Best Practices for Open Sound Control

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Overview:

- What is Open Sound Control?
- What does OSC practice include?
- Definition of audio control data, examples.
- Temporal quality assurance.
- Transport layer considerations.
- Description strategies for control data.
- Programming for audio control.

What is Open Sound Control?

Section 1

Monday, May 3, 2010

What is OSC?

- Open Sound Control (OSC) is a content format for messaging among computers, sound synthesizers, and other multimedia devices that are optimized for modern networking technology.
 - Wikipedia.org

What is OSC?

- A collection of ideas and practice for realtime audio control. Based around a descriptive document of the format and code (the "OSC-Kit") published by Matt Wright at CNMAT circa 2002.
- Now, lots of diverse implementations in applications and embedded software
- OSC, which is pronounced "oh-ess-cee", or sometimes "osk", stands for Open Sound Control.
- Actually not going to be called "Open Show Control" as of April 1st 2010.

What is Open?

- No license requirements
- No patented algorithms
- No conformance certification
- No strict specification of requirements
- Lots of open source code available

What is not Open?

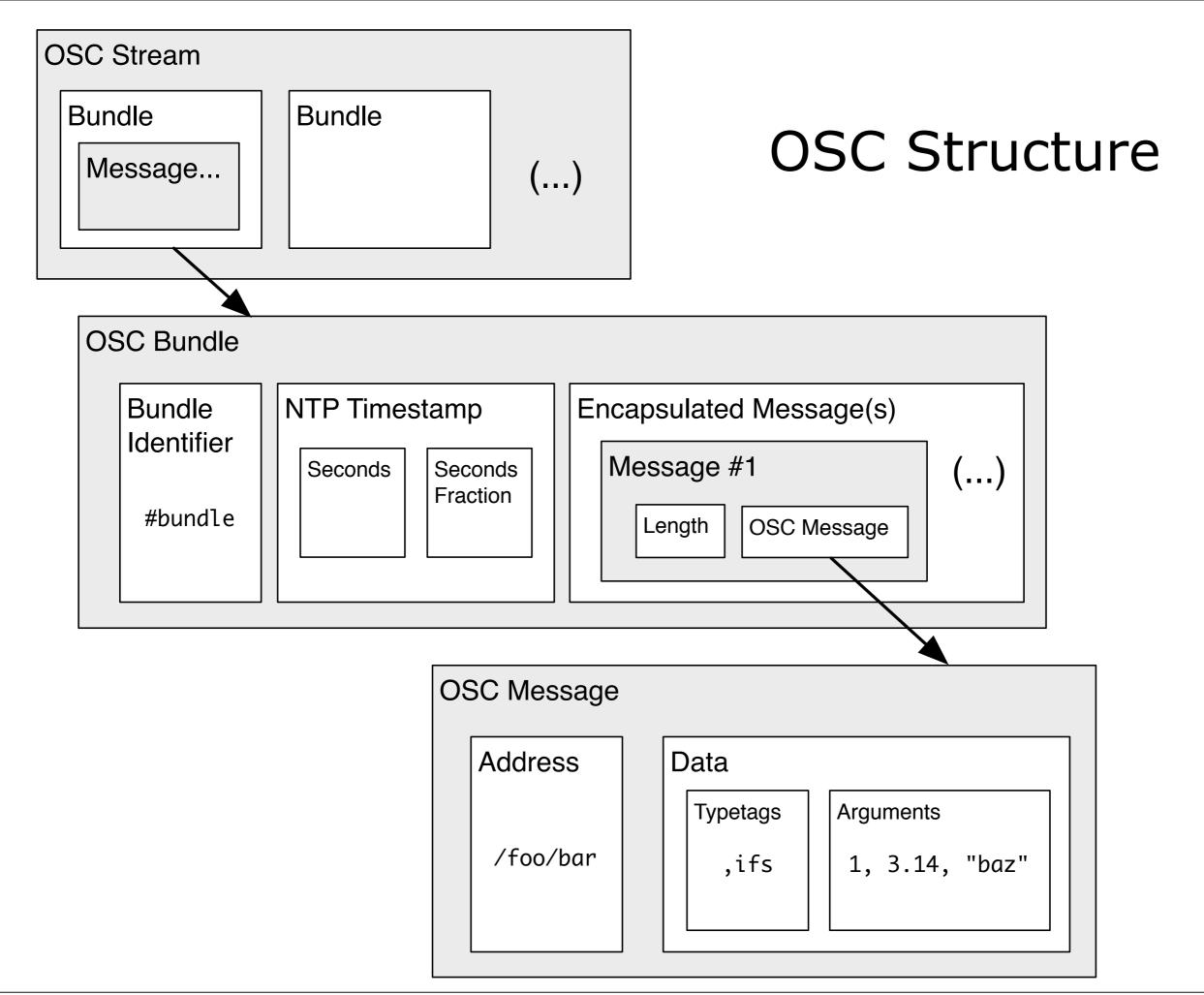
- Design: at the whim of its benevolent dictators
 - Acceptance criteria for a new idea:
 - Appropriate to the scope of OSC definition
 - Established need
- Can be used in closed-source products, provided the implementation has a compatible license.

