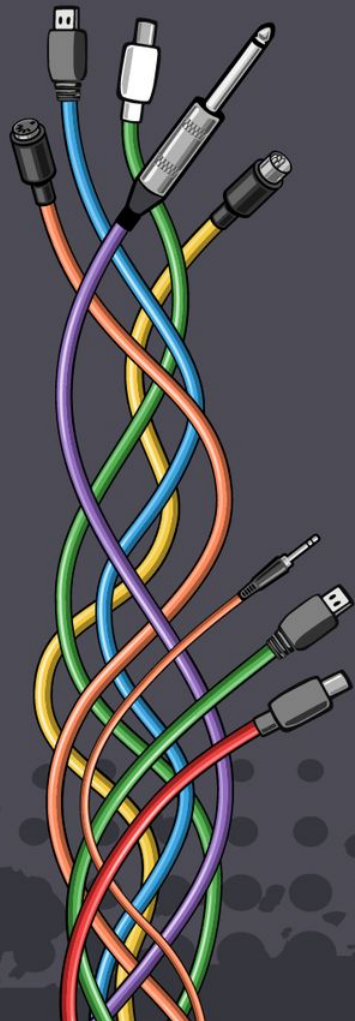




CREATING FROM LEGACY CODE

*A CASE STUDY OF PORTING LEGACY CODE
FROM EXPONENTIAL AUDIO*

HARRIET DRURY



Hello! My name is Harriet



- Software Engineer at Native Instruments
- Working on iZotope Products for 2 years
- DSP/ML Focus
- Guitarist
- Can pronounce Cynefin Correctly

Reverbs, Quickly...



<http://articles.ircam.fr/textes/Jot97b/>

Jean-Marc Jot

Efficient Models for Reverberation and
Distance Rendering in Computer Music
and Virtual Audio Reality

<https://valhalladsp.com/2010/12/21/the-reverb-beard/>

The Reverb Beard - Valhalla DSP

Something that I find rather curious, is
that many of the reverb pioneers sported
some seriously impressive beards.

<https://ccrma.stanford.edu/~jos/pasp/pasp.html>

PHYSICAL AUDIO SIGNAL PROCESSING

FOR VIRTUAL MUSICAL INSTRUMENTS
AND AUDIO EFFECTS

Reverbs, Quickly...



Reverbs, Quickly...



21



Reverbs, Quickly...



Reverbs, Quickly...



Reverb

Shop for used & new music gear...



Sell your gear



Sign up

Log in

Guitars Pedals and Amplifiers Keyboards and Synths Recording Gear Drums DJ and Audio Gear More Categories

Brands News Explore Artist shops Reverb Gives Help Centre



Vintage 1958 EMT 140 valve plate reverb

Second-hand – Good

£4,995

+ £350 Delivery

Buy it now

Add to Basket

Make an offer



Listed: a month ago Views: 195 Watchers: 24 Offers: 0



Vintage things ★★★★★

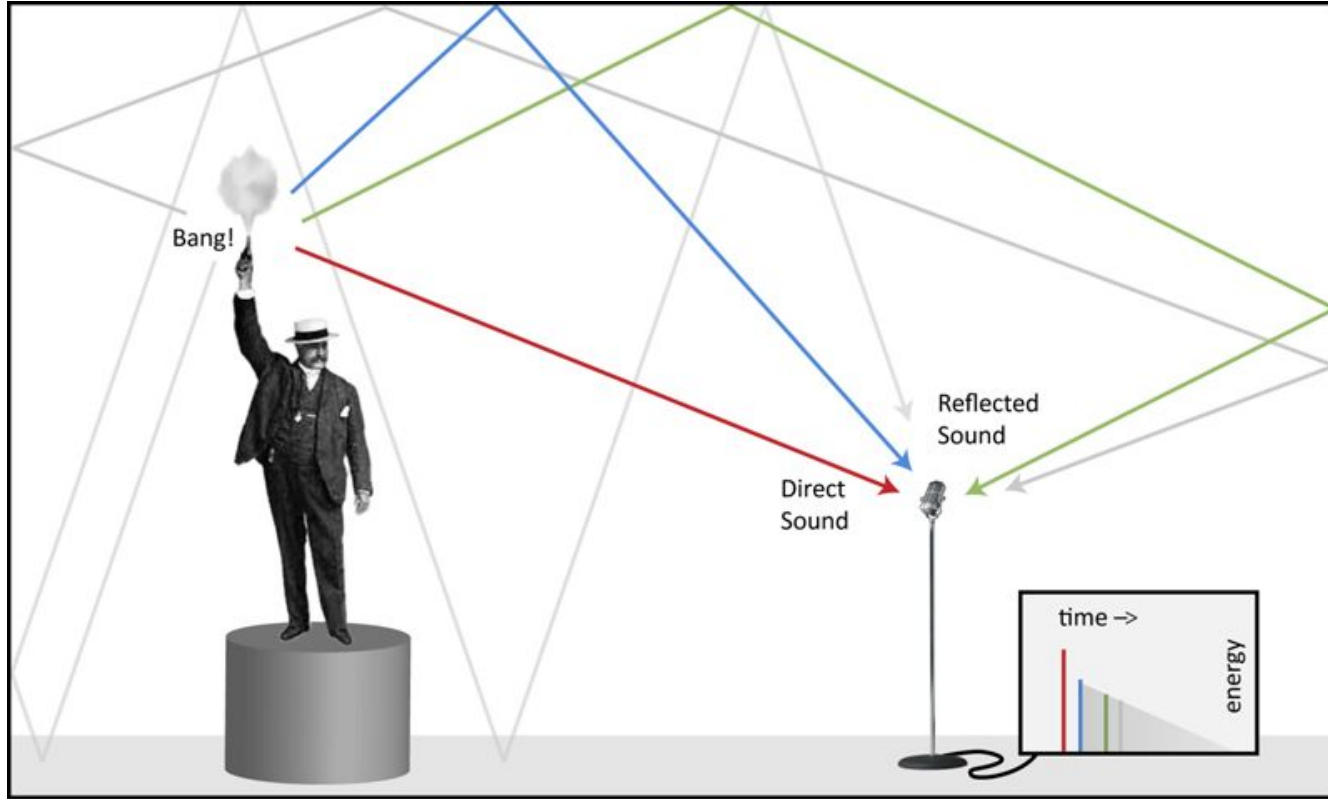
London, United Kingdom

Our users rate Reverb

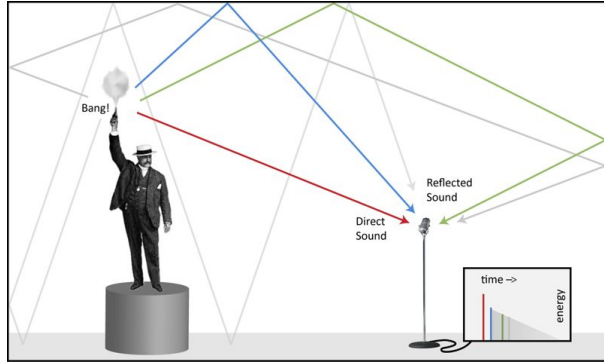
Excellent ★★★★★ Trustpilot



Digital Modelling - Capture Impulse Response (IR)



