navigator, extra optional tools can be installed. Jupyter Notebook was the Integrated Development Environment (IDE) used in this work and was installed by default.

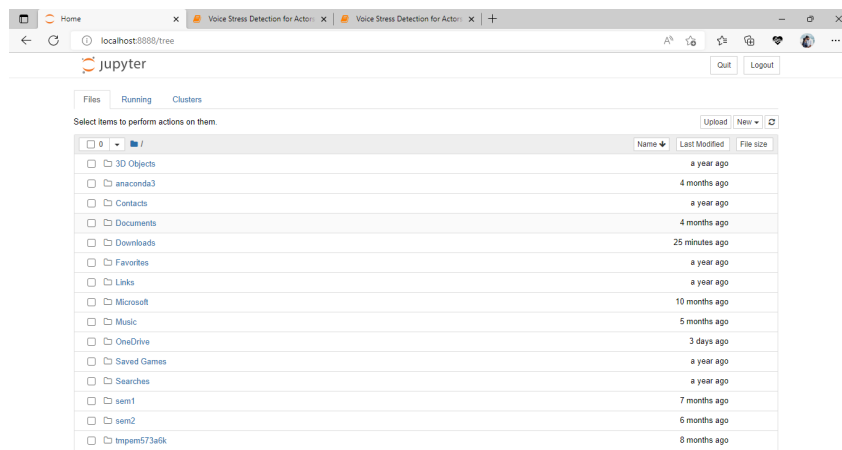


Figure 2: Jupyter Notebook

## 4 Dataset

The dataset taken for analysis was The Ryerson Audio-Visual Database of Emotional Speech and Song (RAVDESS) taken from Steven R. Livingstone (2018). The dataset was taken from an article taken from PLUS One. It was a publically available data with audio as well as video content. For this work only the audio signals are taken into consideration. The article allows to get to know the privacy and licence linked to the content and the recordings.

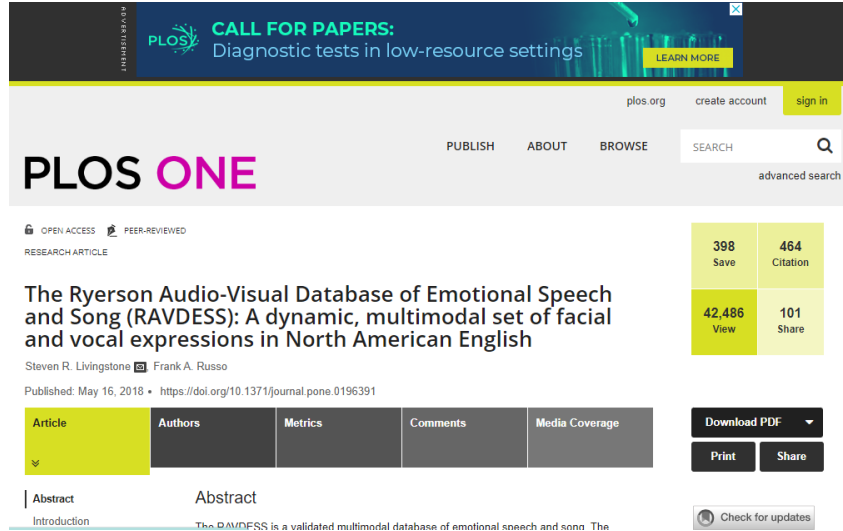


Figure 3: Citation of Dataset

The dataset include recordings of 24 different actors among which 12 are male subjects and 12 are female subjects. The file name convention shows the recording details such as Modality, Vocal channel, Emotion, Emotional intensity, Statement, Repetition and Actor gender. This can distinguish the unisexual recordings. There are 1440 recordings of speech in the dataset. This can be directly downloaded into the local PC and was directly used in the IDE using Python programming.