OSC is a Content Format

- OSC is not a Standard:
 - no conformance certification.
- OSC is not a Protocol:
 - no convention for detection, negotiation, error handling
- OSC is a Content Format:
 - a content format is a structured container of *primitive* data types.

OSC Primitive Types

- strings (human readable)
- numbers: int32, IEEE 754 float single
 - optional types: int64, double, etc
- "blobs" (byte arrays)
- time
 - `t' typetag as NTP time,
 - pair of uint32 {seconds, seconds fraction}



OSC practice

OSI Layer	OSI Layer #	Торіс		
Application	7	Control Semantics, Choreography		
Presentation	6	OSC Structure		
Session	5	Discovery, Enumeration, Authentication		
Transport	4	Quality of Service		
Network	3	Bandwidth Reservation		
Frame	2	Clock Synchronization		
Hardware	1	Cabling, Wireless, Power		

Definition for Audio Control Data and Examples

Section 2

Audio Control Data

- Any time-based information related to an audio stream other than the audio component
 - Non-time-based audiorelated information are static stream properties
 - You can use OSC for this but its not "intended"

Properties of Audio Control

- Temporal errors can produce audible sideeffects,
 - "zipper" noise
 - spatialization aliasing errors
 - low quality interactivity (boring instruments)
- Variable sample rates, mixed rates
- Audio systems may have sensitive or highpower hardware components needing robust control

Examples

- Instrumental gesture data
- Spatial auditory scene parameters
- Spatial rendering engine control
- Audio synthesis engine control

Instrumental Gestures

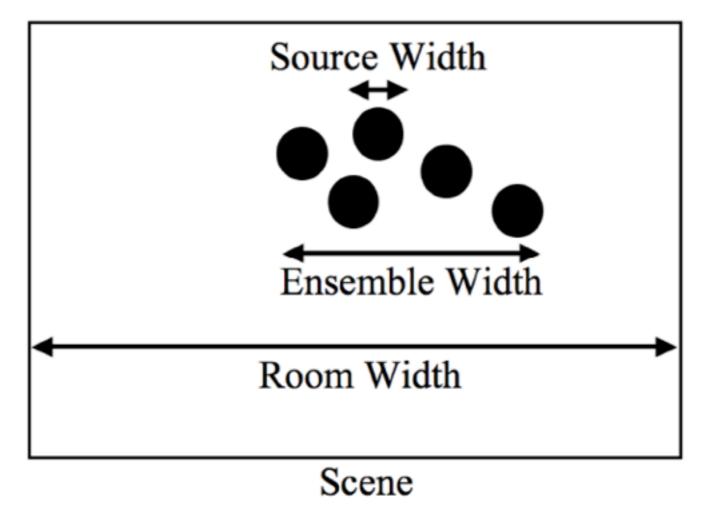
- Constrained by limits of mechanical and neural human body dynamics
- With training, gestures can be repeatable with very high precision in time and space
 - Delay Tolerance in performance: 20msec round trip delay (Chafe et al)
 - Temporal Repeatability: ~10hz continuous motion, 10msec accuracy, 1msec precision, 1000hz SR (Wessel)

Information Rate

$$I = \log_2\left(1 + \frac{\rho}{\sigma}\right) \text{ bits}$$

- Essentially, number of bits that change per second.
- Is the fundamental determinant in Fitt's Law
 - ISO-1941-9 (measurement of information transfer rate in target selection, mouse = ~3 bits/sec)
- Instrumental gestures are ~100bits/sec in time dimension alone. IR in space/force to be determined.