From: <https://www.prosoundweb.com/what-is-an-impulse-response/>



$$y(n) = (h * x)(n)$$

Example Impulse Response Plugin - Altiverb



The Altiverb software interface is displayed, showing a 3D scene of St David's Hall Cardiff. The interface includes a large circular control knob labeled "DECAY" on the left. The main 3D view shows the interior of the hall with green seats and a stage. Below the 3D view, there are controls for "BRIGHTNESS", "BASS", and "TREBLE", along with an "EQ ON" button. A sidebar on the right lists various impulse response locations, including "New York", "Paris", "Berlin", and "Manhattan". The sidebar also includes a search bar and buttons for "SHOWING DEFAULT IRs" and "RECOMMENDED".

altiverb
Binaural Reverb Processor

DECAY

St David's Hall Cardiff
11.4m (2.0 sec)

HOME • COLOR • TIME • LEVELS • SNAPSHOTS

BRIGHTNESS BASS TREBLE

EQ ON

MUSIC
concert hall
religious
studio
scoring stage
opera & theatre
stadium
mausoleum
club
miscellaneous

POST
GEAR
DESIGN
USER
THIRD PARTY

Similar to Kaufman Astoria Similar RT60 of 1.2 sec New Impulses Favorites Last Used recording studio

New York

20th c. Fox scoring stage
Paramount Stage M
Todd-AO Los Angeles

Davout studio A Paris
Kaufman Astoria New York
Teldex Berlin

Clinton recording Manhattan
Filmorchester studio 1 Berlin
Filmorchester studio 2 Berlin

SHOWING DEFAULT IRs RECOMMENDED

Example Convolution Plugin - Trash



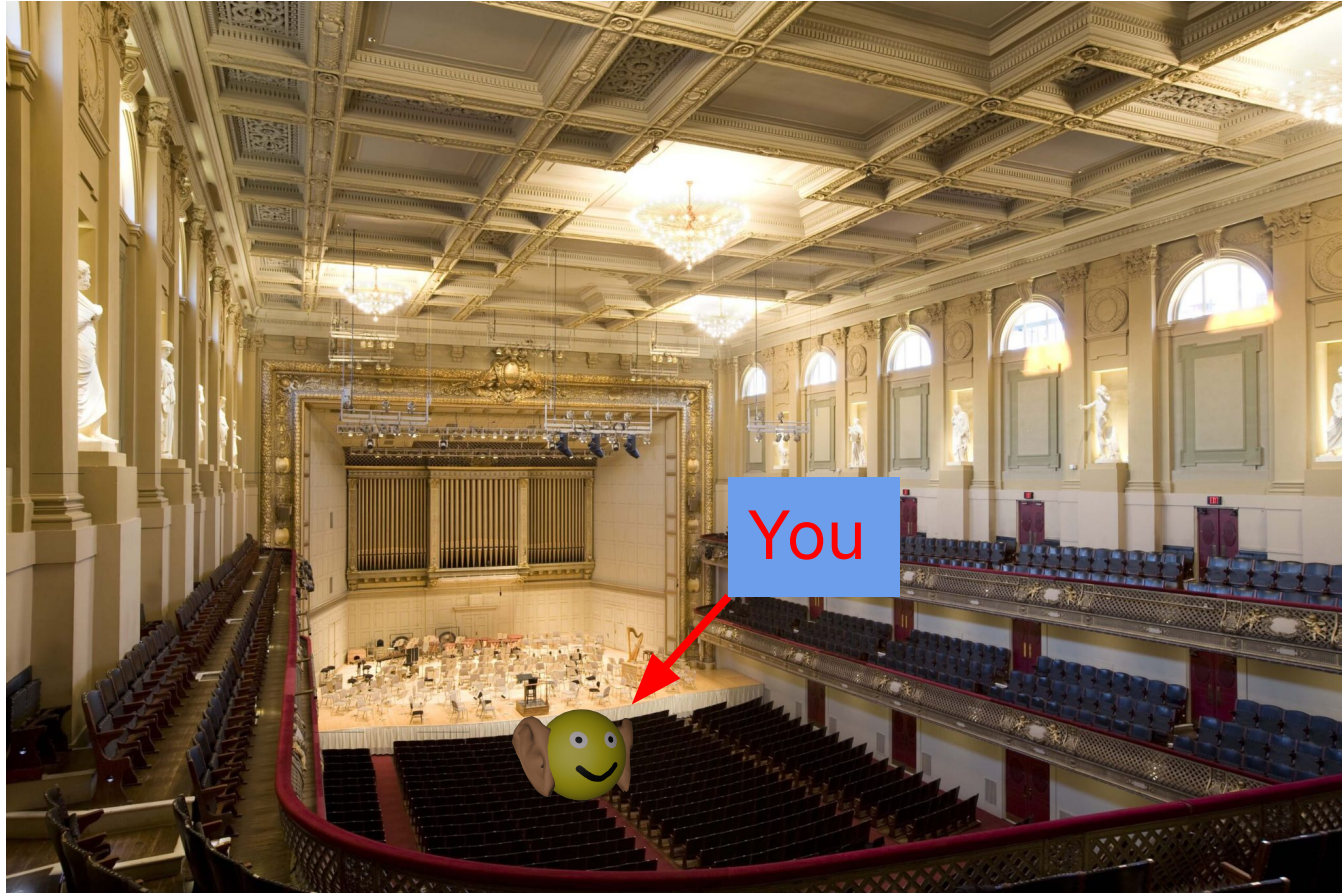
Digital Modelling - Impulse Response



Digital Modelling - Impulse Response



2



Digital Modelling - Impulse Response



2



Digital Modelling - Impulse Response



Digital Modelling - Algorithmic Reverbs



idk, it just sounds like
Boston Symphony
Hall



Digital Modelling - Algorithmic Reverbs

An algorithmic reverb is based on a mathematical model that simulates the behavior of a physical space.

