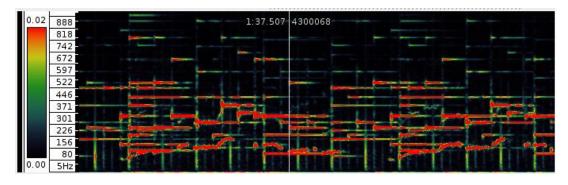
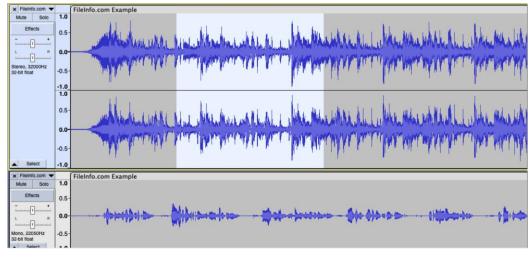


# **TOPIC:**

Programmatic
Simulation of
Digital Filters for
Audio Applications

# Linear Integrated Circuits and Applications – Experiential Learning Phase - I





# **Team Introduction**

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# Introduction

- > Sound eXchange (SoX) is an open source cross-platform audio editing software. It has a command line interface and is written in 'C'.
- > Some of SoX's features include:
  - 1) Reading and writing into various audio file formats like mp3, wav, aiff etc.
  - 2) Recording and playing audio.
  - 3) Editing via concatenate, trim, pad, repeat, reverse, volume, fade etc.
  - 4) Noise removal using frequency profiling.
  - 5) Multi-track mixing
  - 6) Spectrogram analysis
  - 7) Adjustment of speed (pitch and tempo) and sample rate.
- > Bash is a powerful tool for automating system administration tasks and performing other routine tasks in Unix/Linux.
- > Sonic Visualiser and Audacity are two audio editing softwares.

# **Objectives**

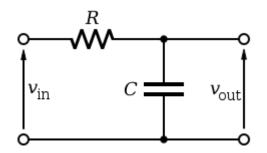
- ➤ Using the "Libsox" library to design digital filter like low-pass, high-pass, band pass and band reject.
- > Creating multiple files using "bash" scripting and remixing them using "sox" commands.
- > Analysing spectrograms of the mixed frequency files using the tool "Sonic Visualiser".
- ➤ Noise is any unwanted signals in the audio and can make it unpleasant to hear. For thisf reason, many youtubers do post audio processing to remove any noise present in their audio. We will be using our filters to remove unwanted noises from different audio files.
- > To compare input and output files on how they sound before and after they pass through various filters.
- > Analysing spectrograms of files to visualise the filters' response.

# **Low-pass Filter**

> A Low-pass Filter is a filter that is designed to pass all frequencies below a given cut-off frequency.

All frequencies higher than the cut-off frequency are attenuated.

#### Simple first order low-pass RC filter:-



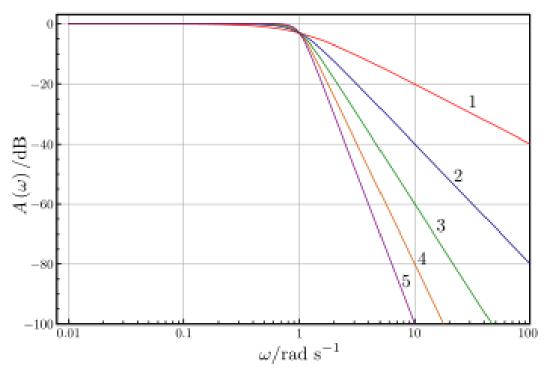
#### **Transfer function:-**

$$H(s) = rac{V_{\mathrm{out}}(s)}{V_{\mathrm{in}}(s)} = rac{\omega_0}{(s+\omega_0)}$$

#### **Cut-off Frequency:-**

$$f_c=rac{1}{2\pi au}=rac{1}{2\pi RC},$$

#### **Frequency Response:-**



Roll-off = -20n dB/decade where n is the filter order

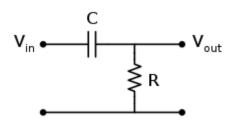


# **High-pass Filter**

> A High-pass Filter is a filter that is designed to pass all frequencies above a given cut-off frequency.