Spatial Scene Parameters



- Auditory Spatial Schemata (Gary Kendall)
 - Source location, width, directivity
 - Diffusion from enclosing geometry (rooms)
- Sub-audible frequency band (0-40hz)

Spatial Rendering Engine

- Examples including driving a distributed array with Ambisonic/Wave-Field-Synthesis filter coefficients:
 - Temporal error is equivalent to transducer positioning error
 - Temporal sync within 5% of sample frame at max controlled frequency
 - 500 microseconds (usec) at 96,000hz
 - AES2003-11 Best Practices for Network Audio

Audio Synthesis Engine

- Data-driven analysis and synthesis algorithms
 - Granular, concatenative, additive synthesis, large filter banks
- Can have very high bandwidth: thousands of entities per second
- Sub-sample accuracy (500 usec is good enough)
 - float32 is good enough for 500usec accuracy (but only just barely).

Temporal Quality Assurance

Section 3

Monday, May 3, 2010

Temporal Quality Assurance

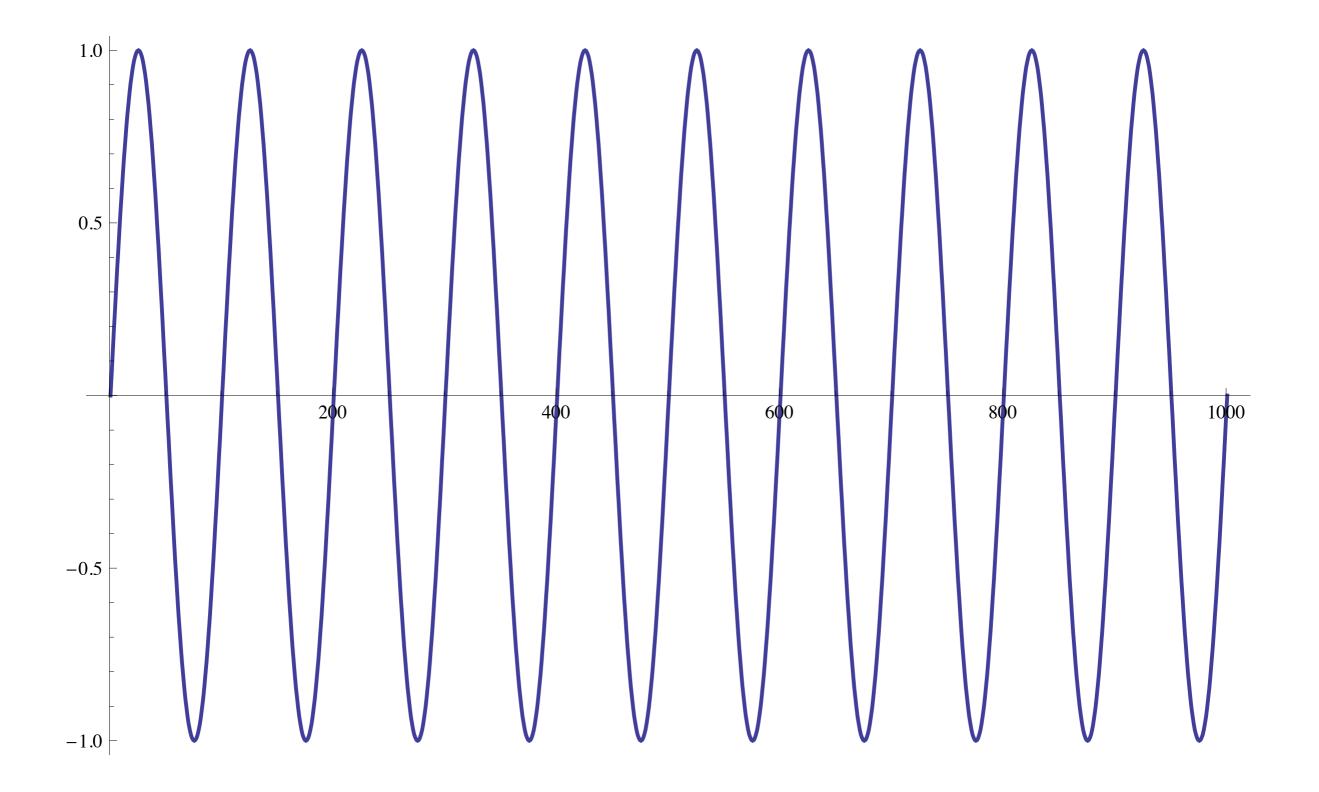
- Bounds on various delay properties:
 - maximum, minimum
 - variance.
 - accuracy and precision of scheduling

