

A stylized audio waveform is visible in the background. The top portion of the waveform is blue, while the bottom portion, which is more prominent and larger, is a reddish-pink color. The waveform consists of many sharp, vertical spikes of varying heights.

# Build Your First Audio Plug-in with JUCE

19/11/2021

The JUCE Team -  
Attila Szarvas, Ed Davies, Reuben Thomas, Tom Poole

# Overview

What is JUCE?

Creating JUCE-based projects

Building audio plug-ins

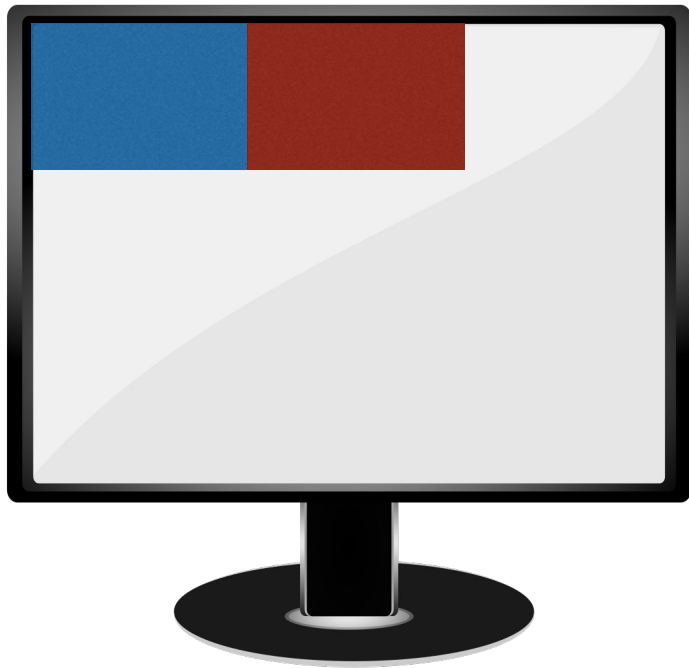
Testing audio plug-ins





```
{  
    setColour (blue);  
    drawRect (0, 0, 100, 50);  
  
    setColour (red);  
    drawRect (100, 0, 100, 50);  
}
```

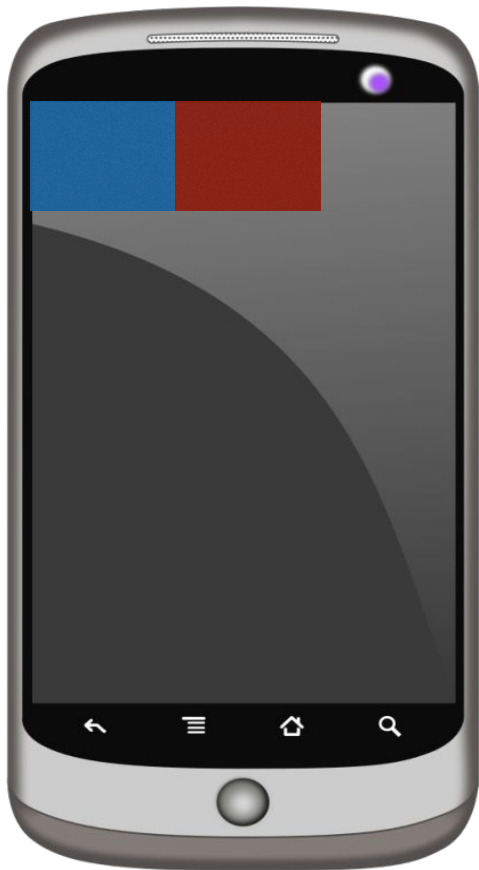




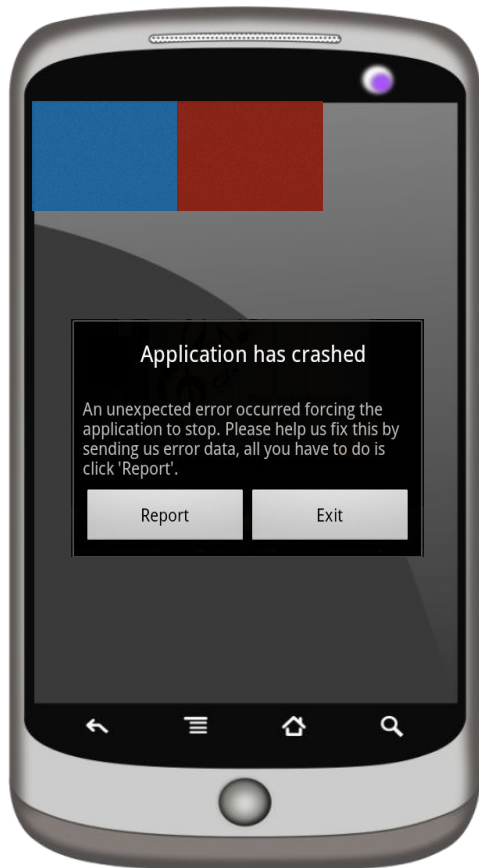
```
{  
    setColour (blue);  
    drawRect (0, 0, 100, 50);  
  
    setColour (red);  
    drawRect (100, 0, 100, 50);  
}
```



```
{  
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    drawRect (0, 0, 100, 50);  
  
    setColour (red);  
    drawRect (100, 0, 100, 50);  
}
```

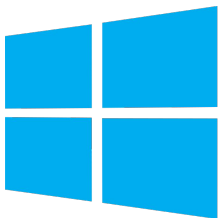


```
{  
    setColour (blue);  
    drawRect (0, 0, 100, 50);  
  
    setColour (red);  
    drawRect (100, 0, 100, 50);  
}
```



```
{  
    setColour (blue);  
    drawRect (0, 0, 100, 50);  
  
    setColour (red);  
    drawRect (100, 0, 100, 50);  
}
```

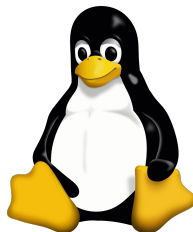
C++



C++



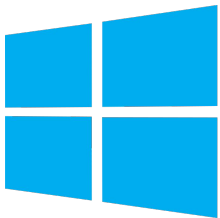
C++



Obj-C++



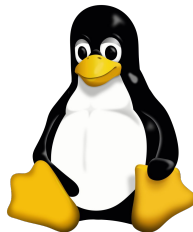
Win32



Native  
Activity



POSIX/  
X11



AppKit/UIKit



## macOS/iOS

```
{
    CGRect rc = { 0, 0, 100, 50 };
    CGContextSetRGBFillColor (context, 0.0f, 0.0f, 1.0f);
    CGContextFillRect (context, &rc);

    rc = { 100, 0, 100, 50 };
    CGContextSetRGBFillColor (context, 1.0f, 0.0f, 0.0f);
    CGContextFillRect (context, &rc);
}
```

# Windows

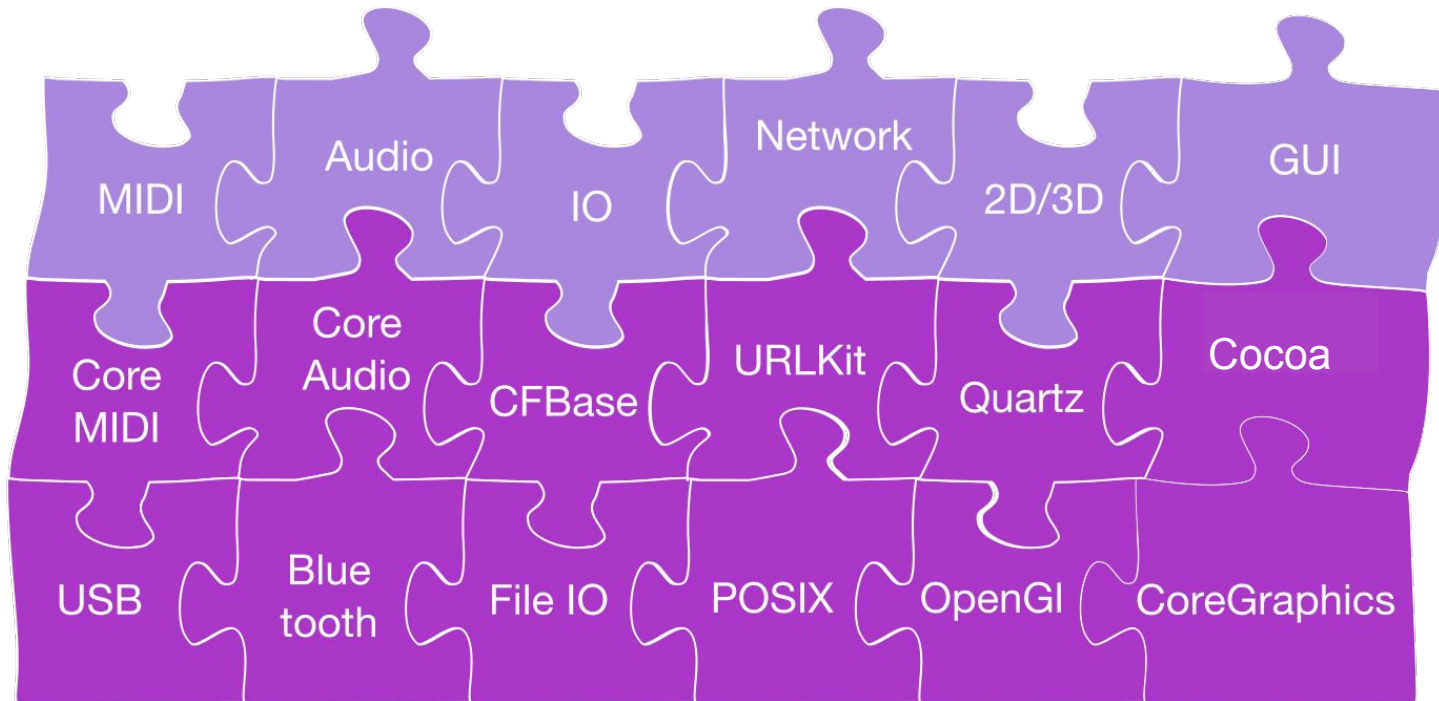
```
{
    Rect rc = { 0, 0, 100, 50 };
    FillRect (context, &rc, RGB (0, 0, 255));

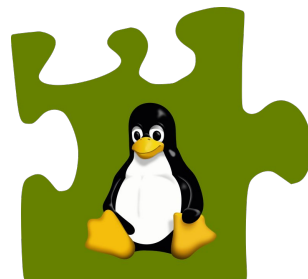
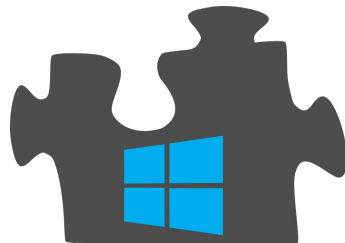
    rc = { 100, 0, 100, 50};
    FillRect (context, &rc, RGB (255, 0, 0));
}
```

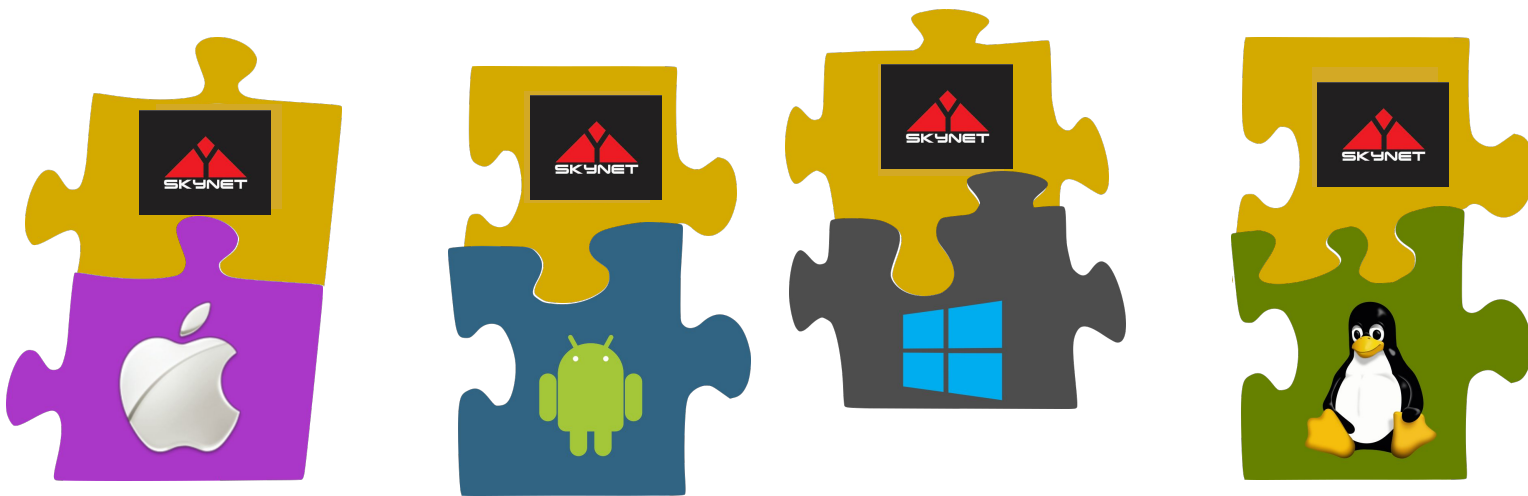


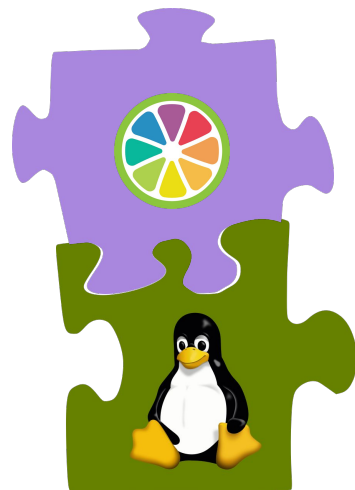
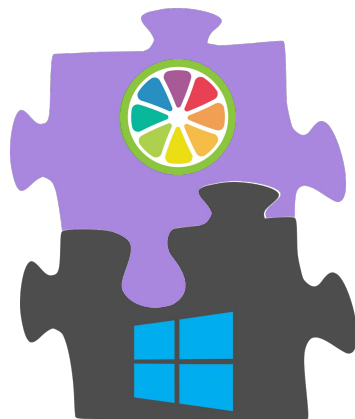
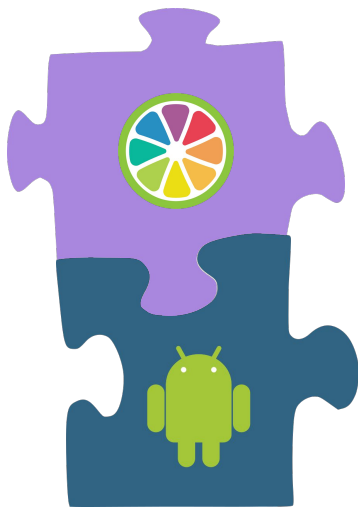


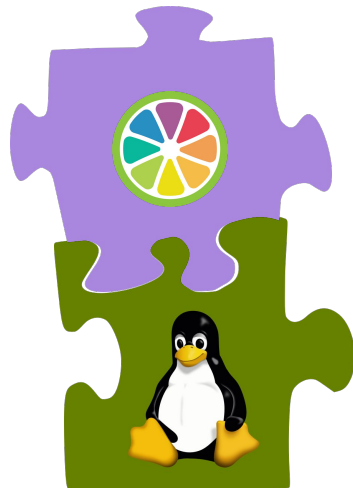
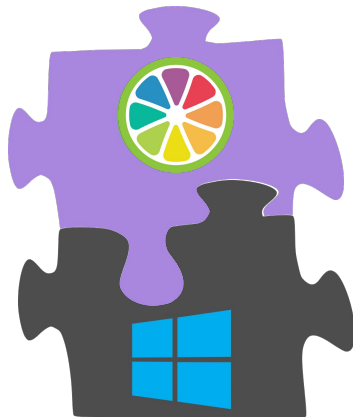