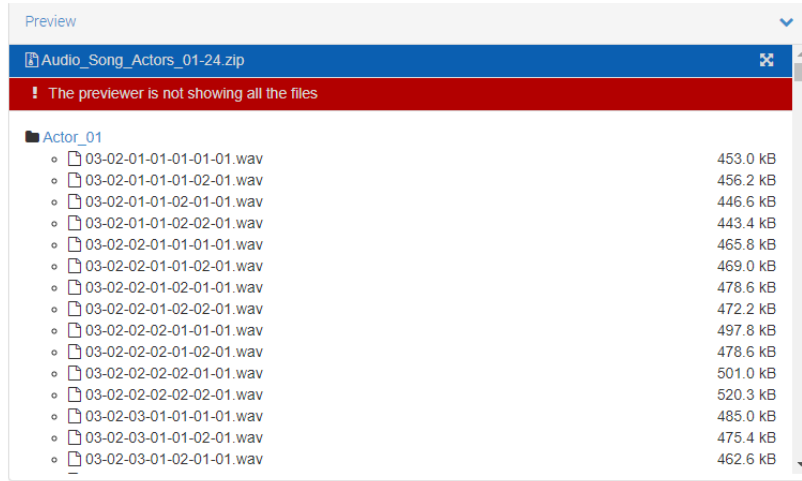


Figure 4: Dataset

## 5 Implementation

As mentioned earlier, the programming language used is Python version 3. Hence, the libraries associated with Python are used in the implementation of the model for analysis of deception by the application of CNN. The libraries used are mentioned below.

- NumPy
- Matplotlib
- Pandas
- Librosa
- Sklearn
- Plotly
- TensorFlow
- Keras

```
import os
import random
import sys

## Package
import glob
import keras
import IPython.display as ipd
import librosa
import librosa.display
import matplotlib.pyplot as plt
import numpy as np
import pandas as pd
import plotly

plotly.graph_objs as go
import plotly.offline as py
import plotly.tools as tls
import seaborn as sns
import scipy.io.wavfile
import tensorflow
py.init_notebook_mode(connected=True)
```

Figure 5: Package Import

```
## Keras
from tensorflow.keras import regularizers
from tensorflow.keras.callbacks import ModelCheckpoint, LearningRateScheduler, EarlyStopping
from tensorflow.keras.callbacks import History, ReduceLROnPlateau, CSVLogger
from tensorflow.keras.models import Model, Sequential
from tensorflow.keras.layers import Dense, Embedding, LSTM
from tensorflow.keras.layers import Input, Flatten, Dropout, Activation, BatchNormalization
from tensorflow.keras.layers import Conv1D, MaxPooling1D, AveragePooling1D
from tensorflow.keras.preprocessing import sequence
from tensorflow.keras.preprocessing.sequence import pad_sequences
from tensorflow.keras.preprocessing.text import Tokenizer
from keras.utils import np_utils
from tensorflow.keras.utils import to_categorical

## Sklearn
from sklearn.metrics import confusion_matrix
from sklearn.preprocessing import LabelEncoder

## Rest
from scipy.fftpack import fft
from scipy import signal
from scipy.io import wavfile
from tqdm import tqdm

input_duration=3
# % pylab inline
```

Figure 6: Package Import

## 5.1 Preprocessing of Data

The preprocessing done on recording include pitch tuning, speed tuning, stretching, silencing and white noise addition. The code performed is given below.

```
Data Making/Processing

In [37]: def plot_time_series(data):
        """
        Plot the Audio Frequency.
        """
        fig = plt.figure(figsize=(14, 8))
        plt.title('Raw wave ')
        plt.ylabel('Amplitude')
        plt.plot(np.linspace(0, 1, len(data)), data)
        plt.show()

        def noise(data):
            """
            Adding White Noise.
            """
            noise_amp = 0.005*np.random.uniform()*np.amax(data)
            data = data.astype('float64') + noise_amp * np.random.normal(size=data.shape[0])
            return data

        def shift(data):
            """
            Random Shifting.
            """
            s_range = int(np.random.uniform(low=-5, high = 5)*500)
            return np.roll(data, s_range)

        def stretch(data, rate=0.8):
            """
            Streching the Sound.
            """
            data = librosa.effects.time_stretch(data, rate)
            return data

        def pitch(data, sample_rate):
            """
            Pitch Tuning.
            """
            bins_per_octave = 12
            pitch_pm = 2
            pitch_change = pitch_pm * 2*(np.random.uniform())
            data = librosa.effects.pitch_shift(data.astype('float64'),
```

