MUSC 208 Winter 2014 John Ellinger Carleton College

Lab 16 Filters

Lab 16 needs to be done on the iMacs.

The only reason we need to use the iMacs is because of the Signal Scope application that will be used towards the end of the lab to display a real time FFT of the audio that's being played. All code created in Processing and ChucK is cross platform and will work on both Mac and Windows. I'm certain there is an equivalent Windows application for the Mac only SignalScope but you'll have to discover that on your own.

Setup

Download the m208Lab16.zip files.

Overview

Finite Impulse Response (FIR) Filters

Moving Average Filter

Low Pass and High Pass filters

Processing GUI to Experiment with

An audio filter can be thought of as a black box that has an input and an output. Digital samples enter through the input. Some type of processing occurs on a single sample or on a group of samples. The processed samples are then sent to the output ready to be played through a speaker. The output may have a strong resemblance the input or have not resemblance at all.

Lab 16 will look at the two basic digital filter types, the Finite Impulse Response filter and the Infinite Impulse Response filter. The FIR filters depend only on current and previous samples while IIR filters also depend on past output. IIR filters can become unstable if the output is fed back into the input and causes an exponential increase in volume similar to holding a live microphone close to an output speaker.

Create a wave file with a fundamental frequency of 100 Hz and 200 harmonics all with equal amplitude that covers the frequency spectrum from 100 Hz - 20,000 Hz. When we apply a filter to this waveform it will be easy to see the results. Enter and execute this code.

```
// 200sines.ck
// John Ellinger, Music 208, Spring2013
200 => int NUM_HARMONICS;
SinOsc sinRA[ NUM_HARMONICS ]; // array of 200 Sine oscillators
Gain g => dac => WvOut w => blackhole; // no audio output
me.sourceDir() + "/sines20K.wav" => w.wavFilename; // filename to save
100 => float f0; // fundamental frequency
1.0 / NUM_HARMONICS => float amp; // scale amplitude
for (\emptyset \Rightarrow int n; n < sinRA.cap(); n++)
{
    f0 * ( n +1 )=> sinRA[n].freq;
    amp => sinRA[n].gain;
     // send output to Gain object that will mix all SinOsc's together
     sinRA[n] => g;
3
1::second => now; // waveform will be one second long
```

When the program exits, open the sines20k.wav file in Audacity and look at the frequency spectrum.

| 000 | Fr | equency | Analysis | | |
|------------|---------------------------------|-----------|--------------------|-------|---------------|
| -48dB- | | | **** | | |
| -51dB- | | | | | |
| -54dB- | | | | | |
| -57dB- | | | | | |
| -60dB- | | | | | |
| -63dB- | | | | | |
| -66dB- | | | | | |
| -69dB- | | | | | |
| -72dB_ | <u> </u> | | | | |
| 1000 | 0Hz 3000Hz 5000Hz 7000Hz | 10000 | Hz 15 | 000Hz | 20000Hz |
| Curs | sor: 1061 Hz (C6) = -46 dB Peak | c: 1240 H | dz (D#6) = -45.6 d | В | |
| Algorithm: | Spectrum ‡ | Size: | 512 | \$ | Export Replot |
| Function: | Rectangular window \$ | Axis: | Linear frequency | \$ | Close Grids |
| | | | | | |

Finite Impulse Response Filter

FIR filters only depend on current and previous samples. The general form of an FIR filter is:

 $output[n] = a_0 sample[n] + a_1 sample[n-1] + ...a_M sample[n-M]$ or

$$output[n] = \sum_{i=0}^{M} a_i sample[n-i]$$

Moving Average Filter

We'll look at a very simple FIR filter, called the moving average filter. The output of the filter is the average of the current and previous sample.

Low Pass

$$output[n] = 0.5 \cdot sample[n] + 0.5 \cdot sample[n-1]$$

High Pass

$output[n] = 0.5 \cdot sample[n] - 0.5 \cdot sample[n-1]$

Create this code.

```
// movingAverageFilter.ck
// John Ellinger Music 208 Winter2014
```

```
SndBufUtilsClass sbc;
Impulse imp => dac => WvOut w => blackhole;
sbc.init( me.sourceDir() + "/sines20k.wav" );
me.sourceDir() + "/lowPassMovingAverage.wav" => w.wavFilename;
0.5 => float a0;
0.5 => float a1;
```

```
for (\emptyset \Rightarrow int n; n < sbc.lenSamples; n++)
{
     if ( n == 0 )
         a0 * sbc.sampRA[0] => imp.next;
     else // low pass uses + between a0 and a1
          (a0 * sbc.sampRA[n] ) + (a1 * sbc.sampRA[n-1] ) =>
sbc.sampRA[n];
     sbc.sampRA[n] => imp.next;
     1::samp \Rightarrow now;
}
sbc.init( me.sourceDir() + "/sines20k.wav" );
 me.sourceDir() + "/highPassMovingAverage.wav" => w.wavFilename;
0.5 => a0;
0.5 => a1;
for (\emptyset \Rightarrow int n; n < sbc.lenSamples; n++)
ł
     if ( n == 0 )
          a0 * sbc.sampRA[0] => imp.next;
     else // high pass uses - between a0 and a1
          (a0 * sbc.sampRA[n] ) - (a1 * sbc.sampRA[n-1] ) =>
sbc.sampRA[n];
     sbc.sampRA[n] => imp.next;
     1::samp \Rightarrow now;
}
null @=> w;
Run it and you'll get an error.
```



There's an updated version of SndBufUtilsClass.ck in the m208Lab16 folder. There are three ways to run the movingAverageFilter.ck code without causing the error. Try all three.

