

# Audio Filtering

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## 1 DIGITAL FILTER

### 1.1 Download the sound file from

[https://github.com/Avyaaz13/Audio-Filtering/blob/main/Audio%20Filtering/codes/Input\\_audio.wav](https://github.com/Avyaaz13/Audio-Filtering/blob/main/Audio%20Filtering/codes/Input_audio.wav)

### 1.2 Below is the Python Code to perform the Audio Filtering:

```
import matplotlib.pyplot as plt
import numpy as np
import soundfile as sf
from scipy import signal

# Read .wav file
input_signal, fs = sf.read('Input_audio.wav')

# Order of the filter
order = 3

# Cutoff frequency 4kHz
cutoff_freq = 4000.0

# Digital frequency
Wn = 2 * cutoff_freq / fs
b, a = signal.butter(order, Wn, 'low')

# Ensure the signal is long enough for the filter
if len(input_signal) < max(3 * (max(len(a), len(b)) - 1), 15):
    raise ValueError("Input signal is too short for the specified filter order and padding.")

print(a)
print(b)

# Filter the input signal with a Butterworth filter
output_signal = signal.filtfilt(b, a, input_signal, method="gust")
```

```
sf.write('ReducedNoise_s181.wav',
        output_signal, fs)
```

### 1.3 The audio file is analyzed using spectrogram using the online platform <https://academo.org/demos/spectrum-analyzer>.

The orange and yellow areas represent frequencies that have high intensities in the sound. Also, the signal is blank for frequencies above 5.1 kHz.

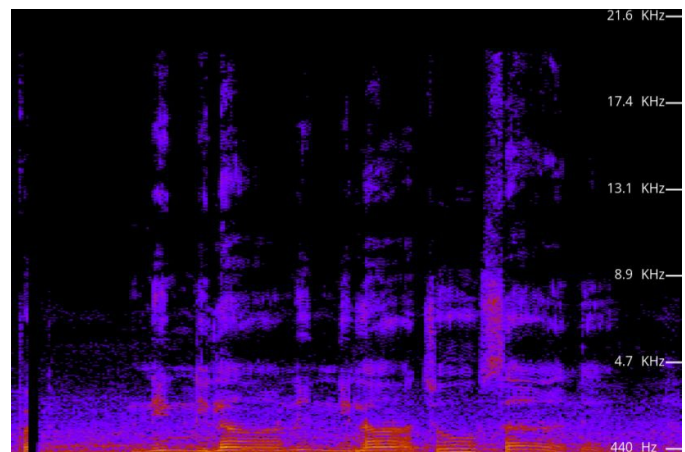


Fig. 1: Spectrogram of Input Audio

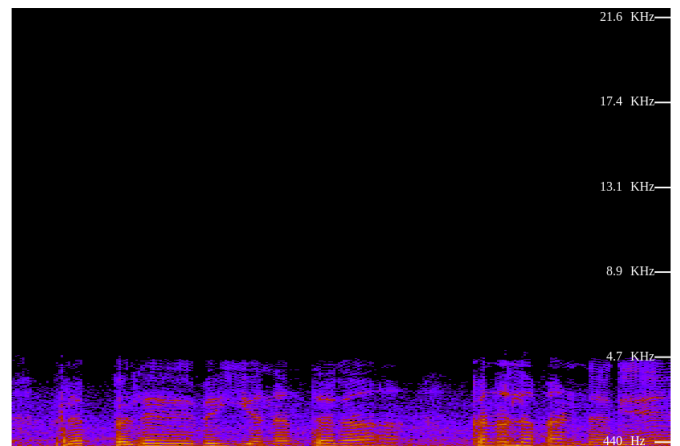


Fig. 2: Spectrogram of Filtered Input Audio

## 2 DIFFERENCE EQUATION

2.1 Let

$$x(n) = \left\{ \underset{\uparrow}{1}, 2, 3, 4, 2, 1 \right\} \quad (2.0.1)$$

Sketch  $x(n)$ .

2.2 Let

$$y(n) + \frac{1}{2}y(n-1) = x(n) + x(n-2),$$

$$y(n) = 0, n < 0 \quad (2.0.2)$$

Sketch  $y(n)$ .