All frequencies lower than the cut-off frequency are attenuated.

#### Simple first order high- pass filter



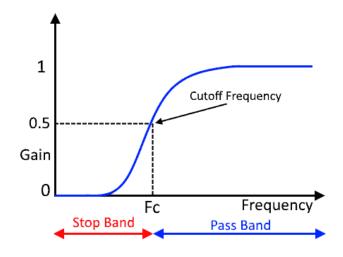
#### **Transfer Function**

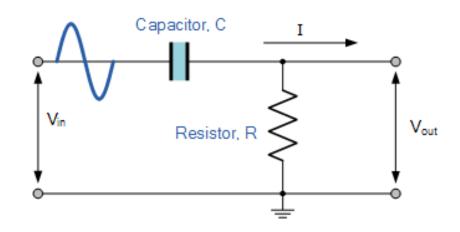
$$rac{V_{\mathrm{out}}(s)}{V_{\mathrm{in}}(s)} = rac{sRC}{1+sRC}.$$

## **Cut-off Frequency**

$$f_c=rac{1}{2\pi au}=rac{1}{2\pi RC},$$

#### **Frequency Response**



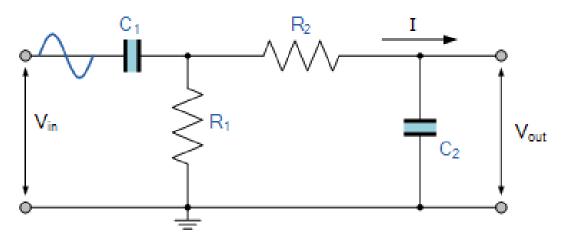




# **Band-pass Filter**

- ➤ A Band-pass Filter is a filter that is designed to pass all frequencies that fall between its cut-off frequencies fc1 and fc2. All frequencies outside the range are attenuated.
- > Can be created by combining low-pass and high-pass filters in series.

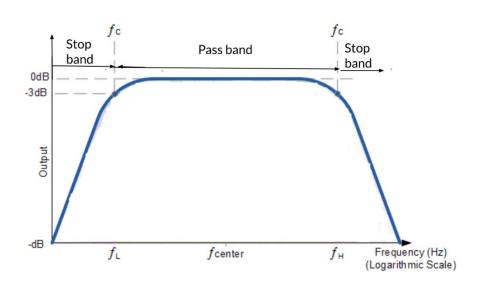
## Simple band-pass filter circuit:-



High pass filter cutoff frequency:  $fc1 = \frac{1}{2\pi R1C1}$ 

**Low-pass filter cutoff frequency:**  $fc2 = \frac{1}{2\pi R_2 C_2}$ 

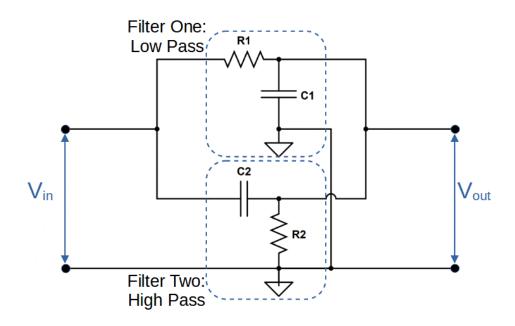
## **Frequency Response:-**



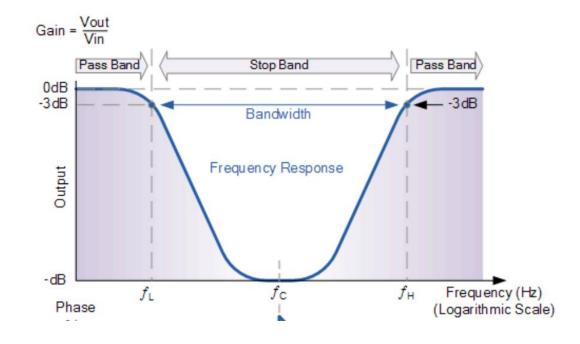
# **Band-stop Filter**

- > A Band-pass Filter is a filter that is designed to block all frequencies that fall between its cut-off frequencies fc1 and fc2. All frequencies outside the range are allowed to pass through.
- > Can be created by combining low-pass and high-pass filters in parallel.