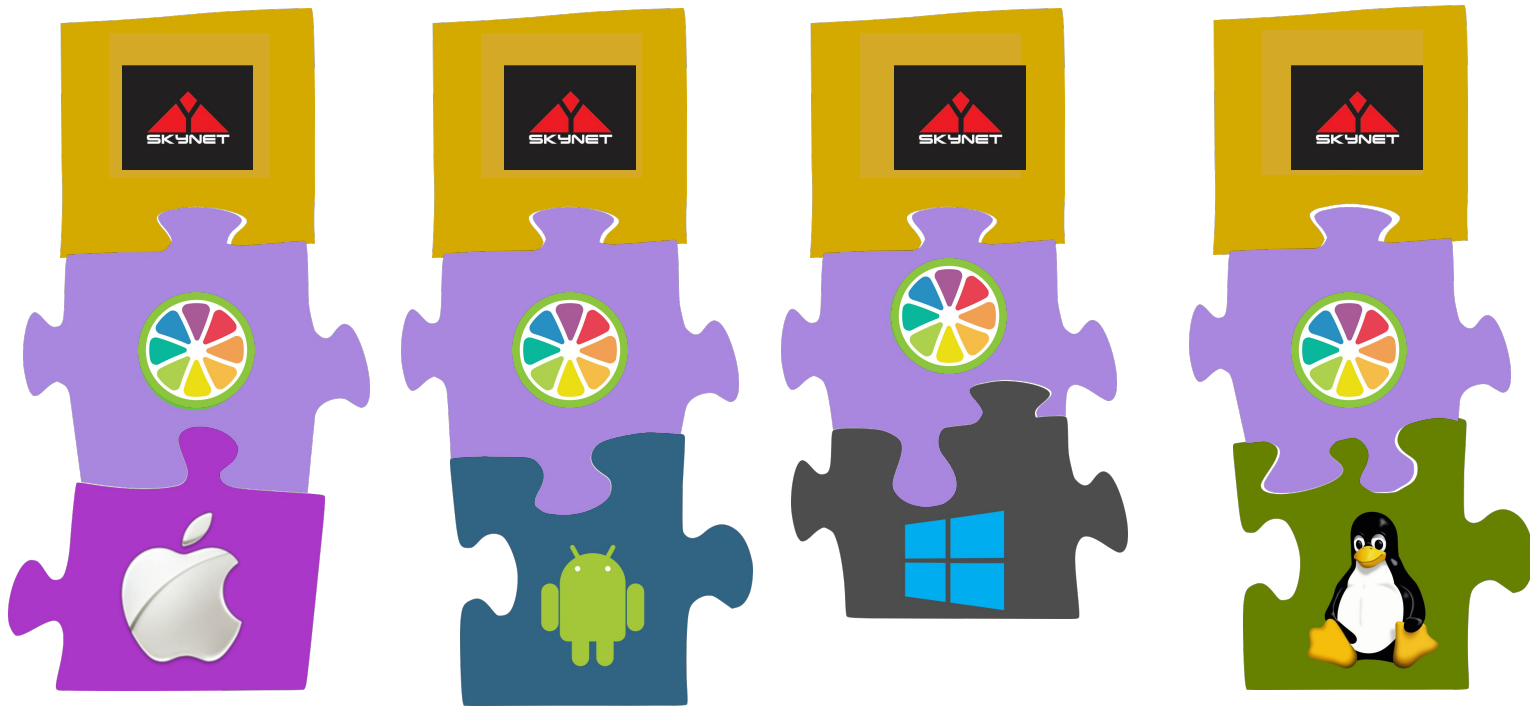








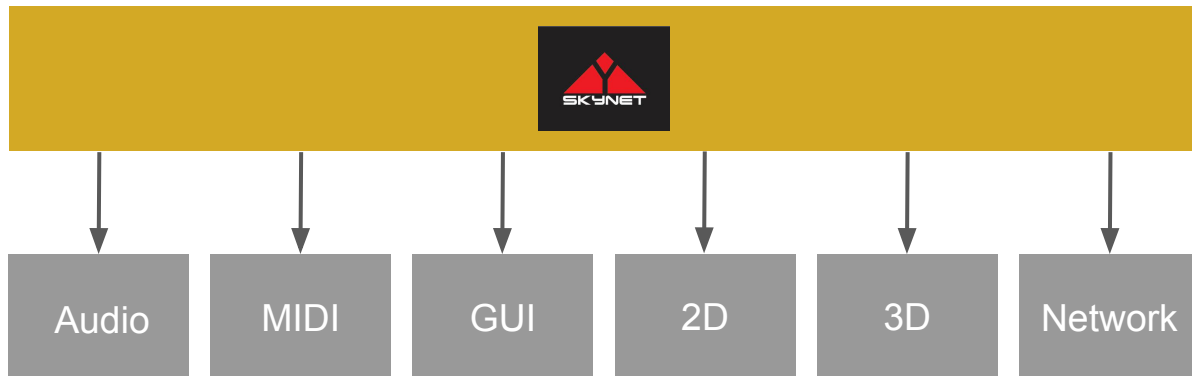


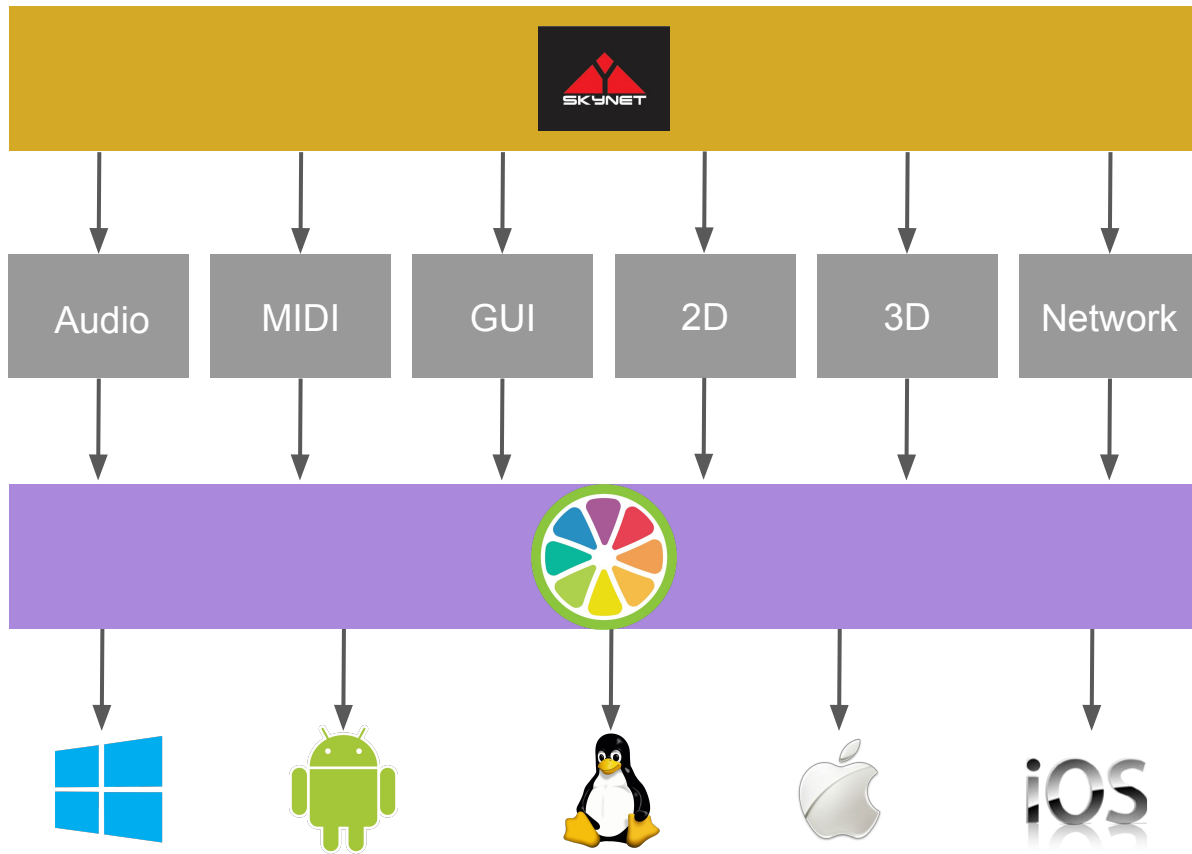


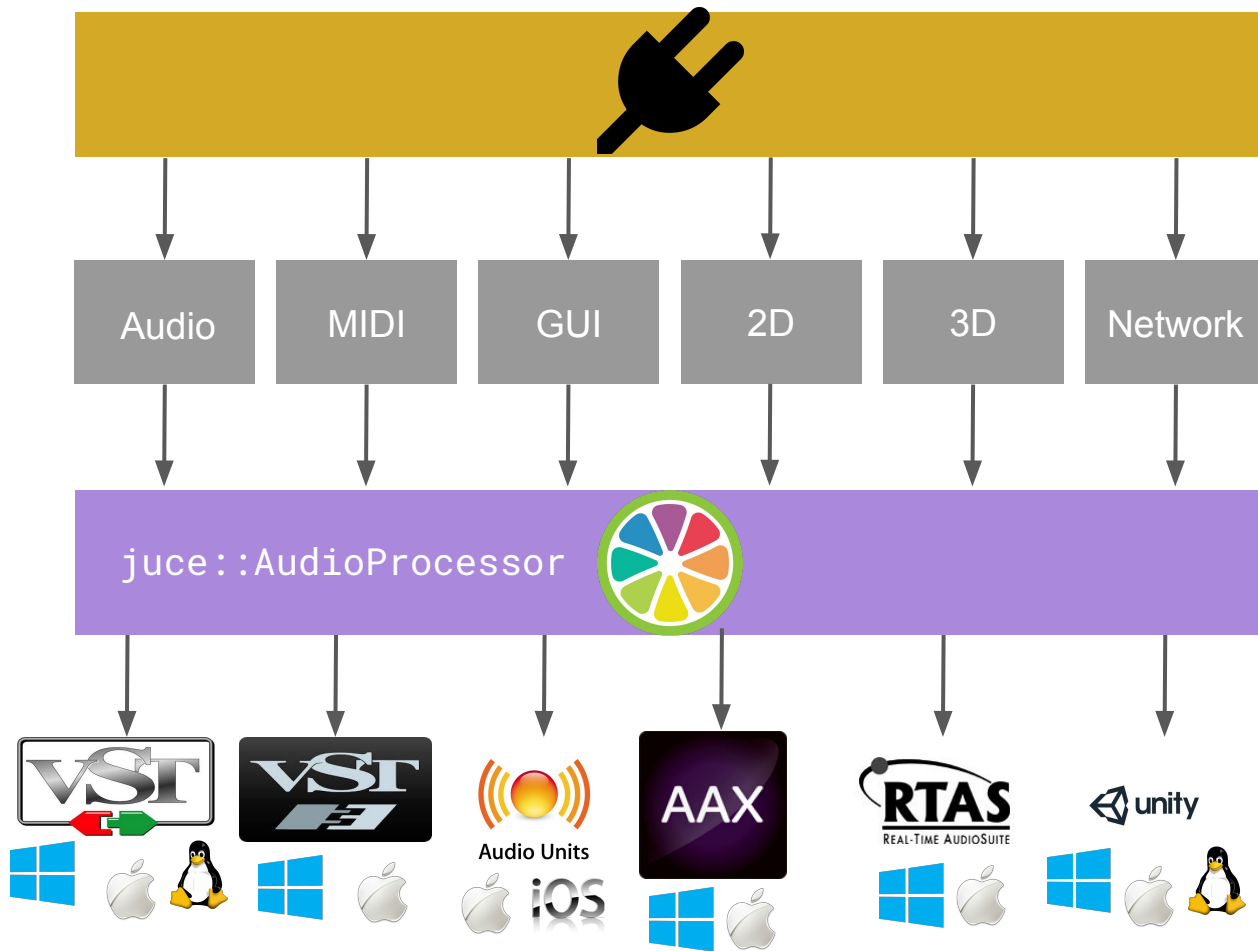


```
{
    g.setColour (juce::Colours::blue);
    g.drawRect (0, 0, 100, 50);

    g.setColour (juce::Colours::red);
    g.drawRect (100, 0, 100, 50);
}
```







# Building an Audio Plug-in

# During the workshop

- All the slides are numbered
- Post questions in the Discord channel
- We'll take a couple of breaks during the session
- The slides are available in the workshop materials, so you can revisit sections

# Creating a plug-in using JUCE

We're going to walk through the creation of a simple delay effect plug-in

- Setting up a JUCE project using the Projucer
- Writing the C++ plug-in code in an IDE
- Compiling the different plug-in formats
- Adding plug-in parameters
- Creating a GUI
- Testing the plug-in

# What won't be covered (but you may want to look into)

- Plug-in configurations - multibus, surround
- Mobile platforms
- Performance optimisations
  - Profiling
  - Compiler settings
- Release process
  - Installers
  - Code signing

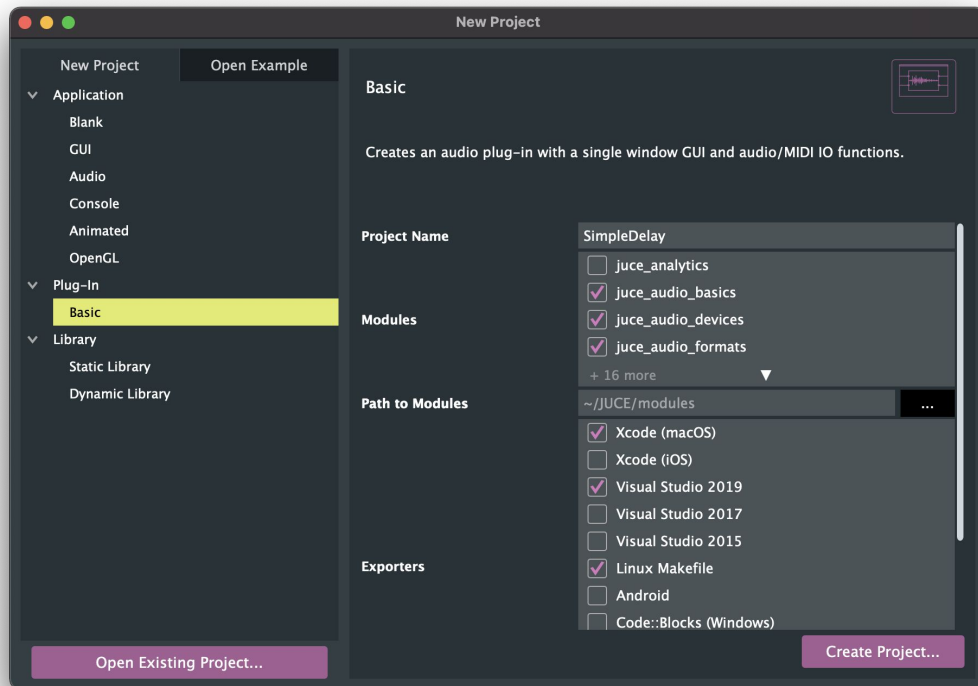


# Workshop materials

## Contents:

- The workspace directory is where we'll build our plug-in
- The Plugins directory is where you will find your compiled plug-ins
- Numbered directories containing source code snapshots corresponding to each section

# Creating a plug-in project



# #01: Creating a plug-in project

We're now using the workshop checkpoints

- The #01 in the title signals that we're using the contents of the 01 directory
- To return to this point in the workshop you can delete the contents of the workspace folder and copy the contents of the 01 folder into the workspace folder
- We've already done this for step 01, so we're all starting from the same place

