National College of Ireland

Configuration Manual

MSc Research Project
MSc. Data Analytics

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MSc Project Submission Sheet

School of Computing

Student Name:	Raj Ravindra Kupekar		
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Programme:	MSc. Data Analytics	**Year:**	2020
Module:	Research Project		
Lecturer:	Prof. Christian Horn		
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Project Title:	Conversational Emotion Recognition using Text and Audio Modalities		
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I hereby certify that the information contained in this (my submission) is information pertaining to research I conducted for this project. All information other than my own contribution will be fully referenced and listed in the relevant bibliography section at the rear of the project.
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Date:	28/09/2020

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Configuration Manual

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1 Hardware Setup

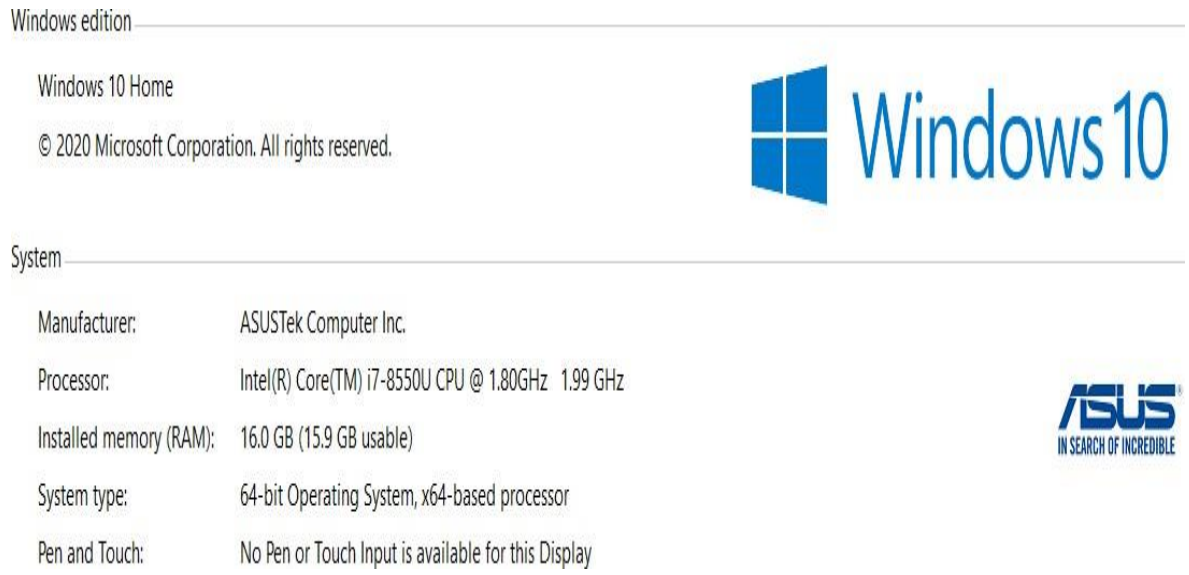

Fig. 1 Computer Specification

The specification of the machine used in this research project is illustrated in above figure. It is having an installed RAM of 16 GB with 64-bit Operating System and an installed Windows 10 configuration. The processor of the machine is Intel® Core ™ i7-8550U CPU with Intel ® UHD Graphics 620.

