# **REPORT**

Zajęcia: Analog and digital electronic circuits Teacher: prof. dr hab. Vasyl Martsenyuk

#### **Lab 5-6**

Date 23.11.2024

**Topic:** "5. Digital Filter Design and Analysis: Implementing FIR and IIR filters in Python.

6. Adaptive Filtering: Applying adaptive filtering algorithms to noise reduction."

Variant 10

Anna Więzik Informatyka II stopień, niestacjonarne, 1 semestr, Gr.1b

#### 1. Problem statement:

Each task requires you to implement all three types of filters: FIR, IIR, and Adaptive LMS, using different parameters and observe the performance for noise reduction.

### 2. Input data:

Variant 10: - Design an FIR filter with the following coefficients and implement it in Python to reduce noise in a noisy sinusoidal signal.

```
FIR Filter Coefficients: b = \{0.4, 0.4, 0.4\}
```

 Design an IIR filter with the following coefficients and implement it in Python to reduce noise in the same noisy sinusoidal signal.

```
IIR Filter Coefficients: b = \{1, -0.6\}, a = \{1, 0.5\}
```

- Implement an adaptive LMS filter in Python with a step size μ = 0.1 and filter length M = 5 to reduce noise in the same noisy sinusoidal signal.

link to remote repozytorium: <a href="https://github.com/AnaShiro/DSP\_2024">https://github.com/AnaShiro/DSP\_2024</a>

### 3. Commands used (or GUI):

FIR

```
import numpy as np
import matplotlib.pyplot as plt

Qodo Gen: Options|Test this function
def fir filter(x, b):

    M = len(b)
    y = np.zeros(len(x))
    for n in range(M, len(x)):
        y[n] = np.dot(b, x[n-M+1:n+1][::-1])
    return y

fs = 1000  # Sampling frequency
t = np.linspace(0, 1, fs)
x = np.sin(2 * np.pi * 5 * t) + 0.5 * np.random.randn(len(t))  # Signal with noise
b = [0.4, 0.4, 0.4]  # FIR coefficients

y = fir_filter(x, b)

plt.figure(figsize=(10, 6))
plt.plot(t, x, Label="Noisy Signal")
plt.plot(t, y, Label="Filtered Signal", Linewidth=2)
plt.legend()
plt.title("FIR Filter")
plt.xlabel("Time [s]")
plt.ylabel("Amplitude")
plt.grid()
plt.show()
```

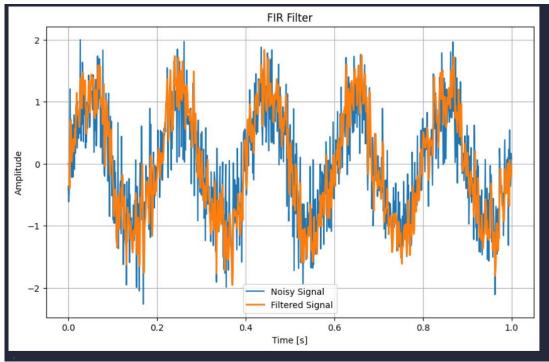
#### IIR

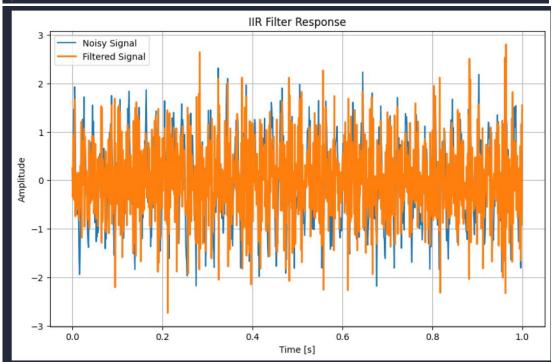
```
≣ ▷ □ □ □
import numpy as np
Qodo Gen: Options | Test this function
def iir_filter(x, b, a):
    M = len(b)  # Length of numerator coefficients (b)
    N = len(a)  # Length of denominator coefficients (a)
    y = np.zeros(len(x))  # Initialize output signal array
               # Ensure we use the correct slice length for the convolution

x_slice = x[max(0, n-M+1):n+1] # Input signal slice

y[n] = np.dot(b[:len(x_slice)], x_slice[::-1]) # Apply reverse convolution for numerator
                       # Ensure we use the correct slice length for the feedback part
y_slice = y[max(0, n-N+1):n] # Output signal slice
y[n] -= np.dot(a[1:min(N, len(y_slice)+1)], y_slice[::-1]) # Apply reverse convolution for feedback
13 = 1000 # Jampeong yequency
t = np.linspace(0, 1, fs) # Time vector
x = np.sin(2 * np.pi * 50 * t) + 0.5 * np.random.randn(len(t))
y = iir_filter(x, b, a)
plt.figure(figsize=(10, 6))
plt.plot(t, x, label="Noisy Signal")
plt.plot(t, y, label="Filtered Signal", linewidth=2)
plt.iegend()
plt.title("IIR Filter Response")
plt.xlabel("Time [s]")
plt.ylabel("Amplitude")
```

# 4. Outcomes:





## 5. Conclusions:

This manual offers a comprehensive overview of the mathematical principles underlying FIR (Finite Impulse Response), IIR (Infinite Impulse Response), and LMS (Least Mean Squares) adaptive filters. It not only provides the theoretical background but also includes practical Python implementations for each filter type. Furthermore, the manual incorporates various visualization techniques to effectively demonstrate the performance and efficiency of these filters in real-world scenarios. This combination of theory, implementation, and visual aids ensures a clear understanding of the filters' functionality and their applications.