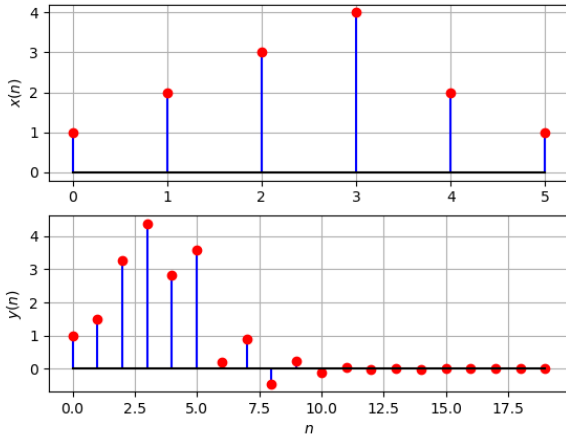
Solve

**Solution:** The C code calculates  $y(n)$  and Python plots the graph.

<https://github.com/Avyaaz13/Audio-Filtering/blob/main/Audio%20Filtering/codes/xnyn.c>

Below are the plots of the  $x(n)$  and  $y(n)$ :

<https://github.com/Avyaaz13/Audio-Filtering/blob/main/Audio%20Filtering/codes/xnyn.py>



3.4 Show that

$$a^n u(n) \xleftrightarrow{z} \frac{1}{1 - az^{-1}} \quad |z| > |a| \quad (3.0.16)$$

**Solution:**

$$a^n u(n) \xleftrightarrow{z} \sum_{n=0}^{\infty} (az^{-1})^n \quad (3.0.17)$$

$$= \frac{1}{1 - az^{-1}} \quad |z| > |a| \quad (3.0.18)$$

3.5 Let

$$H(e^{j\omega}) = H(z = e^{j\omega}). \quad (3.0.19)$$

Plot  $|H(e^{j\omega})|$ . Comment.  $H(e^{j\omega})$  is known as the *Discrete Time Fourier Transform* (DTFT) of  $x(n)$ .

**Solution:** Below is the code which plots the magnitude of Transfer Function:

```
https://github.com/Avyaaz13/Audio-Filtering/
blob/main/Audio%20Filtering/codes/H.py
```

Substituting  $z = e^{j\omega}$  in (3.0.9), we get

$$|H(e^{j\omega})| = \left| \frac{1 + e^{-2j\omega}}{1 + \frac{1}{2}e^{-j\omega}} \right| \quad (3.0.20)$$

$$= \sqrt{\frac{(1 + \cos 2\omega)^2 + (\sin 2\omega)^2}{\left(1 + \frac{1}{2} \cos \omega\right)^2 + \left(\frac{1}{2} \sin \omega\right)^2}} \quad (3.0.21)$$

$$= \frac{4|\cos \omega|}{\sqrt{5 + 4 \cos \omega}} \quad (3.0.22)$$

$$|H(e^{j(\omega+2\pi)})| = \frac{4|\cos(\omega + 2\pi)|}{\sqrt{5 + 4 \cos(\omega + 2\pi)}} \quad (3.0.23)$$

$$= \frac{4|\cos \omega|}{\sqrt{5 + 4 \cos \omega}} \quad (3.0.24)$$

$$= |H(e^{j\omega})| \quad (3.0.25)$$

Therefore, the fundamental period is  $2\pi$ , which implies that DTFT of a signal is always periodic.

#### 4 IMPULSE RESPONSE

4.1 Find an expression for  $h(n)$  using  $H(z)$ , given that

$$h(n) \xleftrightarrow{z} H(z) \quad (4.0.1)$$

and there is a one to one relationship between  $h(n)$  and  $H(z)$ .  $h(n)$  is known as the *impulse*

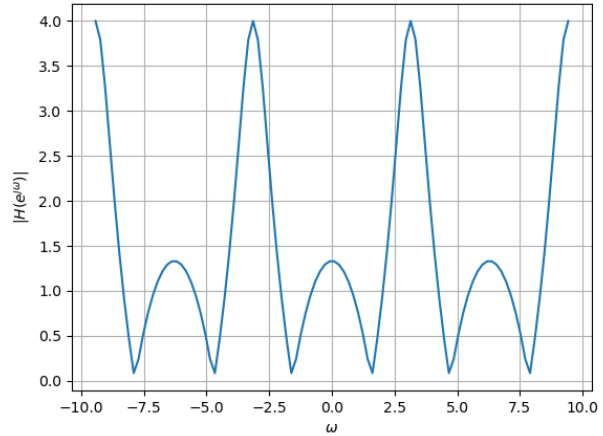


Fig. 4:  $|H(e^{j\omega})|$  vs  $\omega$

response of the system defined by (2.0.2).

