The Next-Gen Dynamic Sound System of Killzone Shadow Fall





Promote experimentation

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- Own your tools to allow it to happen

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- Reduce "Idea to Game" time as much as possible



 If a programmer is in the creative loop, creativity suffers

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- Don't be afraid to give designers a visual programming tool
- Generating code from assets is easy and powerful



Who are Andreas & Anton? Why would you listen to them? Well, we worked on sound for these games...



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We work at Guerrilla, which is one of Sony's 1st party studios. Here's a view of our sound studios





After Killzone 3, we started working on Killzone Shadow Fall, which was a launch title for the PS4. This was planned as a next-gen title from the beginning.



Here's a short video showing how the game looks and sounds.



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Andreas: But what exactly does next-gen mean?

We thought next gen sound (initially) wouldn't be about fancy new DSP but about faster iteration and less optimization. To allow the creation of more flexible sounds that truly can become a part of the game design. We also wanted to make sure that the sound team works with the same basic work flow as the rest of the company.

So our motivation was to build a **cutting edge** sound system and tool set, but **don't build a new synth**.



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- Instantly hearing work in game, never reload

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- Extendable system, reusable components
- High run-time performance
- Don't worry about the synth



Audio is separate from rest

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- Physical: sound proof rooms

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- Mental: closed off sound engine and tools

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- Physical: sound proof rooms
- Mental: closed off sound engine and tools
- Should actively pursue integration



• Gain a community of end users that:

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 - Can help with complex setup

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 - Find bugs

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 - Teach you new tricks
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 - Benefit mutually from improvements



Motivation for Change

Previous generation very risk averse, failing was not possible
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	New Workflow		
Nuendo O Sound Tool		Running Game	







Video of iterating on a single sound by re-exporting its wav files from Nuendo and syncing the changes with the running game.



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• Game is the inspiration

- Game is the inspiration
- Get sound into the game quickly

- Game is the inspiration
- Get sound into the game quickly
- No interruptions

- Game is the inspiration
- Get sound into the game quickly
- No interruptions
- Why audition anywhere else?



• Make sound engine and tools in-house

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- Allows deep integration

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- Immediate changes, based on designer feedback

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- Allows deep integration
- Immediate changes, based on designer feedback
- No arbitrary technical limitations

Extendable, Reusable and High Performance?









Anton: Sound designers are already used to data flow environments such as Max/MSP and puredata.

Andreas: But while that's one good way of thinking about our system, it's also a bit different from them. Our system doesn't work at on the sample level or audio rate, i.e. we're not using our graphs for synthesis, but just for playback of already existing waveform data.



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• We will show code

• But you don't have to be afraid

▲ Technical Details Ahead! ▲

- We will show code
- But you don't have to be afraid
- Sound designers will never have to look at it



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- Execute resulting program every 5.333ms (187Hz)
- Used to play samples and modulate parameters
- NOT running actual synthesis
- Dynamic behavior



The functionality of each node is defined in a **C++ file**, which typically just contains **one function**.

When we need to translate this graph into C++ code, we take those **node definition files** and paste them into a C++ file for the whole graph.



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_	Sound.cpp	
	A.cpp	
	B.cpp C.cpp	

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The most simple graph sound example we can come up with

Lets get to know the system by building a very basic sound











graph_01_Inputs		Wave			
control		control_	outAllVoicesDone		
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	Wave_01

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Andreas: This is the **result of the code generator**, it's fairly readable.

You can see the **Wave node function call** and how the **on_start input** of the graph is passed as **a parameter**. The **sample** is a constant **resource attached to the graph** and can be retrieved with this function, which is implemented in the engine code.

Clang's optimizer creates really tight assembly code from this. It's super fast, so we can run it at a high rate, currently once per synth frame, i.e. every 5.3 ms.

As you can see, every input of the graph, becomes a parameter of the graph function, and each node input is also a parameter of the node function.

