

# Audio Filtering

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## I. DIGITAL FILTER

- I.1 The sound file used for this code is obtained from the below link

[https://github.com/Meowfessor96/  
audio\\_filtering/blob/main/audio\\_filtering/  
codes/audio.wav](https://github.com/Meowfessor96/audio_filtering/blob/main/audio_filtering/codes/audio.wav)

- I.2 Audio filtering is carried out using the following Python Code

```
import soundfile as sf
from scipy import signal

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

#sampling frequency of Input signal
sampl_freq=fs

#order of the filter
order=4

#cutoff frequency 1000Hz
cutoff_freq=1000.0

#digital frequency
Wn=2*cutoff_freq/sampl_freq

# b and a are numerator and denominator
polynomials respectively
b, a = signal.butter(order,Wn, 'low')

#filter the input signal with butterworth filter
output_signal = signal.filtfilt(b, a,
    input_signal)
#output_signal = signal.lfilter(b, a,
    input_signal)

#write the output signal into .wav file
sf.write('filtered_audio.wav', output_signal,
    fs)
```

- I.3 Audio is analysed using a spectrogram found in the following website :

The figures represent the distribution of sound among frequencies in the x axis and similarly the y-axis represents the intensity of the sound

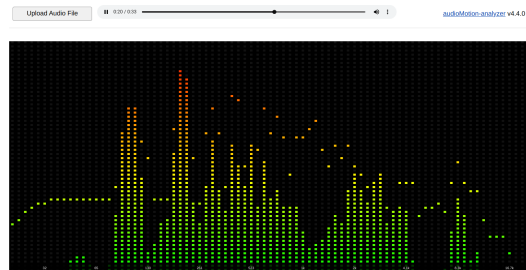


Fig. 1. frequency distribution of the audio file before Filtering(at 0:20)

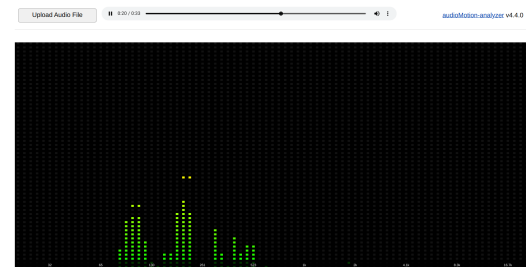


Fig. 2. frequency distribution of the audio file after Filtering(at 0:20)

## II. DIFFERENCE EQUATION

- II.1 Let

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

Sketch  $x(n)$ .

- II.2 Let

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

$$y(n) = 0, n < 0 \quad (2)$$

Sketch  $y(n)$ .

Solve

**Solution:**  $x(n)$  and  $y(n)$  values for different

values of  $n$  are generated in a text file using C file.

[https://github.com/Meowfessor96/  
audio\\_filtering/blob/main/audio\\_filtering/  
codes/2\\_2.c](https://github.com/Meowfessor96/audio_filtering/blob/main/audio_filtering/codes/2_2.c)

The following code plots (1) and (2)

[https://github.com/Meowfessor96/  
audio\\_filtering/blob/main/audio\\_filtering/  
codes/2.2.py](https://github.com/Meowfessor96/audio_filtering/blob/main/audio_filtering/codes/2.2.py)

