## Digits Speech Recognition on Google Speech Commands

This notebook demonstrates how to predict spoken digits from audio files using a neural network model. The process involves the following steps:

- Preprocessing the audio files from dataset to extract MFCC features.
- Loading a trained model.
- Using the model to predict the digit for a given audio file.

Let's get started!

### **Imports and Setup**

We begin by importing the necessary libraries and setting up the environment for audio processing and model prediction.

```
In [ ]: import pathlib
        import seaborn
        from IPython import display
        from sklearn.preprocessing import LabelEncoder
        import numpy as np
        import os
        import tensorflow as tf
        import matplotlib.pyplot as plt
        from sklearn.model_selection import train_test_split
        from tensorflow.keras.models import Sequential
        from tensorflow.keras.layers import Dense, Conv2D, MaxPooling2D, Flatten, Dropout
        from tensorflow.keras.utils import to_categorical
        import librosa
        import librosa.display
In [ ]: from google.colab import drive
```

```
drive.mount('/content/drive')
```

Mounted at /content/drive

```
!kaggle datasets download -d neehakurelli/google-speech-commands --path /content/drive
```

```
Warning: Looks like you're using an outdated API Version, please consider updating (s erver 1.6.14 / client 1.6.12)

Dataset URL: https://www.kaggle.com/datasets/neehakurelli/google-speech-commands
License(s): unknown

Downloading google-speech-commands.zip to /content/drive/MyDrive/SpeechCommands
100% 1.38G/1.38G [00:24<00:00, 50.4MB/s]

100% 1.38G/1.38G [00:24<00:00, 60.0MB/s]
```

```
In [ ]: | import os
        import zipfile
        # Define the path to the zip file and the extraction directory
        zip_path = '/content/drive/MyDrive/SpeechCommands/google-speech-commands.zip'
        extract_dir = '/content/drive/MyDrive/SpeechCommands'
        # Check if the zip file exists
        if not os.path.exists(zip_path):
            print(f"Zip file not found at {zip_path}")
        else:
            try:
                # Attempt to open and extract the zip file
                with zipfile.ZipFile(zip_path, 'r') as zip_ref:
                     zip_ref.extractall(extract_dir)
                print(f"Extraction completed successfully to {extract_dir}")
            except zipfile.BadZipFile:
                 print(f"The file at {zip_path} is not a zip file or it is corrupted.")
            except Exception as e:
                 print(f"An error occurred: {e}")
```

Extraction completed successfully to /content/drive/MyDrive/SpeechCommands

```
In [ ]: extract_dir = '/content/drive/MyDrive/SpeechCommands'
# List the extracted files and folders
extracted_files = os.listdir(extract_dir)
print("Extracted files and folders:")
print(len(extracted_files))
```

Extracted files and folders: 36

## **Data Preprocessing**

This section covers the preprocessing of audio files to extract MFCC features. These features are essential for the model to make accurate predictions.

## Mel-Frequency Cepstral Coefficients (MFCC)

MFCC is a representation of the short-term power spectrum of a sound. It's widely used in speech and audio processing because it effectively captures the timbral aspects of the sound. The process involves:

1. Taking the Fourier transform of a windowed signal.

- 2. Mapping the powers of the spectrum to the mel scale.
- 3. Taking the logarithm of the powers at each mel frequency.
- 4. Taking the discrete cosine transform of the resulting coefficients.

## **Loading and Preprocessing the Dataset**

In this section, we load and preprocess the dataset. The dataset contains audio recordings of spoken digits ("zero" to "nine"). The preprocessing steps include:

- 1. Loading each audio file.
- 2. Extracting MFCC features from the audio.
- 3. Ensuring consistent shape of the MFCC features.

```
In [ ]: DATASET_PATH = '/content/drive/My Drive/SpeechCommands'
        DIGITS = ['zero', 'one', 'two', 'three', 'four', 'five', 'six', 'seven', 'eight', 'nir
        def load data(path, digits):
            x_data, y_data = [], []
            for i, digit in enumerate(digits):
                digit_path = os.path.join(path, digit)
                for filename in os.listdir(digit_path):
                    if filename.endswith('.wav'):
                        file_path = os.path.join(digit_path, filename)
                         y, sr = librosa.load(file_path, sr=16000)
                        mfcc = librosa.feature.mfcc(y=y, sr=sr, n_mfcc=40)
                         if mfcc.shape[1] == 32: # Ensure consistent shape
                            x data.append(mfcc)
                            y_data.append(i)
            return np.array(x_data), np.array(y_data)
        x data, y data = load data(DATASET PATH, DIGITS)
        x_data = x_data[..., np.newaxis] # Add channel dimension
        y_data = to_categorical(y_data, num_classes=len(DIGITS))
        # Split data into training and test sets
        x_train, x_test, y_train, y_test = train_test_split(x_data, y_data, test_size=0.2, ran
```