Interrupt Service Jitter

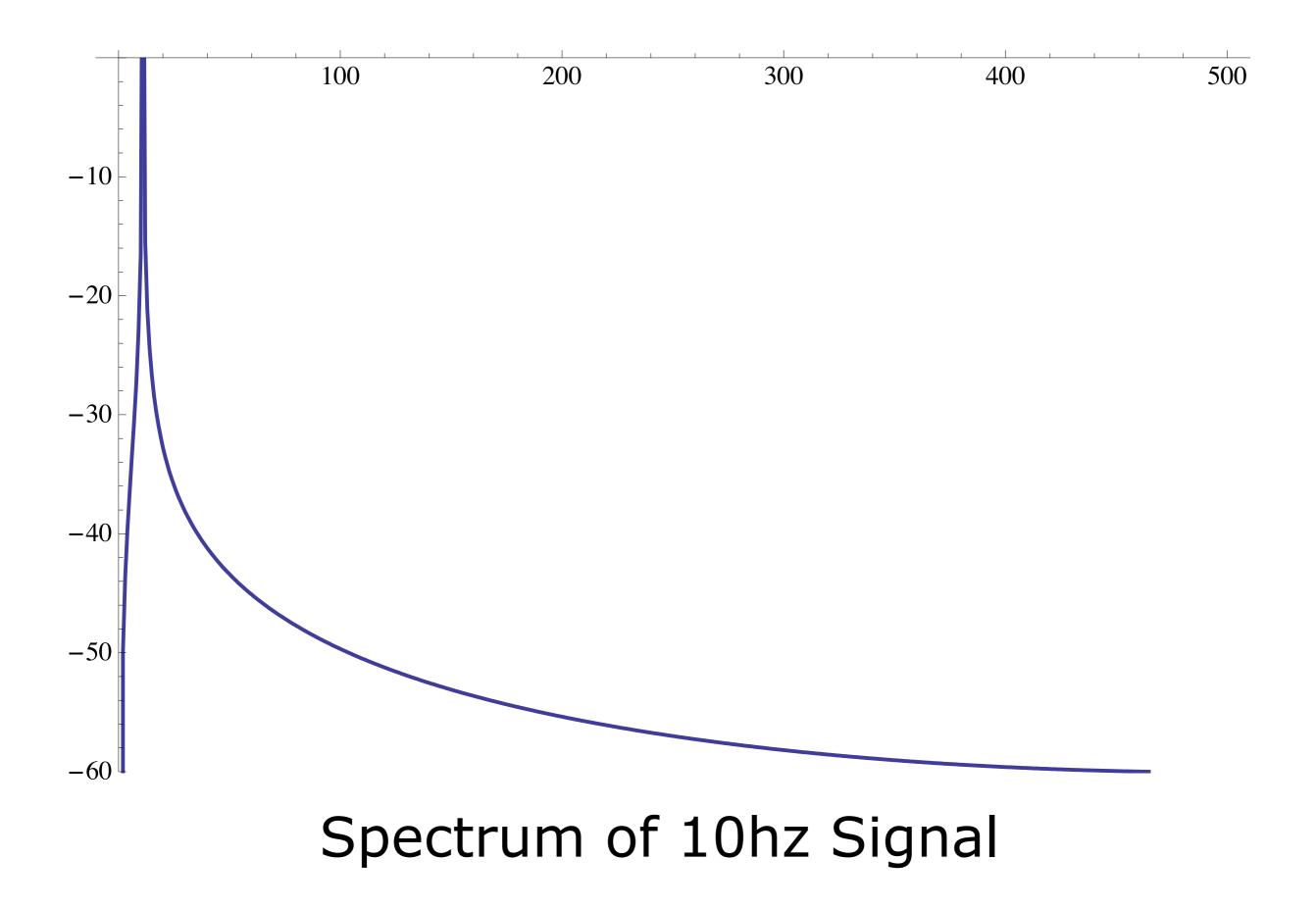
- Hardware or software gets some data and raises and interrupt service request.
- Data goes into a buffer until the interrupt is serviced by the system scheduler, then it gets delivered
- Interrupt servicing has delay distribution of a random wait-time queue