2 Software Information

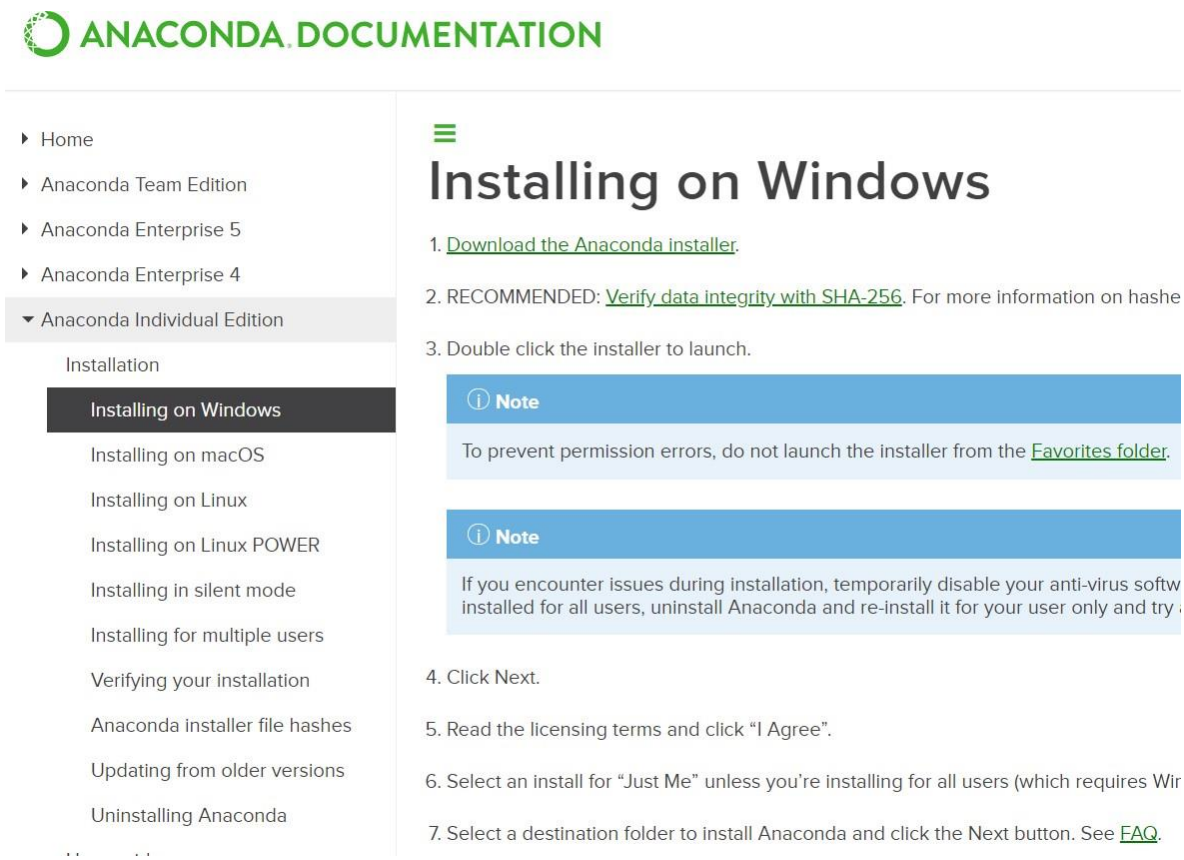

Fig 2.1 Anaconda Software for Python

In this research project, an Anaconda software [2] is installed on machine for accessing the Jupyter environment.

Fig 2.2 Anaconda Installer

Accordingly, the anaconda can be installed for each operating system with the proper operating bit. For this work, a windows 64-Bit anaconda installed

Fig 2.3 Jupyter Environment

A Jupyter notebook can be created by clicking the Python 3 option from new. Accordingly, in this work two separate notebooks are created for text and audio.

3 Data Preparation

Here the data preparation of text and audio modality are been carried out using the appropriate libraries.

```
#importing the required libraries for pre processing
import nltk
nltk.download('punkt')
import string
import re
```

```
[nltk_data] Downloading package punkt to
[nltk_data]     C:\Users\kupekarraj\AppData\Roaming\nltk_data...
[nltk_data]   Package punkt is already up-to-date!
```

```
#defining a function for cleaning the text
def cleantext(x):
    text=re.sub("[^a-zA-Z]"," ",x)
    text=text.lower()
    text=text.split()
    text=" ".join(text)
    return(text)
```

```
#Applying the cleantext function for cleaning the textual data in train, test and validation data
train_text["Utterance"]= train_text["Utterance"].apply(lambda x : cleantext(x))
test_text["Utterance"]=test_text["Utterance"].apply(lambda x : cleantext(x))
val_text["Utterance"]=val_text["Utterance"].apply(lambda x : cleantext(x))
```

Fig 3.1 Pre-processing libraries for text

The above figure shows the pre-processing libraries for text modalities. Accordingly, the nltk libraries are used for pre-processing. The regular expression (re) libraries are used to remove the non-alphabetical words from the raw text.