#### **Model Architecture**

The model used for digit recognition is a Convolutional Neural Network (CNN). Here's a detailed breakdown of each layer:

• Conv2D (32, (3, 3), activation='relu', input\_shape=(40, 32, 1)): The first convolutional layer with 32 filters, a kernel size of (3, 3), ReLU activation, and an input shape matching the MFCC feature dimensions.

• MaxPooling2D ((2, 2)): A max pooling layer with a pool size of (2, 2) to reduce the spatial dimensions.

- **Dropout (0.25):** A dropout layer with a dropout rate of 0.25 to prevent overfitting.
- **Conv2D (64, (3, 3), activation='relu'):** The second convolutional layer with 64 filters and a kernel size of (3, 3), again with ReLU activation.
- MaxPooling2D ((2, 2)): Another max pooling layer with a pool size of (2, 2).
- **Dropout (0.25):** Another dropout layer with a dropout rate of 0.25 to further prevent overfitting.
- **Flatten ():** A flatten layer to convert the 2D matrix into a 1D vector for the fully connected layers.
- **Dense (128, activation='relu'):** A dense (fully connected) layer with 128 units and ReLU activation.
- **Dropout (0.5):** A dropout layer with a higher dropout rate of 0.5 for more aggressive regularization.
- **Dense (len(DIGITS), activation='softmax'):** The output layer with a number of units equal to the number of digit classes and softmax activation to output class probabilities.

The model is compiled with the Adam optimizer, categorical crossentropy loss function, and accuracy as the evaluation metric.

Model: "sequential"

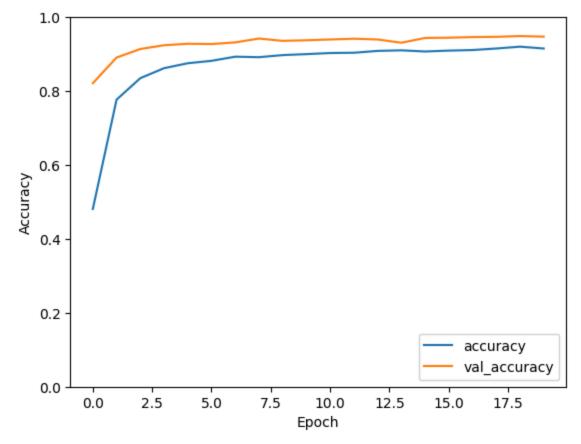
Layer (type)	Output Shape	Param #
conv2d (Conv2D)	(None, 38, 30, 32)	320
<pre>max_pooling2d (MaxPooling2 D)</pre>	(None, 19, 15, 32)	0
dropout (Dropout)	(None, 19, 15, 32)	0
conv2d_1 (Conv2D)	(None, 17, 13, 64)	18496
<pre>max_pooling2d_1 (MaxPoolin g2D)</pre>	(None, 8, 6, 64)	0
dropout_1 (Dropout)	(None, 8, 6, 64)	0
flatten (Flatten)	(None, 3072)	0
dense (Dense)	(None, 128)	393344
dropout_2 (Dropout)	(None, 128)	0
dense_1 (Dense)	(None, 10)	1290

Total params: 413450 (1.58 MB)
Trainable params: 413450 (1.58 MB)
Non-trainable params: 0 (0.00 Byte)

