**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).
	+ **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.

**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.