Method 1. miniAudicle

Stop and restart the miniAudicle Virtual Machine. Run the SndBufUtilsClass.ck first and the movingAverageFilter.ck code second.

Method 2. Terminal

Open Terminal. Set the working directory to the m208Lab16 folder and execute these commands.

```
509,je@jemac:~/Desktop/m208Lab16$ cd ~/Desktop/m208Lab16
510,je@jemac:~/Desktop/m208Lab16$ ls
SndBufUtilsClass.ck movingAverageFilter.ck sines20K.wav
511,je@jemac:~/Desktop/m208Lab16$ chuck SndBufUtilsClass.ck movingAverageFilter.ck
512,je@jemac:~/Desktop/m208Lab16$
```

Method 3. Use the Machine.add() function

Create a new ChucK file called runMovingAverage.ck. Enter and run this code.

```
// runMovingAverage.ck
// John Ellinger Music 208 Winter2014
Machine.add( me.sourceDir() + "/SndBufUtilsClass.ck" );
Machine.add( me.sourceDir() + "/movingAverageFilter.ck" );
Two new wave files should be in the m208Lab16 folder.
512,je@jemac:~/Desktop/m208Lab16$ 1s
SndBufUtilsClass.ck lowPassMovingAverage.wav sines20K.wav
highPassMovingAverage.wav movingAverageFilter.ck
```

Open the two wave files in Audacity and look at their spectrums.

Low Pass Spectrum

The low pass filter allows low frequencies to pass through but attenuates high frequencies.

| 000 | Fro | equency Analysis | | |
|------------|---------------------------------------|----------------------|----------|---------------|
| -48dB- | MAAAAAA | | | |
| -51dB- | | | | |
| -54dB- | · · · · · · · · · · · · · · · · · · · | | | |
| -57dB- | | | * * * * | |
| -60dB- | | | | |
| -63dB- | | | | |
| -66dB- | | | | |
| -69dB- | | | | |
| -72dB- | | | | |
| -75dB- | | | | |
| -78dB- | | | | |
| -81dB- | | | <u> </u> | |
| 1000 | 0Hz 3000Hz 5000Hz 7000Hz | 10000Hz | 15000Hz | 20000Hz |
| Curs | sor: 86 Hz (F2) = -45 dB Peak: 4 | 3 Hz (F1) = -45.9 dB | | |
| Algorithm: | Spectrum + | Size: 512 | \$ | Export Replot |
| Function: | Rectangular window \$ | Axis: Linear frequer | ncy ‡ | Close Grids |
| | | | | 11 |

High Pass Spectrum

The high pass filter allows high frequencies to pass through but attenuates low frequencies.

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More Flexible Low Pass And High Pass Filters

If you vary the coefficients of the moving average filter, the frequency response will change. If you add the samples you'll get a low pass filter. If you subtract the samples you'll get a high pass filter.

```
output[n] = a_0 \cdot sample[n] \pm a_1 \cdot sample[n-1]
where a_0 and a_1 sum to 1.0 to avoid clipping
```

Processing => ChucK => SoundScope

We'll investigate this filter by building a GUI in Processing to manipulate the coefficients, send their values to ChucK where hear the filter effects, and also send the audio to the SoundScope FFT Analyzer application so we can see the results.

 Lab16_FIR

 0.50
 Gain

 0.50
 a0

 Lo - Hi

 0.50
 a1

 Bypass

When completed, the Processing GUI you'll build will look like this.

Open the Processing Application and save your sketch as "Lab16_FIR" in the m208Lab16 folder.

Step 1 Create the Gain Slider

Enter and run this code in Processing.

Imports and global variables.

```
// Lab16_FIR.pde
// MUSC208 John Ellinger Winter2014
import oscP5.*;
import netP5.*;
import controlP5.*;
ControlP5 cp5;
// the controlP5 controls
Slider sliderGain;
```

Next define the Gain control settings.

```
// GUI control layout
void addGUIControls()
{
 // create the font used for for control labels
 // use true/false for anti-aliased/not anti-aliased
 PFont pfont = createFont("Arial", 18, true);
 ControlFont font = new ControlFont (pfont );
 // slider dimensions
 int sliderHt = 20;
                       // slider height
 int sliderWid = 480; // slider width
 // control default colors
 int bkgColor = color(128);
                                       // light gray
  int foreColor = color(255,255,162); // light yellow
  int activeColor = color(255,255,162); // light yellow
   // create slider Gain
 sliderGain = cp5.addSlider( "Gain" )
                                     // 20 pixels in from left, 16 down from top
    .setPosition(20, 16)
    .setSize( sliderWid, sliderHt ) // 480 pixels wide by 20 pixels tall
    .setRange( 0, 1 )
                                     // slider minimum value is 0, max is 1.0
    .setValue( 0.5 )
                                     // initial setting at 0.5
    .setColorCaptionLabel(0)
                                     // caption color is black
                                    // label color is black
    .setColorValueLabel(0)
    .setDecimalPrecision(2)
                                    // display 2 decimal places
    .setColorBackground( bkgColor ) // dark gray
    .setColorForeground( foreColor ) // light yellow
                                   // light yellos
    .setColorActive( activeColor )
    5
 // change control label font and size
  cp5.getController("Gain")
                                  // operate on the Gain control
     .getCaptionLabel()
                                    // get the control label
                                    // set the font to Arial 18
     .setFont(font)
     .toUpperCase(false)
                                    // all upper case is false
                                     // font size is 18
     .setSize(18)
} // end addGUIControls
```

Create the setup() function. The setup function is the first function called by the Processing application. The size() function must be the first line in setup().s