# #01: Creating a plug-in project

Objectives of this section:

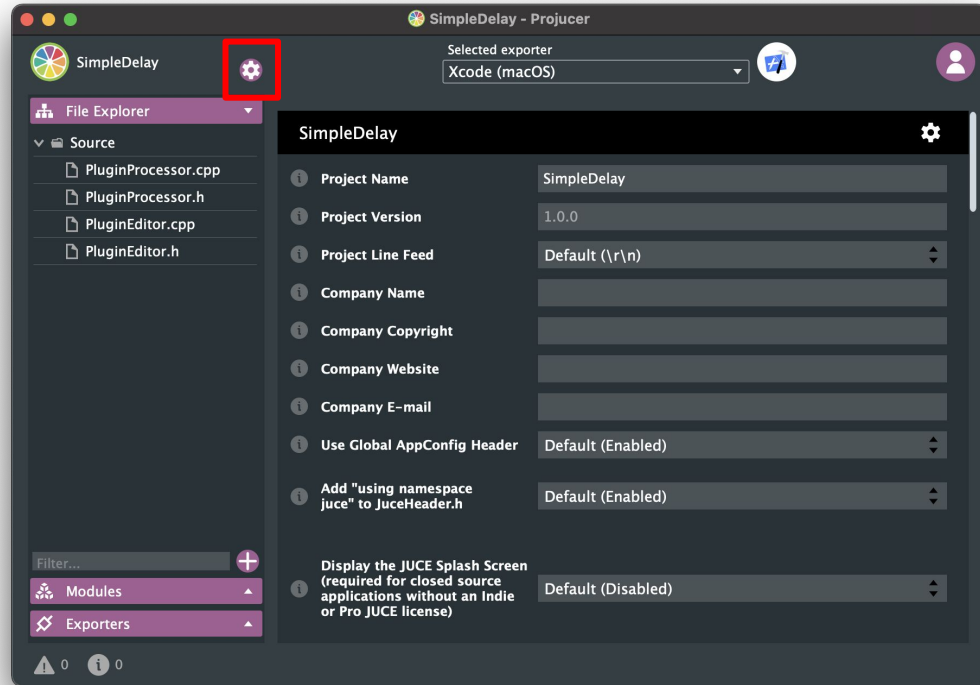
- Go through the common project configurations settings in the Projucer
- Export the project and open it in your IDE of choice
  - Xcode
  - Visual Studio
  - Whichever Linux editor you want to use
- Build the empty template project
- Load the plug-in in the TestHost

In the Projucer, open the workspace/`SimpleDelay.jucer` project.

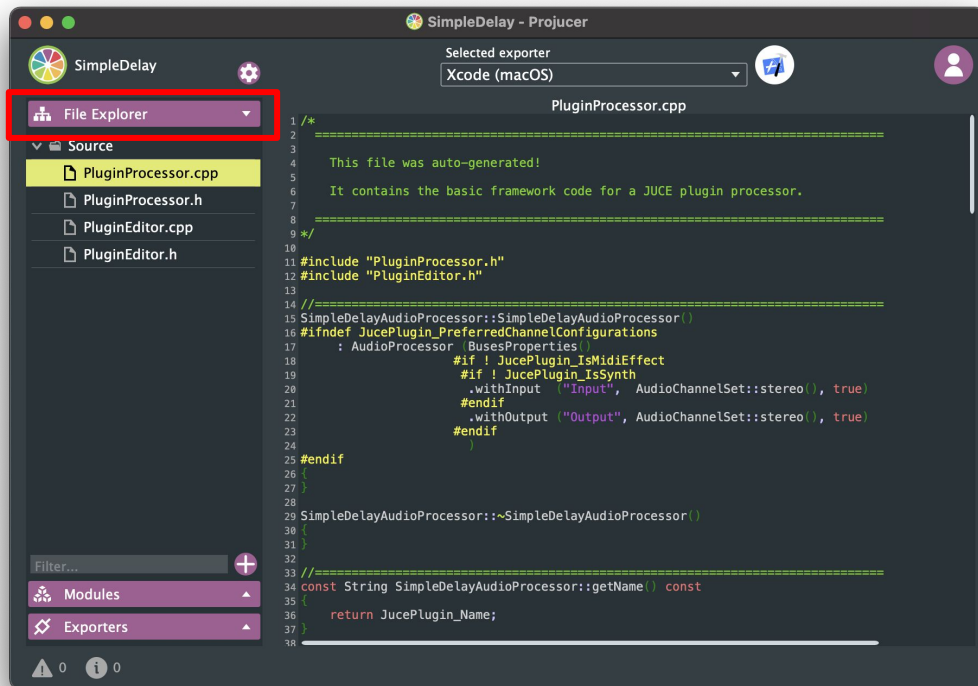
# #01: The project



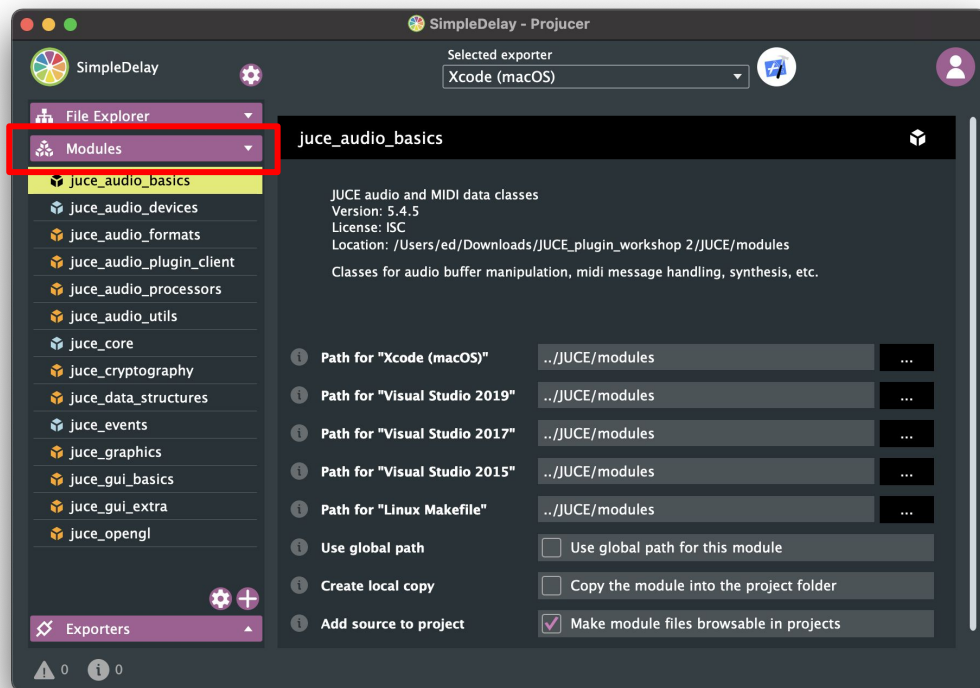
# #01: Project settings



# #01: File explorer

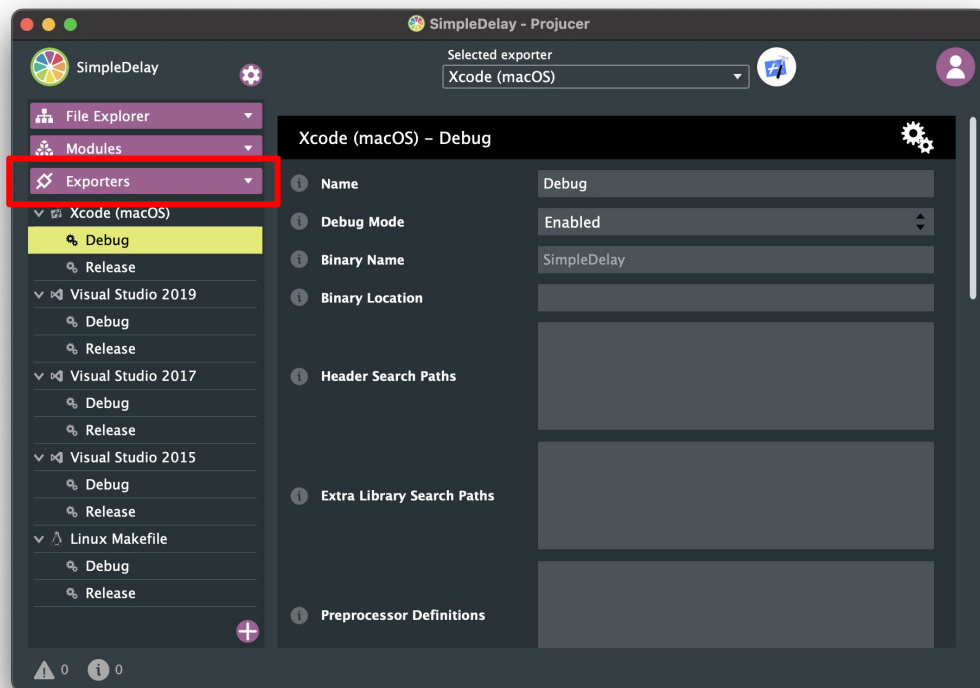


# #01: Module settings





# #01: Exporter Settings



# #01: Generate IDE project files

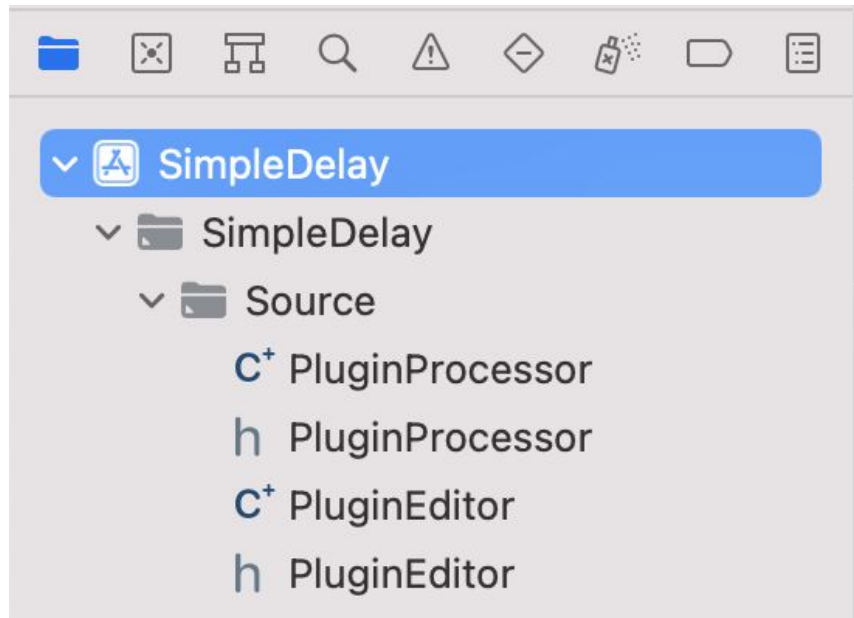


# #01: Generate IDE project files

We've seen a lot of Projucer settings but we don't need to change any for this workshop!

Creating a Linux Makefile is a little different - there are no IDEs associated with Makefiles so after saving your project you need to run make from the command line

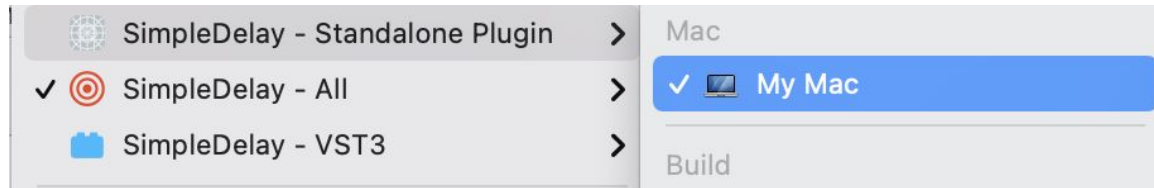
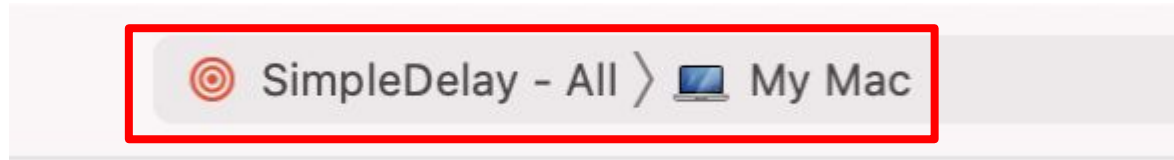
# #01: Project structure



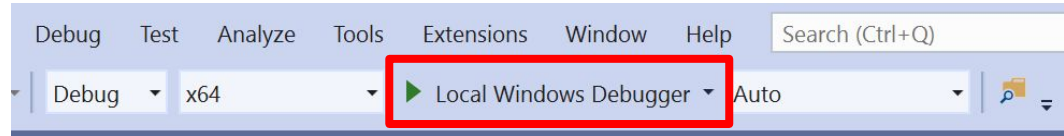
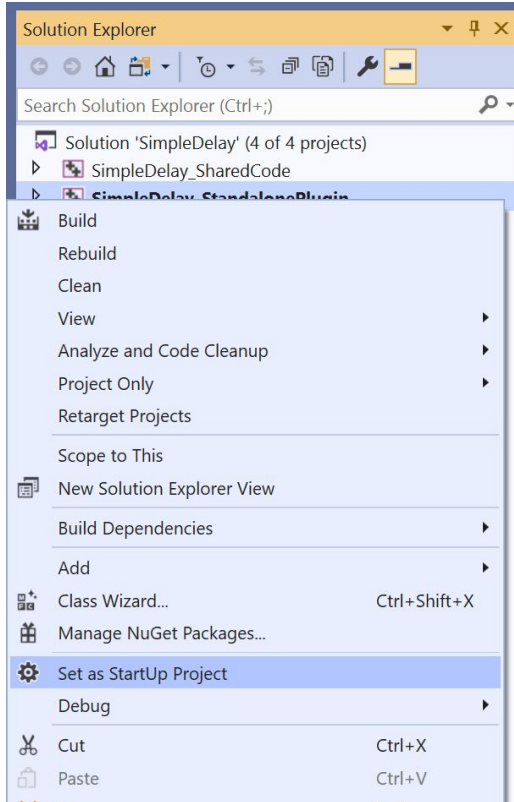
AudioProcessor

AudioProcessorEditor

# #01: Running Standalone (macOS)



# #01: Running Standalone (Windows)



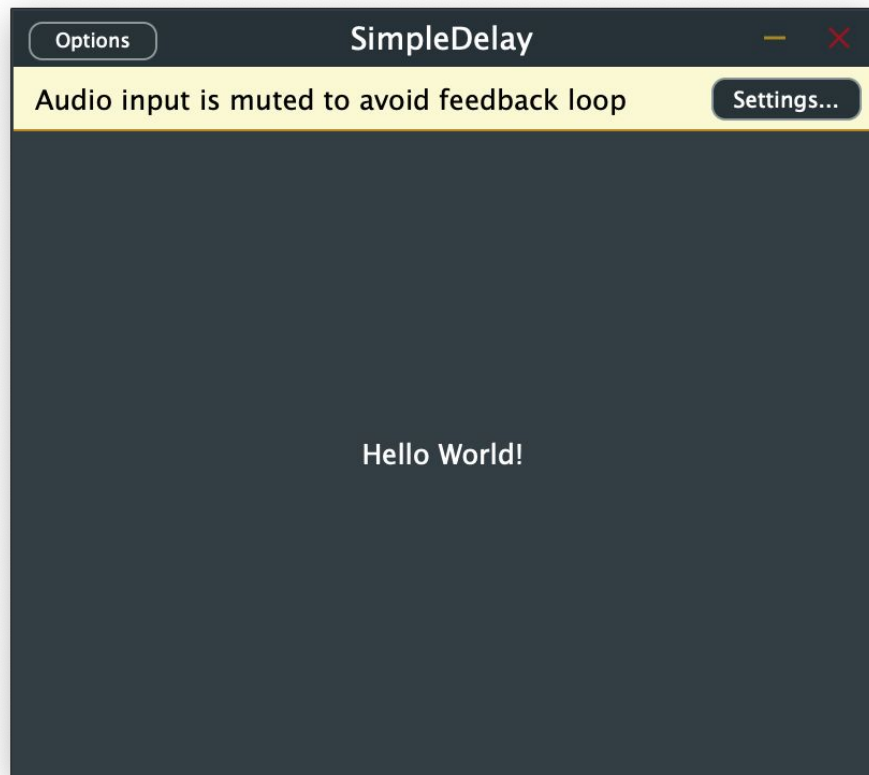
# #01: Running Standalone (Linux)

```
cd workspace/Buils/LinuxMakefile
```

```
make
```