In [ ]: history = model.fit(x\_train, y\_train, epochs=20, batch\_size=32, validation\_data=(x\_tes

```
Epoch 1/20
0.1029 - val_loss: 2.3032 - val_accuracy: 0.0888
Epoch 2/20
540/540 [================== ] - 30s 56ms/step - loss: 2.3029 - accuracy:
0.1043 - val_loss: 2.3035 - val_accuracy: 0.0895
Epoch 3/20
540/540 [=======================] - 29s 54ms/step - loss: 2.2943 - accuracy:
0.1169 - val_loss: 2.2710 - val_accuracy: 0.1194
Epoch 4/20
0.1846 - val_loss: 1.9355 - val_accuracy: 0.3123
Epoch 5/20
0.4349 - val loss: 0.8964 - val accuracy: 0.7090
Epoch 6/20
0.6430 - val_loss: 0.5257 - val_accuracy: 0.8535
Epoch 7/20
0.7375 - val loss: 0.4084 - val accuracy: 0.8831
0.7779 - val_loss: 0.3540 - val_accuracy: 0.9005
Epoch 9/20
0.7956 - val_loss: 0.3282 - val_accuracy: 0.9022
Epoch 10/20
540/540 [======================] - 31s 57ms/step - loss: 0.5698 - accuracy:
0.8157 - val_loss: 0.3112 - val_accuracy: 0.9068
Epoch 11/20
540/540 [======================] - 31s 58ms/step - loss: 0.5188 - accuracy:
0.8318 - val_loss: 0.2810 - val_accuracy: 0.9170
Epoch 12/20
0.8364 - val_loss: 0.2813 - val_accuracy: 0.9172
Epoch 13/20
540/540 [================= ] - 31s 58ms/step - loss: 0.4777 - accuracy:
0.8474 - val_loss: 0.2662 - val_accuracy: 0.9195
Epoch 14/20
0.8525 - val_loss: 0.2500 - val_accuracy: 0.9251
Epoch 15/20
540/540 [=======================] - 31s 57ms/step - loss: 0.4623 - accuracy:
0.8526 - val_loss: 0.2617 - val_accuracy: 0.9200
Epoch 16/20
0.8635 - val_loss: 0.2556 - val_accuracy: 0.9214
Epoch 17/20
0.8688 - val_loss: 0.2367 - val_accuracy: 0.9267
Epoch 18/20
0.8714 - val_loss: 0.2412 - val_accuracy: 0.9267
Epoch 19/20
0.8743 - val_loss: 0.2374 - val_accuracy: 0.9297
Epoch 20/20
540/540 [=======================] - 29s 54ms/step - loss: 0.3934 - accuracy:
0.8747 - val_loss: 0.2427 - val_accuracy: 0.9291
```

Test accuracy: 93.62%



```
In [ ]: model.save('/content/drive/My Drive/digit_speech_recognition_TRAILmodel.h5')
In [ ]: !pip install wandb
```

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Requirement already satisfied: wandb in /usr/local/lib/python3.10/dist-packages (0.1 7.0)

Requirement already satisfied: click!=8.0.0,>=7.1 in /usr/local/lib/python3.10/dist-p ackages (from wandb) (8.1.7)

Requirement already satisfied: docker-pycreds>=0.4.0 in /usr/local/lib/python3.10/dist-packages (from wandb) (0.4.0)

Requirement already satisfied: gitpython!=3.1.29,>=1.0.0 in /usr/local/lib/python3.1 0/dist-packages (from wandb) (3.1.43)

Requirement already satisfied: platformdirs in /usr/local/lib/python3.10/dist-package s (from wandb) (4.2.1)

Requirement already satisfied: protobuf!=4.21.0,<5,>=3.19.0 in /usr/local/lib/python 3.10/dist-packages (from wandb) (3.20.3)

Requirement already satisfied: psutil>=5.0.0 in /usr/local/lib/python3.10/dist-packag es (from wandb) (5.9.5)

Requirement already satisfied: pyyaml in /usr/local/lib/python3.10/dist-packages (from wandb) (6.0.1)

Requirement already satisfied: requests<3,>=2.0.0 in /usr/local/lib/python3.10/dist-p ackages (from wandb) (2.31.0)

Requirement already satisfied: sentry-sdk>=1.0.0 in /usr/local/lib/python3.10/dist-pa ckages (from wandb) (2.2.0)

Requirement already satisfied: setproctitle in /usr/local/lib/python3.10/dist-package s (from wandb) (1.3.3)

Requirement already satisfied: setuptools in /usr/local/lib/python3.10/dist-packages (from wandb) (67.7.2)

Requirement already satisfied: six>=1.4.0 in /usr/local/lib/python3.10/dist-packages (from docker-pycreds>=0.4.0->wandb) (1.16.0)