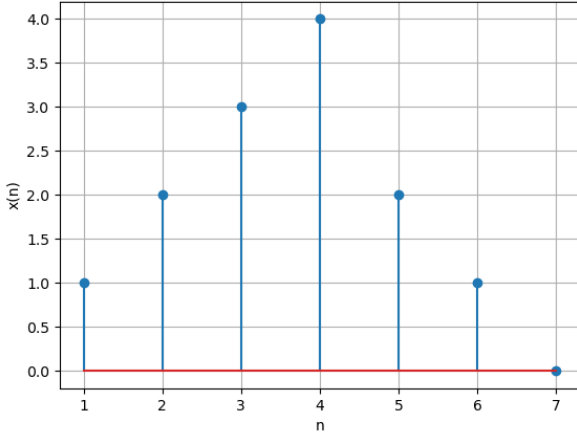


Fig. 3. Plot of  $x(n)$  vs  $n$

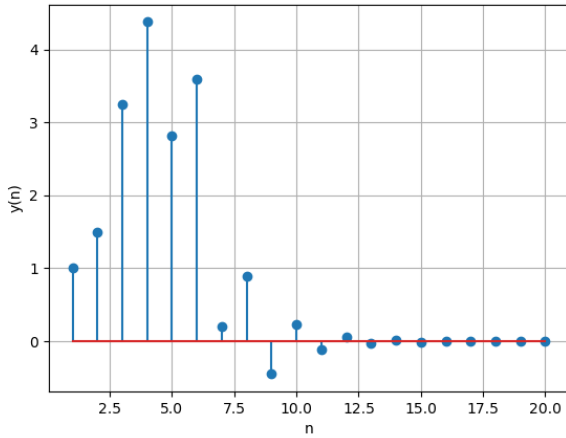


Fig. 4. Plot of  $y(n)$  vs  $n$

### III. Z-TRANSFORM

III.1 The Z-transform of  $x(n)$  is defined as

$$X(z) = \mathcal{Z}\{x(n)\} = \sum_{n=-\infty}^{\infty} x(n)z^{-n} \quad (3)$$

Show that

$$\mathcal{Z}\{x(n-1)\} = z^{-1}X(z) \quad (4)$$

and find

$$\mathcal{Z}\{x(n-k)\} \quad (5)$$

**Solution:** From (3),

$$\mathcal{Z}\{x(n-k)\} = \sum_{n=-\infty}^{\infty} x(n-k)z^{-n} \quad (6)$$

$$= \sum_{n=-\infty}^{\infty} x(n)z^{-n-1} = z^{-1} \sum_{n=-\infty}^{\infty} x(n)z^{-n} \quad (7)$$

$$= z^{-1}X(z) \quad (8)$$

resulting in (4). Similarly, it can be shown that

$$\mathcal{Z}\{x(n-k)\} = z^{-k}X(z) \quad (9)$$

III.2 Find

$$H(z) = \frac{Y(z)}{X(z)} \quad (10)$$

from (2) assuming that the Z-transform is a linear operation.

**Solution:** Applying (9) in (2),

$$Y(z) + \frac{1}{2}z^{-1}Y(z) = X(z) + z^{-2}X(z) \quad (11)$$

$$\Rightarrow \frac{Y(z)}{X(z)} = \frac{1 + z^{-2}}{1 + \frac{1}{2}z^{-1}} \quad (12)$$

III.3 Find the Z transform of

$$\delta(n) = \begin{cases} 1 & n = 0 \\ 0 & \text{otherwise} \end{cases} \quad (13)$$

and show that the Z-transform of

$$u(n) = \begin{cases} 1 & n \geq 0 \\ 0 & \text{otherwise} \end{cases} \quad (14)$$

is

$$U(z) = \frac{1}{1 - z^{-1}}, \quad |z| > 1 \quad (15)$$

**Solution:** It is easy to show that

$$\delta(n) \xleftrightarrow{Z} 1 \quad (16)$$

and from (14),

$$U(z) = \sum_{n=0}^{\infty} z^{-n} \quad (17)$$

$$= \frac{1}{1 - z^{-1}}, \quad |z| > 1 \quad (18)$$

using the formula for the sum of an infinite geometric progression.

III.4 Show that

$$a^n u(n) \xleftrightarrow{Z} \frac{1}{1 - az^{-1}} \quad |z| > |a| \quad (19)$$

**Solution:**

$$a^n u(n) \xleftrightarrow{Z} \sum_{n=0}^{\infty} (a^n z^{-n}) \quad (20)$$

$$= \frac{1}{1 - az^{-1}} \quad |z| > |a| \quad (21)$$

III.5 Let

$$H(e^{j\omega}) = H(z = e^{j\omega}). \quad (22)$$

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

**Solution:** The DTFT of the transfer function is plotted using the following code

```
https://github.com/Meowfessor96/
audio_filtering/blob/main/audio_filtering/
codes/3.5.py
```

Substituting  $z = e^{j\omega}$  in (12), we get

$$\left| H(e^{j\omega}) \right| = \left| \frac{1 + e^{-2j\omega}}{1 + \frac{1}{2}e^{-j\omega}} \right| \quad (23)$$

$$= \frac{4|\cos \omega|}{\sqrt{5 + 4 \cos \omega}} \quad (24)$$

A function  $f(x)$  is said to be periodic of time period  $T$  if  $f(x + nT) = f(x)$ ,  $n \in \mathbb{Z}$

$$\left| H(e^{j(\omega+2\pi)}) \right| = \frac{4|\cos(\omega + 2\pi)|}{\sqrt{5 + 4 \cos(\omega + 2\pi)}} \quad (25)$$