But what about the code?

```
void Sound(bool _on_start_, bool _on_stop_)
{
    bool outAllVoicesDone = false;
    float outDuration = 0;
    Wave(_on_start_, false, false,
        WaveDataPointer, VoiceParameters, ...,
        &outAllVoicesDone, &outDuration);
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	1.0 DinSustain		
	0.0 DinRelease		
	None DinVoiceLimitMode		
	10 inVoiceLimit		
	Wave	_01	

But the nice thing is, that the code is not immediately visible. It's there, and as a programmer you can use it to debug what's going on, but a sound designer would only see the graph representation.

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	WaveResource Resource that represents sound wave sample data

Anton: Now **let's inject the sound into the game** and **play it**, for testing purposes.

Video showing how a sound is created from scratch and injected into the running game.

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Video showing effect of distance-based low pass filtering example



Video showing effect of distance-based low pass filtering example



I guess by now you can see how this system allowed us to build powerful dynamic sounds, just limited by the set of nodes we have available.

But why is it truly extendable?

That's because it's really easy to create a new node in this system. Basically all you need to do is create an asset describing the inputs and outputs of the node (can be done in our editor) and a C++ file containing the implementation.

Our code generator will collect the little snippets of C++ code for each node and create a combined function for each graph. No external libraries are used, even for math we call functions in the game engine. The generated graph programs are not linked with anything, so very small and light weight.



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- Create C++ file with one function
- Create small description file for node
 - List inputs and outputs

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- Create C++ file with one function
- Create small description file for node
 - List inputs and outputs
 - Define the default values

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 - Define the default values
- No need to recompile game, just add assets

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Andreas: Let's look at an **example node for a mathematical problem**, the obvious ones like add or multiply are trivial, but here's a useful one, **converting from musical semitones to a pitch ratio** value. So for instance inputting a value of **12 semitones**, would result in a **pitch ratio of 2**, i.e. an **octave**.



This shows how easy it is to create a new node. The only additional information that needs to be created is a meta data file that describes the input and output values and their default values.





Here's another simple example for comparison of values. Typically you have different parts that depend on the comparison result, so it's useful to have "smaller" and "greater than" available at the same time. Also saves us from using additional nodes.



The implementation is just as simple. All the **comparison results are stored in output parameters**.





This is the sine LFO node, it will output a sine wave at a given frequency and amplitude. This is a slightly more complicated node.

Let's look at how this is implemented. It's using **two functions** that are implemented in our **engine**, one for **calculating the sine** and one for **getting the time step**. If the game time is scaled (for example if we're running the game in slow motion), then the sound time-step will also be adjusted. However it's also possible to have sounds that are not affected by this.

Normally the graph execution is state less, that means that every time it runs, it does exactly the same, because it doesn't store any information, so it can't depend on previous runs. This node however is an example of something that requires state information, and it's possible to do that on a per node basis. So here you can see that it uses a special "phase" parameter, which is stored separately for each instance of the node, and is passed in by reference as a pointer. The state system allows us to store any arbitrary set of data, for example a full C++ object, or just a simple value like in this example.

The Wave node also uses this to store the information about the voices it has created.



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Certain math or logic processing that starts to happen often in sounds can be abstracted into their own nodes, without having to know any programming. Here's an example of a node, built from an actual graph.









Nodes built from Graphs

Generate a random pitch within a range of semitones

RandomRangedPitchRatio

- oinTrigger outPitchRatio
- o inMaxSemitone



When you notice repeated work... ...abstract it!





Inputs		
Start ChanceToPlay		











 If your idea cannot fail, you're not going to go to uncharted territory.

- If your idea cannot fail, you're not going to go to uncharted territory.
- We tried a lot

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- We tried a lot
- Some of it worked, some didn't

- If your idea cannot fail, you're not going to go to uncharted territory.
- We tried a lot
- Some of it worked, some didn't
- Little or no code support all in audio design





• Experiment: How to make battle sound more interesting

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- Experiment: How to make battle sound more interesting
- Idea: Slight variation in fire rate

- Experiment: How to make battle sound more interesting
- Idea: Slight variation in fire rate





• Experiment: Birds should react when you fire gun

- Experiment: Birds should react when you fire gun
- Birds exist as sound only

- Experiment: Birds should react when you fire gun
- Birds exist as sound only
- Use sound-to-sound messaging system

- Experiment: Birds should react when you fire gun
- Birds exist as sound only
- Use sound-to-sound messaging system
- Gun sounds notify bird sounds of shot


















Material Dependent Environmental Reactions (MADDER)

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 Inspiration: Guns have this explosive force in the world when they're fired

Material Dependent Environmental Reactions (MADDER)

- Inspiration: Guns have this explosive force in the world when they're fired
- Things that are near start to rattle



See http://www.youtube.com/watch?v=nMkiDguYtGw