- Reflections

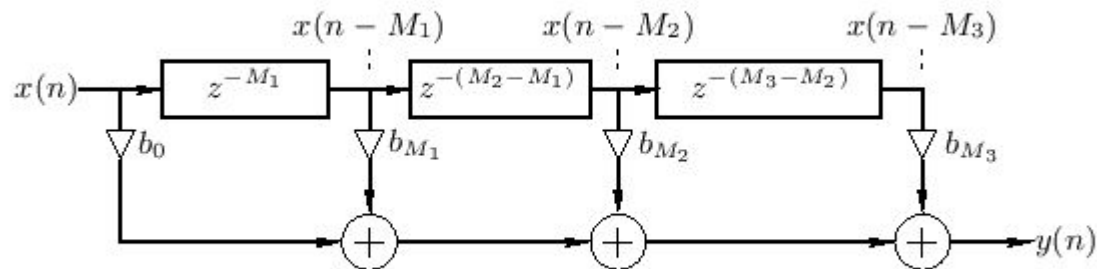
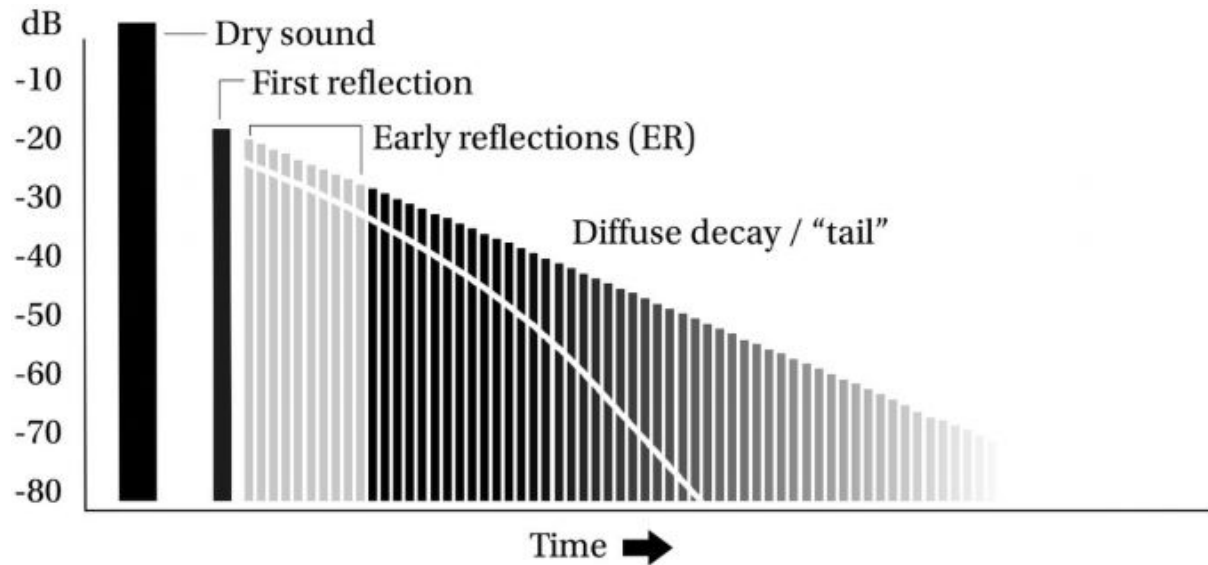
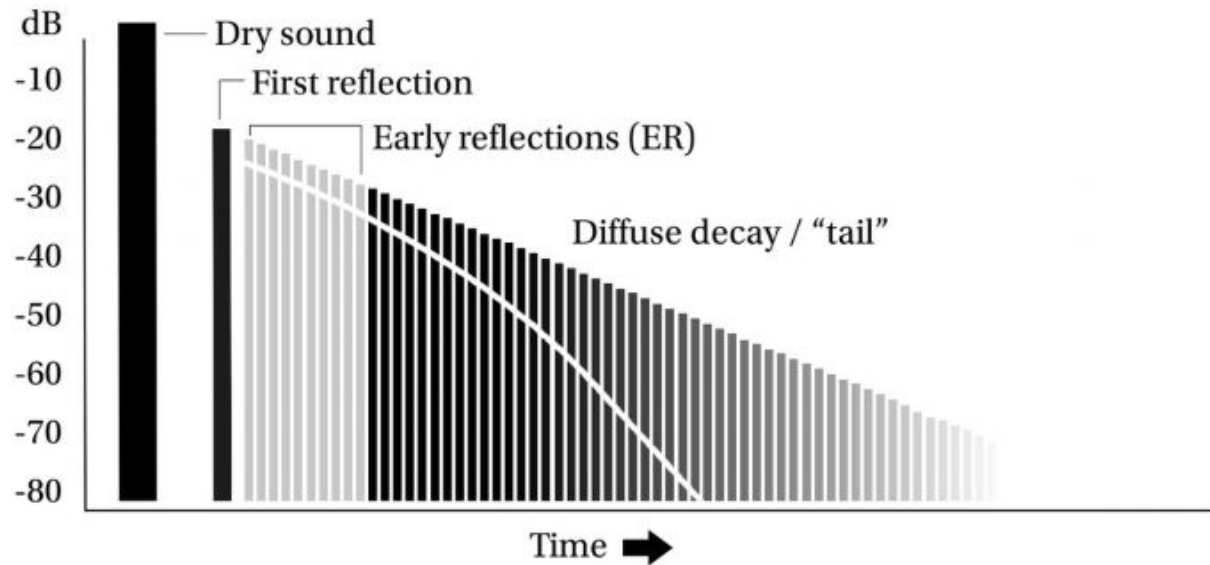


Figure 2.19: Tapped Delay Line (TDL).

- Reflections
- Tail



- Reflections
- Tail
- Diffusion







Stereo & Surround Plugins



Michael Carnes



Musician, Composer, Engineer, Maker of Great Things

Studied at Boston University and M.I.T

Now retired from engineering

**EXPONENTIAL
AUDIO** 

AN **iZOTOPE** COMPANY

Acquired by iZotope in 2019

M1 support added with UI refresh in 2022

1st Generation



2nd Generation - 2017 Onwards





- Early Reflections similar to Phoenixverb/R2

Symphony



Goals

Fat tail, character, modulation as a feature

- Ability to 'Freeze' a buffer, in perpetuity
- Chorusing Junction

Equinox

Released April 2025

Two Tank Based reverb engines (Stratus & Symphony)

Built by incorporating Exponential Audio DSP into a shared lib in iZotope's codebase



Equinox



1. **Main controls:**
Contains the most important settings.

1

2. **Header:** Contains global controls including the Preset selector.

2

3. **Additional controls:**
Provide detailed settings for the reverb engine.

3

4. **EQ section:** Lets you adjust the filtering for the input signal, the early reflections, and the tail of the reverb.

4

5. **Dynamics Controls:**
Lets you apply a saturation, a compression, and a gate to the reverb sound.

5



Code maintained by someone other than the original author

 README



Exponential Audio Check-in

This is the entire Development folder delivered to iZotope on a thumbdrive May 1, 2019.

Code maintained by someone other than the original author

Legacy Code



EXPO-development Repository

Development Tools

- Tools/ Audio Utilities
 - EQ Correction Generator
 - Pan Table Tools
 - Tap Scrambler
 - Build Scripts

- Documentation
 - Doxygen Config
 - API Documentation
 - Developer Notes

Core Infrastructure

- Build System
 - Python Build Scripts
 - Cross-Platform Support
 - Code Signing
 - Package Generation

- DSPLibrary
 - Reverb Algorithms
 - Filters & EQ
 - Delay Effects
 - Parameter Management
 - GUI Components

- SDKs/ Plugin Standards
 - JUCE Framework (Heavily Customised)
 - AAX (Pro Tools)
 - VST/VST3 (Universal)
 - AU (Apple)

Audio Products (11 Professional Plugins)