$$= \frac{4|\cos \omega|}{\sqrt{5 + 4 \cos \omega}} \quad (26)$$

$$= \left| H(e^{j\omega}) \right| \quad (27)$$

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

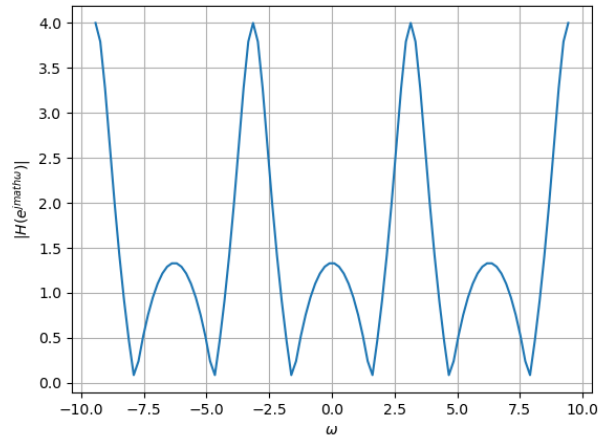


Fig. 5. DTFT,  $|H(e^{j\omega})|$

## IV. IMPULSE RESPONSE

IV.1 Find an expression for  $h(n)$  using  $H(z)$ , given that

$$h(n) \xleftrightarrow{Z} H(z) \quad (28)$$

and there is a one to one relationship between  $h(n)$  and  $H(z)$ .  $h(n)$  is known as the *impulse response* of the system defined by (2).

**Solution:** From (12),

$$H(z) = \frac{1}{1 + \frac{1}{2}z^{-1}} + \frac{z^{-2}}{1 + \frac{1}{2}z^{-1}} \quad (29)$$

$$\Rightarrow h(n) = \left(-\frac{1}{2}\right)^n u(n) + \left(-\frac{1}{2}\right)^{n-2} u(n-2) \quad (30)$$

using (19) and (9).

IV.2 Sketch  $h(n)$ . Is it bounded? Convergent?

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

```
https://github.com/Meowfessor96/
audio_filtering/blob/main/audio_filtering/
codes/4.2.py
```



Fig. 6.  $h(n)$  as the inverse of  $H(z)$

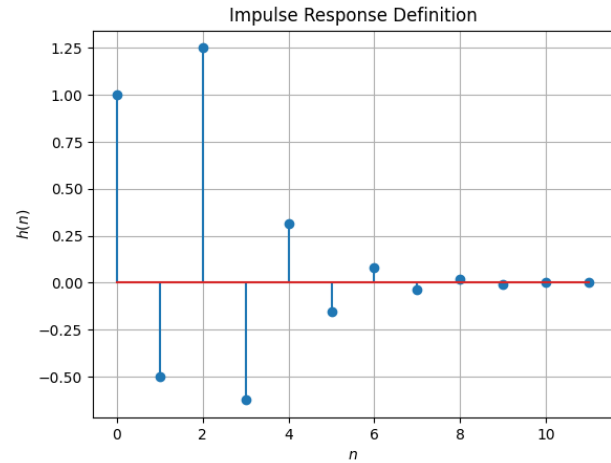


Fig. 7.  $h(n)$  from the definition is same as Fig. 6

IV.3 The system with  $h(n)$  is defined to be stable if

$$\sum_{n=-\infty}^{\infty} h(n) < \infty \quad (31)$$

Is the system defined by (2) stable for the impulse response in (28)?

**Solution:** For stable system (31) should converge.

By using ratio test

$$\lim_{n \rightarrow \infty} \left| \frac{h(n+1)}{h(n)} \right| < 1 \quad (32)$$

$$(33)$$

For large  $n$

$$u(n) = u(n-2) = 1 \quad (34)$$

$$\lim_{n \rightarrow \infty} \left( \frac{h(n+1)}{h(n)} \right) = 1/2 < 1 \quad (35)$$

Therefore it converges. Hence it is stable.

IV.4 Compute and sketch  $h(n)$  using

$$h(n) + \frac{1}{2}h(n-1) = \delta(n) + \delta(n-2), \quad (36)$$

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. 7. Note that this is the same as Fig. 6.

[https://github.com/Meowfessor96/  
audio\\_filtering/blob/main/audio\\_filtering/  
codes/4.4.py](https://github.com/Meowfessor96/audio_filtering/blob/main/audio_filtering/codes/4.4.py)

IV.5 Compute

$$y(n) = x(n) * h(n) = \sum_{k=-\infty}^{\infty} x(k)h(n-k) \quad (37)$$

Comment. The operation in (37) is known as *convolution*.