Create the draw() function. Processing calls the draw() function whenever the screen needs updating.

```
void draw()
{
   // width and height are Processing variables passed to
   // the the size() function in in setup()
   rect(0, 0, width, height);
   background( color(192) ); // a gray window background
}
```

Create the controlEvent() function that handles user interaction with the controlP5 controls you've created.

```
void controlEvent(ControlEvent theEvent)
{
    if ( theEvent.isFrom( sliderGain ) )
      {
        float gane = sliderGain.getValue();
        println( "Gain", gane );
    }
}
```

Run the program. You should see this.

| 000 | jeLab16_FIR_pde | |
|------|-----------------|------|
| 0.50 | | Gain |
| | | |
| | | |
| | | |

Move the slider and watch the output in the Processing console.

| งนะท | 0.00000 | | |
|------|------------|--|--|
| Gain | 0.56875 | | |
| Gain | 0.5708334 | | |
| Gain | 0.57708335 | | |
| Gain | 0.5791667 | | |
| Gain | 0.5833334 | | |

Step2 Create the OSC event that will be sent to ChucK

Add these lines of code to the globals section.

```
ControlP5 cp5;
// the controlP5 controls
Slider sliderGain;
// OscP5 is name of class, oscP5 is a variable instance of that class
OscP5 oscP5;
// ChucK network address
// on the same computer it will be "127.0.0.1"
// or "localhost"
NetAddress ChuckAddress;
```

// GUI control layout
void addGUIControls()

Add the OSC setup commands in the setup() function.

Create the function that will send the OSC message

```
void draw()
ł
  // width and height are Processing variables passed to
  // the the size() function in in setup()
  rect(0, 0, width, height);
  background( color(192) ); // a gray window background
3
void sendOSCgain( float gane)
ł
  // the OscMessage class has handy formatting methods
  // genereate the message title ChucK will use to identify
  // the control that sent the message
  OscMessage oscGain = new OscMessage("/lab16/gain");
  // add one float value to the message
 // the parameter gane is the current value of the slider
  oscGain.add( gane );
  //send the message immediately
  oscP5.send(oscGain, ChuckAddress);
void controlEvent(ControlEvent theEvent)
```

```
Modify the controlEvent so the sendOSCgain() function is called when the slider is moved.
```

```
void controlEvent(ControlEvent theEvent)
{
    if ( theEvent.isFrom( sliderGain ) )
    {
      float gane = sliderGain.getValue();
      sendOSCgain( gane );
      println( "OSC sent gain", gane );
    }
}
```

{

Step 3 Create the OSC receive messages in ChucK

Create a new ChucK file named simpleFIR.ck. Enter this code.

Important: Turn down the computer volume. The noise.wav is loud.

```
// simpleFIR.ck
// John Ellinger Music 208 Winter 2014
```

```
Impulse imp; // define but don't connect yet
SndBufUtilsClass sbc;
sbc.init( me.sourceDir() + "/noise.wav" );
// copy the SndBuf samples from the SndBufUtilsClass
// into a local array of samples called ra.
sbc.getSamples() @=> float ra[];
ra.cap() => int numSamps;
// sanity check
<<< "numSamps", numSamps >>>;
// Set the gain to the initial Processing
0.5 \Rightarrow float gane;
// create our OSC receiver
OscRecv recv:
// listen on port 12346 that was defined in Processing
12346 => recv.port;
// start listening
recv.listen();
// listen for the /lab16/gain event followed by one float
recv.event( "/lab16/gain, f" ) @=> OscEvent oeGain;
// The OSC Listening function for sliderGain
function void updateGain()
{
    while ( true )
    {
        // wait for event to arrive
        oeGain => now;
        // grab the next message from the queue
        // it will be one float - the sliderGain value
        while ( oeGain.nextMsg() != 0 )
         {
             oeGain.getFloat() => gane;
             <<< "sliderGain", gane >>>;
        }
    }
}
// spork the OSC listener
spork ~ updateGain();
```

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```
// connect Impulse to dac
imp => dac;
// our counter
0 => int n;
while (true)
{
    if ( n == 0 )
        gane * ra[0] => imp.next;
    else
        gane * ra[n] => imp.next; // low pass
    1::samp => now;
    n++;
    if ( n == numSamps )
        0 => n;
}
```

Create a new file in ChucK that will run both SndBufUtilsClass and the simpleFIR.

```
// runSimpleFIR.ck
```

```
Machine.add( me.sourceDir() + "/SndBufUtilsClass.ck" );
Machine.add( me.sourceDir() + "/simpleFIR.ck" );
```

Step 4 Implement the A0 slider in Processing.

Add the sliderA0 variable to the global variables list. You might as well add the sliderA1 variable at the same time.

```
ControlP5 cp5;

// the controlP5 controls

Slider sliderGain;

Slider sliderA0;

Slider sliderA1;
```

Duplicate (copy/paste) the sliderGain section of of addGUIControls(). Make a few changes.