# Simple band-stop filter circuit:-



## **Frequency Response:-**

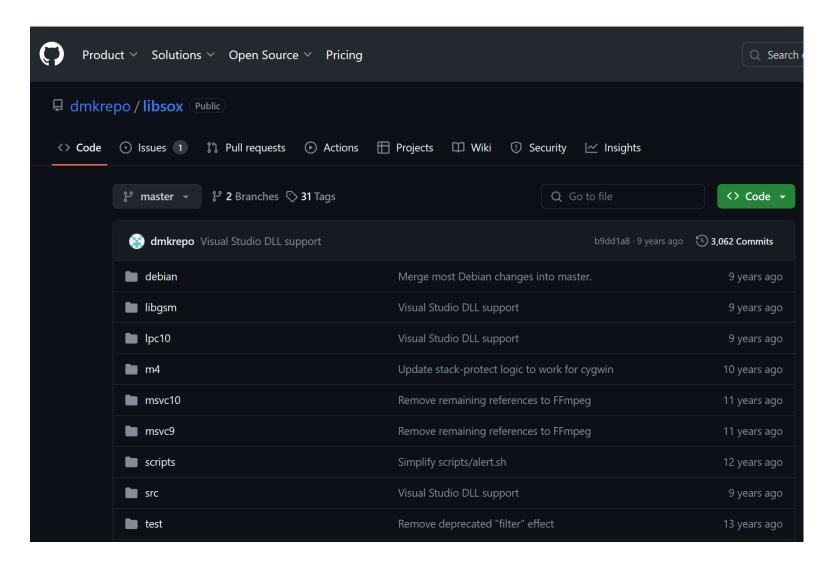




- Implementing the open source library called "LibSoX". Its Source Code is available on Github.
- > Creating our own audio files (at least 100 files) using 'sox' command and bash scripting.
- Mixing multiple files to create a mixed frequency audio files. The files vary in parameters like volume, type of waveform (sin, sawtooth), no. of channels etc.
- > Writing code for digital filters like low pass, high pass, band pass and band reject filters.
- Passing mixed frequency files as input to these digital filters and storing output in separate files.
- Analysing frequency spectrum of these output files on software tool "Sonic Visualizer" and "Audacity".