**Solution:** The following code plots Fig. 8. Note that this is the same as  $y(n)$  in Fig. ??.

[https://github.com/Meowfessor96/  
audio\\_filtering/blob/main/audio\\_filtering/  
codes/4.5.py](https://github.com/Meowfessor96/audio_filtering/blob/main/audio_filtering/codes/4.5.py)

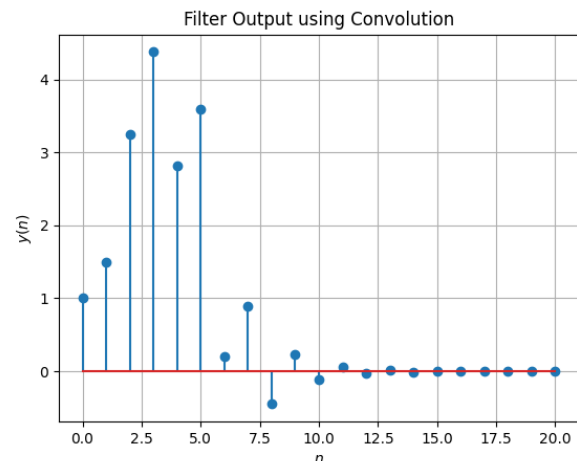


Fig. 8.  $y(n)$  from the definition of convolution

#### IV.6 Show that

$$y(n) = \sum_{k=-\infty}^{\infty} x(n-k)h(k) \quad (38)$$

**Solution:** In (37), we substitute  $k = n - k$  to get

$$y(n) = \sum_{k=-\infty}^{\infty} x(k)h(n-k) \quad (39)$$

$$= \sum_{n-k=-\infty}^{\infty} x(n-k)h(k) \quad (40)$$

$$= \sum_{k=-\infty}^{\infty} x(n-k)h(k) \quad (41)$$

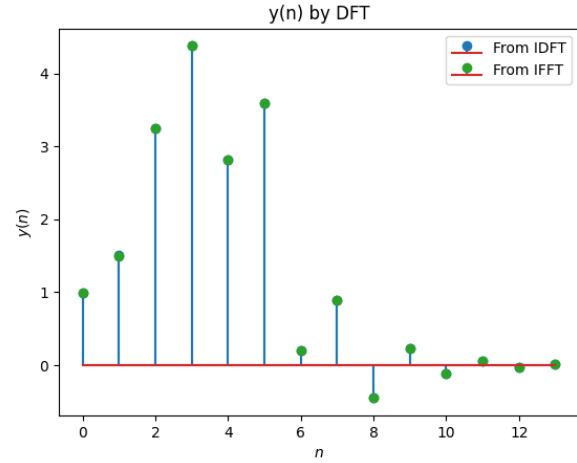


Fig. 9.  $y(n)$  obtained from IDFT and IFFT is plotted and verified

### V. DFT AND FFT

#### V.1 Compute

$$X(k) \triangleq \sum_{n=0}^{N-1} x(n)e^{-j2\pi kn/N}, \quad k = 0, 1, \dots, N-1 \quad (42)$$

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

#### V.2 Compute

$$Y(k) = X(k)H(k) \quad (43)$$

#### V.3 Compute

$$y(n) = \frac{1}{N} \sum_{k=0}^{N-1} Y(k) \cdot e^{j2\pi kn/N}, \quad n = 0, 1, \dots, N-1 \quad (44)$$

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

[https://github.com/Meowfessor96/  
audio\\_filtering/blob/main/audio\\_filtering/  
codes/5\\_sol.py](https://github.com/Meowfessor96/audio_filtering/blob/main/audio_filtering/codes/5_sol.py)

#### V.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/Meowfessor96/  
audio\\_filtering/blob/main/audio\\_filtering/  
codes/5.4.py](https://github.com/Meowfessor96/audio_filtering/blob/main/audio_filtering/codes/5.4.py)

This code verifies the result by plotting the obtained result with the result obtained by IDFT.