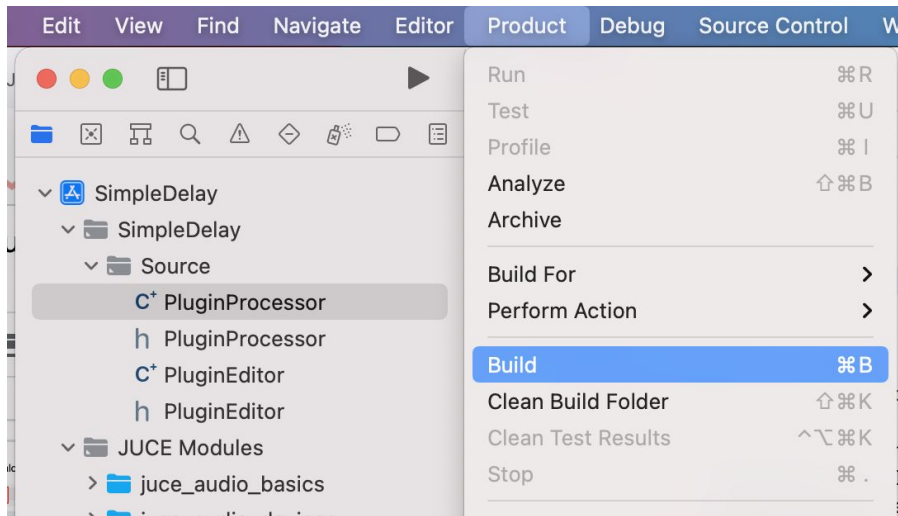
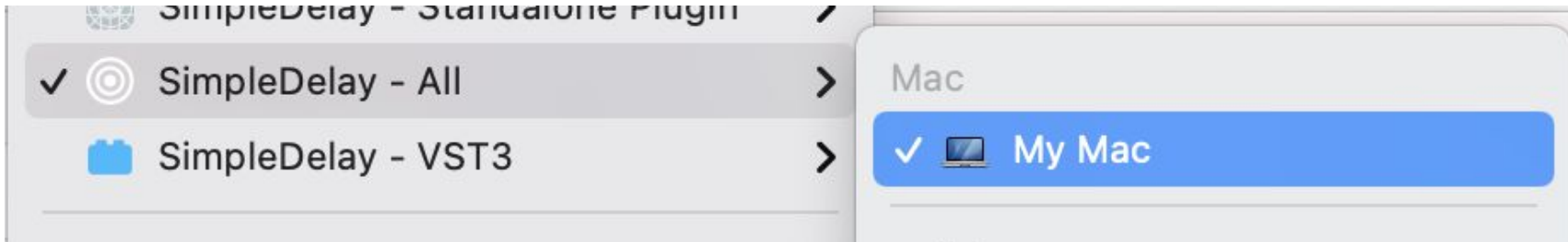
```
./build/SimpleDelay
```

# #01: Running Standalone

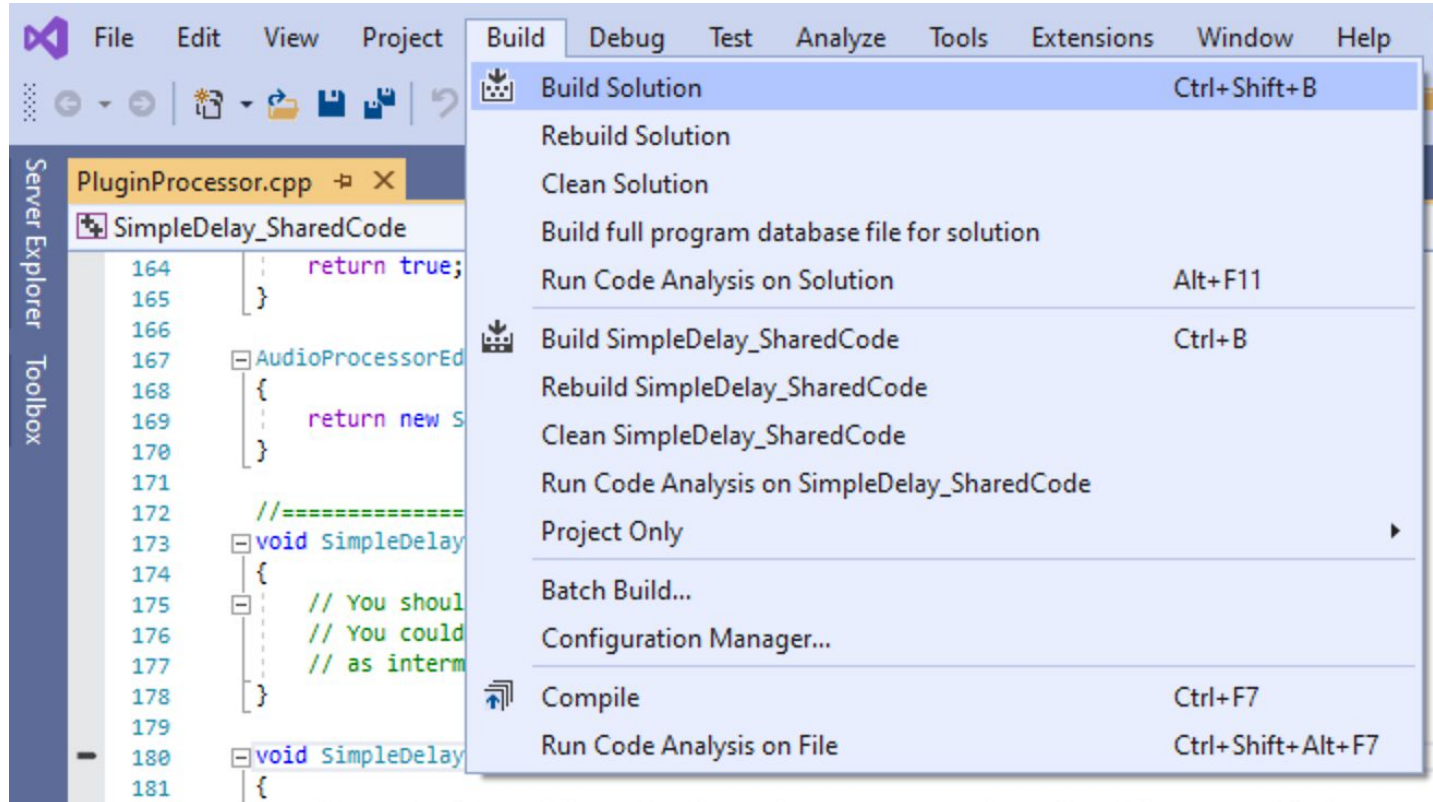




# #01: Building plug-ins (macOS)



# #01: Building plug-ins (Windows)



# #01: Building plug-ins (Linux)

```
cd workspace/Buils/LinuxMakefile
```

```
make
```

# #01: Running in the TestHost

A simple host application for testing our plug-in can be found in the workshop materials:

- TestHost/macOS/TestHost.app
- TestHost/Win64/TestHost.exe
- TestHost/Linux/TestHost

Launch the one appropriate for your platform.

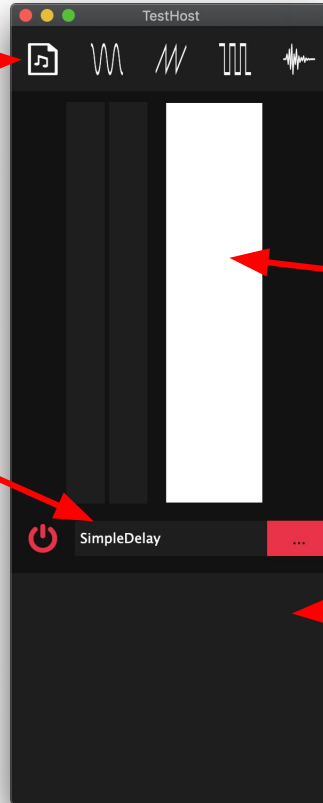
# #01: Running in the TestHost

Audio input sources

Plug-in section

Gain fader & level meters

Oscilloscope



# #01: Running in the TestHost

Find the SimpleDelay plug-in that you've built in the Plugins folder and drop it onto the plug-in section of the TestHost app.

Click on the plug-in name to open the UI.

# Break

## #02: Modifying audio

Objectives of this section:

- Edit source files in your IDE
- Add a very simple gain reduction to the plug-in
- Listen to the results

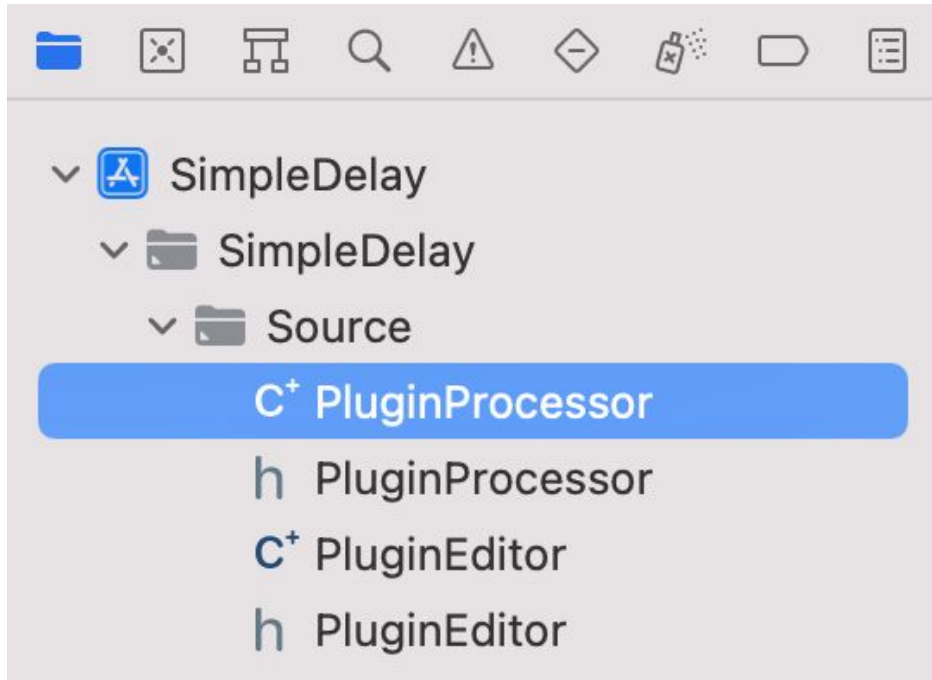
Copy the contents of the 02 directory into workspace.

Open your IDE project file

- `workspace/Builds/VisualStudio2019/SimpleDelay.sln`
- `workspace/Builds/macOS/SimpleDelay.xcodeproj`



## #02: Edit the AudioProcessor



## #02: The SimpleDelayAudioProcessor class

The SimpleDelayAudioProcessor class has a lot of methods

- All of these methods have been automatically written by the template created by the Projucer
- You can change the contents of these methods to change the behavior of the plug-in

The only one needed for this section is processBlock

## #02: processBlock

```

132 void SimpleDelayAudioProcessor::processBlock (juce::AudioBuffer<float>& buffer,
133                                             juce::MidiBuffer& midiMessages)
134 {
135     juce::ScopedNoDenormals noDenormals;
136     auto totalNumInputChannels = getTotalNumInputChannels();
137     auto totalNumOutputChannels = getTotalNumOutputChannels();
138
139     // In case we have more outputs than inputs, this code clears any output
140     // channels that didn't contain input data, (because these aren't
141     // guaranteed to be empty - they may contain garbage).
142     // This is here to avoid people getting screaming feedback
143     // when they first compile a plugin, but obviously you don't need to keep
144     // this code if your algorithm always overwrites all the output channels.
145     for (auto i = totalNumInputChannels; i < totalNumOutputChannels; ++i)
146         buffer.clear (i, 0, buffer.getNumSamples());
147
148     // This is the place where you'd normally do the guts of your plugin's
149     // audio processing...
150     // Make sure to reset the state if your inner loop is processing
151     // the samples and the outer loop is handling the channels.
152     // Alternatively, you can process the samples with the channels
153     // interleaved by keeping the same state.
154     for (int channel = 0; channel < totalNumInputChannels; ++channel)
155     {
156         auto* channelData = buffer.getWritePointer (channel);
157
158         // ..do something to the data...
159     }
160 }

```

## #02: What does processBlock do?

processBlock is called repeatedly by the plug-in host with a chunk of audio to process

For plug-ins the amount of audio to process usually corresponds to the “buffer size” setting of the host (1024, 512, ...)

Be careful here! processBlock will be called on the *audio* thread, which is not the same as that used for drawing a GUI or handling mouse events (more on this a little later)

## #02: `juce::AudioBuffer<float>`

`AudioBuffer` is a class containing non-interleaved channels of audio data represented as floats, and methods to access and modify them

*Non-interleaved*: separate channels of continuous audio data

*Floats*: each sample is represented by a floating point number in the range -1.0 to 1.0

`processBlock` is passed a reference to an `AudioBuffer` containing the incoming audio data. We create an audio effect by modifying this data.