Requirement already satisfied: gitdb<5,>=4.0.1 in /usr/local/lib/python3.10/dist-pack ages (from gitpython!=3.1.29,>=1.0.0->wandb) (4.0.11)

Requirement already satisfied: charset-normalizer<4,>=2 in /usr/local/lib/python3.10/dist-packages (from requests<3,>=2.0.0->wandb) (3.3.2)

Requirement already satisfied: idna<4,>=2.5 in /usr/local/lib/python3.10/dist-package s (from requests<3,>=2.0.0->wandb) (3.7)

Requirement already satisfied: urllib3<3,>=1.21.1 in /usr/local/lib/python3.10/dist-p ackages (from requests<3,>=2.0.0->wandb) (2.0.7)

Requirement already satisfied: certifi>=2017.4.17 in /usr/local/lib/python3.10/dist-p ackages (from requests<3,>=2.0.0->wandb) (2024.2.2)

Requirement already satisfied: smmap<6,>=3.0.1 in /usr/local/lib/python3.10/dist-pack ages (from gitdb<5,>=4.0.1->gitpython!=3.1.29,>=1.0.0->wandb) (5.0.1)

# Using Weights & Biases (wandb) for Experiment Tracking

Weights & Biases (wandb) is a tool for tracking and visualizing machine learning experiments. It provides an easy way to log various metrics, visualize model performance, and compare different runs. Here's a detailed explanation of how wandb is used in this notebook:

- **Tracking Metrics:** wandb allows you to track important metrics like training and validation accuracy, loss, and more in real-time. This helps you monitor the progress of your model training and detect any issues early on.
- **Logging Hyperparameters:** You can log hyperparameters used in your experiments, such as learning rate, batch size, and number of epochs. This makes it easy to keep track of different configurations and their effects on model performance.

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Visualizing Performance: wandb provides interactive plots and charts that help you
visualize the performance of your model over time. This includes plots for accuracy,
loss, and other custom metrics.

- **Comparing Runs:** With wandb, you can compare different training runs side-by-side. This is particularly useful for hyperparameter tuning and understanding the impact of different changes to your model or training process.
- **Collaboration:** wandb facilitates collaboration by allowing you to share your experiments and results with your team. You can easily access and review the results from different runs, making it easier to work together on improving the model.

In this notebook, we use wandb.keras.WandbCallback() as a callback in the model2.fit() function. This callback integrates wandb with Keras, automatically logging metrics and parameters for each epoch during the training process.

Run data is saved locally in /content/wandb/run-20240519\_100205-qbn7fd1w Syncing run swift-galaxy-1 to Weights & Biases (docs)

View project at https://wandb.ai/alexandria-university\_foe/digit-speech-recognition

View run at https://wandb.ai/alexandria-university\_foe/digit-speech-recognition/runs/qbn7fd1w

Out[ ]: Display W&B run

import wandb

In [ ]:

## **Changes in Model Architecture**

Compared to the previous model, model2 incorporates several significant changes to potentially improve performance:

- **Increased Filters:** The number of filters in each convolutional layer has been increased:
  - The first Conv2D layer now has 64 filters instead of 32.
  - The second Conv2D layer now has 128 filters instead of 64.
  - A new third Conv2D layer has been added with 256 filters.
- Additional Convolutional Layer: An additional convolutional layer with 256 filters
  has been added to increase the model's capacity to capture more complex features
  from the input data.

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• **Increased Units in Dense Layer:** The number of units in the fully connected Dense layer has been increased from 128 to 512, allowing the model to learn more complex representations before the final classification layer.

These changes are intended to enhance the model's ability to learn from the data by providing more parameters and depth, which can be beneficial for capturing intricate patterns in the audio features.

```
In [ ]: model2.compile(optimizer='adam', loss='categorical_crossentropy', metrics=['accuracy']
    model.summary()
```

Param #

Output Shape

Model: "sequential"