#### V.5 Wherever possible, express all the above equations as matrix equations.

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

$$\mathbf{W} = \begin{pmatrix} \omega^0 & \omega^0 & \dots & \omega^0 \\ \omega^0 & \omega^1 & \dots & \omega^{N-1} \\ \vdots & \vdots & \ddots & \vdots \\ \omega^0 & \omega^{N-1} & \dots & \omega^{(N-1)(N-1)} \end{pmatrix} \quad (45)$$

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

$$\mathbf{X} = \mathbf{W}\mathbf{x} \quad (46)$$

where

$$\mathbf{x} = \begin{pmatrix} x(0) \\ x(1) \\ \vdots \\ x(n-1) \end{pmatrix} \quad (47)$$

$$\mathbf{X} = \begin{pmatrix} X(0) \\ X(1) \\ \vdots \\ X(n-1) \end{pmatrix} \quad (48)$$

Thus we can rewrite (43) as:

$$\mathbf{Y} = \mathbf{X} \odot \mathbf{H} = (\mathbf{W}\mathbf{x}) \odot (\mathbf{W}\mathbf{h}) \quad (49)$$

where the  $\odot$  represents the Hadamard product which performs element-wise multiplication.

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

[https://github.com/Meowfessor96/audio\\_filtering/  
blob/main/audio\\_filtering/codes/5.5.py](https://github.com/Meowfessor96/audio_filtering/blob/main/audio_filtering/codes/5.5.py)



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

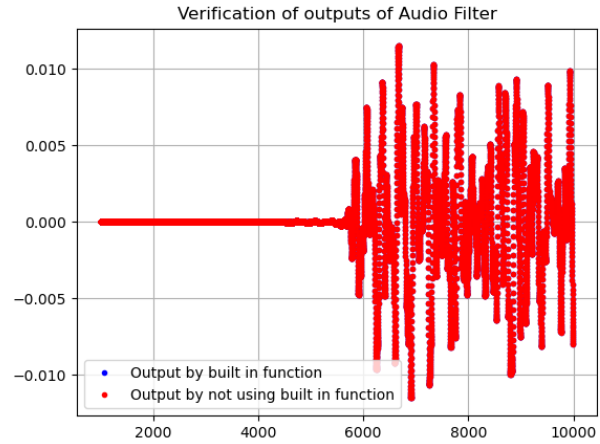


Fig. 11. Both the outputs using and without using function overlap

## VI. EXERCISES

Answer the following questions by looking at the python code in Problem I.2.