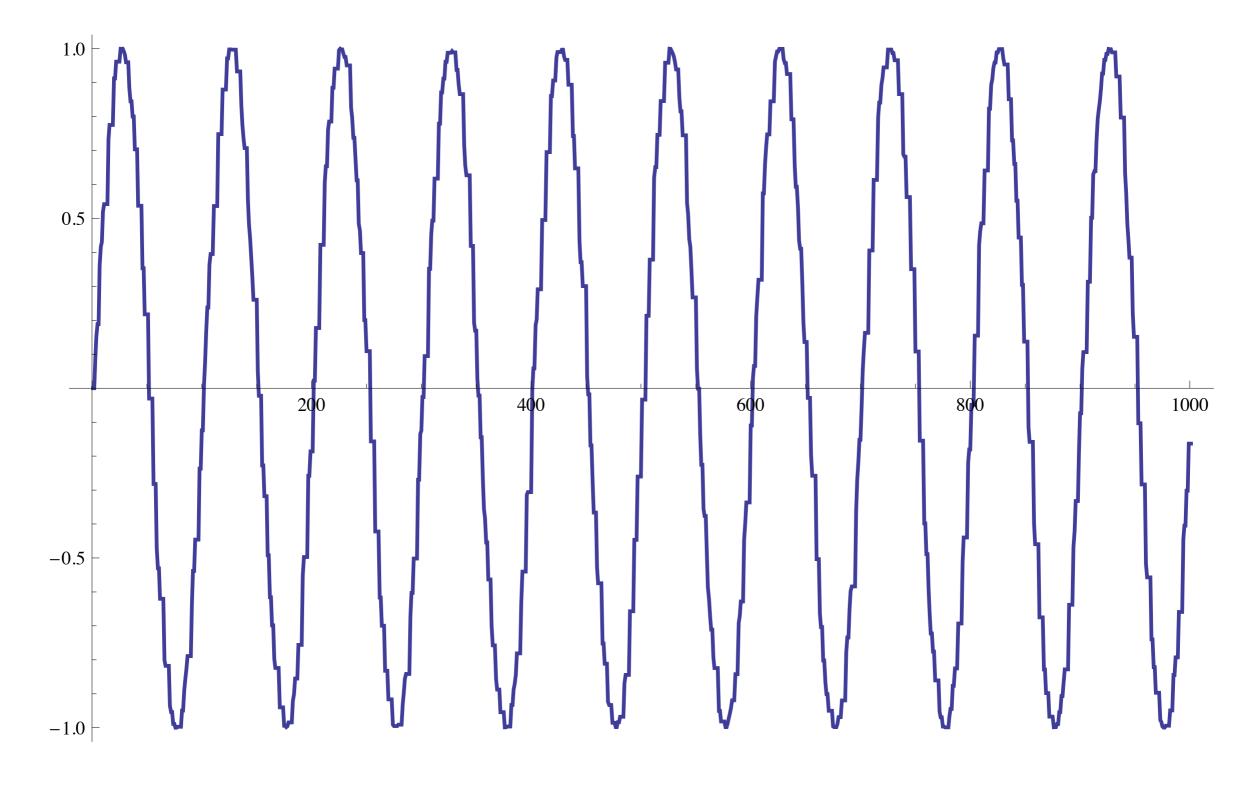
Random delay from buffered I/O 0.10 0.08 0.06 0.04 0.02 0.02 0.04 0.08 0.06 0.10