Figure 7: Data Preprocessing

Spectrograms are the main visualizations used in the analysis of audio signal in order to determine the amplitude and frequency of audio signal. The code of visualizing the spectrograms are discussed below.

```
[9]: # Plotting Wave Form and Spectrogram
     freqs, times, spectrogram = log_specgram(samples, sample_rate)

     fig = plt.figure(figsize=(14, 8))
     ax1 = fig.add_subplot(211)
     ax1.set_title('Raw wave of ' + filename)
     ax1.set_ylabel('Amplitude')
     librosa.display.waveshow(samples, sr=sample_rate)

     ax2 = fig.add_subplot(212)
     ax2.imshow(spectrogram.T, aspect='auto', origin='lower',
               extent=[times.min(), times.max(), freqs.min(), freqs.max()])
     ax2.set_yticks(freqs[::16])
     ax2.set_xticks(times[::16])
     ax2.set_title('Spectrogram of ' + filename)
     ax2.set_ylabel('Freqs in Hz')
     ax2.set_xlabel('Seconds')
```

Figure 8: Spectrogram

## 5.2 Data Labelling

The six different emotions are labelled to positive, negative and neutral deception audio signals. Thus the overall model becomes a three class problem.

```

#3 class: Positive, Neutral & Negative

# Positive: Happy
# Negative: Angry, Fearful, Sad
# Neutral: Calm, Neutral

label3_list = []
for i in range(len(data_df)):
    if data_df.emotion[i] == 1: # Neutral
        lb = "_neutral"
    elif data_df.emotion[i] == 2: # Calm
        lb = "_neutral"
    elif data_df.emotion[i] == 3: # Happy
        lb = "_positive"
    elif data_df.emotion[i] == 4: # Sad
        lb = "_negative"
    elif data_df.emotion[i] == 5: # Angry
        lb = "_negative"
    elif data_df.emotion[i] == 6: # Fearful
        lb = "_negative"
    else:
        lb = "_none"

    # Add gender to the label
    label3_list.append(data_df.gender[i] + lb)

len(label3_list)

```

Figure 9: Data Labelling

### 5.3 Models Implemented

For the model implementation, data from a single actor was used to compare Machine Learning model with Deep Learning model. In this work, Decision Tree and Random Forest model was implemented on single actor data.

- Decision Tree: The implementation of Decision Tree is discussed with the figure as discussed in Bento (2021). As the precision of the model obtained is 57.5%, it is not further taken for analysis.

```

DecisionTreeClassifier(criterion="gini", max_depth=10, max_features="log2",
                       max_leaf_nodes = 10, min_samples_leaf = 3, min_samples_split = 20,
                       n_estimators= 200, random_state= 5)

dt(X_train, y_train)

get_classification(max_depth=10, max_features='log2', max_leaf_nodes=10)

```

Figure 10: Decision Tree

- Random Forest: The implementation of Random Forest model for a single actor audio data is discussed with the figure below and cited in Yiu (2019). As the precision of the model obtained is just 62.34%, it is also not considered for analysis. As the problem is a multiclass classification problem, Decision Trees and Random Forests are the best possible machine learning solutions.



```

]: from sklearn.ensemble import RandomForestClassifier

]: rforest = RandomForestClassifier(criterion="gini", max_depth=10, max_features="log2",
                                max_leaf_nodes = 10, min_samples_leaf = 3, min_samples_split = 20,
                                n_estimators= 200, random_state= 5)

]: rforest.fit(X_train, y_train)

```

Figure 11: Random Forest

- Convolutional Neural Network: The implementation of Convolutional Neural network as covered in Mandal (2021) for the research work is as discussed below. Keras was used for implementation with Sequential model.

