Audio Plugins with JUCE

EECS 4462 - Digital Audio

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Setup

- Create a new project with the Projucer
- The GUI component as well as the audio parameters for the plugin processor are the same as before
- The main difference is in the processBlock method
- We will now work with the AudioBuffer<float>& argument



Important Class: AudioBuffer

- A collection of sample values
- Can be read using pointer arithmetic

```
const float* data =
    buffer.getReadPointer (channel,0);
float s1 = data[5];
```

• Can be written using pointer arithmetic

float* data =

buffer.getWritePointer (channel,0);
float s2 = data[3];
data[7] = s2 + 0.3;



A more realistic example

- Double the value of every sample
- This will make the output of the plugin louder

- Method getNumSamples() returns the number of samples in the buffer
 - In processBlock, typically the value is 1024



Warning

- From the API of getWritePointer:
- For speed, this doesn't check whether the channel number or index are out of range, so be careful when using it!
- Common practice in audio to avoid checks
- It's up to the programmer to make sure all values will be valid!



Storing audio data

- You can create your own AudioBuffer objects
 - In fact, you will need to do that for A2
- You most likely want to use the constructor AudioBuffer (int channels, int samplesToAllocate)
- You can change the size of an AudioBuffer with setSize (int channels, int samplesToAllocate)



Starter code for A2

AudioBuffer<float> delayBuffer;

```
How to set this to 2 sec?
delayBuffer.setSize(2, delayBufferLength);
delayBuffer.clear();
.....
float* delayData =
delayBuffer.getWritePointer(ch);
```