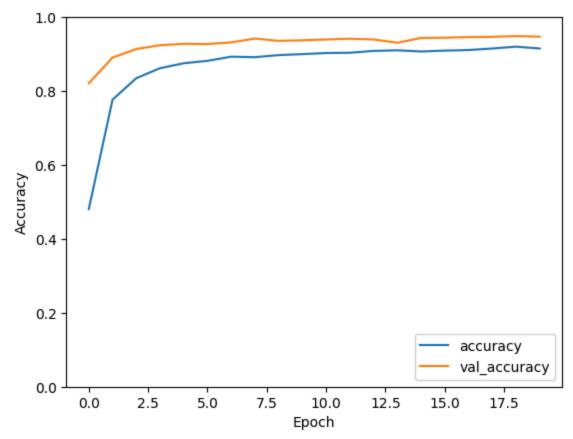
Layer (type)

```
______
        conv2d (Conv2D)
                                (None, 38, 30, 32)
                                                      320
        max pooling2d (MaxPooling2 (None, 19, 15, 32)
                                                      0
        D)
        dropout (Dropout)
                                (None, 19, 15, 32)
        conv2d_1 (Conv2D)
                                (None, 17, 13, 64)
                                                      18496
        max_pooling2d_1 (MaxPoolin (None, 8, 6, 64)
        g2D)
        dropout_1 (Dropout)
                                (None, 8, 6, 64)
                                                      0
        flatten (Flatten)
                                (None, 3072)
        dense (Dense)
                                (None, 128)
                                                      393344
        dropout_2 (Dropout)
                                (None, 128)
        dense 1 (Dense)
                                (None, 10)
                                                      1290
       ______
       Total params: 413450 (1.58 MB)
       Trainable params: 413450 (1.58 MB)
       Non-trainable params: 0 (0.00 Byte)
In [ ]: history = model2.fit(
          x_train, y_train,
          epochs=20,
          batch_size=32,
          validation_data=(x_test, y_test),
           callbacks=[wandb.keras.WandbCallback()]
       )
       Epoch 1/20
       /usr/local/lib/python3.10/dist-packages/keras/src/engine/training.py:3103: UserWarnin
       g: You are saving your model as an HDF5 file via `model.save()`. This file format is
       considered legacy. We recommend using instead the native Keras format, e.g. `model.sa
       ve('my_model.keras')`.
         saving_api.save_model(
       wandb: Adding directory to artifact (/content/wandb/run-20240519 100205-qbn7fd1w/file
       s/model-best)... Done. 0.1s
       0.4810 - val_loss: 0.5848 - val_accuracy: 0.8212
       Epoch 2/20
       539/540 [==========================>.] - ETA: 0s - loss: 0.6819 - accuracy: 0.7767
       /usr/local/lib/python3.10/dist-packages/keras/src/engine/training.py:3103: UserWarnin
       g: You are saving your model as an HDF5 file via `model.save()`. This file format is
       considered legacy. We recommend using instead the native Keras format, e.g. `model.sa
       ve('my_model.keras')`.
         saving_api.save_model(
       wandb: Adding directory to artifact (/content/wandb/run-20240519 100205-qbn7fd1w/file
       s/model-best)... Done. 0.1s
```

```
540/540 [======================] - 102s 188ms/step - loss: 0.6819 - accuracy:
0.7767 - val loss: 0.3380 - val accuracy: 0.8906
Epoch 3/20
/usr/local/lib/python3.10/dist-packages/keras/src/engine/training.py:3103: UserWarnin
g: You are saving your model as an HDF5 file via `model.save()`. This file format is
considered legacy. We recommend using instead the native Keras format, e.g. `model.sa
ve('my model.keras')`.
 saving_api.save_model(
wandb: Adding directory to artifact (/content/wandb/run-20240519_100205-qbn7fd1w/file
s/model-best)... Done. 0.1s
0.8348 - val loss: 0.2700 - val accuracy: 0.9137
Epoch 4/20
/usr/local/lib/python3.10/dist-packages/keras/src/engine/training.py:3103: UserWarnin
g: You are saving your model as an HDF5 file via `model.save()`. This file format is
considered legacy. We recommend using instead the native Keras format, e.g. `model.sa
ve('my model.keras')`.
 saving_api.save_model(
wandb: Adding directory to artifact (/content/wandb/run-20240519_100205-qbn7fd1w/file
s/model-best)... Done. 0.1s
0.8618 - val_loss: 0.2448 - val_accuracy: 0.9240
Epoch 5/20
/usr/local/lib/python3.10/dist-packages/keras/src/engine/training.py:3103: UserWarnin
g: You are saving your model as an HDF5 file via `model.save()`. This file format is
considered legacy. We recommend using instead the native Keras format, e.g. `model.sa
ve('my model.keras')`.
 saving_api.save_model(
wandb: Adding directory to artifact (/content/wandb/run-20240519_100205-qbn7fd1w/file
s/model-best)... Done. 0.1s