## #02: Modifying audio

```

132 void SimpleDelayAudioProcessor::processBlock (juce::AudioBuffer<float>& buffer,
133                                             juce::MidiBuffer& midiMessages)
134 {
135     juce::ScopedNoDenormals noDenormals;
136     auto totalNumInputChannels = getTotalNumInputChannels();
137     auto totalNumOutputChannels = getTotalNumOutputChannels();
138
139     // In case we have more outputs than inputs, this code clears any output
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143     // when they first compile a plugin, but obviously you don't need to keep
144     // this code if your algorithm always overwrites all the output channels.
145     for (auto i = totalNumInputChannels; i < totalNumOutputChannels; ++i)
146         buffer.clear (i, 0, buffer.getNumSamples());
147
148     // This is the place where you'd normally do the guts of your plugin's
149     // audio processing...
150     // Make sure to reset the state if your inner loop is processing
151     // the samples and the outer loop is handling the channels.
152     // Alternatively, you can process the samples with the channels
153     // interleaved by keeping the same state.
154     for (int channel = 0; channel < totalNumInputChannels; ++channel)
155     {
156         auto* channelData = buffer.getWritePointer (channel);
157
158         // ..do something to the data...
159     }
160 }

```

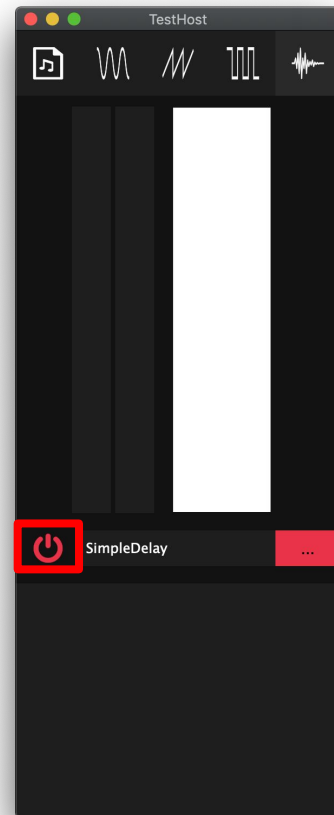
## #02: Fixed gain reduction

```
for (int channel = 0; channel < totalNumInputChannels; ++channel)
{
    auto* channelData = buffer.getWritePointer (channel);

    for (int i = 0; i < buffer.getNumSamples(); ++i)
        channelData[i] *= 0.2f;
}
```

## #02: Testing the fixed gain reduction

- Load the plug-in in the TestHost
- Toggle the bypass
- Hear the gain reduction when audio is going through the plug-in





## #03: Plug-in parameters

Objectives of this section:

Create a parameter to control the gain reduction

- Add a parameter to your plug-in interface
- Use the parameter to change how the audio is processed
- Change the parameter value in a (auto-generated) user interface
- Listen to how the audio is changed in real time

## #03: What is a parameter?

Parameters are how plug-in hosts control plug-ins

- They are exposed as part of the plug-in format's interface
  - Hosts can query plug-ins to find out what parameters they offer
- Parameters can be changed via the plug-in's GUI
- Parameters can be changed via automation
  - Though they can also be marked as non-automatable

## #03: Parameter types

All derived from the `AudioProcessorParameter` class

- Methods for getting and setting parameter values and properties
- Added to, and then managed by, your `AudioProcessor`
- JUCE provides some basic types to get your started
  - `AudioParameterFloat`
  - `AudioParameterBool`
  - `AudioParameterChoice`
  - `AudioParameterInt`

## #03: Adding a parameter

```

12 //=====
13 SimpleDelayAudioProcessor::SimpleDelayAudioProcessor()
14 #ifndef JucePlugin_PreferredChannelConfigurations
15     : AudioProcessor (BusesProperties()
16         #if ! JucePlugin_IsMidiEffect
17         #if ! JucePlugin_IsSynth
18             .withInput  ("Input",  juce::AudioChannelSet::stereo(), true)
19         #endif
20             .withOutput ("Output", juce::AudioChannelSet::stereo(), true)
21         #endif
22     )
23 #endif
24 {
25 }
26

```

## #03: Adding a parameter

```

12 //=====
13 SimpleDelayAudioProcessor::SimpleDelayAudioProcessor()
14 #ifndef JucePlugin_PreferredChannelConfigurations
15     : AudioProcessor (BusesProperties()
16         #if ! JucePlugin_IsMidiEffect
17         #if ! JucePlugin_IsSynth
18             .withInput  ("Input",  juce::AudioChannelSet::stereo(), true)
19         #endif
20             .withOutput ("Output", juce::AudioChannelSet::stereo(), true)
21         #endif
22     )
23 #endif
24 {
25     addParameter (new juce::AudioParameterFloat ("gain", "Gain", 0.0f, 1.0f, 1.0f));
26 }
27

```

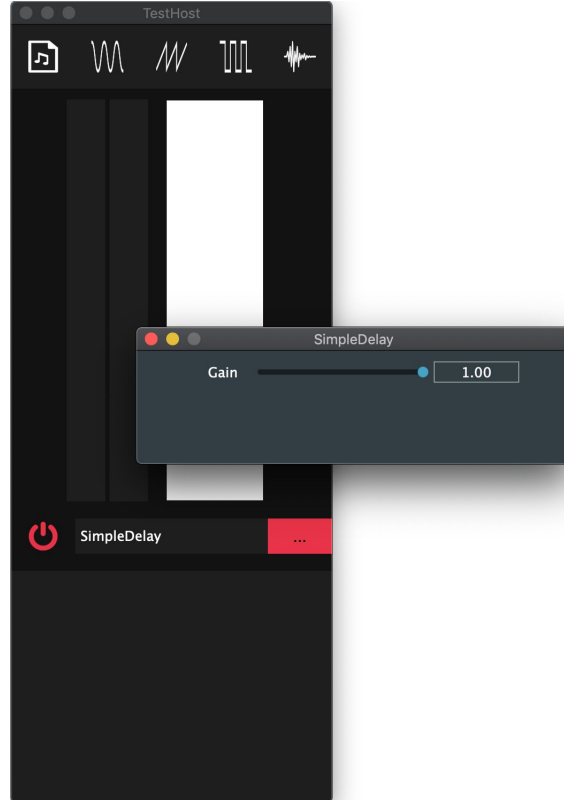
## #03: Using a parameter in processBlock

```
juce::AudioProcessorParameter* gainParameter = getParameters()[0];  
float gain = gainParameter->getValue();  
  
for (int channel = 0; channel < totalNumInputChannels; ++channel)  
{  
    float* channelData = buffer.getWritePointer (channel);  
  
    for (int i = 0; i < buffer.getNumSamples(); ++i)  
        channelData[i] *= gain;  
}
```

## #03: Use an automatically generated user interface

```
173 juce::AudioProcessorEditor* SimpleDelayAudioProcessor::createEditor()  
174 {  
175     //return new SimpleDelayAudioProcessorEditor (*this);  
176     return new juce::GenericAudioProcessorEditor (*this);  
177 }
```

## #03: Compile the plug-in and load it





## #04: Creating the delay effect

Objectives of this section:

Add some less trivial audio processing

- Add some more parameters
- Configure our audio processing algorithm in `prepareToPlay`
- Implement a basic delay effect
- Play with the effect in `TestHost`

This section requires quite a lot of typing to complete. Start from the contents of the 05 folder and review the changes rather than typing along.

Begin compiling the plug-in formats now!

## #04: What is a delay effect?

An echo of the incoming audio

The audio at time  $t$  is combined with a recursively attenuated audio at time  $(t - nD)$  where  $D$  is the delay time and  $n = 1, 2, 3, 4, \dots$

Increasing the delay time  $D$  increases the time between echos

Increasing the attenuation decreases the time taken for the echos to fade away

Our SimpleDelay plug-in will feature a fixed delay time, but a variable feedback (opposite of attenuation) with a dry/wet mix control

## #04: Add some more parameters

```

12 //=====
13 SimpleDelayAudioProcessor::SimpleDelayAudioProcessor()
14 #ifndef JUCE_PLUGIN_PREFERRED_CHANNEL_CONFIGURATIONS
15     : AudioProcessor (BusesProperties())
16         #if ! JUCE_PLUGIN_IS_MIDI_EFFECT
17         #if ! JUCE_PLUGIN_IS_SYNTH
18             .withInput ("Input", juce::AudioChannelSet::stereo(), true)
19         #endif
20             .withOutput ("Output", juce::AudioChannelSet::stereo(), true)
21         #endif
22     )
23 #endif
24 {
25     addParameter (new juce::AudioParameterFloat ("gain", "Gain", 0.0f, 1.0f, 1.0f));
26     addParameter (new juce::AudioParameterFloat ("feedback", "Delay Feedback", 0.0f, 1.0f, 0.35f));
27     addParameter (new juce::AudioParameterFloat ("mix", "Dry / Wet", 0.0f, 1.0f, 0.5f));
28 }
29

```

## #04: The delay algorithm

A simple circular buffer of audio history

- The size of the circular buffer will determine the (fixed) delay
- Each call to `processBlock` we will add incoming audio to the circular buffer
- The output audio will be a combination of the input and the contents of the circular buffer

## #04: Add a circular buffer to PluginProcessor.h

```
56 private:
57     //=====
58     int delayBufferPos = 0;
59     juce::AudioBuffer<float> delayBuffer;
60
61     JUCE_DECLARE_NON_COPYABLE_WITH_LEAK_DETECTOR (SimpleDelayAudioProcessor)
62 };
```

## #04: Configure the buffer in prepareToPlay

```
void SimpleDelayAudioProcessor::prepareToPlay (double sampleRate, int /*samplesPerBlock*/)
{
    // Use this method as the place to do any pre-playback
    // initialisation that you need..

    int delayMilliseconds = 200;
    auto delaySamples = (int) std::round (sampleRate * delayMilliseconds / 1000.0);
    delayBuffer.setSize (2, delaySamples);
    delayBuffer.clear();
    delayBufferPos = 0;
}
```

# #04: Implement delay algorithm in processBlock

```

auto& parameters = getParameters();
float gain      = parameters[0]->getValue();
float feedback  = parameters[1]->getValue();
float mix       = parameters[2]->getValue();

int delayBufferSize = delayBuffer.getNumSamples();

for (int channel = 0; channel < totalNumInputChannels; ++channel)
{
    float* channelData = buffer.getWritePointer (channel);
    int delayPos = delayBufferPos;

    for (int i = 0; i < buffer.getNumSamples(); ++i)
    {
        float drySample = channelData[i];

        float delaySample = delayBuffer.getSample (channel, delayPos) * feedback;
        delayBuffer.setSample (channel, delayPos, drySample + delaySample);

        delayPos++;
        if (delayPos == delayBufferSize)
            delayPos = 0;

        channelData[i] = (drySample * (1.0f - mix)) + (delaySample * mix);
        channelData[i] *= gain;
    }
}

delayBufferPos += buffer.getNumSamples();
if (delayBufferPos >= delayBufferSize)
    delayBufferPos -= delayBufferSize;

```

## #04: Implement delay algorithm in processBlock

```
auto& parameters = getParameters();  
float gain       = parameters[0]->getValue();  
float feedback   = parameters[1]->getValue();  
float mix        = parameters[2]->getValue();
```



# #04: Implement a delay algorithm in processBlock

```

for (int channel = 0; channel < totalNumInputChannels; ++channel)
{
    float* channelData = buffer.getWritePointer (channel);
    int delayPos = delayBufferPos;

    for (int i = 0; i < buffer.getNumSamples(); ++i)
    {
        float drySample = channelData[i];

        float delaySample = delayBuffer.getSample (channel, delayPos) * feedback;
        delayBuffer.setSample (channel, delayPos, drySample + delaySample);

        delayPos++;
        if (delayPos == delayBufferSize)
            delayPos = 0;

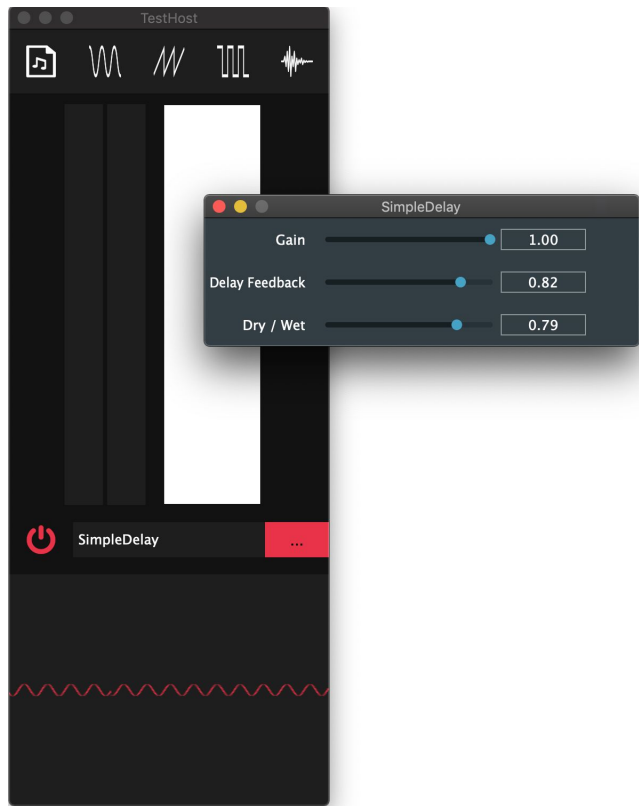
        channelData[i] = (drySample * (1.0f - mix)) + (delaySample * mix);
        channelData[i] *= gain;
    }
}

```

## #04: Implement delay algorithm in processBlock

```
delayBufferPos += buffer.getNumSamples();  
if (delayBufferPos >= delayBufferSize)  
    delayBufferPos -= delayBufferSize;
```

## #04: Have a play with the plug-in



## #05: Parameter management and state

Objectives of this section:

Use an `AudioProcessorValueTreeState` to manage your parameters

- See improved parameter handling
- Serialise and deserialise your plug-ins state

This section requires quite a lot of typing to complete. Start from the contents of the 06 folder and review the changes rather than typing along.

## #05: Parameter management

```
auto& parameters = getParameters();  
float gain      = parameters[0]->getValue();  
float feedback  = parameters[1]->getValue();  
float mix       = parameters[2]->getValue();
```

## #05: Adding an AudioProcessorValueTreeState

```

56 private:
57     //=====
58     juce::AudioProcessorValueTreeState state;
59     int delayBufferPos = 0;
60     juce::AudioBuffer<float> delayBuffer;
61
62     JUCE_DECLARE_NON_COPYABLE_WITH_LEAK_DETECTOR (SimpleDelayAudioProcessor)
63 };
64

```

## #05: Adding an AudioProcessorValueTreeState

```

12 //=====
13 SimpleDelayAudioProcessor::SimpleDelayAudioProcessor()
14 #ifndef JUCE_PLUGIN_PREFERRED_CHANNEL_CONFIGURATIONS
15     : AudioProcessor (BusesProperties()
16         .withInput ("Input", juce::AudioChannelSet::stereo(), true)
17         .withOutput ("Output", juce::AudioChannelSet::stereo(), true)
18     )
19 #endif
20
21 #endif
22
23 , state (*this, nullptr, "STATE", {
24     std::make_unique<juce::AudioParameterFloat> ("gain", "Gain", 0.0f, 1.0f, 1.0f),
25     std::make_unique<juce::AudioParameterFloat> ("feedback", "Delay Feedback", 0.0f, 1.0f, 0.35f),
26     std::make_unique<juce::AudioParameterFloat> ("mix", "Dry / Wet", 0.0f, 1.0f, 0.5f)
27 })
28 {
29 }
30 }

```

## #05: Parameter management

```
float gain      = state.getParameter ("gain")->getValue();  
float feedback = state.getParameter ("feedback")->getValue();  
float mix       = state.getParameter ("mix")->getValue();
```



## #05: Saving and restoring plug-in state

When plug-in hosts save and load projects, each plug-in must save and restore its state

- `getStateInformation (juce::MemoryBlock& destData)`
- `setStateInformation (const void* data, int sizeInBytes)`

The plug-in's state must be serialised to, and deserialised from, a block of memory managed by the host.

