# MIDAS - IIITD (Internship Task, 2021) Task - 1 (Speech Processing) - Detailed Explanation

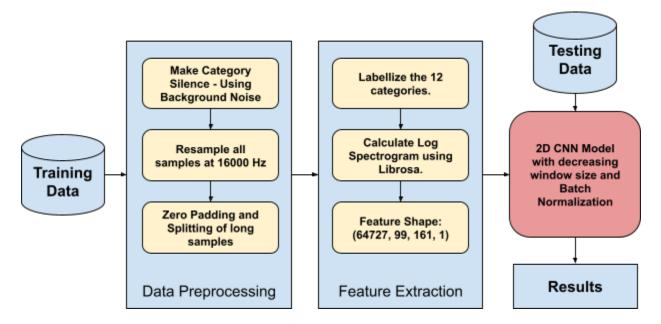
# 1. Task Introduction

In this competition, you're challenged to use the Speech Commands Dataset to build an algorithm that understands simple spoken commands. Dataset details are present at <u>Kaggle</u>.

# 2. Introduction to Proposed Solution

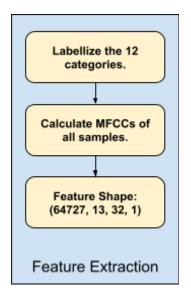
The following two flow charts represent the three proposed and implemented methodologies for the task.

#### 2a) Using Log Scaled Normalized Spetrograms:

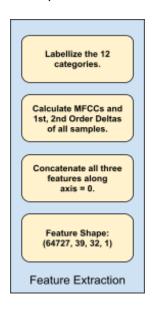


The other two methods differ only in the feature extraction methodologies shown below:

# 2b) Using MFCCs (First 13):



# 2c) Using MFCCs and 1st, 2nd Order Deltas (First 13):

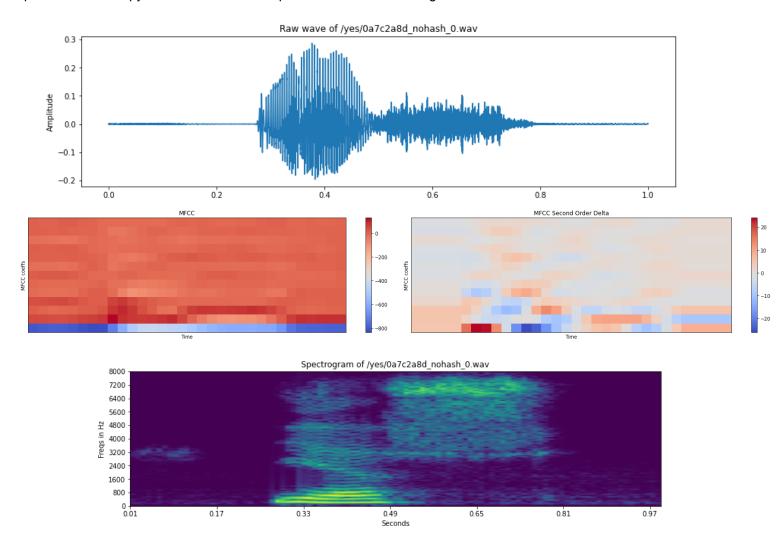


# 3. Exploratory Analysis of Dataset

To correctly describe and represent an audio sample, the following features were identified:

- Normalized Log Scaled Spectrogram of Audio Samples [1]
- Mel Scaled Cepstral Coefficients and their higher order deltas.
- Fourier Transform of Samples. [1]

The same are shown below for a random sample of audio, whose raw plot, sampled at 16000 Hz is shown below. Other plots are in the Jupyter Notbook which compare the same for all categories.