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MADDER

• Inputs required:

• Distance to closest surface

MADDER

- Inputs required:
 - Distance to closest surface
 - Direction of closest surface

MADDER

- Inputs required:
 - Distance to closest surface
 - Direction of closest surface
 - Material of closest surface










































Video showing MADDER effect for 4 different materials



Video showing MADDER effect for 4 different materials



Video showing initial MADDER prototype with 4 different materials in scene.



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- Divide in four quadrants around listener
- One sweeping raycast in each quadrant



MADDER Four-angle raycast

- Divide in four quadrants around listener
- One sweeping raycast in each quadrant
- Closest hit point in each quadrant is taken















Video showing final 4-angle MADDER with 4 materials placed in the scene



Video showing final 4-angle MADDER with 4 materials placed in the scene



Capture from final game with MADDER disabled



Capture from final game with MADDER disabled



Capture from final game with MADDER enabled



Capture from final game with MADDER enabled









• Experiment with existing data

- Experiment with existing data
 - Wall raycast

- Experiment with existing data
 - Wall raycast
 - Inside/Outside

- Experiment with existing data
 - Wall raycast
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 - Rounds Fired

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What kind of **debugging support** did we have? Obviously a **sound designer is creating** much more **complex logic** now, and that means **bugs will creep in**. We need to be able to **find problems** such as incorrect behaviour **quickly**.

Manually Placed Debug Probes

SineLFO		
0.5 O inFrequency 0.5 O inAmplitude 1 O inOffset	outValue	
0.0 inPhase		

Anton: These are **placed by designers** into their sounds, to **query individual outputs** of nodes. The values are shown as a **curve on screen**, useful for debugging a single value and how it changes over time. **Not very useful for quickly changing things**, such as boolean events. Since the sound **graphs execute multiple times per game frame**, but the **debug** visualisation **is only drawn at the game frame rate**, a **quick change can be missed**. In the final version of the game, these nodes are completely ignored.

Screenshot!
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Automatically Embedded Debug Probes

- All values are recorded, nothing is lost
- Simple in-game debugger to show data

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- All values are recorded, nothing is lost
- Simple in-game debugger to show data
- Scrub through recording

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So what when wrong and what went right?







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Andreas: We started working on PS4 technology very early, many parts of the PS4 hardware were still undefined when we started. At one point we even expected the hardware to do more than decoding, i.e. we assumed that there might be a way of executing our own DSP code on it. We thought we might have to write custom DSP code for that chip. At that time there were no middleware vendors disclosed yet, so we just couldn't really talk to them about any of this.

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New system was up and running within 6 months!

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"You're making sound designers do programming work, that'll be a disaster! The game will crash all the time!"

This didn't really happen. We had **a few buggy nodes**, but in general, **things worked** out fine. What we did is to make sure that if **new nodes are peer-reviewed** like any other code, especially **if** it's **written by** a **non-programmer**. Also since nodes are so simple and limited in their scope, it's hard to make something that truly breaks the game.

"This is an unproven idea, why don't we just use some middleware solution?"

There were **doubts** that we can **create new tech and a toolset** with a user friendly workflow in time. The safer thing would've been to **license a middleware** solution, but at the time we had to make this decision, **no 3rd parties** were **disclosed** about PS4 yet, and we couldn't be sure if they will properly support the PS4 audio hardware.

Also we were keen on having a **solution** that's **integrated with** our normal **asset pipeline**, which allowed us to **share the tech** built for audio with other disciplines.

Sound designers using the same workflow as artists and game designers is a benefit that middleware can't give us. New nodes created by e.g. game programmers are immediately useful for sound designers, and vice versa. All of these reasons led us to push forward with our own tech, to make something that really fits the game and our way of working.

Anton: Also we wouldn't have been able to do the kind of deep tool integration that we have now.





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- Used for procedural rigging

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On PS3 most sounds were ADPCM, a few were PCM, only streaming sounds used MP3 with various bit rates. We had about 20 MB budgeted for in-memory samples.

This generation we ended up using a combination of PCM and MP3 for in-memory sounds, with a total of about 300MB of memory used at run-time, that's roughly 16x as much as on PS3, but still the same percentage of the whole memory (roughly 4%). We didn't use any ADPCM samples anymore (good riddance). The split between PCM and MP3 is roughly 50:50. We used PCM for anything that needs to loop or seek with sample accuracy, and MP3 for larger sounds that don't require precise timing, such as the tails of guns, for example. Obviously we relied on the audio coprocessor of the PS4 to decode our MP3 data.



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- Create compute shaders from graphs

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• Improve workflow even further

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- Time to Game could be immediate
- Integrate asset creation side even more

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Integrated with our asset pipeline and workflow

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- Generate C++ and compile to native code



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Any questions?