

Classification for Basic Voice Commands

HOW CAN DEEP LEARNING IMPACT SPEECH RECOGNITION?

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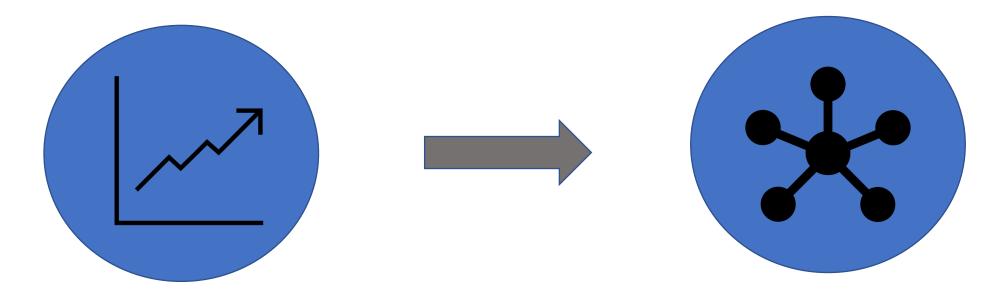
1. INTRODUCTION



ORIGINAL MESSAGE:

Have you tried scaling your data using a MinMaxScaler?

2. PROBLEM STATEMENT: Context

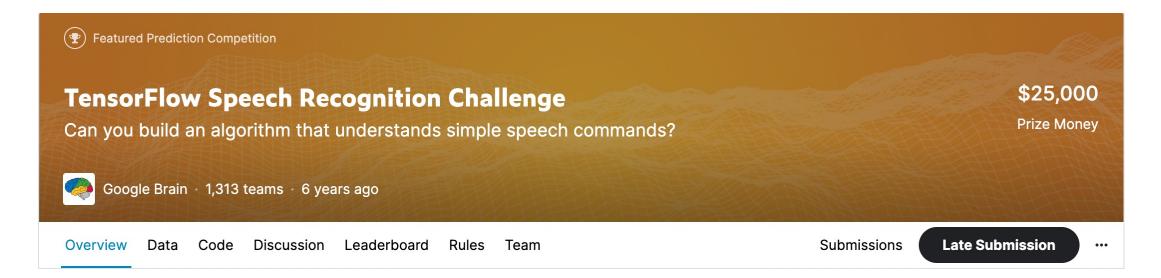


Rising demand for smart devices and voice-controlled applications

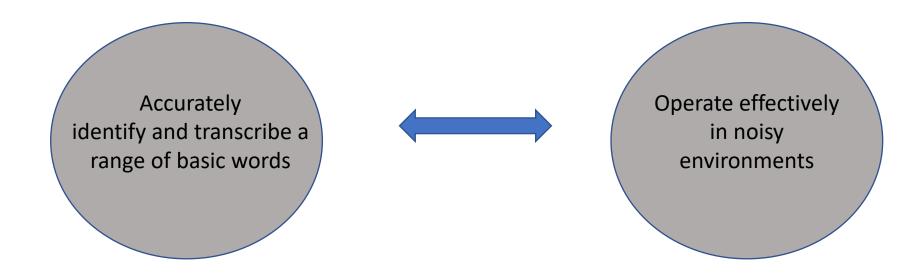
Need for **efficient** and **accurate** speech recognition technology

3. OUR PROJECT

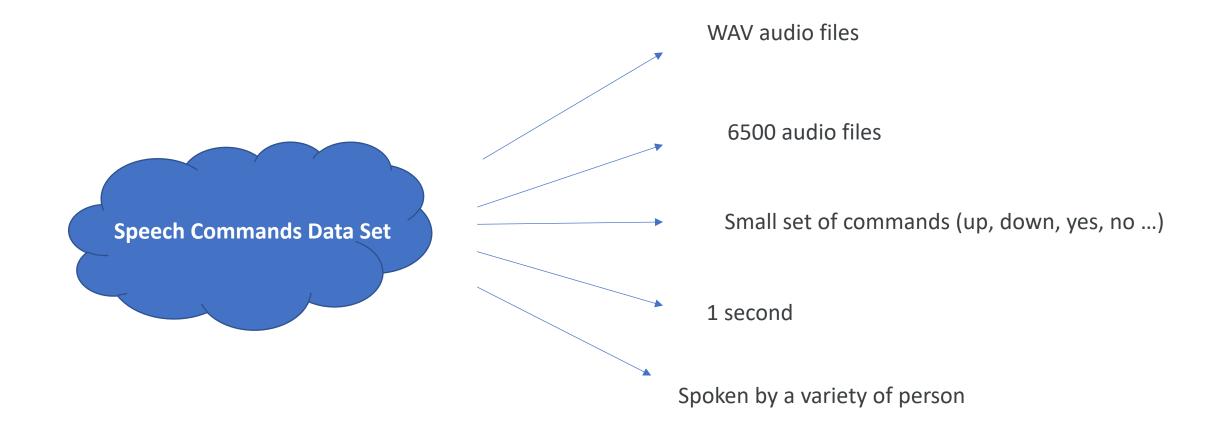
Kaggle Competition (ended Jan - 2018)