0.8752 - val_loss: 0.2400 - val_accuracy: 0.9279
Epoch 6/20
540/540 [=======================] - 94s 174ms/step - loss: 0.3765 - accuracy:
0.8817 - val_loss: 0.2415 - val_accuracy: 0.9272
Epoch 7/20
/usr/local/lib/python3.10/dist-packages/keras/src/engine/training.py:3103: UserWarnin
g: You are saving your model as an HDF5 file via `model.save()`. This file format is
considered legacy. We recommend using instead the native Keras format, e.g. `model.sa
ve('my_model.keras')`.
 saving api.save model(
wandb: Adding directory to artifact (/content/wandb/run-20240519_100205-qbn7fd1w/file
s/model-best)... Done. 0.1s
0.8929 - val_loss: 0.2224 - val_accuracy: 0.9316
Epoch 8/20
/usr/local/lib/python3.10/dist-packages/keras/src/engine/training.py:3103: UserWarnin
g: You are saving your model as an HDF5 file via `model.save()`. This file format is
considered legacy. We recommend using instead the native Keras format, e.g. `model.sa
ve('my model.keras')`.
 saving api.save model(
wandb: Adding directory to artifact (/content/wandb/run-20240519_100205-qbn7fd1w/file
s/model-best)... Done. 0.1s
```

```
540/540 [==================] - 95s 176ms/step - loss: 0.3462 - accuracy:
0.8916 - val loss: 0.1895 - val accuracy: 0.9420
Epoch 9/20
0.8972 - val_loss: 0.2137 - val_accuracy: 0.9358
Epoch 10/20
540/540 [================= ] - 92s 171ms/step - loss: 0.3301 - accuracy:
0.8999 - val_loss: 0.2142 - val_accuracy: 0.9374
Epoch 11/20
0.9028 - val loss: 0.1973 - val accuracy: 0.9395
Epoch 12/20
0.9036 - val_loss: 0.1983 - val_accuracy: 0.9416
Epoch 13/20
0.9086 - val_loss: 0.1985 - val_accuracy: 0.9395
Epoch 14/20
540/540 [=======================] - 95s 177ms/step - loss: 0.3083 - accuracy:
0.9102 - val_loss: 0.2212 - val_accuracy: 0.9307
Epoch 15/20
540/540 [=================] - 94s 174ms/step - loss: 0.3146 - accuracy:
0.9070 - val_loss: 0.1907 - val_accuracy: 0.9437
Epoch 16/20
0.9094 - val_loss: 0.1954 - val_accuracy: 0.9444
Epoch 17/20
/usr/local/lib/python3.10/dist-packages/keras/src/engine/training.py:3103: UserWarnin
g: You are saving your model as an HDF5 file via `model.save()`. This file format is
considered legacy. We recommend using instead the native Keras format, e.g. `model.sa
ve('my_model.keras')`.
 saving api.save model(
wandb: Adding directory to artifact (/content/wandb/run-20240519_100205-qbn7fd1w/file
s/model-best)... Done. 0.1s
0.9110 - val_loss: 0.1821 - val_accuracy: 0.9460
Epoch 18/20
539/540 [==========================>.] - ETA: 0s - loss: 0.2862 - accuracy: 0.9151
/usr/local/lib/python3.10/dist-packages/keras/src/engine/training.py:3103: UserWarnin
g: You are saving your model as an HDF5 file via `model.save()`. This file format is
considered legacy. We recommend using instead the native Keras format, e.g. `model.sa
ve('my model.keras')`.
 saving_api.save_model(
wandb: Adding directory to artifact (/content/wandb/run-20240519_100205-qbn7fd1w/file
s/model-best)... Done. 0.1s
0.9151 - val_loss: 0.1745 - val_accuracy: 0.9467
Epoch 19/20
/usr/local/lib/python3.10/dist-packages/keras/src/engine/training.py:3103: UserWarnin
g: You are saving your model as an HDF5 file via `model.save()`. This file format is
considered legacy. We recommend using instead the native Keras format, e.g. `model.sa
ve('my_model.keras')`.
 saving_api.save_model(
wandb: Adding directory to artifact (/content/wandb/run-20240519_100205-qbn7fd1w/file
s/model-best)... Done. 0.1s
```

```
540/540 [=================] - 99s 184ms/step - loss: 0.2710 - accuracy:
       0.9201 - val_loss: 0.1697 - val_accuracy: 0.9488
       Epoch 20/20
       540/540 [=======================] - 99s 182ms/step - loss: 0.2938 - accuracy:
       0.9151 - val_loss: 0.2013 - val_accuracy: 0.9471
In [ ]: loss, accuracy = model.evaluate(x_test, y_test)
       print(f"Test accuracy: {accuracy*100:.2f}%")
       9362
       Test accuracy: 93.62%
       plt.plot(history.history['accuracy'], label='accuracy')
In [ ]:
       plt.plot(history.history['val_accuracy'], label = 'val_accuracy')
       plt.xlabel('Epoch')
       plt.ylabel('Accuracy')
       plt.ylim([0, 1])
       plt.legend(loc='lower right')
       plt.show()
```