```
// create slider A0
sliderA0 = cp5.addSlider( "a0" ) // a0 is the control label
  .setPosition(20, XX)
                                 // XX means you figure it out
  .setSize( sliderWid, sliderHt ) // 480 pixels wide by 20 pixels tall
  .setRange(0, 1)
                                 // slider minimum value is 0, max is 1.0
  .setValue( 0.5 )
                                 // initial setting at 0.5
  .setColorCaptionLabel(0)
                                 // caption color is black
                                 // label color is black
  .setColorValueLabel(0)
  .setDecimalPrecision(2)
                                 // display 2 decimal places
  .setColorBackground( bkgColor ) // dark gray
  .setColorForeground( foreColor ) // light yellow
  .setColorActive( activeColor ) // light yellos
// change control label font and size
cp5.getController("a0")
                               // operate on the Gain control
   .getCaptionLabel()
                                 // get the control label
   .setFont(font)
                                 // set the font to Arial 18
   .toUpperCase(false)
                                 // all upper case is false
                                   // font size is 18
   .setSize(18)
   5
```

Step 5 Implement the A1 slider in Processing

Do the same thing to create sliderA1.

Step 6 Create the OSC message function for the two sliders

```
// the sendOSCa0 function will be used for both sliderA0 and sliderA1
// they're linked so that they always sum to 1.0
void sendOSCa0( float a0 )
{
    //the OscMessage class has handy formatting methods
    OscMessage oscA0 = new OscMessage("/lab16/a0"); //greate the address
    oscA0.add( a0 );//append these items
    oscA0.add( 1 - a0 );//append these items
    oscP5.send(oscA0, ChuckAddress);//send the message
}
```

Step 7 Handle the controlEvent when the sliders are moved

```
void controlEvent(ControlEvent theEvent)
{
 if ( theEvent.isFrom( sliderGain ) )
  {
    float gane = sliderGain.getValue();
    sendOSCgain( gane );
   println( "OSC sent gain", gane );
  }
  else if ( theEvent.isFrom( sliderA0 ) )
  {
    float a0 = sliderA0.getValue();
    sliderA1.setValue( 1-a0 );
    sendOSCa0( a0 );
   println( "a0_a0a1", a0, 1-a0 );
 }
 else if ( theEvent.isFrom( sliderA1 ) )
  {
    float a1 = sliderA1.getValue();
    sliderA0.setValue( 1-a1 );
    float a0 = sliderA0.getValue();
    sendOSCa0( a0 );
   println( "a1_a0a1", a0, 1-a0 );
 }
}
```

Step 8 Create the ChucK code to process the OSC events

Create variables for the a0 and a1 filter coefficients and add a new receive event.

```
// Set the gain to the initial Processing
0.5 => float gane;
0.5 => float a0;
0.5 => float a1;
// create our OSC receiver
OscRecv recv;
// listen on port 12346 that was defined in Processing
12346 => recv.port;
// start listening
recv.listen();
// listen for the /lab16/gain event followed by one float
recv.event( "/lab16/gain, f" ) @=> OscEvent oeGain;
// listen for the /lab16/a0 event followed by two floats
recv.event( "/lab16/a0, f f" ) @=> OscEvent oeA0;
```

Add the OSC listening function for the the a0 event and spork it.

```
function void updateSliders()
{
    while ( true )
    {
         // wait for event to arrive
         oeA0 \implies now;
         // grab the next message from the queue.
         // it will be two floats - the values of a0 and a1
         while ( oeA0.nextMsg() != 0 )
         £
             oeA0.getFloat() => a0;
             oeA0.getFloat() => a1;
             <<< "a0", a0, "\ta1", a1 >>>;
         }
    }
3
// spork the OSC listener
spork ~ updateGain();
spork ~ updateSliders();
// connect Impulse to dac
imp \Rightarrow dac;
```

Write the code for the simpleFIR filter.

```
// connect Impulse to dac
imp => dac;
// our counter
0 => int n;
while (true)
{
    if ( n == 0 )
        gane * ra[0] => imp.next;
    else
        gane * ( a0 * ra[n] + a1 * ra[n-1] ) => imp.next; // low pass
    1::samp => now;
    n++;
    if ( n == numSamps )
        0 => n;
}
```

Run the Processing Controls GUI and then runSimpleFIR.ck. Move the sliders and listen for changes in the sound. I did not hear much difference with the low pass filter but it might just be aging ears. I definitely heard a difference with the high pass filter that we'll do next. Stop ChucK.

Change one line to create the high pass filter and run the programs again.

```
if ( n == 0 )
    gane * ra[0] => imp.next;
else
    gane * ( a0 * ra[n] - a1 * ra[n-1] ) => imp.next; // high pass
```

Step 9 GUI Enhancemets

We'll add a two Toggle controls to our Processing GUI, one to switch between low pass and high pass filters and one to bypass the filter and play the original samples.

| 000 | jeLab16_FIR_pde | | |
|------|-----------------|----------|---|
| 0.50 | | Gain | |
| 0.50 | | a0 Lo-Hi | |
| 0.50 | | a1 | |
| | | Bypass | ; |

Add these variables to the global controls.

ControlP5 cp5; // the controlP5 controls Slider sliderGain; Slider sliderA0; Slider sliderA1; Toggle lohiToggle; Toggle bypassToggle;

Add this code to the addGUIControls function. First the Lo-Hi switch.

```
// create a toggle
 lohiToggle =cp5.addToggle("Lo - Hi")
    .setPosition(XX,YY) // XX YY means you figure out placement in window
    .setSize(XX,YY)
                        // XX YY means you figure out width height in window
    .setValue(true)
    .setMode(ControlP5.SWITCH) // left/right switch mode
    .setColorBackground(color(128))
    .setColorForeground(color(255,255,162))
    .setColorActive(color(255,255,162))
  cp5.getController("Lo - Hi")
    .getCaptionLabel()
    .setFont(font)
    .toUpperCase(false)
    .setSize(XX) // XX means you figure out font size
    .setColor(0)
    5
```

Then the Bypass button.

```
// create a toggle
 bypassToggle =cp5.addToggle("Bypass")
     .setPosition(XX,YY) // XX YY means you figure out placement in window
                        // XX YY means you figure out width height in window
     .setSize(XX,YY)
     .setValue(false)
     .setColorBackground(color(128))
     .setColorForeground(color(128))
     .setColorActive(color(255,255,162))
   cp5.getController("Bypass")
     .getCaptionLabel()
     .setFont(font)
     .toUpperCase(false)
     .setSize(XX) // XX means you figure out font size
     .setColor(0)
} // end addGUIControls
```

Add the OSC message functions.