3. OUR PROJECT: Goals



4. DATA COLLECTION AND PREPROCESSING



4. DATA COLLECTION AND PREPROCESSING

13 commands

(yes, no, one, two, three, four, five, six, seven, eight, nine, up, down)



500 audio files for each command



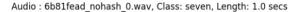
Added Background Noise

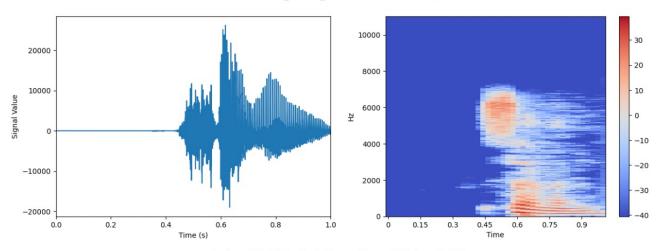
4. DATA COLLECTION AND PREPROCESSING: RAW AUDIO



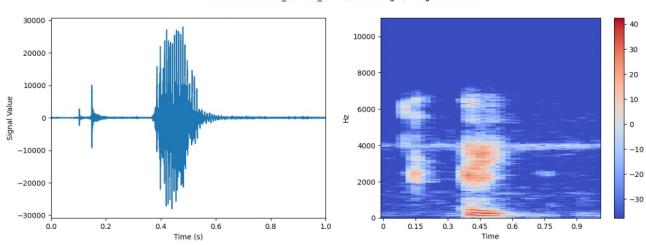
4. DATA COLLECTION AND PREPROCESSING: Waveform and Spectogram

STFT (Short-Time Fourier Transform)



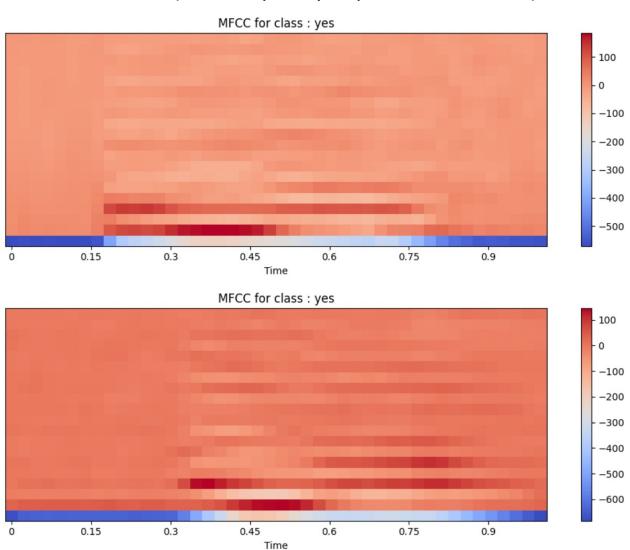


Audio: c634a189_nohash_3.wav, Class: eight, Length: 1.0 secs



4. DATA COLLECTION AND PREPROCESSING: MFCC

MFCC (Mel Frequency Cepstral Coefficients)



4. DATA COLLECTION AND PREPROCESSING: Challenges

PROBLEM



Initial Pre-processing

SOLUTION



LIBROSA library

5. MODEL DEVELOPMENT: TRAINING PROCESS

GENERAL INFO:

- Training Set, Validation Set, Test Set
- 50 epochs
- Accuracy Metric
- Adam Optimizer
- Learning Rate: 0.001

OUR MODEL:

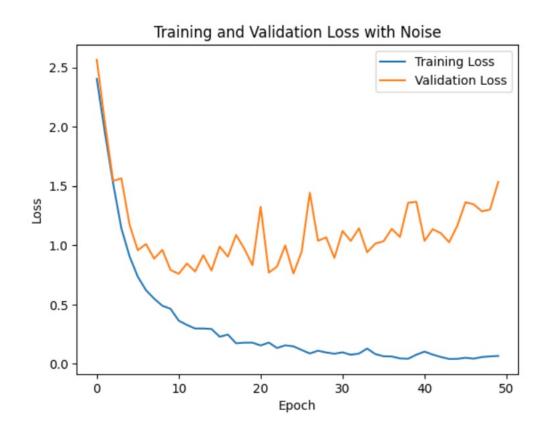
- Sequential model architecture
- Reshape, Conv2D, MaxPooling2D, Flatten, Dense, and Dropout layers
- Smaller number of layers compared to AlexNet
- Fewer trainable parameters compared to AlexNet

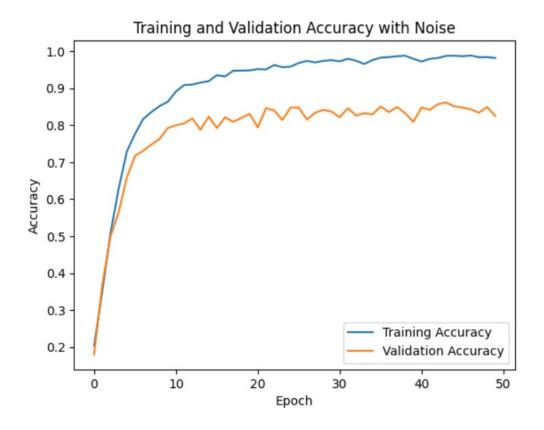
Model: "sequential 4"

Layer (type) ====================================	Output Shape	Param #
reshape_4 (Reshape)		0
conv2d_12 (Conv2D)	(None, 18, 30, 32)	320
max_pooling2d_9 (MaxPooling 2D)	(None, 9, 15, 32)	0
conv2d_13 (Conv2D)	(None, 7, 13, 64)	18496
max_pooling2d_10 (MaxPoolin g2D)	(None, 3, 6, 64)	0
conv2d_14 (Conv2D)	(None, 1, 4, 128)	73856
flatten_4 (Flatten)	(None, 512)	0
dense_12 (Dense)	(None, 128)	65664
dropout_8 (Dropout)	(None, 128)	0
dense_13 (Dense)	(None, 64)	8256
dropout_9 (Dropout)	(None, 64)	0
dense_14 (Dense)	(None, 13)	845

Total params: 167,437 Trainable params: 167,437 Non-trainable params: 0

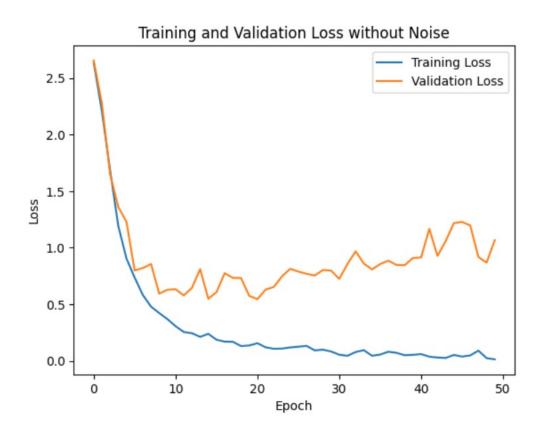
6. RESULTS AND EVALUATION: MFCC with noise

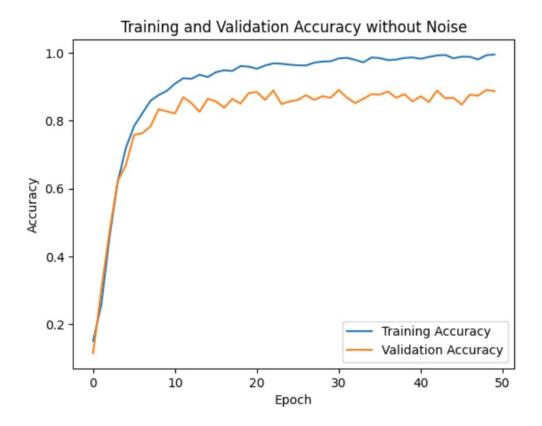




Validation Accuracy achieved: 0.86

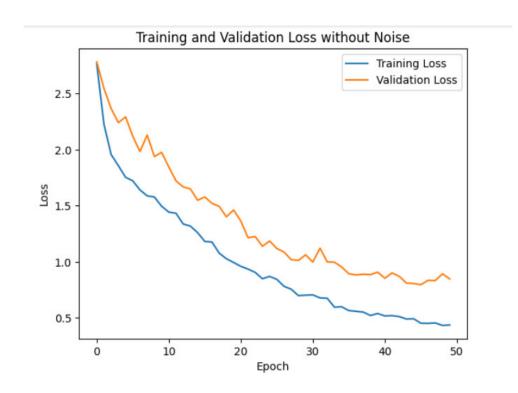
6. RESULTS AND EVALUATION: MFCC without noise