In [ ]: model.save('/content/drive/MyDrive/SavedSpeechModel/digit\_speech\_recognition\_Finalmode

#### **Audio Preprocessing and Digit Prediction**

This section loads the previous pre-trained digit recognition model and defines functions to preprocess audio files and predict the corresponding digit:

- **Preprocess Audio:** Loads an audio file, extracts MFCC features, resizes them to a consistent shape, and prepares them for model input.
- **Predict Digit:** Uses the preprocessed audio to predict the digit with the trained model.

```
In [ ]: import numpy as np
        import tensorflow as tf
        import librosa
        import librosa.display
        # Load the trained model
        model = tf.keras.models.load_model('/content/drive/MyDrive/Lab6_number0 to 9 using CNN
        # Function to preprocess audio
        def preprocess_audio(audio_path, sample_rate=16000, expected_shape=(40, 32, 1)):
            y, sr = librosa.load(audio_path, sr=sample_rate)
            mfcc = librosa.feature.mfcc(y=y, sr=sr, n_mfcc=40)
            # Check the current shape
            print(f"Original MFCC shape: {mfcc.shape}")
            # Resize the MFCC to the expected shape
            if mfcc.shape[1] < expected_shape[1]:</pre>
                pad_width = expected_shape[1] - mfcc.shape[1]
                mfcc = np.pad(mfcc, ((0, 0), (0, pad_width)), mode='constant')
            else:
                mfcc = mfcc[:, :expected_shape[1]]
            # Check the new shape
            print(f"Resized MFCC shape: {mfcc.shape}")
            mfcc = mfcc[np.newaxis, ..., np.newaxis] # Add batch and channel dimension
            return mfcc
        # Function to predict digit
        def predict_digit(audio_path):
            preprocessed audio = preprocess audio(audio path)
            prediction = model.predict(preprocessed_audio)
            predicted_digit = np.argmax(prediction)
             return predicted_digit
```

#### **Predictions**

Finally, we predict the digits for a few test audio files and print the results.

```
In [ ]: # Predict the digit for the uploaded audio file
    audio_file = 'test_audio9.wav' # Path to the uploaded audio file
    predicted_digit = predict_digit(audio_file)
    print("Predicted Digit:", predicted_digit)
```

```
Original MFCC shape: (40, 75)
       Resized MFCC shape: (40, 32)
       1/1 [=======] - 0s 99ms/step
       Predicted Digit: 9
In [ ]: # Predict the digit for the uploaded audio file
       audio_file = 'test_audio1.wav' # Path to the uploaded audio file
       predicted_digit = predict_digit(audio_file)
       print("Predicted Digit:", predicted_digit)
       Original MFCC shape: (40, 57)
       Resized MFCC shape: (40, 32)
       1/1 [=======] - 0s 71ms/step
       Predicted Digit: 1
In [ ]: # Predict the digit for the uploaded audio file
       audio_file = 'test_audio7.wav' # Path to the uploaded audio file
       predicted_digit = predict_digit(audio_file)
       print("Predicted Digit:", predicted_digit)
       Original MFCC shape: (40, 59)
       Resized MFCC shape: (40, 32)
       1/1 [=======] - 0s 150ms/step
       Predicted Digit: 7
```