**Solution:** From (3.0.9),

$$H(z) = \frac{1}{1 + \frac{1}{2}z^{-1}} + \frac{z^{-2}}{1 + \frac{1}{2}z^{-1}} \quad (4.0.2)$$

$$\Rightarrow h(n) = \left(-\frac{1}{2}\right)^n u(n) + \left(-\frac{1}{2}\right)^{n-2} u(n-2) \quad (4.0.3)$$

using (3.0.16) and (3.0.6).

4.2 Sketch  $h(n)$ . Is it bounded? Convergent?

**Solution:** The following code plots  $h(n)$

```
https://github.com/Avyaaz13/Audio-Filtering/
blob/main/Audio%20Filtering/codes/h.py
```

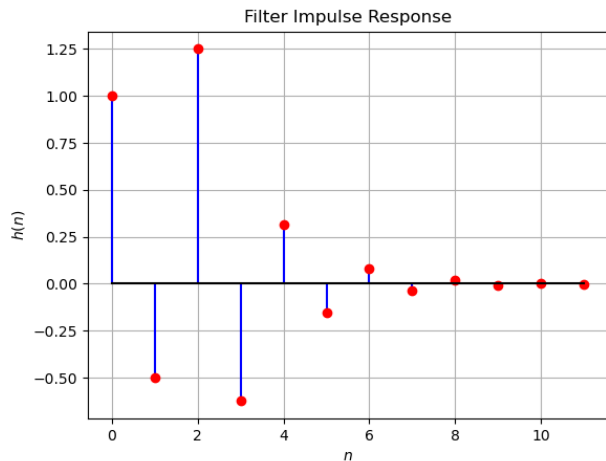


Fig. 5:  $h(n)$  vs  $n$

4.3 The system with  $h(n)$  is defined to be stable if

$$\sum_{n=-\infty}^{\infty} h(n) < \infty \quad (4.0.4)$$

Is the system defined by (2.0.2) stable for the impulse response in (4.0.1)?

**Solution:** For stable system (4.0.4) should converge.

By using ratio test for convergence:

$$\lim_{n \rightarrow \infty} \left| \frac{h(n+1)}{h(n)} \right| < 1 \quad (4.0.5)$$

$$(4.0.6)$$

For large  $n$

$$u(n) = u(n-2) = 1 \quad (4.0.7)$$

$$\lim_{n \rightarrow \infty} \left( \frac{h(n+1)}{h(n)} \right) = 1/2 < 1 \quad (4.0.8)$$

Hence it is stable.

4.4 Compute and sketch  $h(n)$  using

$$h(n) + \frac{1}{2}h(n-1) = \delta(n) + \delta(n-2), \quad (4.0.9)$$

This is the definition of  $h(n)$ .

**Solution:**

Definition of  $h(n)$ : The output of the system when  $\delta(n)$  is given as input.

The following code plots Fig. 6. Note that this is the same as Fig. 5.

<https://github.com/Avyaaz13/Audio-Filtering/blob/main/Audio%20Filtering/codes/hndef.py>

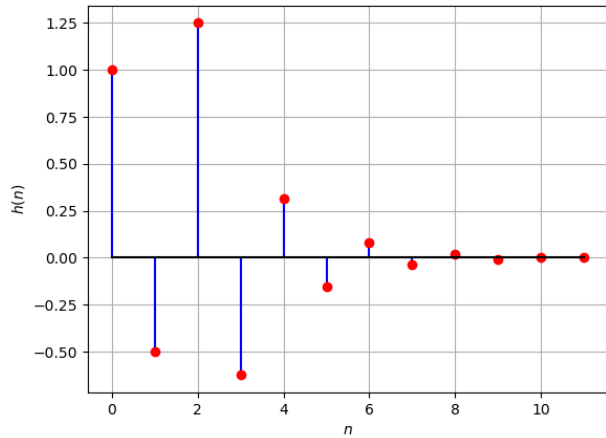


Fig. 6:  $h(n)$  vs  $n$  using definition

4.5 Compute

$$y(n) = x(n) * h(n) = \sum_{k=-\infty}^{\infty} x(k)h(n-k) \quad (4.0.10)$$

Comment. The operation in (4.0.10) is known as *convolution*.