## #05: Saving plug-in state

```

209 //=====
210 void SimpleDelayAudioProcessor::getStateInformation (juce::MemoryBlock& destData)
211 {
212     // You should use this method to store your parameters in the memory block.
213     // You could do that either as raw data, or use the XML or ValueTree classes
214     // as intermediaries to make it easy to save and load complex data.
215
216     if (auto xmlState = state.copyState().createXml())
217         copyXmlToBinary (*xmlState, destData);
218 }

```

## #05: Restoring plug-in state

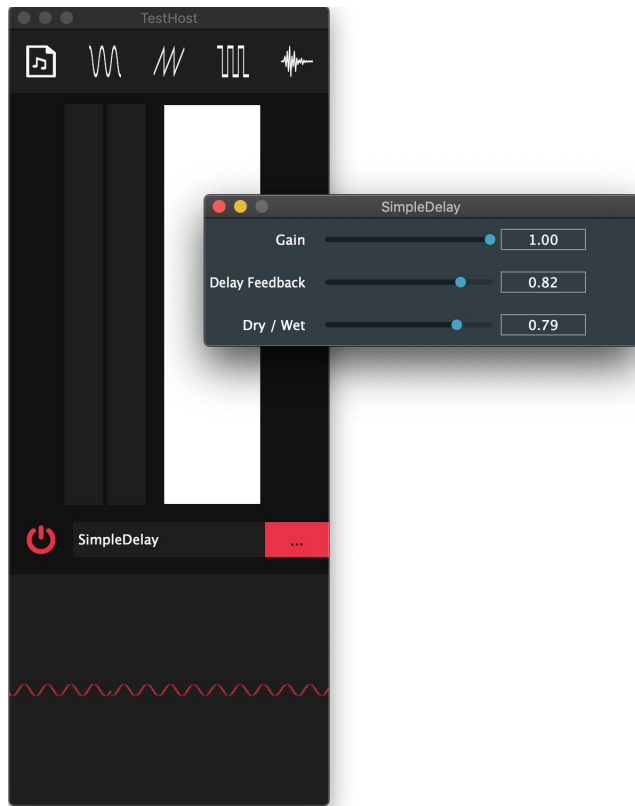
```
220 void SimpleDelayAudioProcessor::setStateInformation (const void* data, int sizeInBytes)
221 {
222     // You should use this method to restore your parameters from this memory block,
223     // whose contents will have been created by the getStateInformation() call.
224
225     if (auto xmlState = getXmlFromBinary (data, sizeInBytes))
226         state.replaceState (juce::ValueTree::fromXml (*xmlState));
227 }
228
```

## #05: See it working!

- Load the plug-in in the TestHost
- Tweak some parameters
- Unload the plug-in
- Re-load and see the restored state!

Host is calling `getStateInformation` to retrieve and store the state and passes it to the new plug-in instance via `setStateInformation`

## #05: Have a play with the plug-in



Break

## #06: Adding a GUI

Objectives of this section:

Display a custom GUI and draw some graphics

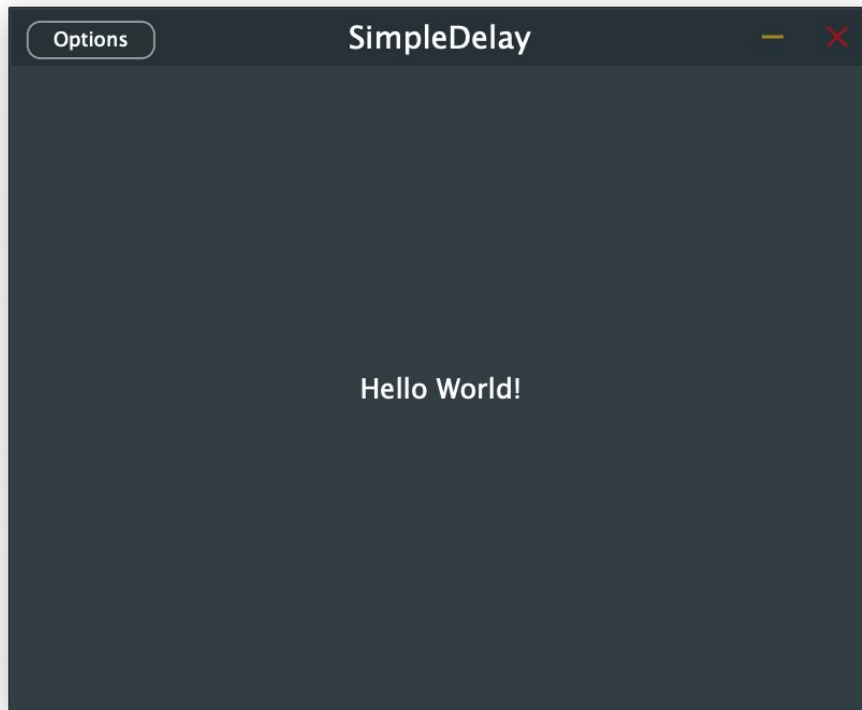
- Basic shapes
- Text

## #06: Remove the generic GUI

```
203  juce::AudioProcessorEditor* SimpleDelayAudioProcessor::createEditor()  
204  {  
205      return new SimpleDelayAudioProcessorEditor (*this);  
206      //return new juce::GenericAudioProcessorEditor (*this);  
207  }
```



## #06: Our custom GUI



## #06: Drawing in paint

```

25 //=====
26 void SimpleDelayAudioProcessorEditor::paint (juce::Graphics& g)
27 {
28     // (Our component is opaque, so we must completely fill the background with a solid colour)
29     g.fillAll (getLookAndFeel().findColour (juce::ResizableWindow::backgroundColourId));
30
31     g.setColour (juce::Colours::white);
32     g.setFont (15.0f);
33     g.drawFittedText ("Hello World!", getLocalBounds(), juce::Justification::centred, 1);
34 }
35

```

## #06: Threads

Be careful here!

- The GUI is rendered on the “main” (GUI, message) thread
- `processBlock` is called on the audio thread

It's very easy to create race conditions, where one thread is modifying a bit of memory whilst another thread is reading it.

This is easily the most complicated aspect of working with real-time audio.

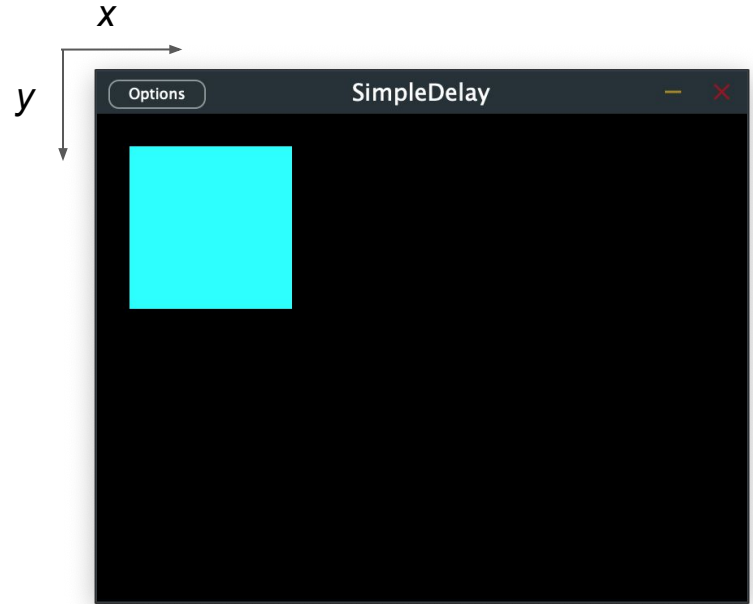
Using JUCE's `AudioParameter` classes and an `AudioProcessorValueTreeState` makes things much simpler.

## #06: Draw a square

```

25 //=====
26 void SimpleDelayAudioProcessorEditor::paint (juce::Graphics& g)
27 {
28     g.fillAll (juce::Colours::black);
29
30     g.setColour (juce::Colours::cyan);
31
32     //          x    y    width  height
33     g.fillRect (20, 20, 100,  100);
34 }

```



## #06: Lots of GUI code incoming

From now until the next break there will be a lot of code to add as we put together a GUI that's more than a few basic shapes

Don't worry about keeping up!

During and after the break there will be time to experiment with your own GUIs

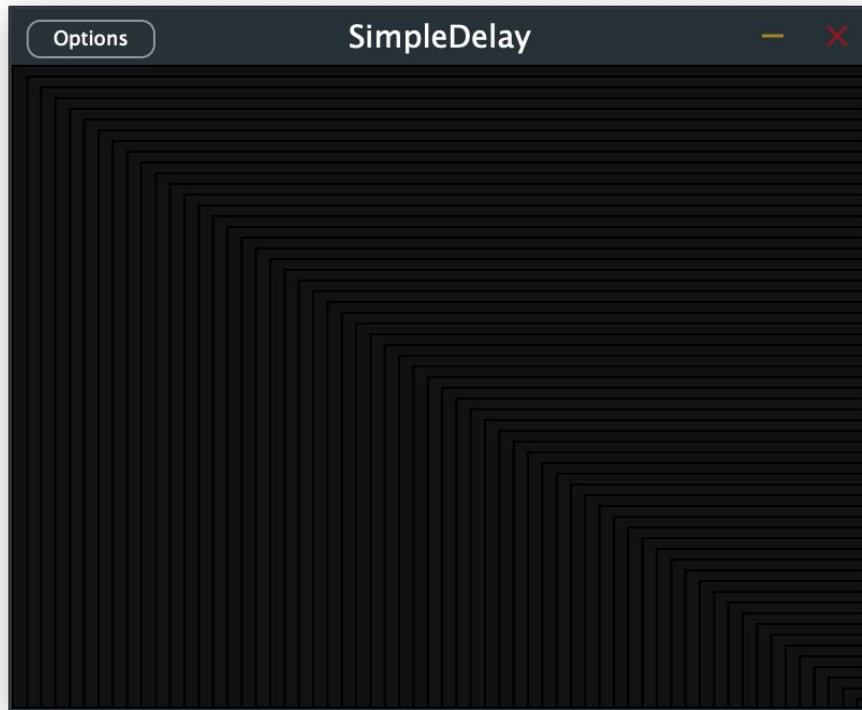
## #06: Something more advanced

```
g.fillAll (juce::Colour (0xff121212));

g.setColour (juce::Colours::black);
juce::Rectangle<float> backgroundRect = getLocalBounds().toFloat();
int numBackgroundRects = 60;
juce::Point<float> offset = backgroundRect.getBottomRight() / numBackgroundRects;

for (int i = 0; i < numBackgroundRects; ++i)
{
    g.drawRect (backgroundRect);
    backgroundRect += offset;
}
```

## #06: Something more advanced



## #06: Something more advanced

```
juce::Rectangle<int> bounds = getLocalBounds();
juce::Rectangle<int> textArea = bounds.removeFromLeft ((bounds.getWidth() * 2) / 3)
                                   .removeFromBottom (bounds.getHeight() / 2)
                                   .reduced (10);

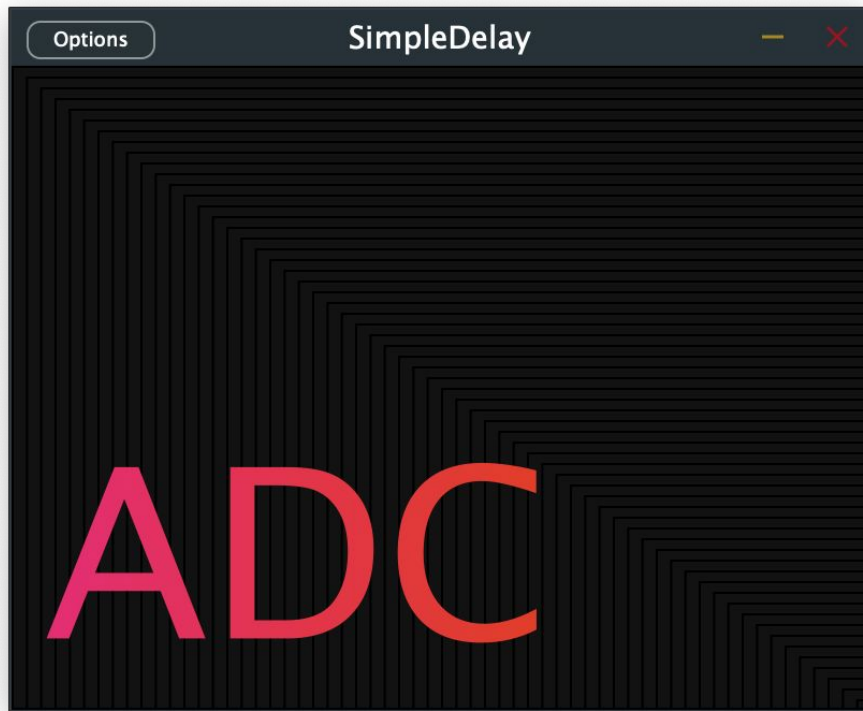
juce::ColourGradient gradient (juce::Colour (0xffe62875),
                               textArea.toFloat().getTopLeft(),
                               juce::Colour (0xffe43d1b),
                               textArea.toFloat().getTopRight(),
                               false);

g.setGradientFill (gradient);

g.setFont (textArea.toFloat().getHeight());
g.drawFittedText ("ADC", textArea, juce::Justification::centred, 1);
```



## #06: Something more advanced



## #07: Components

Objectives of this section:

Create some interactive GUI elements

- Add some sliders
- Handle layout changes in `resized`

## #07: JUCE Components

JUCE GUIs are trees of components

- You can create your own; the `AudioProcessorEditor` is a Component
- Parent components are responsible for laying out child components
- Mouse events and keyboard focus can be passed between them
- JUCE has a selection of common widget components you can use

## #07: Add some sliders to PluginEditor.h

```
27 private:
28     // This reference is provided as a quick way for your editor to
29     // access the processor object that created it.
30     SimpleDelayAudioProcessor& audioProcessor;
31
32     juce::Slider gainSlider, feedbackSlider, mixSlider;
33
34     JUCE_DECLARE_NON_COPYABLE_WITH_LEAK_DETECTOR (SimpleDelayAudioProcessorEditor)
35 };
36
```

# #07: Configure the sliders in PluginEditor.cpp

```

12 //=====
13 SimpleDelayAudioProcessorEditor::SimpleDelayAudioProcessorEditor (SimpleDelayAudioProcessor& p)
14     : AudioProcessorEditor (&p), audioProcessor (p)
15 {
16     gainSlider    .setSliderStyle (juce::Slider::SliderStyle::LinearVertical);
17     feedbackSlider.setSliderStyle (juce::Slider::SliderStyle::Rotary);
18     mixSlider     .setSliderStyle (juce::Slider::SliderStyle::Rotary);
19
20     for (auto* slider : { &gainSlider, &feedbackSlider, &mixSlider })
21     {
22         slider->setTextBoxStyle (juce::Slider::TextBoxBelow, true, 200, 30);
23         addAndMakeVisible (slider);
24     }
25
26     // Make sure that before the constructor has finished, you've set the
27     // editor's size to whatever you need it to be.
28     setSize (400, 300);
29 }
  
```

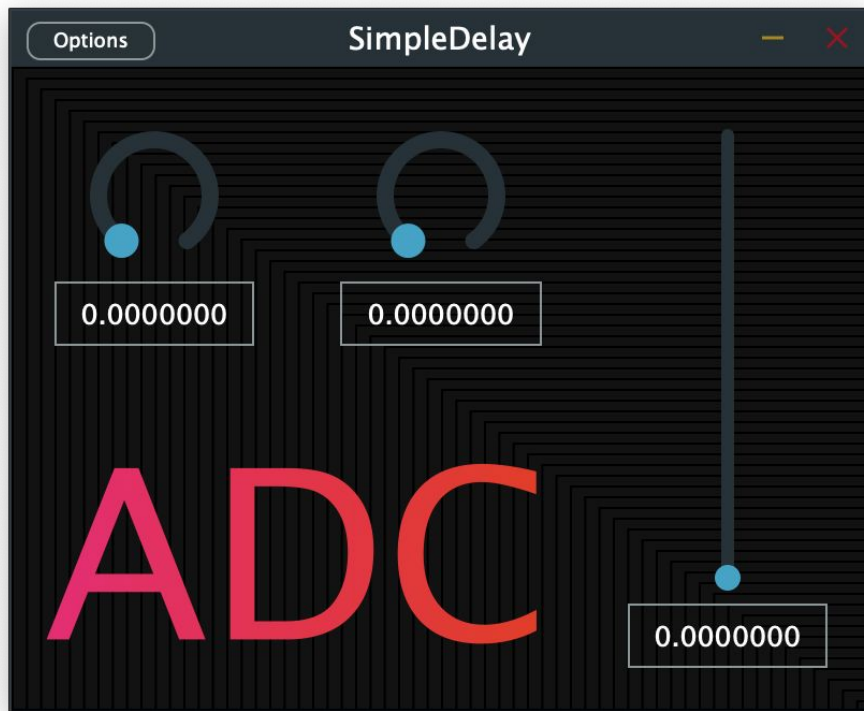
## #07: Layout the sliders in resized

```

67 void SimpleDelayAudioProcessorEditor::resized()
68 {
69     // This is generally where you'll want to lay out the positions of any
70     // subcomponents in your editor..
71
72     juce::Rectangle<int> bounds = getLocalBounds();
73     int margin = 20;
74
75     juce::Rectangle<int> gainBounds = bounds.removeFromRight (getWidth() / 3);
76     gainSlider.setBounds (gainBounds.reduced (margin));
77
78     juce::Rectangle<int> knobsBounds = bounds.removeFromTop (getHeight() / 2);
79     juce::Rectangle<int> feedbackBounds = knobsBounds.removeFromLeft (knobsBounds.getWidth() / 2);
80     feedbackSlider.setBounds (feedbackBounds.reduced (margin));
81     mixSlider.setBounds (knobsBounds.reduced (margin));
82 }
83