Building the Word2Vec Embedding Model

```
#importing the required libraries for building word2vec model
from gensim.utils import simple_preprocess
import gensim
from gensim.models import Word2Vec
import time
#importing the required libraries for keras pre-processing
from keras.preprocessing.text import Tokenizer
from keras.preprocessing.sequence import pad_sequences
from keras.utils import to_categorical
```

Fig 3.2 Feature extraction libraries for text

For text, the features are extracted using the above displayed libraries. The genism library is used for implemented the Word2Vec vector representation of the pre-processed text. The keras libraries are also used for the additional preprocessing of the of the processed text. Accordingly, the to_categorical library is used for converting the dependent variable into a single array matrix.

```
#Importing the required libraries
import numpy as np
import pandas as pd
import seaborn as sns
import librosa
import librosa.display
import matplotlib.pyplot as plt
import glob
import os
```

Fig 3.3 Pre-processing and feature extraction libraries for audio

Similarly, the audio files are pre-processed, and feature extracted using a Librosa library. The Librosa library comes with an MFCC function for extracting the MFCC coefficients and Librosa display for plotting the image representation of MFCCs.

4 Implementation

In this section, the required libraries for the model implementation are been displayed. For model building of text, a Bi-LSTM and a CNN model are built using the keras libraries displayed in Fig 4.1.

```
#Importing the required libraries for building the BiLSTM model
from keras.layers import Dropout, Dense, Embedding, LSTM, Bidirectional, Input, Conv1D
from keras.models import Sequential
from keras.callbacks import EarlyStopping
from keras.optimizers import Adam
#importing the required libraries for CNN model
from keras.layers import Dense, Input, GlobalMaxPooling1D
from keras.layers import Conv1D, MaxPooling1D, Embedding
from keras.models import Model
from keras.layers import Input, Dense, Embedding, Conv2D, MaxPooling2D, Dropout,concatenate
from keras.layers.core import Reshape, Flatten
from keras.callbacks import EarlyStopping
from keras.optimizers import Adam
from keras.models import Model
from keras import regularizers
```

Fig 4.1 Required libraries for model building of text

Accordingly, the LSTM and baseline machine learning models are implemented for audio emotion recognition. The following figure shows the required libraries for model building of audio analysis.

```
#Building an LSTM model with the required libraries
from keras.layers import Dropout, Dense, Embedding, LSTM
from keras.optimizers import SGD
from keras.models import Sequential
from keras.callbacks import EarlyStopping
from keras.optimizers import Adam
#Support Vector Machine Classifier
from sklearn.svm import SVC
model_SVM= SVC(kernel="rbf")
#Multinominal Naive Bayes
from sklearn.naive_bayes import MultinomialNB
model_naive=MultinomialNB()
#Decision Tree Classifier
from sklearn.tree import DecisionTreeClassifier
model_tree= DecisionTreeClassifier()
#Random Forest Classifier
from sklearn.ensemble import RandomForestClassifier
model_rm=RandomForestClassifier(n_estimators = 150)
#Adaboost Ensemble Technique
from sklearn.ensemble import AdaBoostClassifier
model_ada= AdaBoostClassifier()
```