Next Generation

Symphony
Premium Reverb

Symphony3D
3D Extension

Stratus
Surround Reverb

Stratus3D
3D Extension

Multi-Effects

Excalibur
4-Voice Delay + FX

Core Reverbs

PhoenixVerb
Pure Reverb

PhoenixVerbSurround
Surround

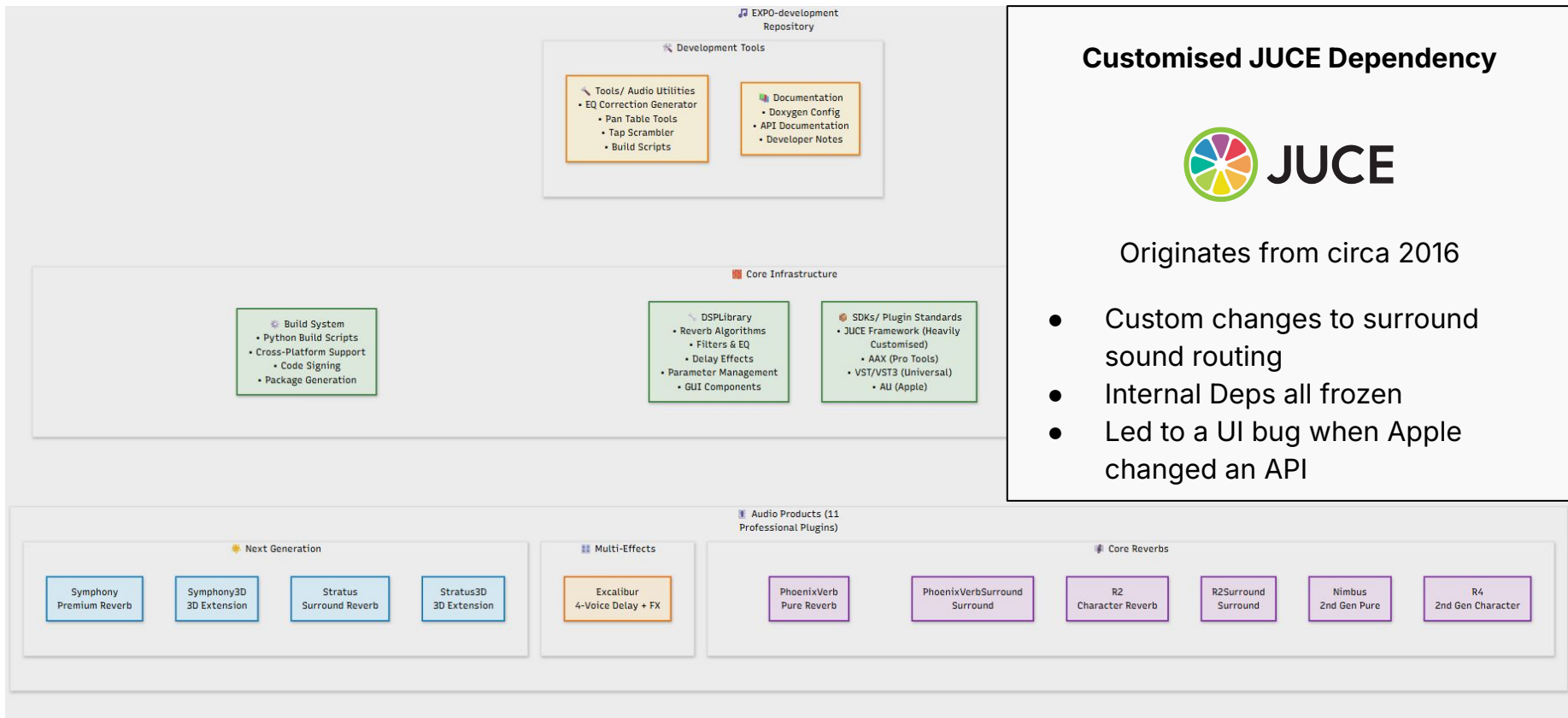
R2
Character Reverb

R2Surround
Surround

Nimbus
2nd Gen Pure

R4
2nd Gen Character

Legacy Code



Customised JUCE Dependency



Originates from circa 2016

- Custom changes to surround sound routing
- Internal Deps all frozen
- Led to a UI bug when Apple changed an API

Legacy Code



Overview

DSPLibrary - Shared Components

Algorithm

- cSurroundReverbNextGenBaseClass
- cMCAgorithm
- PresetDescriptor

Parameter

- Basic Parameters
- Reverb Parameters
- Surround Format Parameters
- Early Pattern Parameters
- Balance Parameters
- Zoom Parameters

Reverb

- cSimpleMultitapReverbSectionVariable
- cSimpleAllpass
- cEarlyReflectionNextGenSurround
- cSimpleReverbJunctionNextGenBaseClass
- Custom Spread Envelopes

Tools

- cCoefficientMultiGlider
- Math utilities
- Random
- Numbers

Editor

- cNewSurroundEditorBase
- Control Objects
- Display Objects
- Parameter Pages
- Preset Management

Delay

- cDelayBaseClass
- cSimpleDelay
- cCircularBuffer
- cChorusMasterNextGen
- cReverbJunctionVariable

Next Gen Products

Stratus

- cStratus :
- cSurroundReverbNextGenBaseClass
- cStratusTank (4x instances)
- cMainEditor :
- cNewSurroundEditorBase
- cSurroundLevelMeterNextGen3D
- EditorInclude.h, EditorValues. h

Stratus3D

- Extends Stratus + Height Channels
- Same UI as Stratus

Symphony

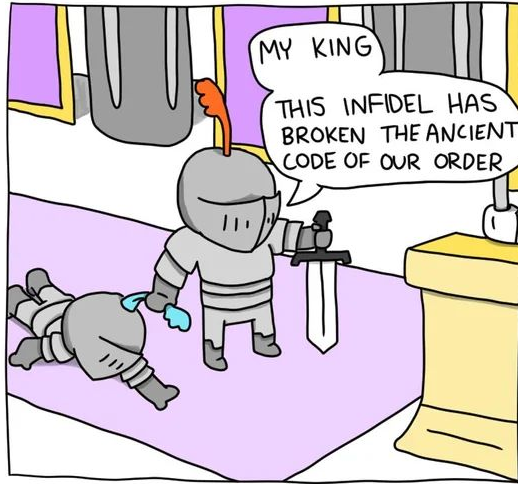
- cSymphony :
- cSurroundReverbNextGenBaseClass
- cSymphonyTank (4x instances)
- cChorusMasterNextGen
- cMainEditor :
- cNewSurroundEditorBase
- cNextGenChorusSubpage
- cNextGenGateSubpage

Symphony3D

- Extends Symphony + Height Channels
- Same UI as Symphony
- Import: R2, R4

- One Monolithic Repo
- Customised Framework Dependencies
- DSP shared between products

Legacy Code



Source:
https://www.reddit.com/r/ProgrammerHumor/comments/9xat04/the_ancient_code/

How can we future proof the DSP written by Michael Carnes?

Legacy Code



1. Continue Building as is

1. Continue Building as is

Pros

- Keeps revenue stream alive
- Branding stays consistent
- Respected code - if it ain't broke...

Cons

- Slow release cycles
- obsolete technology
- Bugs/ instability
- There are only certain team members who understand it

Legacy Code



2. Build as a Shared Lib?

2. Build as a Shared Lib?

Pros

- Quick(ish)
- Clear API Boundary

Cons

- Version Hell
- You have to deploy a DLL
- There is no 'clean' DSP Layer
- No User Benefit

Legacy Code



3. Extract DSP, into a new, shared repo

3. Extract DSP, into a new, shared repo

Pros

- Preserve existing DSP/ Value
- Future Proof the Technology
- Cross Product Potential
- Performance Optimisation Opportunities
- Looked at By More Devs

Cons

- The Most Time Consuming Option
- Risk: Untangling 15+ years of coupled code introduces bugs
- API Design Complexity - how *do* you cleanly abstract an inherited class?