```

## #07: GUI with Sliders



# #08: Connecting GUI controls to plug-in parameters

Objectives of this section:

Control the plug-in from the GUI

- Use the `AudioProcessorValueTreeState` attachment classes to link `Sliders` to plug-in parameters



## #08: Add some attachments to PluginEditor.h

```
27 private:
28     // This reference is provided as a quick way for your editor to
29     // access the processor object that created it.
30     SimpleDelayAudioProcessor& audioProcessor;
31
32     juce::Slider gainSlider, feedbackSlider, mixSlider;
33     juce::AudioProcessorValueTreeState::SliderAttachment gainAttachment,
34                                                         feedbackAttachment,
35                                                         mixAttachment;
36
37     JUCE_DECLARE_NON_COPYABLE_WITH_LEAK_DETECTOR (SimpleDelayAudioProcessorEditor)
38 };
39
```

## #08: We need to pass the state to our editor

```

56     // Encapsulate this better in production code!
57     juce::AudioProcessorValueTreeState state;
58
59     private:
60         //=====
61         int delayBufferPos = 0;
62         juce::AudioBuffer<float> delayBuffer;
63
64         JUCE_DECLARE_NON_COPYABLE_WITH_LEAK_DETECTOR (SimpleDelayAudioProcessor)
65     };
66 
```

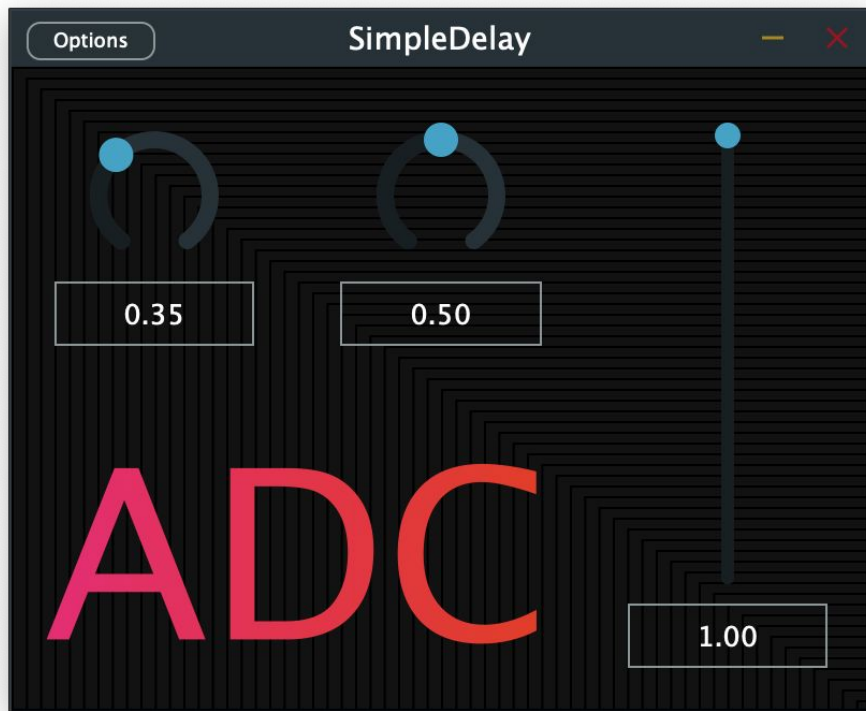
## #08: Linking Slider to plug-in parameters

```

12 //=====
13 SimpleDelayAudioProcessorEditor::SimpleDelayAudioProcessorEditor (SimpleDelayAudioProcessor& p)
14     : juce::AudioProcessorEditor (&p), audioProcessor (p),
15       gainAttachment      (p.state, "gain",      gainSlider),
16       feedbackAttachment  (p.state, "feedback", feedbackSlider),
17       mixAttachment       (p.state, "mix",       mixSlider)
18 {

```

## #09: The interactive plug-in



# Break

# TESTING IN HOSTS

# Debugging

What is a debugger?

- The most useful tool in a programmer's arsenal!
- GDB, LLDB, Microsoft Visual Studio Debugger
- CLI/GUI
- Examine program state, pause when conditions are met

What is a breakpoint?

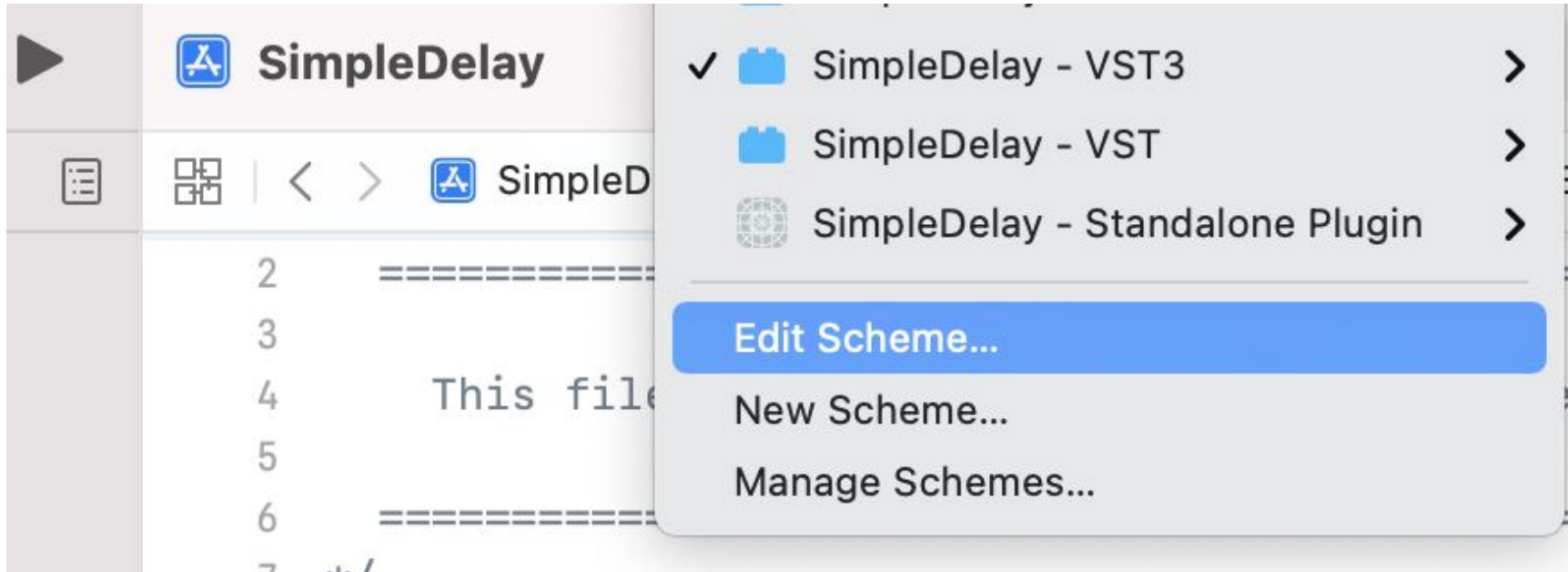
- Can be set via the CLI or GUI
- Program execution pauses when hit

# Debugging

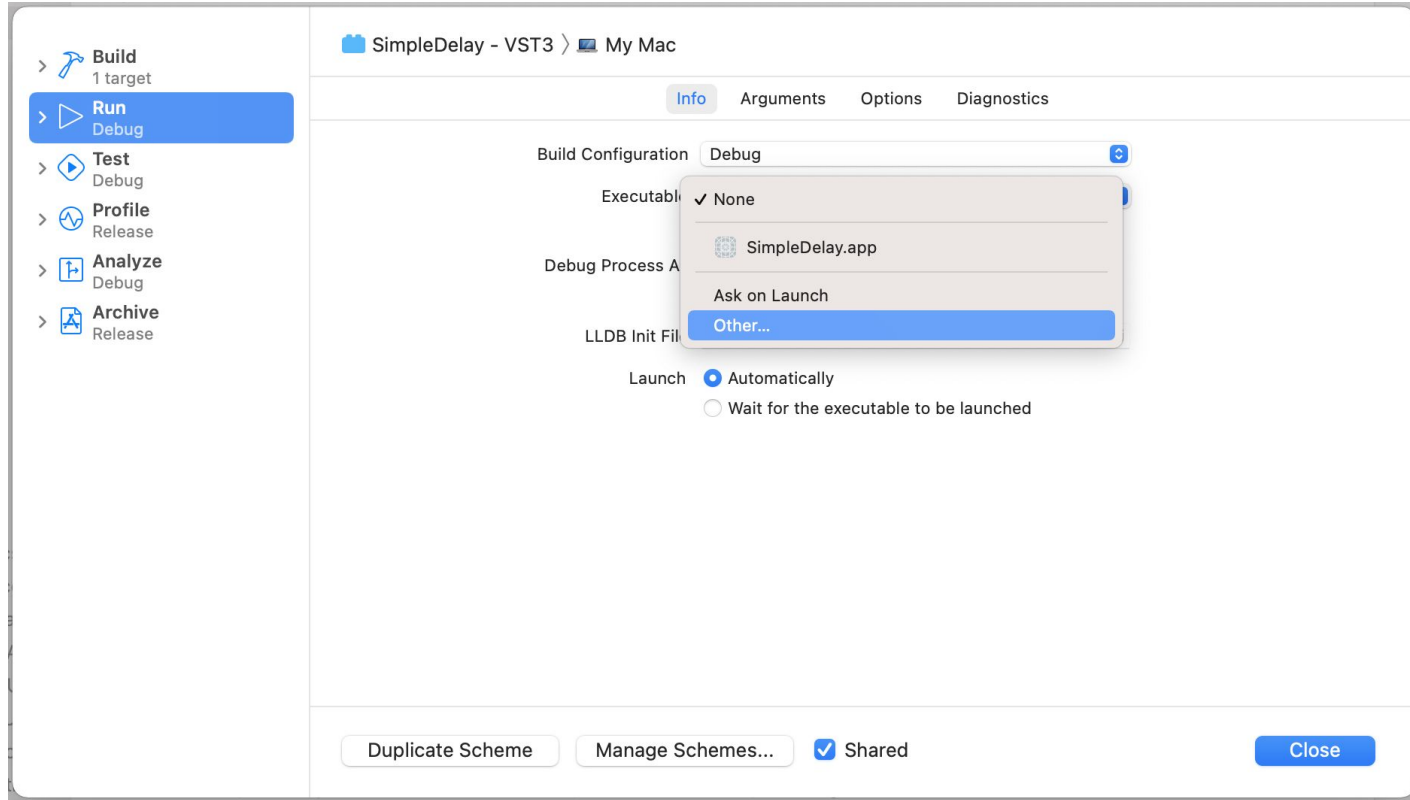
- Debugging standalone plug-in is simple as we control the whole process
- Debugging the plug-in inside an actual host is slightly more complicated as we are running inside a different process (the host)
- Debuggers can *attach* to a separate process to allow you to debug and set breakpoints in your code