Fig 4.2 Required libraries for model building of audio

5 Keras for deep learning

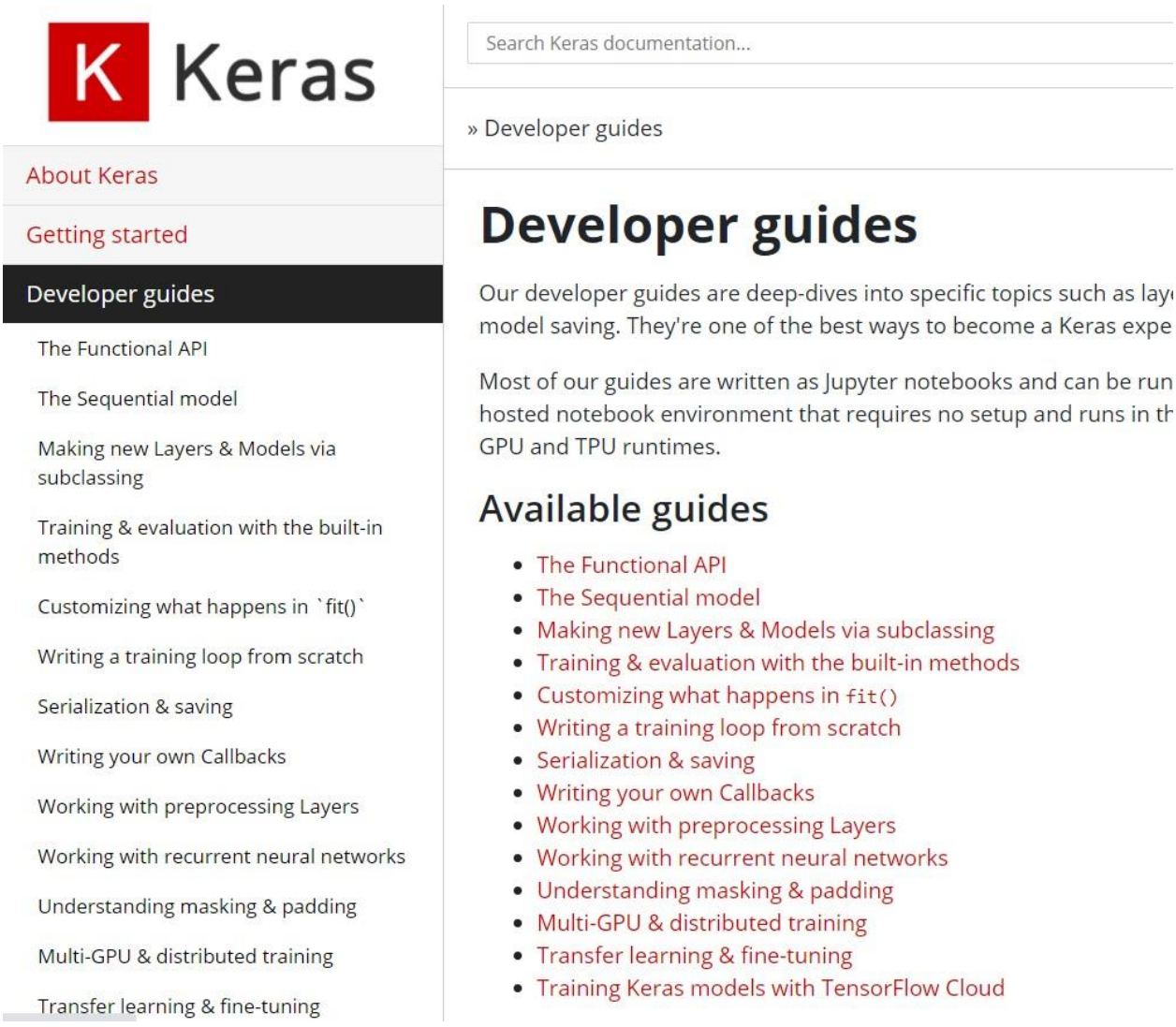

Fig 5. Keras Documentation

The keras model are built by following the guidelines from the Keras documentation [1] on the given official site. The documentation covers the model building for both Bi-LSTM, CNN, and LSTM models along with the hyperparameters and optimizer information.

References

[1] https://keras.io/guides/

[2] https://docs.anaconda.com/anaconda/install/windows/