Typical ISR variable delay 3-10msec

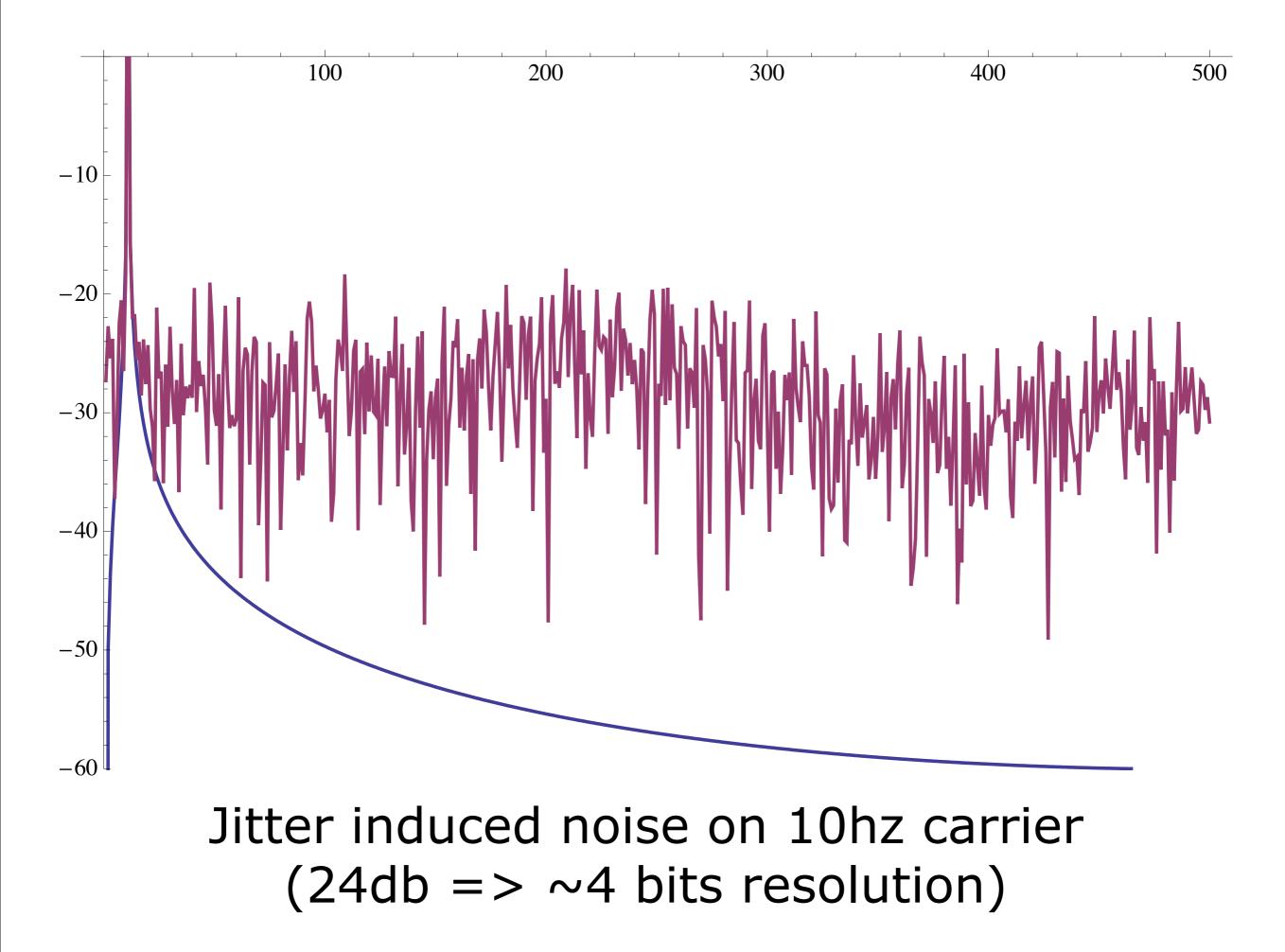


10hz Signal

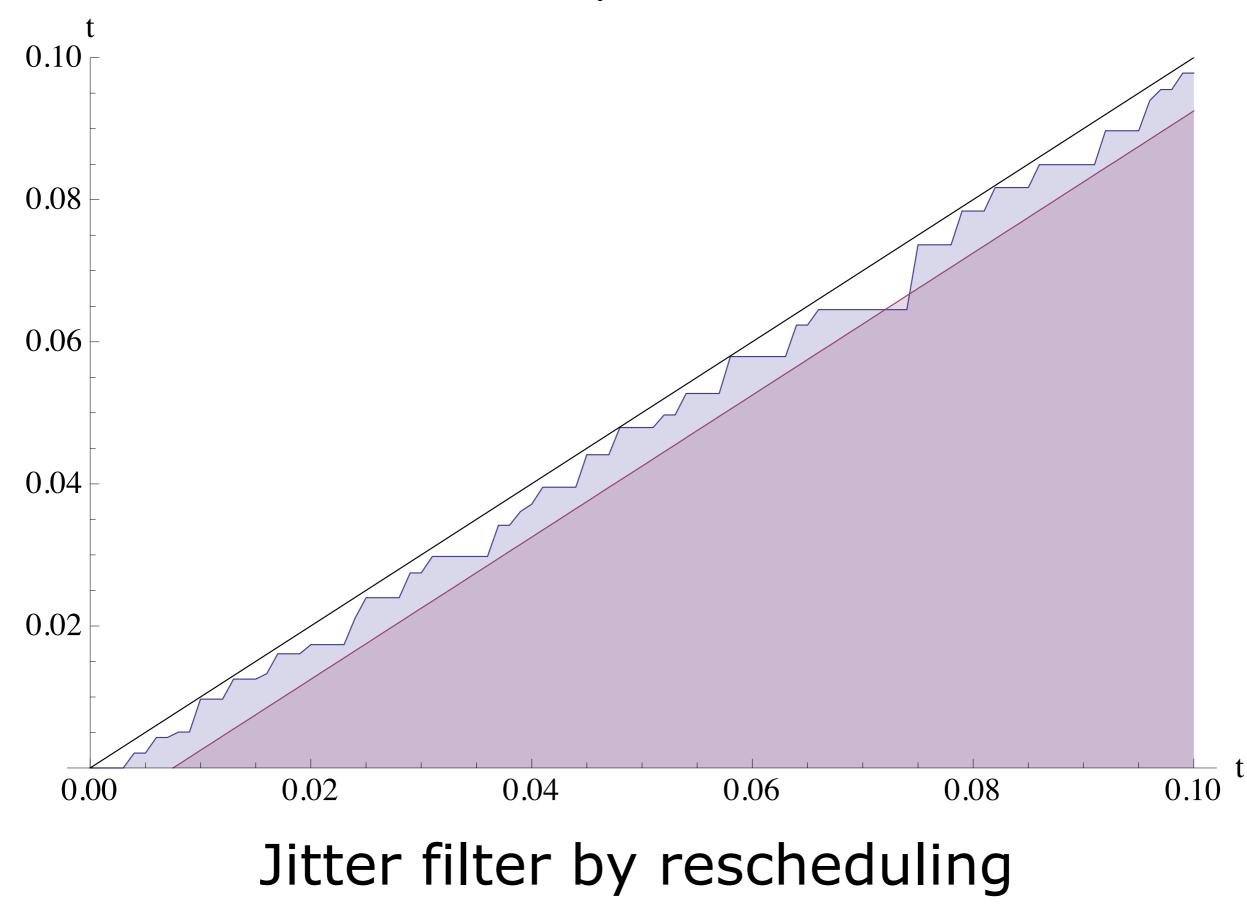


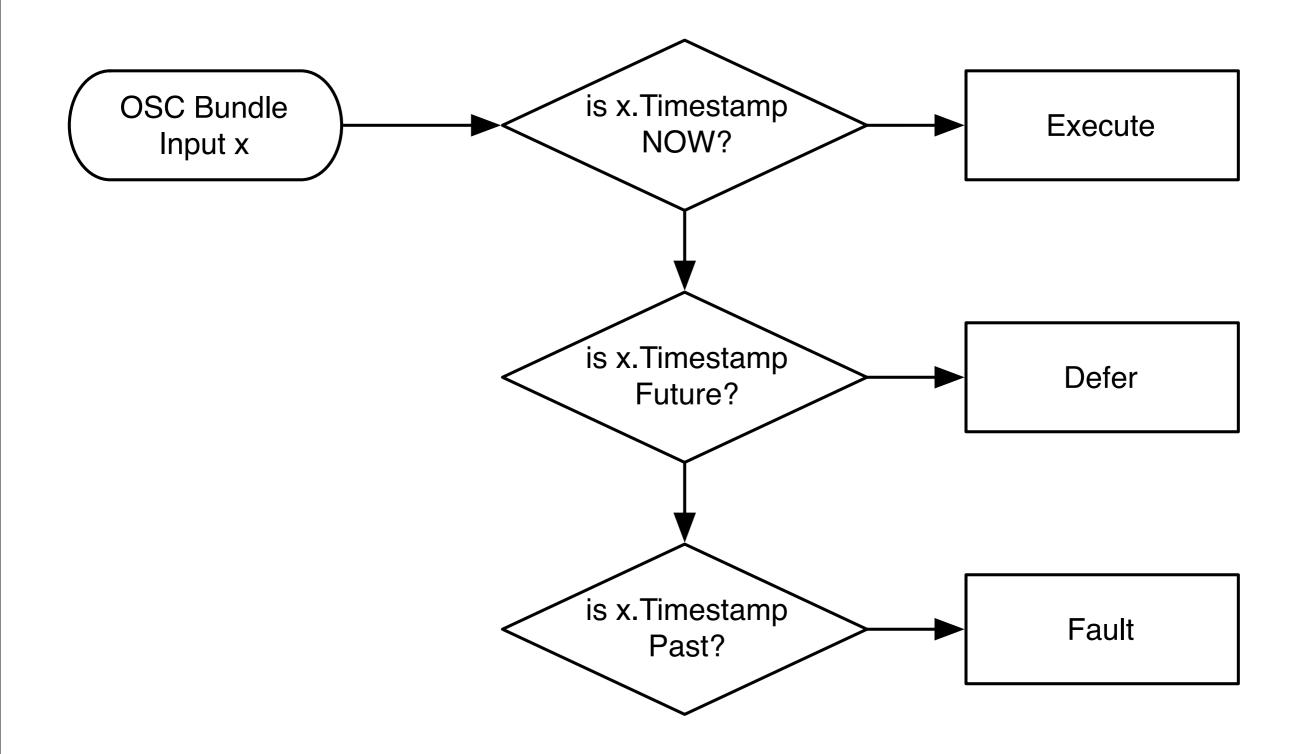


10hz Signal under 1-3msec variable delay



Random delay from buffered I/O





Forward Synchronization Scheduler (implementation is a priority queue)

	0.01 msec	0.1 msec	1.msec	2.msec	4.msec
0.5 Hz	100.806	80.942	60.5853	54.4588	48.2834
1.Hz	89.4672	69.2973	49.5129	<u>42.7719</u>	<u>37.1899</u>
2 Hz	83.5256	64.1865	<u>44.4936</u>	<u>37.811</u>	<u>32.166</u>
4 Hz	77.8606	58.3905	<u>38.2024</u>	<u>32.4498</u>	<u>25.4497</u>
8 Hz	72.3401	52.0053	<u>31.2989</u>	<u>25.7653</u>	<u>20.1786</u>
16 Hz	66.1133	<u>45.8497</u>	<u>25.8291</u>	<u>19.7408</u>	<u>14.3312</u>