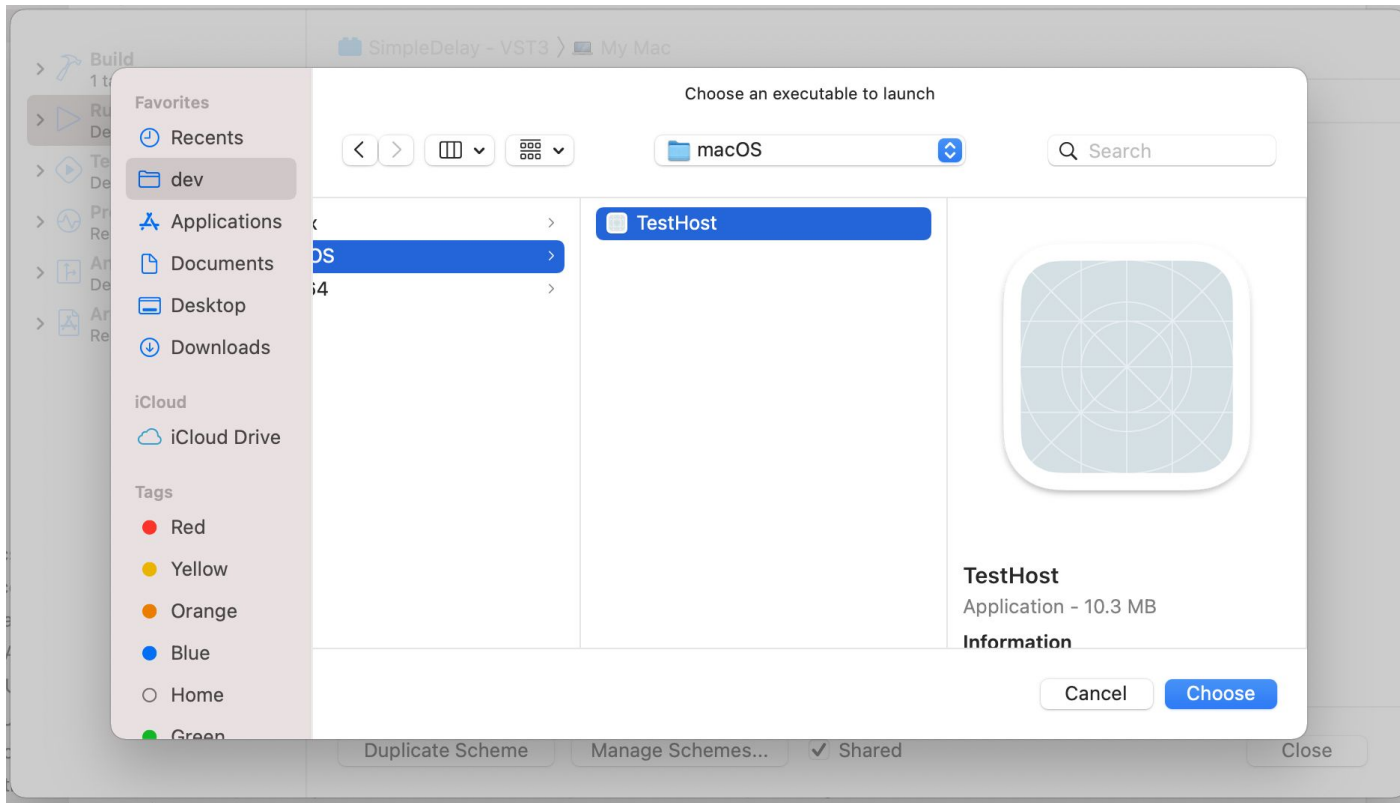
# macOS



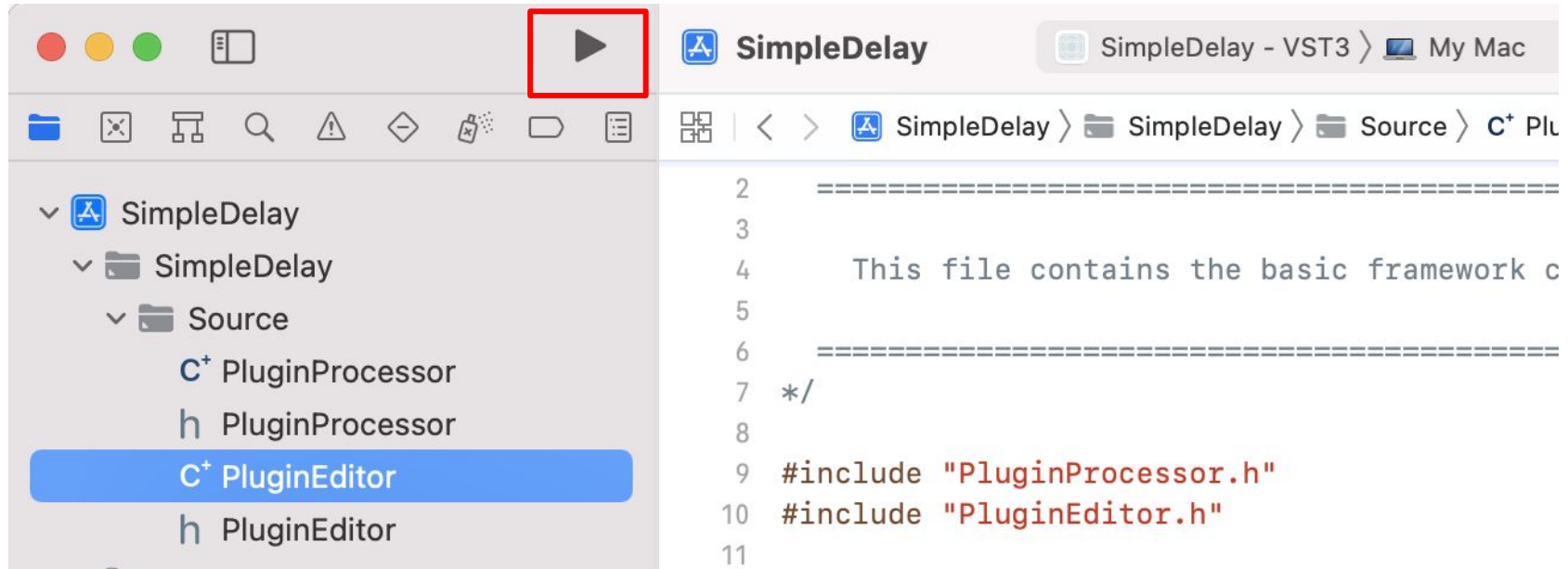
# macOS



# macOS



# macOS



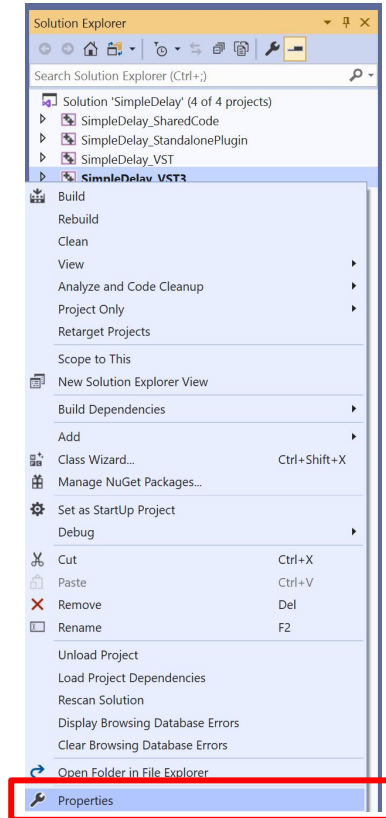
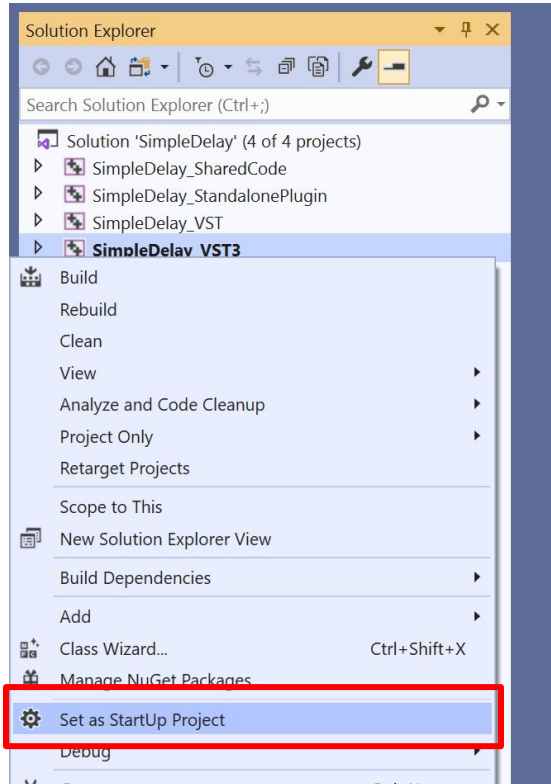
# macOS

```

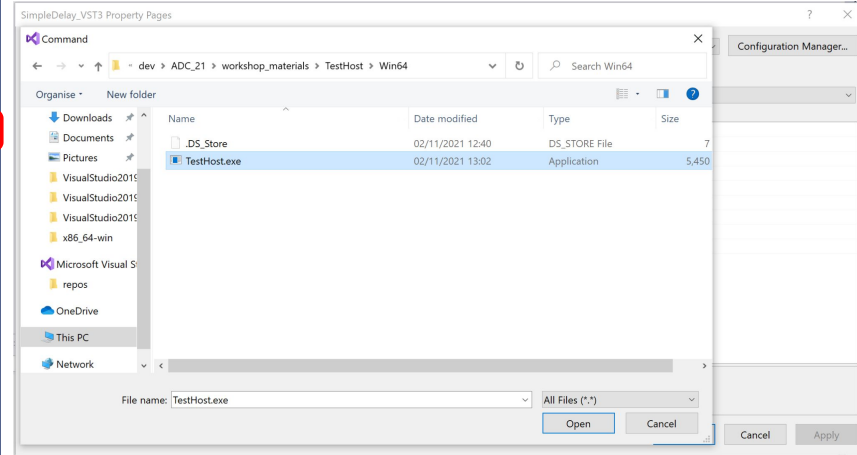
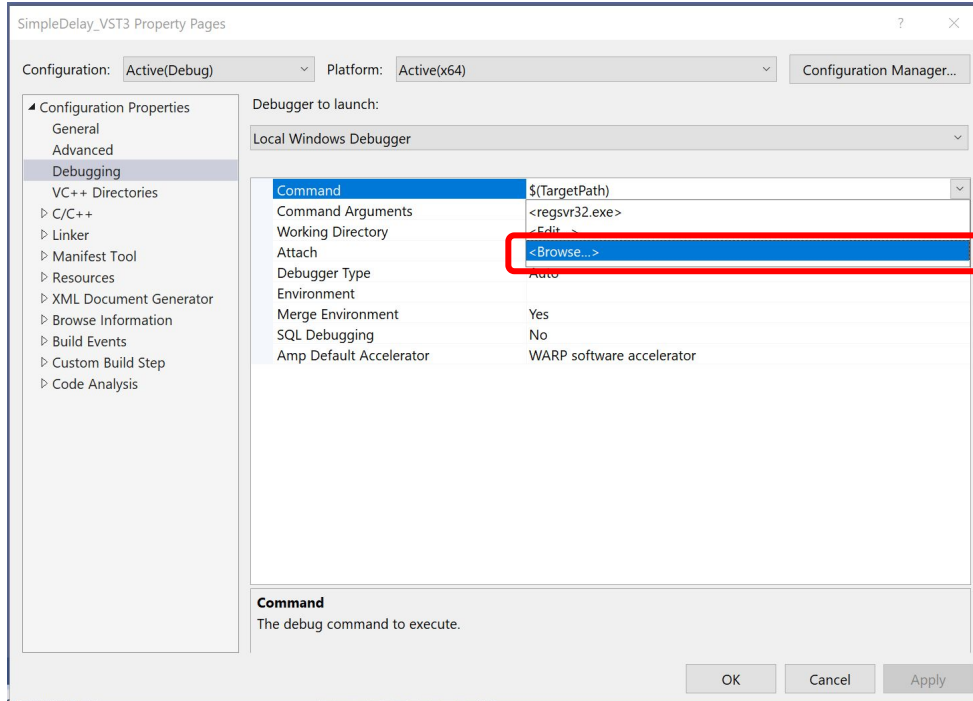
142
143 void SimpleDelayAudioProcessor::processBlock (juce::AudioBuffer<float>& buffer,
144                                             juce::MidiBuffer& midiMessages)
145 {
146     juce::ScopedNoDenormals noDenormals;
147     auto totalNumInputChannels = getTotalNumInputChannels();
148     auto totalNumOutputChannels = getTotalNumOutputChannels();
149
150     // In case we have more outputs than inputs, this code clears any output
151     // channels that didn't contain input data, (because these aren't
152     // guaranteed to be empty - they may contain garbage).
153     // This is here to avoid people getting screaming feedback
154     // when they first compile a plugin, but obviously you don't need to keep
155     // this code if your algorithm always overwrites all the output channels.
156     for (auto i = totalNumInputChannels; i < totalNumOutputChannels; ++i)
157         buffer.clear (i, 0, buffer.getNumSamples());
158

```

# Windows



# Windows



# Windows

SimpleDelay\_VST3 Property Pages

Configuration: Active(Debug) Platform: Active(x64) Configuration Manager...

- Configuration Properties
  - General
  - Advanced
  - Debugging
  - VC++ Directories
  - C/C++
  - Linker
  - Manifest Tool
  - Resources
  - XML Document Generator
  - Browse Information
  - Build Events
  - Custom Build Step
  - Code Analysis

Debugger to launch: Local Windows Debugger

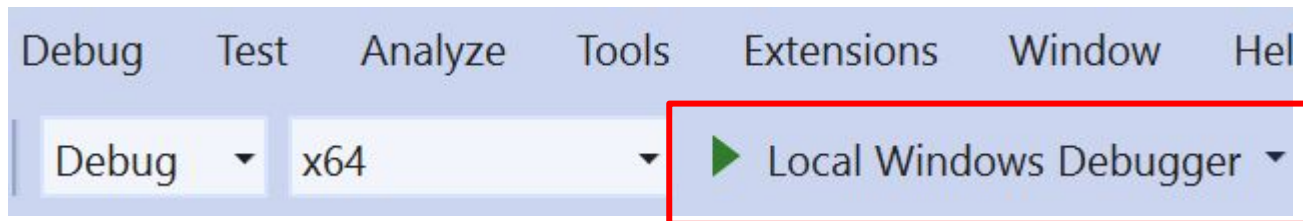
Command	Z:\dev\ADC_21\workshop_materials\TestHost\Win64\TestHost.exe
Command Arguments	
Working Directory	\$(ProjectDir)
Attach	No
Debugger Type	Auto
Environment	
Merge Environment	Yes
SQL Debugging	No
Amp Default Accelerator	WARP software accelerator

**Command**  
The debug command to execute.

OK Cancel Apply



# Windows



```

132 void SimpleDelayAudioProcessor::processBlock(juce::AudioBuffer<float>& buffer,
133 .....juce::MidiBuffer& midiMessages)
134 {
135     juce::ScopedNoDenormals noDenormals;
136     auto totalNumInputChannels = getTotalNumInputChannels();
137     auto totalNumOutputChannels = getTotalNumOutputChannels();
138
139     .....// In case we have more outputs than inputs, this code clears any output
140     .....// channels that didn't contain input data, (because these aren't
141     .....// guaranteed to be empty - they may contain garbage).
142     .....// This is here to avoid people getting screaming feedback
143     .....// when they first compile a plugin, but obviously you don't need to keep
144     .....// this code if your algorithm always overwrites all the output channels.
145     for (auto i = totalNumInputChannels; i < totalNumOutputChannels; ++i)
146     .....buffer.clear(i, 0, buffer.getNumSamples());
147

```

# Linux

```
cd workspace/Buils/LinuxMakefile
```

```
make CONFIG=Debug
```

```
gdb ./TestHost/Linux/TestHost
```

```
break SimpleDelayAudioProcessor::processBlock
```

```
run
```

# Linux

```
ed@ubuntu: /mnt/hgfs/dev/ADC_21/workshop_materials/09/Buils/LinuxMakefile
For help, type "help".
Type "apropos word" to search for commands related to "word"...
Reading symbols from ./build/SimpleDelay...
(gdb) b SimpleDelayAudioProcessor::processBlock(juce::AudioBuffer<float>&, juce::MidiBuffer&)
Breakpoint 1 at 0xa6a41: file ../../Source/PluginProcessor.cpp, line 144.
(gdb) r
Starting program: /mnt/hgfs/dev/ADC_21/workshop_materials/09/Buils/LinuxMakefile/build/SimpleDelay
[Thread debugging using libthread_db enabled]
Using host libthread_db library "/lib/x86_64-linux-gnu/libthread_db.so.1".
JUCE v6.1.2
[New Thread 0x7ffff227b640 (LWP 45117)]
[Thread 0x7ffff227b640 (LWP 45117) exited]
[New Thread 0x7ffff227b640 (LWP 45118)]
[Thread 0x7ffff227b640 (LWP 45118) exited]
[New Thread 0x7ffff227b640 (LWP 45119)]
[New Thread 0x7ffff1a7a640 (LWP 45120)]
[New Thread 0x7ffff1279640 (LWP 45121)]
[New Thread 0x7ffff0a78640 (LWP 45122)]
[Switching to Thread 0x7ffff1279640 (LWP 45121)]

Thread 6 "JUCE ALSA" hit Breakpoint 1, SimpleDelayAudioProcessor::processBlock (this=0x555555f838a0, buffer=..., mi
diMessages=...) at ../../Source/PluginProcessor.cpp:144
144  {
(gdb) █
```

# Out-of-process loading

- Some hosts load plug-ins in a separate process
- Bitwig, Reaper (with some settings), AUv3s
- Need to attach to the plug-in process not the host process:
  - Xcode: Debug->Attach to Process by PID or Name...
  - Visual Studio: Debug->Attach to Process...
  - GDB: `attach <PID>`

# macOS Notarised Hosts

- Since macOS Catalina (10.15), apps distributed outside the App Store must be notarised
- Apps can only be debugged with if they have the `com.apple.security.get-task-allow` entitlement set to `true`
- Must be `false` for notarisation to succeed
- However it is possible to re-sign host binaries!

# macOS Notarised Hosts

- Get app's entitlements using:

```
codesign -d --entitlements :- /path/to/host.app
```

- Modify them to include

```
<key>com.apple.security.get-task-allow</key>
<true/>
```

- Set the new entitlements using:

```
codesign --force --options runtime --sign - --entitlements
/path/to/plist "/path/to/app"
```

# TESTING WITH TOOLS

# auval

- Apple's command line AU verification tool
- Tests basic AudioUnit functionality

```
auval -v aufx DLAY JUCE
```

```
* * PASS
```

```
-----  
AU VALIDATION SUCCEEDED.  
-----
```



# pluginval

- Cross-platform plug-in validation tool for testing AU/VST/VST3
- <https://github.com/Tracktion/pluginval> for source code and tagged releases
- Can be run on the command line or with a GUI

# pluginval

```
pluginval --strictness-level 10 --validate SimpleDelay.vst3
```

All tests completed successfully

Finished validating: SimpleDelay.vst3  
ALL TESTS PASSED

## Possible plug-in improvements

- Parameter change smoothing
- A variable delay length
- Fractional delay lengths
- A LookAndFeel1 to style the widgets



<https://github.com/juce-framework/JUCE>

<https://juce.com/>

<https://juce.com/learn/tutorials>

<https://twitter.com/JUCElibrary>