# Final Project Notes

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### Contents

1	General Information		
	1.1	Description	
	1.2	Tools	
	1.3	Videos about Iplug2	
	1.4	Tutorials about Juce	
2		ital Signal Processing Fast Fourier Transform Algorithm	
	$\frac{2.1}{2.2}$		
	2.3	Wave Shaping	
	2.4	Convolution	
	2.5	Keyboard State	
3	Ack	knoledgments	

# 1 General Information

### Project Link

### 1.1 Description

Fiddle about with trying to develop a realistic simulation for uilleann pipes or violins, or some unique instrumental sound of your own imagination.

### 1.2 Tools

- $\bullet \ \ \mathrm{iPlug2} \ \mathrm{GitHub} \Rightarrow \mathrm{For} \ \mathrm{Creating} \ \mathrm{both} \ \mathrm{plug\text{-}ins} \ \mathrm{and} \ \mathrm{stand\text{-}alone}$
- iPlug2 Wiki
- Juce  $\Rightarrow$  More mature and has more tutorials
- ACM Digital Library

• Physical Modelling

### 1.3 Videos about Iplug2

#### Abandoned

- Oliver Larkin: Faust in iPlug 2
- iPlug2: Desktop Plug-in Framework Meets Web Audio Modules by Oliver Larkin

#### 1.4 Tutorials about Juce

- Juce String Model
- Maximilian Sound Library

## 2 Digital Signal Processing

- Juce DSP
- Digital Signal Processing (DSP) Tutorial

### 2.1 Fast Fourier Transform Algorithm

Faster version of the Discrete Fourier transform.

- Transforms waves into its components or formula
- The inverse can be used to create sound waves from

Can use to get functions.

#### 2.2 Waves

- Sin Wave  $\Rightarrow$  std::sin (x)
- Saw Tooth  $\Rightarrow$  map  $-\pi \pi$  to -1 1 (juce::MathConstants<double>::pi)
- Triangle  $\Rightarrow$  map  $-\pi$  0 to -1 1 and 0  $\pi$  to 1 -1

### 2.3 Wave Shaping

• dsp::WaveShaper

Transforming one signal into another using a transfer function.

- sin(x) can be converted to a square wave using signum transfer function sgn(sin(x))
- This creates a too perfect function and thus we use a hyperbolic tangent transfer function tanh(sin(x))

• To create a square, boost the singal into clipping before using the function tanh(n\*sin(x))

### 2.4 Convolution

• dsp::Convolution

Simulating the reverberation characteristics of a certain space by using a pre-recorded impulse response that describes the properties of the space in question. This process allows us to apply any type of acoustic profile to an incoming signal by convolving, essentially multiplying every sample of data against the impulse response samples to create the combined output.

### 2.5 Keyboard State

 $\bullet$  MidiKeyboardState

Used by the keyboard component to get midi input. Hosted in the plugin processor and accessed by the keyboard component using the processor class.

```
// gets midi buffer from keyboar state and inserts it into current
buffer
keyboardState.processNextMidiBuffer(midiMessages, 0,
buffer.getNumSamples(),true);
```

# 3 Acknoledgments

- The Audio Programmer, Juce Documentation
- Subtractive Synthesis Modelling of the Irish Uilleann Pipes
- https://github.com/leventt/davul/blob/master/Source/Main.cpp finaly getting sample to work.