```
void sendOSCloHi( float lo )
ł
  //the OscMessage class has handy formatting methods
  OscMessage oscLoHi = new OscMessage("/lab16/lowhi"); //greate the address
  if ( lo == 1.0 )
    oscLoHi.add( 1.0 ); // plus for low pass
   else
    oscLoHi.add( -1.0 ); // minus for high pass
 oscP5.send(oscLoHi, ChuckAddress);//send the message
3
void sendOSCbypass( float byp )
₹.
  //the OscMessage class has handy formatting methods
  OscMessage oscBypass = new OscMessage("/lab16/bypass"); //greate the address
  if ( byp == 1.0 )
    oscBypass.add( 1.0 );//append these items
  else
    oscBypass.add( 0.0 );//append these items
   oscP5.send(oscBypass, ChuckAddress);//send the message
3
```

Add the control event handling code.

```
void controlEvent(ControlEvent theEvent)
ł
  if ( theEvent.isFrom( sliderGain ) )
  ł
    float gane = sliderGain.getValue();
    sendOSCgain( gane );
    println( "OSC sent gain", gane );
  }
  else if ( theEvent.isFrom( sliderA0 ) )
  {
    float a0 = sliderA0.getValue();
    sliderA1.setValue( 1-a0 );
    sendOSCa0( a0 );
   println( "a0_a0a1", a0, 1-a0 );
  }
  else if ( theEvent.isFrom( sliderA1 ) )
  {
    float a1 = sliderA1.getValue();
    sliderA0.setValue( 1-a1 );
    float a0 = sliderA0.getValue();
    sendOSCa0( a0 );
    println( "a1_a0a1", a0, 1-a0 );
  3
 else if ( theEvent.isFrom( lohiToggle ) )
  £
    float lo = lohiToggle.getValue();
    sendOSCloHi( lo );
    println( "lohi ", lo );
  3
  else if ( theEvent.isFrom( bypassToggle ) )
  Ł
    float byp = bypassToggle.getValue();
    sendOSCbypass( byp );
    println( "bypass ", byp );
 }
```

Test the controls in Processing.

Step 10 Create the ChucK code to process the OSC events

Create variables for the bypass and low high buttons.

// Set the gain to the initial Processing 0.5 => float gane; 0.5 => float a0; 0.5 => float a1; 1.0 => float lowhi; 0.0 => float bypass;

Add a new receive events.

```
// listen for the /lab16/gain event followed by one float
recv.event( "/lab16/gain, f" ) @=> OscEvent oeGain;
// listen for the /lab16/a0 event followed by two floats
recv.event( "/lab16/a0, f f" ) @=> OscEvent oeA0;
recv.event( "/lab16/lowhi, f" ) @=> OscEvent oeLowHi;
recv.event( "/lab16/bypass, f" ) @=> OscEvent oeBypass;
```

Add the OSC listening function for the the low high event and spork it. function void updateLowPassHiPass()

```
{
    while ( true )
    {
        // wait for event to arrive
        oeLowHi => now;
        // grab the next message from the queue.
        while ( oeLowHi.nextMsq() != 0 )
        {
            oeLowHi.getFloat() => lowhi;
            if ( lowhi == 1.0 )
                <<< "low pass", lowhi >>>;
            else if ( lowhi == -1.0 )
                 <<< "high pass", lowhi >>>;
            else
                <<< "bad pass" >>>;
        }
    }
// spork the OSC listener
spork ~ updateGain();
spork ~ updateSliders();
spork ~ updateLowPassHiPass();
```

Add the OSC listening function for the the bypass event and spork it.

```
function void updateBypass()
{
    while ( true )
    {
        // wait for event to arrive
        oeBypass => now;
        // grab the next message from the queue.
        while ( oeBypass.nextMsg() != 0 )
        {
            oeBypass.getFloat() => bypass;
            <<< "bypass", bypass >>>;
        }
    }
}
// spork the OSC listener
spork ~ updateGain();
spork ~ updateSliders();
spork ~ updateLowPassHiPass();
spork ~ updateBypass();
```

Update the code for the simpleFIR filter.