# 4. Details of Experiments Performed

# 4a) Using Normalized Log Spectrograms:

- 1. First, all samples were zero padded to 1 second samples to make sure that all samples are of equal dimensions. [2]
- 2. It was noticed that **many samples were greater than 1 second.** To tackle this, an idea was borrowed from [3] to split longer samples into new samples. This increases the training size also.
- 3. Normalized Log Specs, a computation of the FFT computed on overlapping windowed segments of the signals, are calculated using Librosa for all features.
- 4. The y-axis is converted to a log scale, and the color dimension is converted to decibels. **This is because humans** can only perceive a very small and concentrated range of frequencies and amplitudes. [4]

# 4b) Using 13 MFFC's:

- 1. It was noticed that the spectrograms did not give good results.
- 2. This could be attributed to the fact that the sample durations are too short for valuable feature extraction over a continuous domain.
- 3. MFCCs were selected since these are known to be similar to the perceived frequency scale in humans [5].
- 4. Only the first thirteen are selected, which is the norm in current research [6].
- 5. These gave much better results and are also computationally inexpensive.

# 4b) Using 13 MFFC's and 1st and 2nd Order Deltas:

- 1. To further improve the accuracy, the deltas of MFCCs, the first and second order derivatives, were concatenated to the MFCCs.
- 2. The benefit of deltas over MFCC features is that they are used to represent the temporal information. [7]
- 3. They all are concatenated to increase dimension size.

# 5. Model Details

- 2 Dimensional CNNs are used.
- Window sizes are varied from high to low while filters are increased.
- Binary Crossentropy Loss is used for training.
- Batch Normalization is added with dropouts to prevent overfitting.
- Models are shown below:

Model: "model"		
Layer (type)	Output Shape	Param #
	[(None, 99, 161, 1)]	0
batch_normalization_1 (Batch	(None, 99, 161, 1)	4
conv2d_4 (Conv2D)	(None, 95, 157, 8)	208
max_pooling2d_1 (MaxPooling2	(None, 47, 78, 8)	0
dropout_1 (Dropout)	(None, 47, 78, 8)	0
conv2d_5 (Conv2D)	(None, 45, 76, 16)	1168
conv2d_6 (Conv2D)	(None, 43, 74, 16)	2320
max_pooling2d_2 (MaxPooling2	(None, 21, 37, 16)	0
dropout_2 (Dropout)	(None, 21, 37, 16)	0
conv2d_7 (Conv2D)	(None, 20, 36, 32)	2080
conv2d_8 (Conv2D)	(None, 19, 35, 32)	4128
max_pooling2d_3 (MaxPooling2	(None, 9, 17, 32)	0
dropout_3 (Dropout)	(None, 9, 17, 32)	0
flatten (Flatten)	(None, 4896)	0
dense (Dense)	(None, 32)	156704
batch_normalization_2 (Batch	(None, 32)	128
dense_1 (Dense)	(None, 64)	2112
batch_normalization_3 (Batch	(None, 64)	256
dense_2 (Dense)	(None, 12)	780

Total params: 169,888 Trainable params: 169,694 Non-trainable params: 194 Model: "model\_1"

Layer (type)	Output Shape	Param #	
input_2 (InputLayer)	[(None, 13, 32, 1)]	0	
batch_normalization_6 (Batch	(None, 13, 32, 1)	4	
conv2d_18 (Conv2D)	(None, 9, 28, 8)	208	
dropout_6 (Dropout)	(None, 9, 28, 8)	0	
conv2d_19 (Conv2D)	(None, 7, 26, 16)	1168	
conv2d_20 (Conv2D)	(None, 5, 24, 16)	2320	
dropout_7 (Dropout)	(None, 5, 24, 16)	0	
conv2d_21 (Conv2D)	(None, 4, 23, 32)	2080	
conv2d_22 (Conv2D)	(None, 3, 22, 32)	4128	
dropout_8 (Dropout)	(None, 3, 22, 32)	0	
flatten_1 (Flatten)	(None, 2112)	0	
dense_3 (Dense)	(None, 32)	67616	
batch_normalization_7 (Batch	(None, 32)	128	
dense_4 (Dense)	(None, 64)	2112	
batch_normalization_8 (Batch	(None, 64)	256	
dense_5 (Dense)	(None, 12)	780	