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Cons

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- Risk: Untangling 15+ years of coupled code introduces bugs
- API Design Complexity - how *do* you cleanly abstract an inherited class?

Legacy

3. Extra

hold up



of
gs
w

Neoverb

1-MIDI - Neoverb

Reverb

Reflections 31 Space

Plate 1.10 Space

Hall 2.50 Space

-3.0 dB

-3.0 dB

20 Hz

Pre EQ

Reverb EQ

Hz

100 300 1k 3k 10k

Freq 100 Hz Gain 0.0 dB Q 0.7

Aurora

Solo Reverb Bypass Presets

Hall

0

Reverb

Pre-Delay 10 Decay 2.00

Width

Dark Tone Bright

dB

0 -1 -2 -3 -4 -5 -6 -7 -8 -9 -10 -11 -12

Unmask

10 ms Attack

300 ms Release

50 % Sensitivity

20 Hz 50 100 200 500 1k 2k 5k 10k

Dialogue Match

Snapshots

REFERENCE

1 Select a reference and click **Capture** or **Load a reference profile**

2 Select a clip to apply processing to and click **Capture**

EQ

100 Amount

Reverb

-Inf Wet 0 Dry

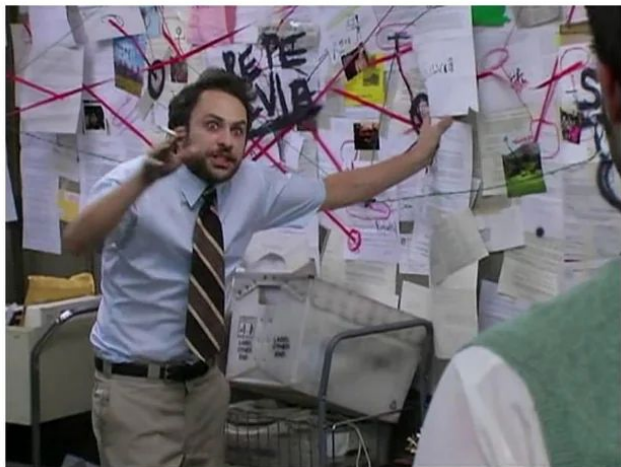
Ambience

0 Trim

Advanced

PhoenixVerb

Me explaining my variable naming scheme to the other devs



In this repo, we have reimplemented Exponential Audio Reverbs with minimal changes using iZotope DSP APIs.

Named after the first reverb to be ported, phoenixverb

DM-3: Commit Initial PhoenixVerb boilerplate #1

 Merged

evan-lynch_ntvinst merged 5 commits into `master` from
`DM-3-setup-shared-dsp-phoneixverb-repo`  on May 9, 2019

Active development since 2019

Contributions from many NI devs over the last
6 years

Phoenixverb, the DSP repo



Increase
maintainability
of ported
expo DSP

Usable
across
product
repos

Testable,
being able to
use iZotope
testing
frameworks

Create an
interface to expo
DSP that matches
pre existing iZo
ideas

Replace
old school,
C styled
code

Utilise
modern
C++ best
practices

Create a
bypass on
the iZo
side

Aux
outputs

What did we
have to Audit
to add expo
features?

Legacy code
for 1 person
vs the
company code

Why Port?



Old JUCE Version (Lead to a UI Bug)

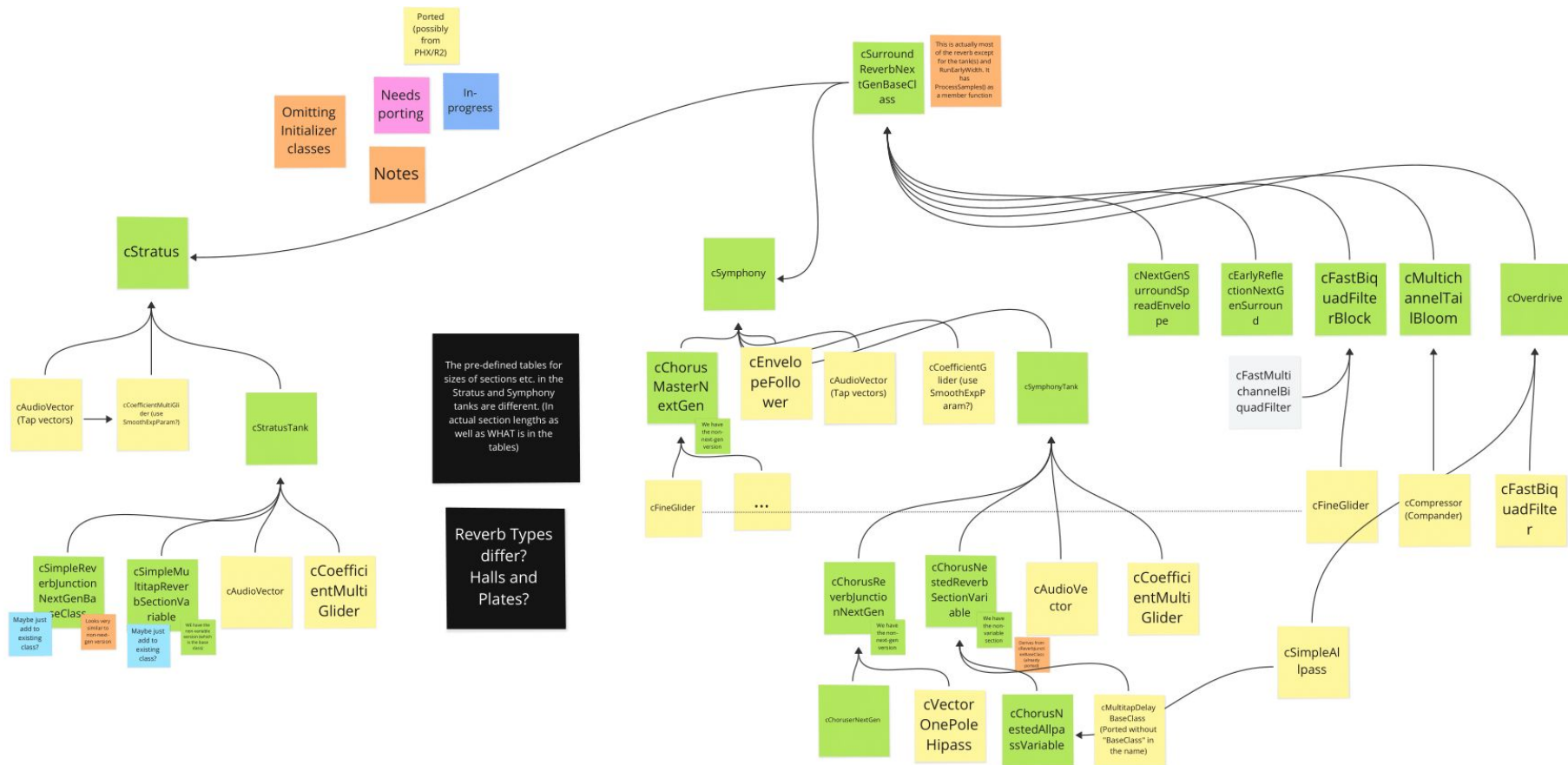
Monolithic Repo of all Exponential Audio
Code

Pre C++17



Harriet Drury 3:51 PM

Happy Monday! I guess today is day 0 of the izo-expo work.