```
// connect Impulse to dac
imp \Rightarrow dac;
// our counter
0 => int n;
while (true)
{
    if ( n == 0 )
         gane * ra[0] => imp.next;
    else
         gane * ( (a0 * ra[n] ) + lowhi * (a1 * ra[n-1] ) ) => imp.next;
    if ( bypass )
         gane * ra[n] => imp.next;
    1:::samp \Rightarrow now;
    n++;
    if ( n == numSamps )
         0 => n;
}
```

Run the Processing Controls GUI and then runSimpleFIR.ck. Test the Low Pass High Pass switch and the Bypass button.

Step 11 Display the real time FFT in Signal Scope

First Open /Applications/Audio MIDI Setup. Use the Window menu to display the Audio Devices window. We're going to create a Multiple Output Device so we can send ChucK audio through SoundFlower to SignalScope and be able to hear the audio at the same time.

Click the plus sign at the bottom of the Audio Devices window and choose Create Multi-Output Device.

| 0 | 0 | | | | uis | | | | i | | | | | | | | | | 1 | | | | | | | | | | | A | uc | di | 0 | 0 | De | 21 | vi | c | e | 25 | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | |
|------|-------------------------------------|----------|-------|------------|-----|-----|----|-----|----|----|---|---|---|---|----|---|---|---|---|---|---|----|---|---|----|----|----|----|-------------|-----|-----|----|---|---|----|----|----|----|---|----|---|---|---|----|---|---|----|----|----|---|----|----|---|---|----|----|---|---|---|---|---|---|---|---|---|----|----|---|----|---|---|----|----|---|----|-----|---|---|
| | Built-in Microphone 2 in/ 0 out | 2 | Buil | t-in Mi | lic | icr | cr | ici | C | c | • | i | i | i | ic | • | • | c | c | | r | 1 | c | • | 0 | pp | pł | ho | or | ıe | • | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | |
| H | Built-in Input 2 in/ 0 out | Ŷ | Cloc | k sourc | rce | ce: | e | ce | e | :e | | 0 | C | 5 | | | • | e | e | e | | | | | 1 | D | De | ef | aı | ult | t | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | (| ? |) | |
| Ŕ | Built-in Output 0 in/ 2 out | (| | | | _ | | _ | | | | | | | | | | | | | | | | | | | | | _ | _ | _ | | | | | | | | | | | | | lı | n | 2 | וכ | u | 11 | t | | | | C | Di | ut | p | u | t | } | _ | | _ | _ | _ | _ | | | | - | | | - | | | _ | | |
| | Soundflower (2ch) 2 in/ 2 out | | S | ource: | Ir | In | In | In | Ir | h | 1 | 1 | 1 | 1 | 1 | 1 | 1 | h | I | r | n | ıt | t | t | te | eı | rr | na | al | m | nic | cr | 0 | p | h | 0 | or | 16 | e | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | |
| | Soundflower (64ch) 64 in/ 64 out | | Fo | ormat: | 4 | 4 | 4 | 4 | 4 | | | | [| | | | [| • | • | 4 | 4 | 4 | 4 | 4 | 4 | 1 | 10 | 00 |) .(| 0 | н | z | | Ŧ | - | | (| 2 | 2 | cl | h | | 2 | 2 | 4 | b | bi | it | t | 1 | In | ۱t | e | c | je | er | _ | _ | _ | _ | _ | _ | _ | _ | _ | _ | _ | _ | _ | | _ | | | | ÷ |) | | |
| | | | Ch | Volu | um | ime | ne | m | m | m | m | n | n | n | n | n | m | r | n | n | e | e | | | | _ | | | | _ | | | | | | | | | | | | | | | | | | | | | _ | | _ | | _ | | | _ | | | | | _ | _ | | Va | lu | 2 | dB | , | | Mu | te | T | Th | nru | , | 1 |
| | | | Maste | r — | | _ | _ | - | | | | | | | | | | | | | | | | | | | _ | _ | _ | - | _ | _ | | _ | | | | | | | | 0 |) |) | | | | | | | | | | | | _ | _ | | | _ | _ | _ | - | - | - | | 0. | 5 | | 0 | 0 | | | (| | | | |
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| | | | 2 | \bigcirc | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | - | | | | | | 6 | | | | | | | |
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| + - | ÷ | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | |
| Crea | ate Aggregate Devi | ce | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | |
| Crea | ate Multi-Output D | evice | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | | |

Next Select the Multi-Output Device in the left panel and click the pop-up menu that looks like a gear and choose Use this device for sound input.

| 0 | 0 | | Audio Devices | |
|----------|---|---------------------|--|----------------------------|
| Ð | Built-in Microphone 2 in/ 0 out | | Multi-Output Device | |
| | Built-in Input 2 in/ 0 out | Ŷ | Master Device: Soundflower (2ch) | ? |
| de la | Built-in Output 0 in/ 2 out | Ð | Devices in this multi-output group will output audio simultaneously. Addit | ional devices may be added |
| | Soundflower (2ch) 2 in/ 2 out | | Use Audio Device | Drift Correction |
| | Soundflower (64ch) 64 in/ 64 out | | ✓ Built-in Output ✓ Soundflower (2ch) | |
| V | Multi-Output Device 0 in/ 2 out | 1 | Soundflower (64ch) | |
| | Built-in Output 0 in/ 2 out | \sim | | |
| | Soundflower (2ch) 2 in/ 2 out | ٥ | | |
| + - | - ☆ ▼ | | | Configure Speakers |
| | Configure device Configure speakers | | | |
| | Use this device for | or sound | d input | |
| | Use this device for Play alerts and so | or sound und eff | d output fects through this device | |

Choose Soundflower 2(ch) as the Master device and check the Built-in Output and Soundflower as the Audio devices. Check Drift Compensation for Built-in Output.

| 00 | 0 | | | | Audio | Devices | | |
|-----|-------------------------------------|----------|-----------|------------------------------|---------------------|-----------------------------|-------------------|-------------------------------|
| ÷ | Built-in Microphone 2 in/ 0 out | | Mul | i-Output | Device | | _ | |
| ₽ | Built-in Input 2 in/ 0 out | ŧ | Mas | er Device: | Soundflo | ower (2ch) 🛟 | | ? |
| D | Built-in Output 0 in/ 2 out | - | Sa Dev | mple Rate: ices in this n | 44100.0 | Hz 💌 | ultaneously. Add | litional devices may be added |
| | Soundflower (2ch) 2 in/ 2 out | | to t | nis device gr | oup by clicki | ng the corresponding buttor | 1 in the Use colu | Imn below. |
| | Soundflower (64ch) 64 in/ 64 out | | | Built-in (Soundflo | output wer (2ch) | | | |
| V | Multi-Output Device 0 in/ 2 out | (| C | Soundflo | wer (64ch) | I | | |
| | Built-in Output | _ | | | | | | |
| | 0 in/ 2 out | N. | | | | | | |
| | Soundflower (2ch) 2 in/ 2 out | ٥ | | | | | | |
| + - | - & - | | | | | | | Configure Speakers |

Open miniAudicle Preferences and choose the Multi-Ouput Device for Audio output.

| O O O Preferences |
|--|
| Audio Editing ChuGins Miscellaneous |
| Enable audio Accept network VM commands |
| Audio output: Apple, Inc.: Multi-Output Device 🗘 |
| Audio input: Apple Inc.: Built-in Input \$ |
| Output channels: 2 ‡ Input channels: 2 ‡ |
| Sample rate: 44100 ‡ Buffer size: 256 ‡ |
| Probe Audio Interfaces |
| Virtual machine stall timeout: 2.0 seconds |
| Changes will not take effect until the virtual machine is restarted. |
| Restore defaults Cancel OK |

Quit and restart miniAudicle.

Run the Processing GUI.

Run simpleFIR.ck.

Open /Applications/SignalScope