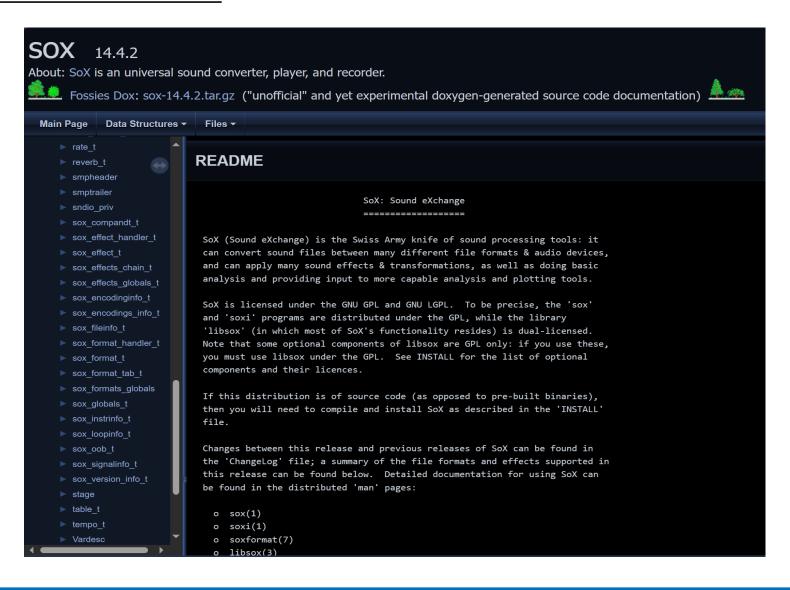


#### **GITHUB REPOSITORY OF "LIBSOX":-**





#### **DOCUMENTATION OF "LIBSOX" LIBRARY:-**



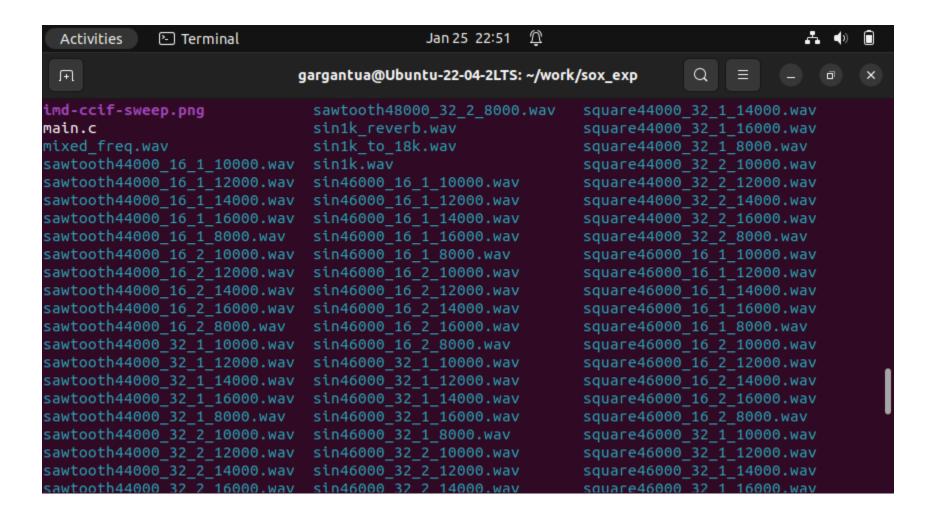


#### **BASH SCRIPT FOR CREATING MULTIPLE FILES:**

```
Terminal
                                          Jan 25 22:20
 Activities
                            gargantua@Ubuntu-22-04-2LTS: ~/work/sox_exp
Ŧ
1 #! /usr/bin/bash
2 sample_rate=48000
3 channels=2
4 duration=10
5 no_of_bits_per_sample=16
6 for rate in {44000..48000..2000}
     for bit per sample in {16..32..16}
 9
     do
        for channels in {1..2}
10
        do
11
12
           for freq in {8000..16000..2000}
13
           do
14
              C=" '
15
              file_name=$rate$c$bit per sample$c$channels$c$freq
              sox -V -r $rate -n -b $bit per sample -c $channels sawtooth$file name.wav synt
  h 10 sawtooth $freq vol 20 dB
              echo "sawtooth$file name.wav, sawtooth, $rate, $bit per sample, $channels, 10,
    20dB" >> sound files.csv; \
           done
18
                                                                            7,0-1
                                                                                          Top
```

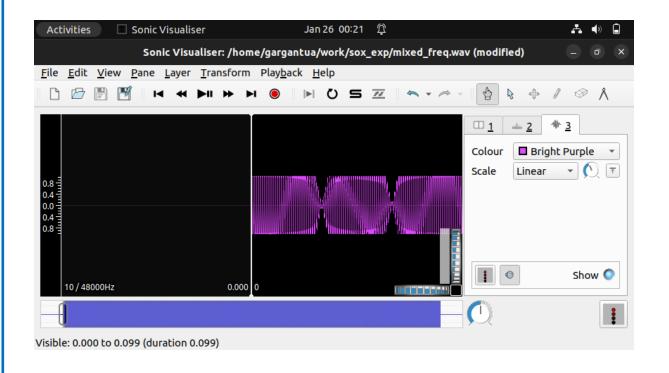


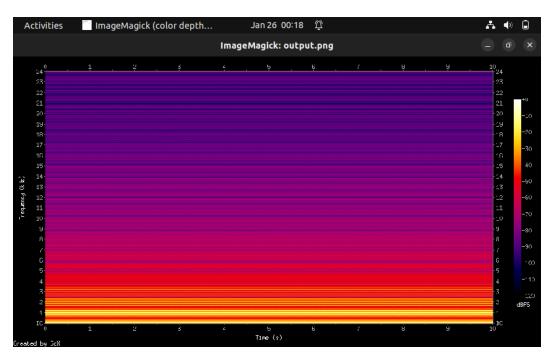
#### **AFTER RUNNING THE BASH SCRIPT:-**