- Get Stratus Ported First
- Divide Work by Class
- Discuss Shared Code as a Team & Decide on Basic Class Layouts
- Add TDSP/ test Coverage
- Sit Back, Relax, Enjoy no Bugs or Misfortunes

Commit d7acb70



harriet-drury_ntvinst committed on Jan 2

Initial Commit of MC's Code



- Initial Commit
- Namespace the Class
- Rename Members/ Variables/ Class
- Modernise Loop Unrolling (Vectorize Block Processors)
- Collapse Multi Stage Init

Commit `d7acb70`

 **harriet-drury_ntvinst** committed on Jan 2

Initial Commit of MC's Code



NOTE: We are not trying to change the class structure of Stratus & Symphony, the goal is to keep all the specific quirks & features of the code.

Commit d7acb70

 harriet-drury_ntvinst committed on Jan 2

Initial Commit of MC's Code



NOTE: We are not trying to change the class structure of Stratus & Symphony, the goal is to keep all the specific quirks & features of the code.

In fact acted as a engine to discussion around design patterns & code design

Me, unrolling a loop in C++ to optimize performance



Code Adjustment - Loop Unrolling/ Auto Vectorization



Loop unrolling is an optimization technique where the body of a loop is duplicated multiple times within a single iteration, reducing the number of loop control operations (like incrementing the index and checking the loop condition). This can improve performance by:

- Decreasing the overhead of loop control instructions.
- Allowing the compiler to better optimize code, especially for SIMD/vector instructions.
- Improving instruction-level parallelism and cache usage.

Me, unrolling a loop in C++ to optimize performance



**The compiler looking at my code:
"I am 4 parallel universes ahead of you."**

```
switch (GlobalPluginEnvironment.VectorSize) {  
case SMALL_VECTOR_SIZE:  
    for (int i = 0; i < SMALL_VECTOR_SIZE; ++i)  
        process(i);  
    break;  
case DOUBLE_VECTOR_SIZE:  
    for (int i = 0; i < DOUBLE_VECTOR_SIZE; ++i)  
        process(i);  
    break;  
}
```



```
template <int VectorSize>
void RunAllpass(float* input, float* output) {
    for (int i = 0; i < VectorSize; ++i)
        process(i);
}
```

Code Adjustment - Loop Unrolling



Aspect	Original Code	Updated Vectorized Code
Loop Structure	Switch/case for each vector size	Template-based, block processing
Code Duplication	High (repeated loops)	Low (single loop, reused via templates)
Buffer Size Handling	Tied to DAW/host buffer size	Fixed internal block size, DAW-agnostic
Maintenance	Harder (add/change in multiple places)	Easier (change in one place)
Compiler Optimization	Fixed-size loops, but repetitive code	Fixed-size, template, unrolled

// Unrolls & Vectorises when using (MSVC) /O2 & (GCC/
Clang) -O3

```
#include <vector>
void xyz(std::vector<float> a, std::vector<float> b)
{
    for (size_t i = 0; i < 3; i++) {
        a[i] += b[i];
    }
};
```

<https://godbolt.org/z/8hxzx88T6>

August - March - Porting & Equinox Plugin Bringup

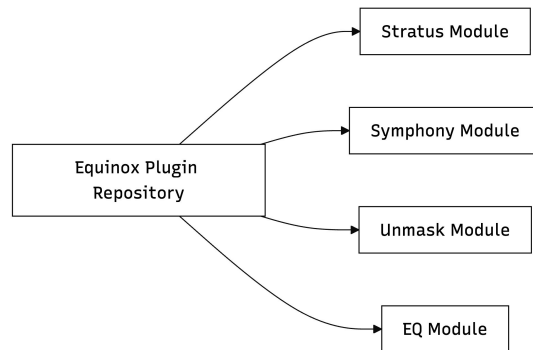


[🎨 ART] Expo-232 create stratus plug in with suites bp #1

Merged

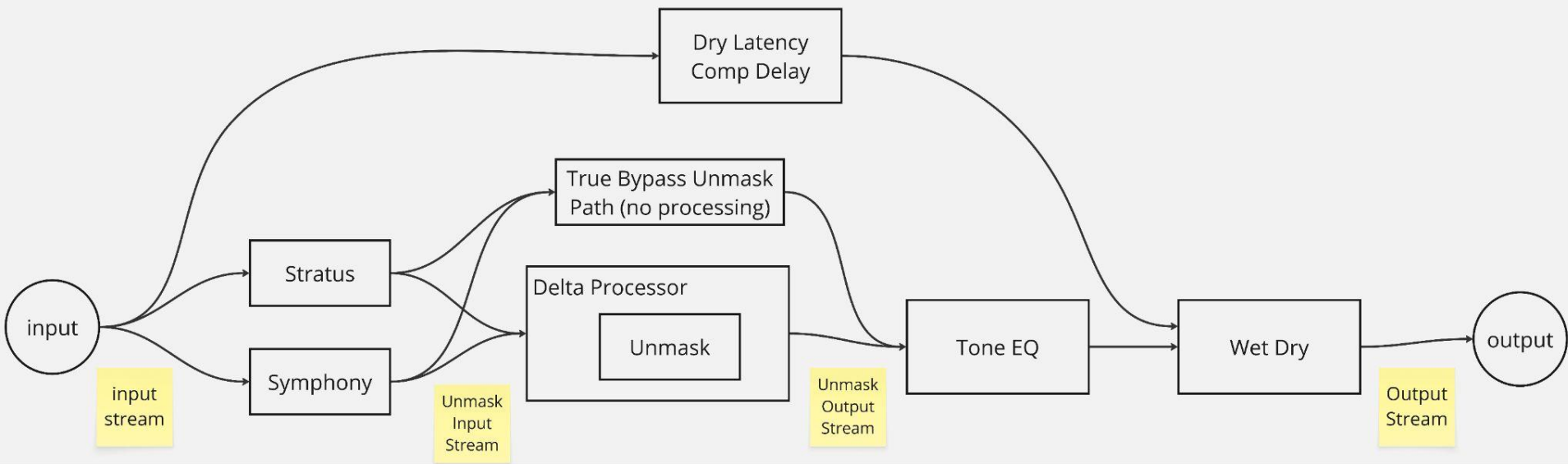
harriet-drury_nt... merged 97 commits into [develop](#) from [EXP0-232-Create-Stratus-Plug-in-with-SuitesBP](#) on Aug 23, 2024

March - We Ship a Beta with Adaptive Unmasking & EQ Added!



- We have Stratus & Symphony DSP in one plugin
- We have started adding our DSP to the Plugin (Adaptive Unmasking & EQ)
- A Placeholder UI Appears!
- The bugs start.....

March - We Ship a Beta with Adaptive Unmasking & EQ Added!



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- A Placeholder UI Appears!
- The bugs start.....

Bug Report! Our Tail is Crackling...