Click the Try It button

| ● ○ ○ Product | t Activation Required |
|---|------------------------------|
| | |
| Some features are disal SignalScope. | bled in the trial version of |
| Days remaining: | 28 |
| Buy on the Mac A | pp Store |
| Buy Now | |
| Enter Serial Nu | mber Try It |

Click the New button

| ● ○ ○ SignalScope | 2 |
|---|-----------------|
| | Recent Projects |
| Welcome to SignalScope Version 3.1.6 (1348) | |
| Create a new project | |
| Open Other | New |

Click the Plus sign to the left of FFT Analizer and choose Soundflower as the input device.

| 000 | untitled.sproproj |
|---------------------------------|---|
| Device IO | Start Capture Export Tools |
| Input Device: Soundflower (2ch) | ÷ |
| Project Toolbox | |
| 🛨 🛛 😡 FFT Analyzer | Start All Hide All Capture All Export All |
| ─ ► , FFT Analyzer | Start Hide Capture Export Arm |
| 🛨 🔻 ญ Oscilloscope | |
| 🕂 🔻 🌐 Spectrogram | |
| + 🔻 😿 XY Plotter | |
| | |
| | |
| | |
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| | |

|) | 0 0 | | | | FFT Ar | nalyzer | | | | |
|------|------------|-------------------|--------------|--------|-----------|------------|-----------|---------|-----------------------|-------------------------------------|
| ~ | | | \mathbf{O} | | • | C | + 🕲 🖞 | Ð | 0 | Signal Measurements: |
| 1000 | 0.0 | | | | | | | | | dF: 10.000 Hz FFT Size: 4410 |
| | -10.0 | | | | | | | | | Avgs: 0 |
| | -20.0 | | | | | | | | ╂───╂ | Analyzer 1: 25.000 kV peak (abs) |
| | -30.0 | | | | | | | +++++ | ╂───╂ | 0.000 V mis |
| | -40.0 | | | | | | | | $\parallel \parallel$ | |
| | -50.0 | | | ++- | | | | | \parallel | |
| | -60.0 | | +++ | | | | | | ╂──╂ | |
| | -70.0 | | +++ | | | | | | ╂──┤ | |
| | -80.0 | | | | | | | | ╂∦ | |
| | -90.0 | | | | | | | | $\parallel $ | |
| | -100.0 | 20 | 50 | 100 | 200 | 500 1 | k 2k | 5k | 10k 20k | Options |
| | | | Live Inpu | ts FF1 | Displ | ay Cursors | Triggerir | Ig | | |
| I | nput De | evice: Sound | flower (20 | :h) | | - | | | + | shu campPA[n]: |
| ſ | Cha Ana | annei alvzer 1 | | Input | Inpi 1 | ut 🛓 | Trigger 1 | Trigger | : - | > SDU.SOMPKALNJ; |
| | | | | mpar | - | | | | • | |
| | | | | | | | | | | |

The FFT Analyzer window will appear. Click the Run button.

FFT Tab

| Live Inputs | FFT Display | Cursors Triggering |
|--------------------------|-----------------|------------------------------|
| Frequency Span: | Spectral Lines: | Guardbanding |
| 22050.0 Hz \$ | | 442 🔻 🗌 Digital Antialiasing |
| Window Type: | Overlap: | |
| Uniform (Rectangular) \$ | | 0.0% |
| Average Type: | Averages: | |
| Exponential \$ | | 25 |

Display Tab

| (| Live Inputs FFT Display Cursors Triggering | |
|---|---|-------|
| Vertical Scale Type: dB Apply changes in | Level: Frequency Scale Type: Peak the table to all rows Frequency Scale Type: | |
| Channel | Draw Freeze Vertical Scale Max Value | Auto |
| 📕 Analyzer 1 | ☑ | Off 🗘 |
| | | |

Cursors Tab

| | Live Inputs FFT | Display Cursors | s Triggering | |
|------------------|-------------------|-----------------|--------------|----|
| Ho | orizontal Bar: | | Peak Track: | |
| 🗹 Cursor 1 🛛 🗍 | Analyzer 1 | \$ | Analyzer 1 | \$ |
| armonic Cursors | : 0ff = | | | |
| 🗹 Cursor 2 🛛 📝 | Analyzer 1 | \$ | Off | \$ |
| Harmonic Cursors | Off 1 | | | |

Triggering Tab

| | (| Live Inputs | FFT Display C | ursors Trig | gering | | | | | |
|---------------------|--|-------------|---------------|-------------|--------|-----|--|--|--|--|
| 🗌 Ena | Enable Triggering Apply changes in the table to all rows | | | | | | | | | |
| | Trigger | Source | Threshold | Slope | Mode | | | | | |
| ✓ | Trigger 1 | Input 1 | 0.000 V | Positive | ‡ Auto | ÷ — | | | | |
| | | | | | | | | | | |
| | | | | | | | | | | |
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GUI Settings 0.5 0.5 0.5



000 FFT Analyzer + 🕲 🗨 1 Signal Measurements: dF: 50.000 Hz -120.0 FFT Size: 882 Avgs: 25 -122.0 Analyzer 1: 44.712 μV peak (abs) 4.626 μV rms -124.0 -126.0 ₿ -128.0 -130.0 -134.0 -136.0 -138.0 -140.0 0.0k Options 4.4k 13.2k 17.6k 22.1k 8.8k Frequency (Hz) pdf 45 Live Inputs FFT Display Cursors Enable Triggering Apply changes in the table to all rows + Trigger Threshold Mode Source Slope • 0.000 V Trigger 1 Auto ÷ sndL 294 x 5

Freeze or Stop Low Pass Picture

High Pass

Settings 0.5 0.5 0.5



| 0 | 00 | | FFT Anal | lyzer | | | |
|-------|-------------|--------------|--------------|----------------|----------|-------|-------------------------------------|
| 6 | | | | + 🕲 | € | 1 | Signal Measurements: |
| | -120.0 | | | | | _ | dF: 50.000 Hz FFT Size: 882 |
| | -122.0 | | | | | | Avgs: 25 |
| | -124.0 | | | | | | Analyzer 1: 42.980 µV peak (abs) |
| | -126.0 | | | | | | 4.201 µV rms Cursor 1: |
| BV) | -128.0 | | | | | | X: 19.900 kHz Y: -124.2 dBV pk |
| de (d | -130.0 | | | | | | Cursor 2: X: 10.800 kHz |
| gnitu | -132.0 | | | | | | Y: -127.3 dBV pk Cursor Delta: |
| Ma | -124.0 | | | | | | df: -9.100 kHz dy: 3.079 dBV pk |
| | -134.0 | | | | | | |
| | -136.0 | | | | | | |
| | -138.0 | | | | | | |
| | -140.0 0.0k | 4.4k | 8.8k Fred | 13.2k | 17.6k | 22.1k | Options |
| | | | riequ | | | | pd1 45 |
| 1 | | Live Inputs | FFT Display | Cursors Trigge | ring | | |
| | Cursor 1 | Analyzer 1 | | Analyze | r 1 | \$ | |
| | Harmonic Cu | rsors: Off ‡ | | | | | |
| | Cursor 2 | Analyzer 1 | | ¢ Off | | ÷ | sndl 294 x 5 |
| | Harmonic Cu | rsors: Off + | | | | | |
| L | | | | | | | |

High Pass Picture

Experiment with other a0 and a1 settings.

Group 70 30

| 00 | Lab16filters | |
|------|--------------|------------|
| 0.50 | | Gain |
| 0.70 | | a0 Lo - Hi |
| 0.30 | | a1 |



High Pass

m208w2014





Using OSC in Chuck

http://booki.flossmanuals.net/chuck/_full/

by Rebecca Fiebrink

To send OSC

Host Decide on a host to send the messages to. E.g., "splash.local" if

sending to computer named "Splash," or "localhost" to send to the same machine that is sending.

Port Decide on a port to which the messages will be sent. This is an integer, like 1234.

Message "address" For each type of message you're sending, decide on a way to identify this type of message, formatted like a web URL e.g., "conductor/downbeat/beat1" or "Rebecca/message1"

Message contents Decide on whether the message will contain data, which can be 0 or more ints, floats, strings, or any combination of them.

To set up a OSC sender in ChucK you'll need code like the following:

//Create an OscSend object: OscSend xmit; //Set the host and port of this object: xmit.setHost("localhost", 1234); For every message you want to send, start the message by supplying the address and format of contents, where "f" stands for float, "i" stands for int, and "s" stands for string:

```
//To send a message with no contents:
xmit.startMsg("conductor/downbeat");
//To send a message with one integer:
xmit.startMsg("conductor/downbeat, i");
//To send a message with a float, an int, and another float:
xmit.startMsg("conductor/downbeat, f, i, f");
For every piece of information in the contents of each message, add this
information to the message:
```

```
//to add an int:
xmit.addInt(10);
//to add a float:
xmit.addFloat(10.);
//to add a string:
xmit.addString("abc");
Once all parts of the message have been added, the message will
```

automatically be sent.

To receive OSC

Decide what port to listen on. This must be the same as the port number of the sender(s) you want to listener to receive messages from. Message address and format of contents: This must also be the same as what the sender is using; i.e., the same as in the sender's startMsg function.

The following code shows how to setting up an OSC receiver with ChucK.

```
//Create an OscRecv object:
OscRecv orec;
//Tell the OscRecv object the port:
1234 => orec.port;
//Tell the OscRecv object to start listening for OSC messages on
that port:
orec.listen();
For each type of message, create an event that will be used to wait on that
type of message, using the same argument as the sender's startMsg
function:
```

```
orec.event("conductor/downbeat, i") @=> OscEvent myDownbeat;
To wait on an OSC message that matches the message type used for a
particular event e, do
```

```
e => now;This is just like waiting for regular Events in ChucK.
```

To process the message first it's necessary to grab the message out of the queue. In our example this can be achieved using e.nextMsg(). After we called this, we can use other methods on e to get the information we're interested in out of the message. We must call these functions in order, according to the formatting string we set up above.

```
e.getInt() => int i;
e.getFloat() => float f;
e.getString() => string s;
If you expect you may receive more than one message for an event at once,
```

you should process every message waiting in the cue:

```
while (e.nextMsg() != 0) {
   //process message here (no need to call nextMsg again
}
```