### VI.1 The command

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

in Problem I.2 is executed through the following difference equation

$$\sum_{m=0}^M a(m) y(n-m) = \sum_{k=0}^N b(k) x(n-k) \quad (50)$$

where the input signal is  $x(n)$  and the output signal is  $y(n)$  with initial values all 0. Replace **signal.filtfilt** 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/Meowfessor96/
  audio_filtering/blob/main/audio_filtering/
  codes/6.1.py
```

### VI.2 Repeat all the exercises in the previous sections for the above $a$ and $b$ .

**Solution:** From the output of  $a$  and  $b$  from code in Problem I.2, and from 50

$$\begin{aligned} & a(0)y(n) + a(1)y(n-1) + a(2)y(n-2) + a(3)y(n-3) \\ & + a(4)y(n-4) = b(0)x(n) + b(1)x(n-1) + b(2)x(n-2) \\ & + b(3)x(n-3) + b(4)x(n-4) \end{aligned} \quad (51)$$

Difference Equation is given by :

$$\begin{aligned} & y(n) - (3.66)y(n-1) + (5.05)y(n-2) - (3.099)y(n-3) + (0.715)y(n-4) \\ & = (1.45 \times 10^{-5})x(n) + (5.74 \times 10^{-5})x(n-1) + (8.62 \times 10^{-5})x(n-2) \\ & + (5.74 \times 10^{-5})x(n-3) + (1.43 \times 10^{-5})x(n-4) \end{aligned} \quad (52)$$

From (50)

$$H(z) = \frac{b_0 + b_1z^{-1} + b_2z^{-2} + \dots + b_Mz^{-M}}{a_0 + a_1z^{-1} + a_2z^{-2} + \dots + a_Nz^{-N}} \quad (53)$$

$$H(z) = \frac{\sum_{k=0}^N b(k)z^{-k}}{\sum_{k=0}^M a(k)z^{-k}} \quad (54)$$

$M = N = 5$

Looking at scipy documentation at [https://docs.scipy.org/doc/scipy/reference/generated/scipy](https://docs.scipy.org/doc/scipy/reference/generated/scipy.signal.partial_fractions.html) the partial fraction on (54) can be generalised as:

$$H(z) = \sum_i \frac{r(i)}{1 - p(i)z^{-1}} + \sum_j (j)z^{-j} \quad (55)$$

Whose values can be obtained using the `scipy.signal.residuez` function. Now,

$$r^n u(n) \xleftrightarrow{z} \frac{1}{1 - rz^{-1}}, \quad |z| > |r| \quad (56)$$

$$\delta(n - k) \xleftrightarrow{z} z^{-k} \quad (57)$$

Taking inverse z transform of (55) by using (56) and (57)

$$h(n) = \sum_i r(i)[p(i)]^n u(n) + \sum_j k(j)\delta(n - j) \quad (58)$$

[https://github.com/Meowfessor96/  
audio\\_filtering/blob/main/audio\\_filtering/  
codes/6.2.py](https://github.com/Meowfessor96/audio_filtering/blob/main/audio_filtering/codes/6.2.py)

$r(i)$	$p(i)$	$k(i)$
$0.0590681 - 0.14379042j$	$0.8869974 + 0.04376584j$	$2.00 \times 10^{-5}$
$0.0590681 + 0.14379042j$	$0.8869974 - 0.04376584j$	—
$-0.05907096 + 0.02466936j$	$0.94551962 + 0.11263131j$	—
$-0.05907096 - 0.02466936j$	$0.94551962 - 0.11263131j$	—

TABLE I  
VALUES OF  $r(i)$ ,  $p(i)$ ,  $k(i)$

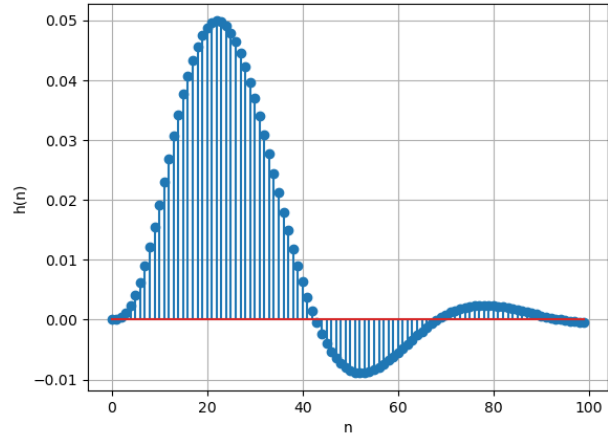


Fig. 12. Plot of  $h(n)$

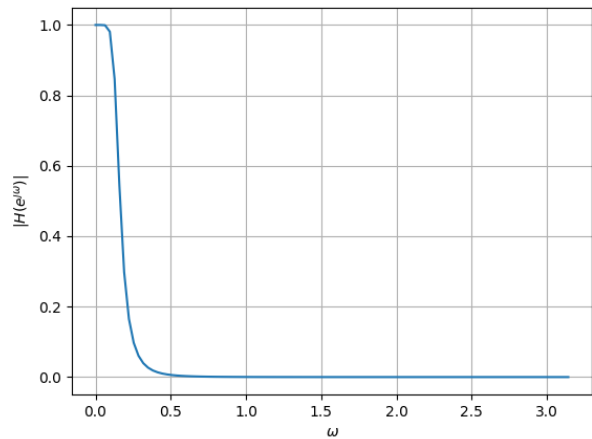


Fig. 13. Frequency Response of Audio Filter

### Stability of $h(n)$ :

According to (31)

$$H(z) = \sum_{n=0}^{\infty} h(n) z^{-n} \quad (59)$$

$$H(1) = \sum_{n=0}^{\infty} h(n) = \frac{\sum_{k=0}^N b(k)}{\sum_{k=0}^M a(k)} < \infty \quad (60)$$

As both  $a(k)$  and  $b(k)$  are finite length sequences they converge.

The below code plots Filter frequency response

[https://github.com/Meowfessor96/  
audio\\_filtering/blob/main/audio\\_filtering/  
codes/6\\_filter\\_response.py](https://github.com/Meowfessor96/audio_filtering/blob/main/audio_filtering/codes/6_filter_response.py)

VI.3 What is the sampling frequency of the input signal?

**Solution:** The Sampling Frequency is 44.1KHz

VI.4 What is type, order and cutoff-frequency of the above butterworth filter

**Solution:** The given butterworth filter is low-pass with order=4 and cutoff-frequency=1kHz.

VI.5 Modify the code with different input parameters and get the best possible output.

**Solution:** A better filtering was found on setting the order of the filter to be 5.