#### **REMIXING THE .WAV FILES, VISUALISING WAVEFORM AND SPECTOGRAM:-**





#### SAMPLE CODE FOR VOLUME EFFECT:-

```
Jan 26 00:30 🖺
 Activities
            Terminal
                                                                                     — ())
                           gargantua@Ubuntu-22-04-2LTS: ~/work/filters_EL
Æ
5 #include "sox.h"
 6 #include <stdlib.h>
7 #include <stdio.h>
8 #include <assert.h>
9 int main(int argc, char * argv[])
     static sox_format_t * in, * out; /* input and output files */
    sox effects chain t * chain;
    sox effect t * e;
    char * args[10];
    assert(argc == 3);
    assert(sox init() == SOX SUCCESS);
    assert(in = sox open read(argv[1], NULL, NULL, NULL));
    assert(out = sox open write(argv[2], &in->signal, NULL, NULL, NULL, NULL));
    chain = sox create effects chain(&in->encoding, &out->encoding);
21
    e = sox create effect(sox find effect("input"));
    args[0] = (char *)in, assert(sox effect options(e, 1, args) == SOX SUCCESS);
    assert(sox add effect(chain, e, &in->signal, &in->signal) == SOX SUCCESS);
                                                                           10,1
                                                                                         17%
```



```
Jan 26 00:31
 Activities
            Terminal
                                                                                    - ()
                           gargantua@Ubuntu-22-04-2LTS: ~/work/filters_EL
                                                                       Q
ſŦ
     assert(sox add effect(chain, e, &in->signal, &in->signal) == SOX SUCCESS);
25
    free(e);
26
     e = sox_create_effect(sox_find_effect("vol"));
    args[0] = "3dB", assert(sox_effect_options(e, 1, args) == SOX_SUCCESS);
     assert(sox add effect(chain, e, &in->signal, &in->signal) == SOX SUCCESS);
    free(e);
30
31
    e = sox_create_effect(sox_find_effect("output"));
     args[0] = (char *)out, assert(sox_effect_options(e, 1, args) == SOX_SUCCESS);
     assert(sox add effect(chain, e, &in->signal, &in->signal) == SOX SUCCESS);
    free(e);
35
36
    sox_flow_effects(chain, NULL, NULL);
    sox_delete_effects_chain(chain);
    sox_close(out);
    sox_close(in);
    sox_quit();
42 return 0;
                                                                          43,1
                                                                                         Bot
```

# **Applications**

#### **LOW-PASS:-**

- ➤ Used to remove high frequency noise from biomedical signals, such as ECG signals ensuring accurate and reliable monitoring of physiological parameters.
- ➤ Often used in power supply circuits to filter out high frequency noise to provide a stable and clean DC output voltage.

#### **HIGH-PASS:-**

- ➤ In image-processing, high pass filters are used to detect edges in images. It allows the higher frequency components associated with the edges to pass through. Moreover, using high pass filters can be used to make an image look more sharper.
- > In geophysics, high pass filters are used in seismic signal processing to remove low frequency noise and focus on high frequency seismic signals

#### **BAND-PASS:-**

- ➤ Used in speech processing applications to focus on specific frequency ranges associated with human speech.
  This is important in applications like voice recognition.
- ➢ Help in channelizing and separating signals to avoid interference in wireless communication systems.

# References

- > GITHUB REPOSITORY OF "LIBSOX" LIBRARY: <a href="https://github.com/dmkrepo/libsox">https://github.com/dmkrepo/libsox</a>
- > SOX LIBRARY DOCUMENTATION: <a href="https://fossies.org/dox/sox-14.4.2/libsox\_8c.html">https://fossies.org/dox/sox-14.4.2/libsox\_8c.html</a>
- ➤ MAN PAGES (LINUX) OF "LIBSOX": <a href="https://linux.die.net/man/3/libsox">https://linux.die.net/man/3/libsox</a>
- ➤ LIBSOX TUTORIAL: <a href="https://www.audiosciencereview.com/forum/index.php?threads/howto-sox-audio-tool-as-a-signal-generator.4242/">https://www.audiosciencereview.com/forum/index.php?threads/howto-sox-audio-tool-as-a-signal-generator.4242/</a>

# Thank You!