Projects /  Exponential Audio /  Add parent /  EXPO-491

Natural algorithms crackling at high reverb times with large or small size

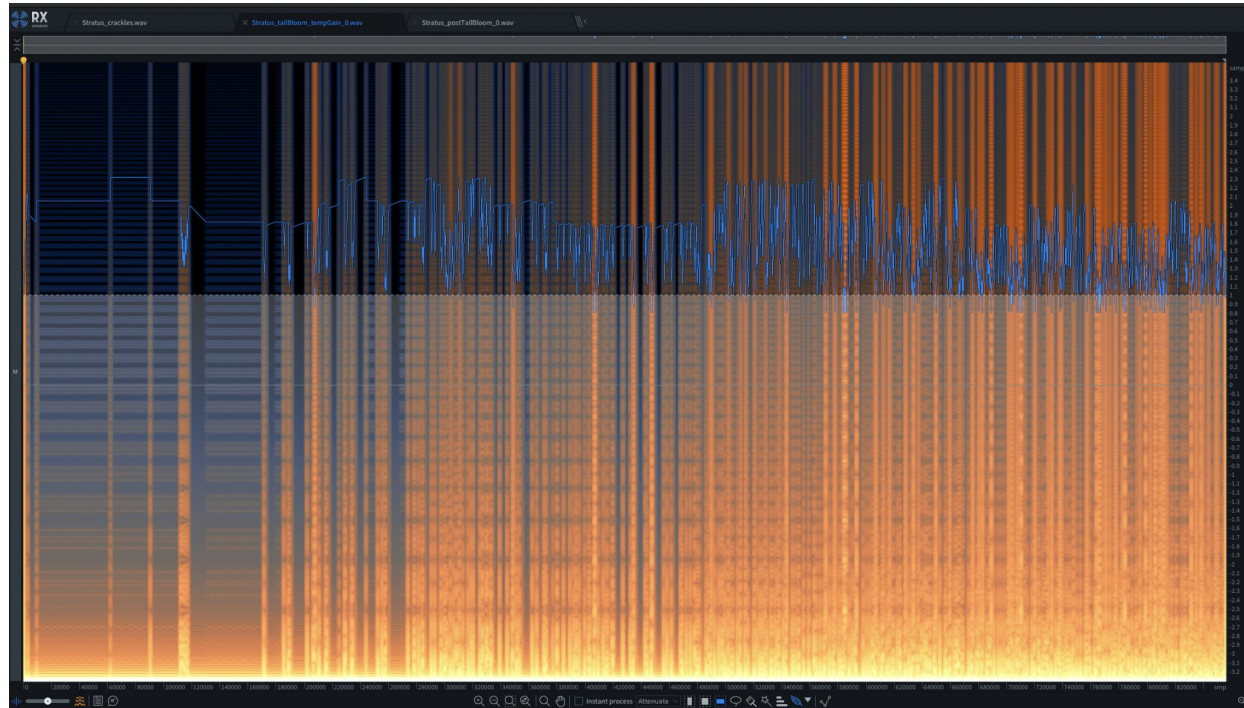


What did we do to change this behaviour?

What we knew:

- Stratus Specific
- Occurs Reliably with Reverb Size set to max (or min), & Reverb Time set to max
- Turning down the Early Reflections level makes crackling in the tail more obvious

Bug Report! Our Tail is Crackling...

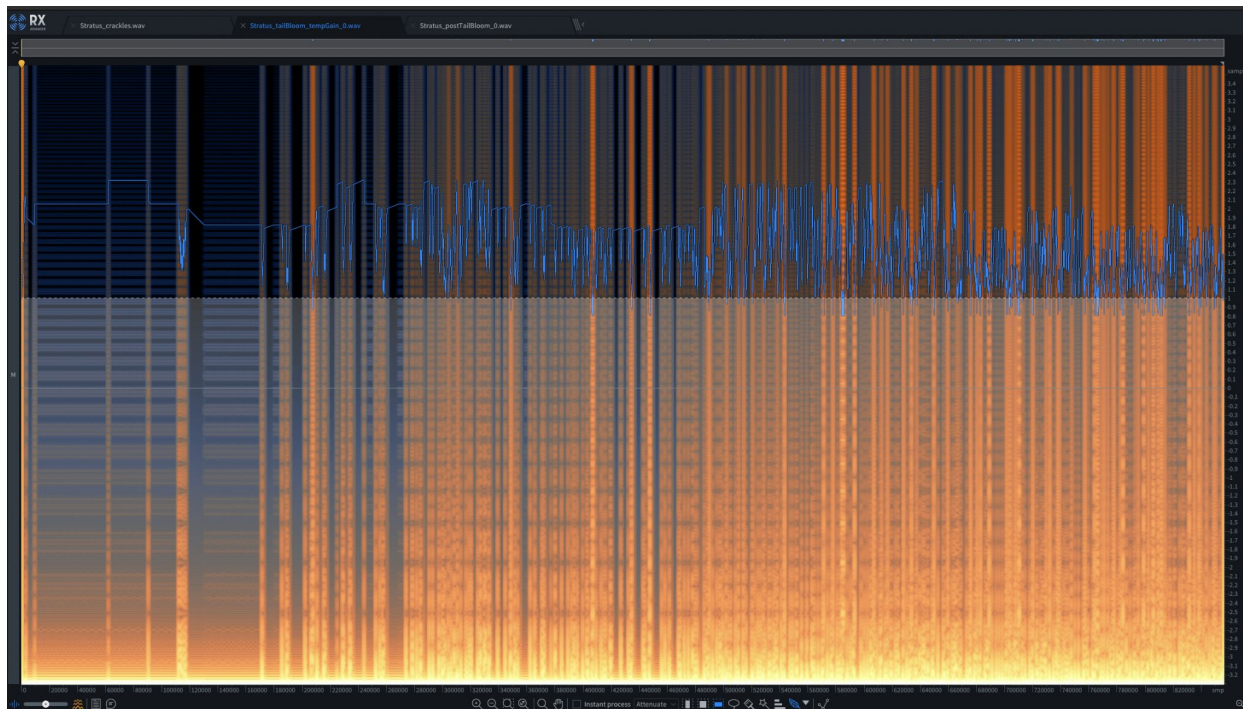


Bug Report! Our Tail is Crackling...