Total params: 80,800 Trainable params: 80,606 Non-trainable params: 194

# 6. Results

Firstly, the training settings and parameters are shown for all three experiments in the below table:

Experiment	Validation Size	Training Convergence Time	Batch Size	Epochs
Spectrogram	15%	3500 Seconds	16	5 Epochs
MFCCs	15%	500 Seconds	32	10 Epochs
MFCCs and Higher Order Deltas	15%	2400 Seconds	32	20 Epochs

The saved models were tested in batches to preserve RAM and the results were as follows:

<sup>\*</sup> The Rank is calculated manually since the competition has expired and does not provide ranks.

Experiment	Validation Accuracy	Validation Loss	Accuracy (Public)	Accuracy (Private)	Rank *
Spectrogram	90.010%	0.0412	68.009%	69.023%	948
MFCCs	93.15%	0.0317	72.039%	72.136%	889
MFCCs and Higher Order Deltas	93.22%	0.0342	72.066%	73.017%	843

The below screenshot shows the results on Kaggle:

Submission1_Exp3_Deltas.csv	0.73017	0.72066	
12 minutes ago by Prabhav Singh			
add submission details			
submission_mfccs13.csv	0.72136	0.72039	
a day ago by Prabhav Singh			
MFCCs			
submission_logspec.csv	0.69023	0.68009	
a day ago by Prabhav Singh			
add submission details			

# 6. Discussion

- 1. It is clear from the results that, the use of MFCCs in conjugation with higher order deltas, gives the best results for the dataset.
- 2. This could be attributed to the **following reasons:** 
  - a. The samples are short and a continuous representation including windowing, like spectrograms, are not suitable for that.
  - b. The Deltas of MFCCs can represent temporal data well enough and hence outperform MFCCs alone.
  - c. MFCCs work upon the human peripheral auditory system. They possess human perception sensitivity with respect to frequencies and hence it is best to understand human speech. [7]
  - d. MFCCs are in general, a more compressible feature in respect to spectrograms. [7]
- 3. Further, one major advantage of MFCC is Since it is much smaller in feature dimensions, the model can be trained for a longer time with a higher batch size to facilitate better learning.
- 4. One flaw noticed is that the validation accuracy is much higher than the testing accuracy. This was due to a large imbalance in data points of the "UNKNOWN" class.

# 6. Other Possible Methods

- One method that was experimented with, was the use of Higher Statistical Features (HSFs) of Prosodic
  Features like spectral rollof, bandwidth, RMSE etc [8]. Used independently, they did not give good results but they
  could be used in multimodal fashion with MFCCs to improve learning.
- 2. Measures to prevent overfitting on the UNKNOWN class should be looked into. L2 Regularization and Oversampling could be used.
- 3. The use of Multimodal Input is worth exploring. For this, **Automatic Speaker Recognition (ASR)** could be used to extract the words itself and this could be used in conjugation with the MFCCs to give better results.

# 7. Reference In Text

- [1] David S., Speech representation and Data Exploration
- [2] Zero Padding and its Advantages
- [3] Splitting Samples into Smaller Segments, Qing Wei
- [4] Understanding the Mel Spectrogram, Leland Roberts
- [5] <u>Understanding MFCCs</u>, <u>Prateestha Nair</u>
- [6] MFCCs and its Features, Springer (Appendix)
- [7] Speaker Recognition Using MFCC and Delta Delta MFCCs with ANNs, Singh et al., IJARSE 2016
- [8] Significance of prosodic features for automatic emotion recognition, John et al., AIP 2020

# 7. Reference Used for Development of Code

- [1] Music Feature Extraction in Python, Saket Doshi
- [2] Musical Instrument Sound Classification using CNN, Muhammad Ardi