**Solution:** Below code plots Fig. 7. Note that this is the same as  $y(n)$  in Fig. 3.

<https://github.com/Avyaaz13/Audio-Filtering/blob/main/Audio%20Filtering/codes/ynconv.py>

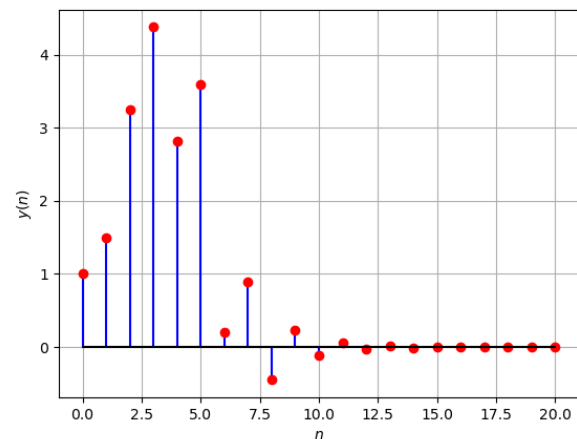


Fig. 7:  $y(n)$  from the definition of convolution

4.6 Show that

$$y(n) = \sum_{k=-\infty}^{\infty} x(n-k)h(k) \quad (4.0.11)$$

**Solution:** In (4.0.10), we substitute  $k = n - k$  to get

$$y(n) = \sum_{k=-\infty}^{\infty} x(k) h(n-k) \quad (4.0.12)$$

$$= \sum_{n-k=-\infty}^{\infty} x(n-k) h(k) \quad (4.0.13)$$

$$= \sum_{k=-\infty}^{\infty} x(n-k) h(k) \quad (4.0.14)$$

## 5 DFT AND FFT

### 5.1 Compute

$$X(k) \triangleq \sum_{n=0}^{N-1} x(n) e^{-j2\pi kn/N}, \quad k = 0, 1, \dots, N-1 \quad (5.0.1)$$

and  $H(k)$  using  $h(n)$ .

### 5.2 Compute

$$Y(k) = X(k)H(k) \quad (5.0.2)$$

### 5.3 Compute

$$y(n) = \frac{1}{N} \sum_{k=0}^{N-1} Y(k) \cdot e^{j2\pi kn/N}, \quad n = 0, 1, \dots, N-1 \quad (5.0.3)$$

**Solution:** The above three questions are solved using the code below.

<https://github.com/Avyaaz13/Audio-Filtering/blob/main/Audio%20Filtering/codes/5sol.py>

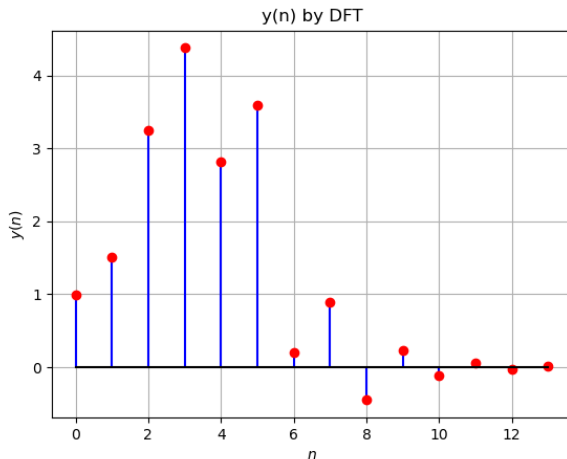


Fig. 8:  $y(n)$  obtained from DFT

### 5.4 Repeat the previous exercise by computing $X(k), H(k)$ and $y(n)$ through FFT and IFFT.

**Solution:** The solution of this question can be found in the code below.

[https://github.com/Avyaaz13/Audio-Filtering/blob/main/Audio%20Filtering/codes/IFFT\\_IDFT.py](https://github.com/Avyaaz13/Audio-Filtering/blob/main/Audio%20Filtering/codes/IFFT_IDFT.py)

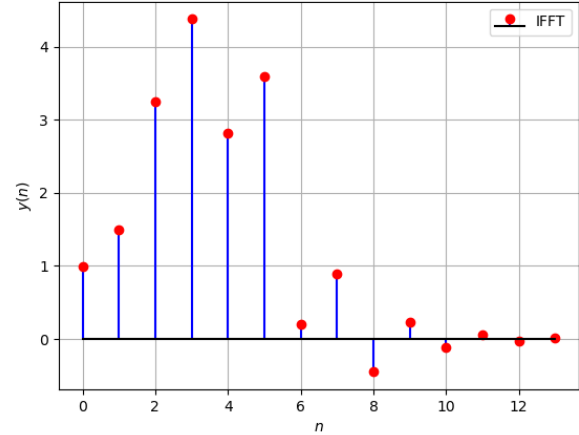


Fig. 9:  $y(n)$  obtained from IFFT

### 5.5 Wherever possible, express all the above equations as matrix equations.

**Solution:** The DFT matrix is defined as :

$$\mathbf{W} = \begin{pmatrix} \omega^0 & \omega^0 & \dots & \omega^0 \\ \omega^0 & \omega^1 & \dots & \omega^{N-1} \\ \vdots & \vdots & \ddots & \vdots \\ \omega^0 & \omega^{N-1} & \dots & \omega^{(N-1)(N-1)} \end{pmatrix} \quad (5.0.4)$$

