#### Class:

# **Downsampling Code Explanation – For the Teacher**

**Downsampling** is a technique used to reduce the number of data points in a dataset. This is particularly useful when dealing with large datasets, such as EMG signals, to make them more manageable and to speed up the analysis. Downsampling reduces the **sampling frequency** by selecting fewer data points while maintaining the integrity of the overall pattern or trend. This is often done by skipping over certain samples in the original dataset.

# Why is Downsampling Important?

- Data Reduction: It reduces the data size, making it easier to process and analyze.
- **Noise Reduction**: Downsampling can help reduce noise by smoothing out small variations that may not be relevant for the analysis.
- **Faster Processing**: Smaller datasets require less computational power, which is beneficial when performing analyses such as machine learning or statistical testing.

# **How Downsampling Works**

When we downsample, we essentially keep data points from the original signal at a reduced frequency. For example, if we start with a signal sampled at 44,100 Hz (samples per second), and we downsample it to 8,000 Hz, we keep every fifth data point to match the new frequency.

# **Downsampling Script Explanation**

The following Python script demonstrates how to downsample data using **SciPy** and **NumPy**. The students will be working with this script to apply the downsampling technique to their EMG data.

# Python Script for Downsampling (In the Jupyter Notebook)

```
import numpy as np
from scipy.io import wavfile
import matplotlib.pyplot as plt
# Downsample to 8000Hz
downsampled_data = resample(data, int(len(data) * (8000 /
sample rate))) # Resample the data to 8000 Hz
duration downsampled = len(downsampled data) / 8000 # Calculate the
duration of the downsampled audio
time axis downsampled = np.linspace(0, duration downsampled,
len(downsampled data)) # Create a time axis for the downsampled data
df_downsampled = pd.DataFrame({'Time (seconds)': time_axis_downsampled,
'Amplitude': downsampled data}) # Create a DataFrame with downsampled
time and amplitude
df_downsampled.to csv('/content/drive/MyDrive/Colab
Notebooks/outputFile downsampled.csv', index=False) # Save the
downsampled DataFrame as a CSV file
```





## Explanation of the Script:

```
downsampled_data = resample(data, int(len(data) * (8000 / sample_rate)))
# Resample the data to 8000 Hz
```

- **Explanation**: This line uses the *resample* function (likely from scipy.signal or another library) to downsample the data.
  - **data**: This is the original signal that was loaded (e.g., a muscle signal from the .wav file).
  - **sample\_rate**: The original sampling rate of the data (for example, 44,100 Hz).
  - o len(data): This returns the number of data points (samples) in the original signal.
  - (8000 / sample\_rate): This is the ratio of the target sampling rate (8000 Hz) to the original sampling rate. This ratio determines how many samples should be kept in the downsampled signal.
  - int(len(data) \* (8000 / sample\_rate)): The total number of samples to keep after downsampling. This calculates the new length of the signal based on the target rate of 8000 Hz.
  - **resample()**: This function resamples the original data to the target number of samples, effectively reducing the frequency.
  - **Result**: The downsampled\_data now holds the resampled signal at 8000 Hz.

duration\_downsampled = len(downsampled\_data) / 8000 # Calculate the duration of the downsampled audio

- **Explanation**: This line calculates the **duration** (in seconds) of the downsampled audio signal.
  - **len(downsampled\_data)**: This returns the number of samples in the downsampled signal.
  - **8000**: This is the new sampling rate (8000 samples per second).
  - len(downsampled\_data) / 8000: The duration of the downsampled audio is calculated by dividing the number of samples in the downsampled signal by the sampling rate (8000 samples per second).
  - **Result**: The *duration\_downsampled* variable now contains the total time (in seconds) of the downsampled audio.

```
time_axis_downsampled = np.linspace(0, duration_downsampled,
len(downsampled_data))  # Create a time axis for the downsampled data
```

- **Explanation**: This line generates a **time axis** for the downsampled data to plot the signal or to represent the time associated with each data point.
  - np.linspace(0, duration\_downsampled, len(downsampled\_data)): This generates an array of evenly spaced time values from 0 to duration\_downsampled, with the same number of points as there are in the downsampled data.
    - **0**: The start time of the signal (0 seconds).
    - duration\_downsampled: The end time of the signal, calculated in the previous line.





- len(downsampled\_data): This ensures the time array has the same number of elements as the downsampled data, so each data point is associated with a corresponding time value.
- **Result**: *time\_axis\_downsampled* is an array that represents the time in seconds for each sample in the downsampled data.

```
df_downsampled = pd.DataFrame({'Time (seconds)': time_axis_downsampled,
'Amplitude': downsampled_data})  # Create a DataFrame with downsampled
time and amplitude
```

- **Explanation**: This line creates a **Pandas DataFrame** that combines the time and amplitude of the downsampled signal into a tabular format.
  - **pd.DataFrame()**: This function creates a DataFrame (a table-like structure in Python) from the provided dictionary.
    - 'Time (seconds)': This is the name of the column that will store the time values.
    - time\_axis\_downsampled: The time values (generated in the previous line).
    - **'Amplitude'**: This is the name of the column that will store the amplitude (signal strength) values.
    - **downsampled\_data**: The amplitude values of the downsampled signal.
  - Result: *df\_downsampled* is a DataFrame where the first column contains the time values, and the second column contains the corresponding amplitude values of the downsampled signal.

```
df_downsampled.to_csv('/content/drive/MyDrive/Colab
Notebooks/outputFile_downsampled.csv', index=False) # Save the
downsampled DataFrame as a CSV file
```

- **Explanation**: This line saves the **downsampled data** in .csv format to Google Drive (or another location).
  - df\_downsampled.to\_csv(): This function saves the DataFrame (df\_downsampled) as a .csv file.
    - '/content/drive/MyDrive/Colab Notebooks/outputFile\_downsampled.csv': This is the file path where the .csv file will be saved. In this case, it is saved to Google Drive within a folder named Colab Notebooks. Students can modify the path to save the file to a different location if desired.
    - **index=False**: This prevents Pandas from saving the index (row numbers) as a separate column in the .csv file.
  - Result: The downsampled data is saved as a .csv file named outputFile\_downsampled.csv that students can open in a program such as Excel or import into other analysis tools.





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## Summary of what happens in the script:

- Downsampling: The original signal is downsampled from its original rate to 8000 Hz.
- Time Axis Creation: A time axis is generated to correspond to the downsampled data.
- Data Storage: A Pandas DataFrame is created, combining the time and amplitude of the downsampled signal.
- Saving the Data: The DataFrame is saved as a .csv file, making it accessible for further analysis or visualization.

# Patterns in Downsampled Data

## • What Students Will See:

- **Simplified Signal**: The downsampled data will retain the key features of the signal but with fewer data points.
- Major Trends: Students can identify broad patterns of activity without the distraction of very fine-grained detail.

# • Relation to Muscle Activity and Movement:

- Downsampling highlights overall trends in muscle activity, making it easier to compare across datasets or detect long-term behaviors (e.g., fatigue over time).
- It may slightly obscure finer details, but the focus remains on significant movements.