In some reverb modes,
the gain could surpass
1.0f

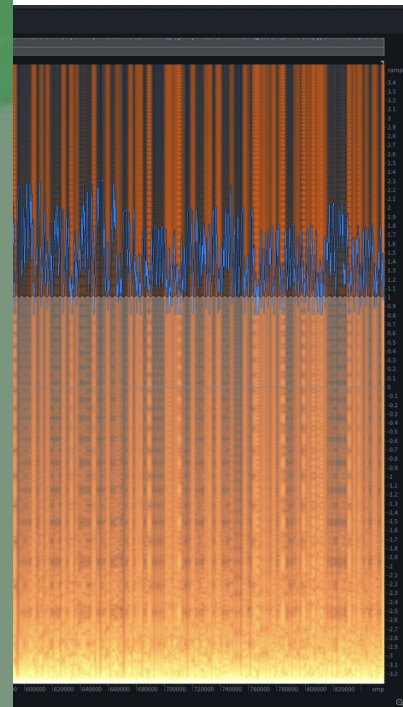
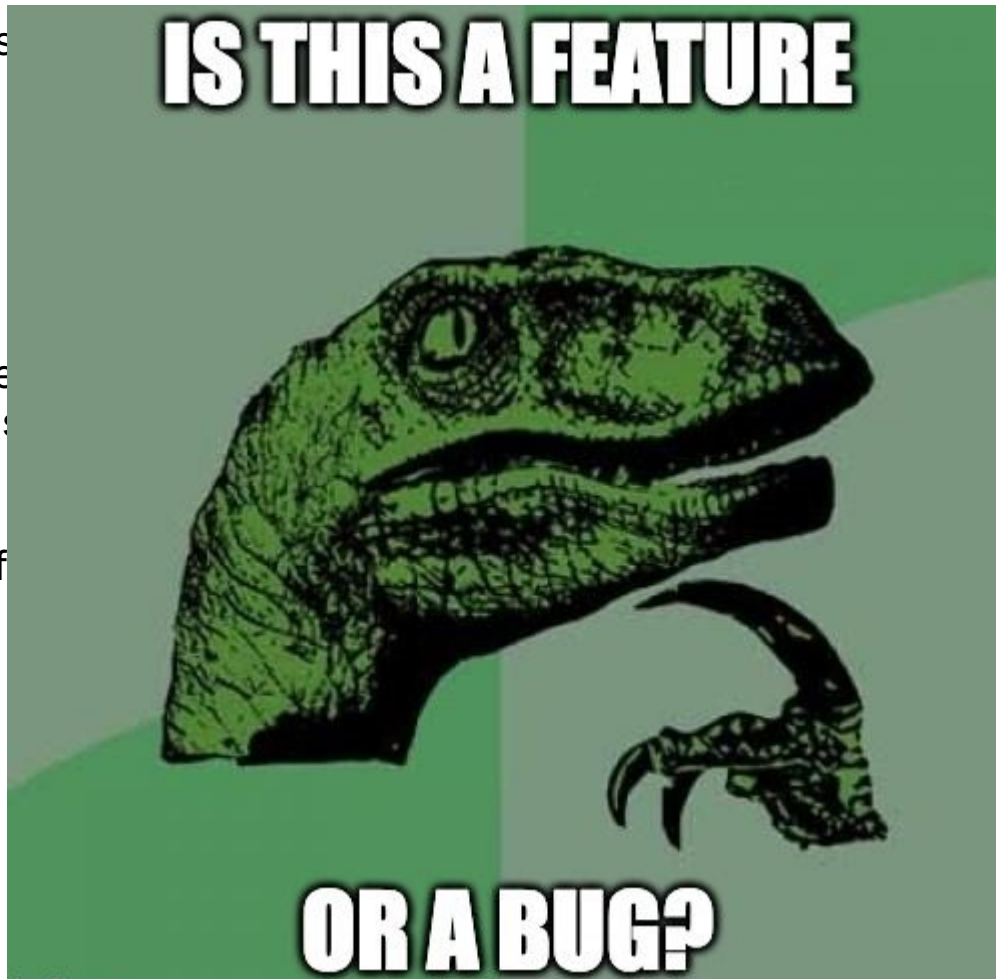
This does not differ from
the original code



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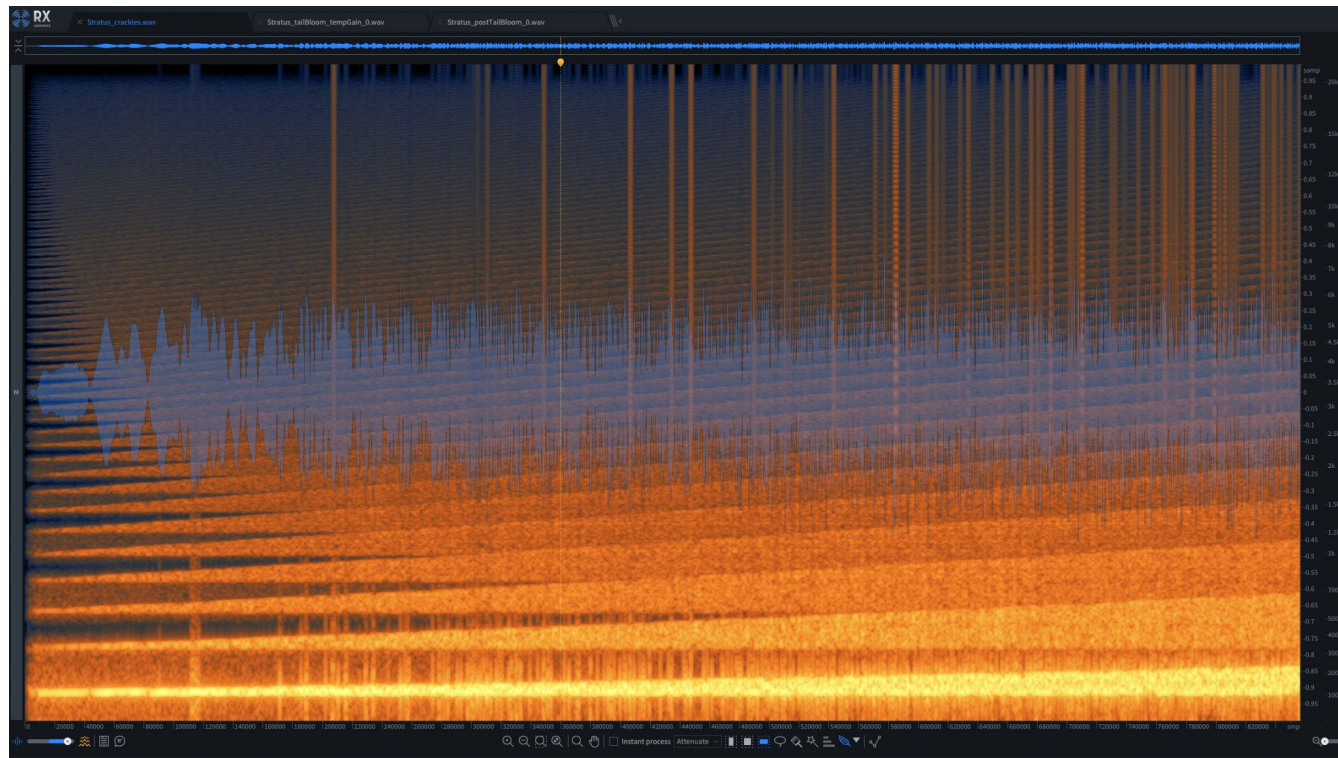


Solution, std::max



For this TailBloom class,
we clamped gains. The
high energy remains in
places

...as a feature, not a bug



April 2025 - Equinox Launches



Conclusion - We Built a Reverb Plugin!

Between August 2024 - April 2025, a team of 6 developers and 3 QAs:

- Ported Legacy Reverb Engines and implemented a seamless UI for switching between the two.
- Created a new preset system for thousands of presets.
- Extended Multi-Channel Config Support (7.x.2, 7.x.6, 9.x.4, etc.).
- Included EQ & Adaptive Unmasking tech.
- Numerous other improvements (Seeded Deterministic Randomness, Stabilised Exploding Filters, Legacy Session Conversion, etc.).
- Retired Stratus & Symphony, keeping the DSP alive in a future proofed DSP repository



Thank you!

Thanks to my Talk Collaborators:

Roth Michaels

Alex Fink

Jefferson Hobbs

All of the ART Team & Audio Production Team! 

Emma Fitzmaurice 

Harriet Drury

harriet.drury@native-instruments.com