where  $\omega = e^{-j\frac{2\pi}{N}}$ . Now any DFT equation can be written as

$$\mathbf{X} = \mathbf{W}\mathbf{x} \quad (5.0.5)$$

where

$$\mathbf{x} = \begin{pmatrix} x(0) \\ x(1) \\ \vdots \\ x(n-1) \end{pmatrix} \quad (5.0.6)$$

$$\mathbf{X} = \begin{pmatrix} X(0) \\ X(1) \\ \vdots \\ X(n-1) \end{pmatrix} \quad (5.0.7)$$

Thus we can rewrite (5.0.2) as:

$$\mathbf{Y} = \mathbf{X} \cdot \mathbf{H} = (\mathbf{W}\mathbf{x}) \cdot (\mathbf{W}\mathbf{h}) \quad (5.0.8)$$

The below code computes  $y(n)$  by DFT Matrix and then plots it.

<https://github.com/Avyaaz13/Audio-Filtering/blob/main/Audio%20Filtering/codes/matrix.py>

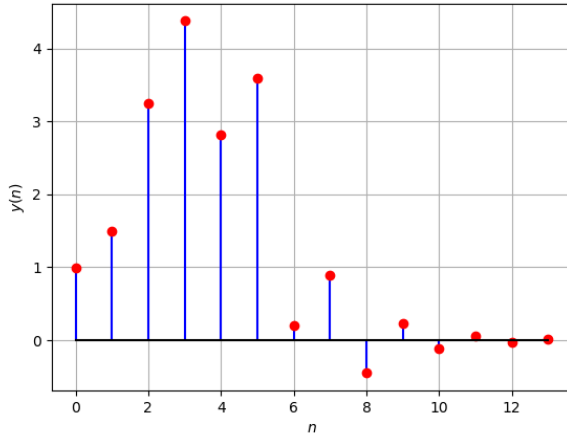


Fig. 10:  $y(n)$  from DFT Matrix

## 6 EXERCISES

Answer the following questions by looking at the python code in Problem 1.2.

6.1 The command

```
output_signal = signal.lfilter(b, a,
                               input_signal)
```

in Problem 1.2 is executed through the following difference equation

$$\sum_{m=0}^M a(m) y(n-m) = \sum_{k=0}^N b(k) x(n-k) \quad (6.0.1)$$

where the input signal is  $x(n)$  and the output signal is  $y(n)$  with initial values all 0. Replace **signal.lfilter** with your own routine and verify.

**Solution:** The below code gives the output of an Audio Filter without using the built in function `signal.lfilter`.

<https://github.com/Avyaaz13/Audio-Filtering/blob/main/Audio%20Filtering/codes/filtfilt.py>

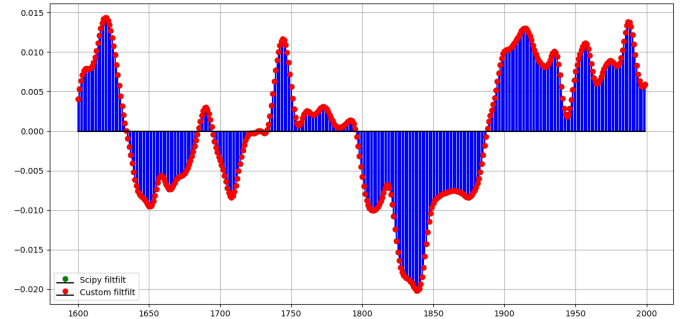


Fig. 11: Both the outputs using and without using function overlap

6.2 Repeat all the exercises in the previous sections for the above  $a$  and  $b$ .

**Solution:** The code in 1.2 generates the values of  $a$  and  $b$  which can be used to generate a difference equation.

And,

$$M = 4 \quad (6.0.2)$$

$$N = 4 \quad (6.0.3)$$

From 6.0.1

$$\begin{aligned} a(0)y(n) + a(1)y(n-1) + a(2)y(n-2) + a(3)y(n-3) \\ = b(0)x(n) + b(1)x(n-1) + b(2)x(n-2) + b(3)x(n-3) \end{aligned}$$