Layer (type)	Output Shape	Param #
conv1d_64 (Conv1D)	(None, 252, 256)	2384
activation_70 (Activation)	(None, 252, 256)	0
conv1d_65 (Conv1D)	(None, 252, 256)	524544
batch_normalization_16 (Batch Normalization)	(None, 252, 256)	1024
activation_71 (Activation)	(None, 252, 256)	0
dropout_16 (Dropout)	(None, 252, 256)	0
max_pooling1d_16 (MaxPooling1D)	(None, 31, 256)	0
conv1d_66 (Conv1D)	(None, 31, 128)	262272
activation_72 (Activation)	(None, 31, 128)	0
conv1d_67 (Conv1D)	(None, 31, 128)	131200
activation_73 (Activation)	(None, 31, 128)	0
conv1d_68 (Conv1D)	(None, 31, 128)	131200
activation_74 (Activation)	(None, 31, 128)	0
conv1d_69 (Conv1D)	(None, 31, 128)	131200
batch_normalization_17 (Batch Normalization)	(None, 31, 128)	512
activation_75 (Activation)	(None, 31, 128)	0
dropout_17 (Dropout)	(None, 31, 128)	0
max_pooling1d_17 (MaxPooling1D)	(None, 3, 128)	0
conv1d_70 (Conv1D)	(None, 3, 64)	65000
activation_76 (Activation)	(None, 3, 64)	0
conv1d_71 (Conv1D)	(None, 3, 64)	32832
activation_77 (Activation)	(None, 3, 64)	0
Flatten_8 (Flatten)	(None, 192)	0
dense_6 (Dense)	(None, 7)	1351
activation_78 (Activation)	(None, 7)	0

Total params: 1,286,810  
 Trainable params: 1,283,271  
 Non-trainable params: 768

Figure 12: CNN

## 6 Evaluation

The evaluation was done by mainly using the loss graph as the dataset considered is discontinuous. The evaluation metrics like precision and recall was also identified in the results section to explain the integrity of the work.

## 7 Overleaf for Documentation

For the documentation purpose, the overleaf or latex was used. This is allows ease in alignment as per the referencing style suggested by NCI.

```
[61]: # Plotting the Train Valid Loss Graph

plt.plot(cnnhistory.history['loss'])
plt.plot(cnnhistory.history['val_loss'])
plt.title('model loss')
plt.ylabel('loss')
plt.xlabel('epoch')
plt.legend(['train', 'test'], loc='upper left')
plt.show()
```

Figure 13: Loss Plot

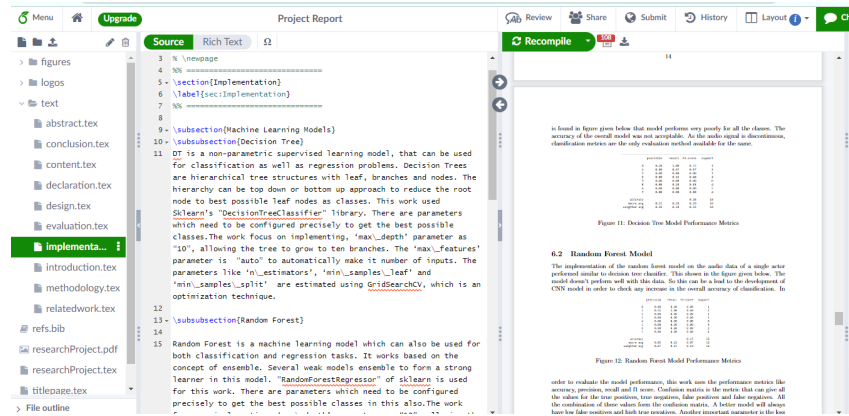


Figure 14: UI of Overleaf

## References

- Bento, C. (2021). Decision tree classifier explained in real-life: picking a vacation destination, *Towards Data Science* .
- Mandal, M. (2021). Introduction to convolutional neural networks (cnn), *Data Science Blogathon* .
- Steven R. Livingstone, F. A. R. (2018). The ryerson audio-visual database of emotional speech and song (ravdess): A dynamic, multimodal set of facial and vocal expressions in north american english, *PLUS One 13(5): e0196391* .
- Yiu, T. (2019). Understanding random forest how the algorithm works and why it is so effective, *Towards Data Science* .