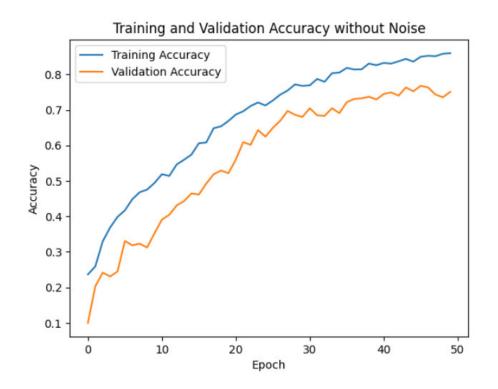




Validation Accuracy achieved: 0.88

6. RESULTS AND EVALUATION: Spectograms





Validation Accuracy achieved: 0.70

7. DEMO



http://172.16.0.57:8501/

8. REAL-WORLD APPLICATIONS



VOICE ASSISTANT



SMART HOME



CARS



GAMING

9. CONCLUSION

RESULTS

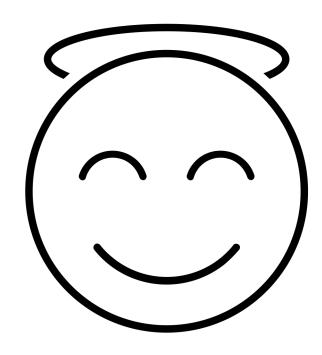
- Achieved very good accuracy with noise BUT overfits ...
- Very inspiring project!

NEXT STEPS

- Increase the different commands
- Try with sentences
- Increase background noise

Last Word!

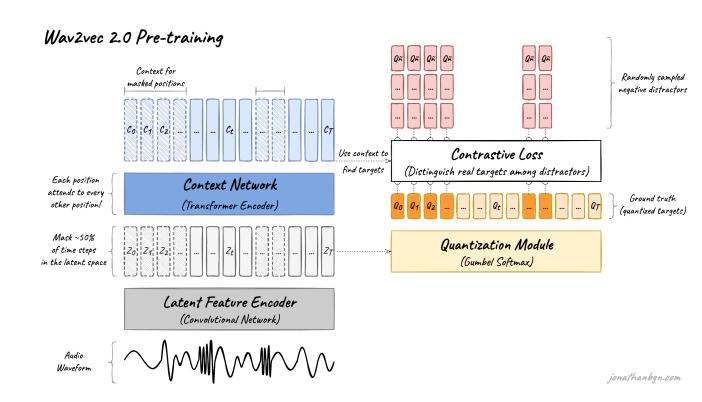
Wav2Vec (META)



ANNEXES

MODEL DEVELOPMENT: WAV2VEC

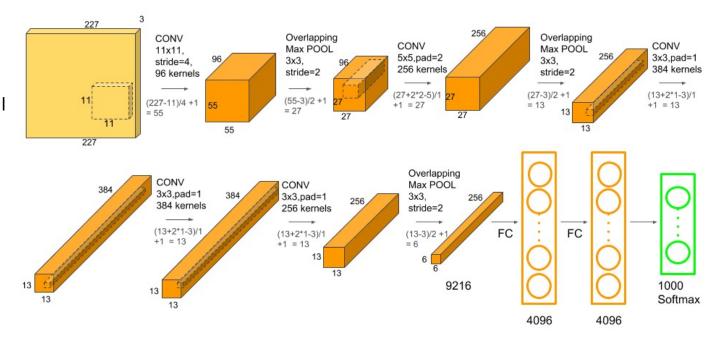
- Deep learning model for speech recognition and speech-related tasks
- State-of-the-art results in speech recognition benchmarks
- Handles raw audio data directly, no manual feature extraction needed
- Uses transformers to process CNN output for feature extraction
- Transformer models capture long-range dependencies and contextual information in audio sequences
- Pre-trained: 72% results



KEY METRIC: Levenshtein

MODEL DEVELOPMENT: AlexNet CNN

- Deep convolutional neural network architecture
- Multiple layers: convolutional, max-pooling, and fully connected
- Eight layers in total, with the first five being convolutional
- Convolutional layers extract low-level features
- Max-pooling layers downsample feature maps
- Fully connected layers serve as classifier
- ReLU activation functions used
- Dropout regularization implemented



KEY METRIC: Accuracy