

Generating Sound

EECS 4462 - Digital Audio

Click to edit Master text styles

Second level

Third level

Fourth level

Fifth level

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Plugins vs Apps

- So far, we have implemented plugins that receive input from a host and provide output back to the host
- With JUCE, one can also create standalone apps
- We will look at how to do that in the context of generating sound

Important Class: AudioAppComponent

- Our app must inherit from AudioAppComponent
- AudioAppComponent takes care of connecting to the audio inputs and outputs of your computer
- Need to implement three methods that should sound familiar

prepareToPlay()
releaseResources()
getNextAudioBlock()

Two important methods to call

- To get access to two audio inputs and outputs, we need to call

setAudioChannels(2,2);

- This call will also start the loop of calling **getNextAudioBlock()**
- When finished, we need to call

shutdownAudio();

White noise generation

- Let's examine the code in the white noise generator tutorial
- Important: **getNextAudioBlock()** receives an **AudioSourceChannelInfo** as an argument
 - Just a struct that contains an audio buffer and two ints: The first sample to write at, and how many samples to write

Sine wave generation

- Let's examine the code in the sine wave generator tutorial (3 different versions)
- V1 creates a sampled version of the sine function
- V2 adds smooth transitioning to the new frequency when the slider is changed
- V3 adds a level slider

Virtual Instruments

- A virtual instrument is a piece of software that receives MIDI events as input, and produces audio samples as output
- This can be quite complicated if we want to produce sounds rich in frequency content
 - Let's listen to some examples...
- We will use JUCE to create a sine wave based virtual instrument

Important Class: Synthesiser

- The base class for virtual instruments in JUCE
- Contains a collection of **SynthesiserSound**
 - Each sound can apply to specific notes or specific MIDI channels
- Contains a collection of **SynthesiserVoice**
 - Each voice can sound independently
 - When playing multiple notes at the same time, each note is a different voice
 - All audio rendering happens in method **renderNextBlock** of **SynthesiserVoice**

MIDI Synthesiser Tutorial

- Let's examine the code in the MIDI Synthesiser tutorial
- The main app makes a **MidiKeyboardComponent** visible, and delegates all audio rendering to a subclass of **AudioSource** called **SynthAudioSource**
- **AudioSource** is a superclass of **AudioAppComponent** and is the one that declares methods

```
prepareToPlay()  
releaseResources()  
getNextAudioBlock()
```

MIDI Synthesiser Tutorial

- **SynthAudioSource** contains a **Synthesiser** object
- Four voices and one sound are added to the synthesiser
- **SineWaveVoice** inherits from **SynthesiserVoice**
- **SineWaveSound** inherits from **SynthesiserSound**
- **getNextAudioBlock** receives a **MidiBuffer** from the keyboard and passes it to the **renderNextBlock** function of the synthesiser, which in turn calls the **renderNextBlock** method of each voice

SynthesiserVoice::renderNextBlock

- **renderNextBlock** creates sample values as before
- Rather than setting direct values to samples, use the **addSample** method
- If the voice has finished producing sound, call **clearCurrentNote** to free the voice for the next note
- Can choose to have sound trail off slowly by continuing to produce sound with decreasing level (see code)

SynthesiserVoice (other methods)

- **canPlaySound** determines if the voice can play a particular sound
- **startNote** initializes class attributes for the next note to be rendered
- **stopNote** indicates what happens when a note has ended
- **pitchWheelMoved**, **controllerMoved** etc. react to the corresponding MIDI events

Wavetable Synthesis

- Real, complex sounds are composed of hundreds of frequencies
- Creating these by calculating and adding as many sine waves is very computation-intensive
 - See v1 of the wavetable synthesis tutorial
- Using wavetables, i.e. precomputed signals, we can accelerate computation by interpolating on the precomputed signal rather than computing directly
 - See v2 of the wavetable synthesis tutorial