32 Hz	60.2471	<u>39.6844</u>	<u>19.7202</u>	<u>13.546</u>	8.26448
64 Hz	53.9285	33.8882	<u>13.9203</u>	<u>7.90135</u>	<u>1.7457</u>

Carrier frequency vs jitter magnitude, **BOLD** => less than 8-bits headroom

Summary of Jitter

- For typical control frequencies in the sub-audible bandwidth 0-40hz, typical transport jitter of a few milliseconds is unacceptable
 - Best effort is not good enough.
- Forward sync scheduling can remove some jitter problems (maybe to 0.1 msec)
- For audio apps a better solution is to synchronize physical-time with sample-time
 - Using a DLL filter, interpolation strategies etc.

Transport Considerations

Section 4

Transport Topics

- Types of transports and their properties, e.g.:
 - UDP
 - TCP
 - Serial (USB, RS232, file pointers)

Ethernet AVB

- A solution for the endpoint-discovery and connection management problem (Bonjour/mDNS + AVBC)
- Uses bandwidth reservation protocols to ensure network availability with bounded delay (2msec, Class A)
- Solves clock synchronization problem at the ethernet frame layer, (500 usec per AES2003-11)
- OSC can be sent over AVB streams using a MIME type identifier (1722.1 working group)

Describing Audio Control Data

Section 5

Describing control data

- Four design patterns for describing a software interface:
 - RPC: Remote Procedure Call
 - REST: Representational State Transfer
 - OOP: Object Oriented Programming
 - RDF: Resource Description Framework

By Example...

Channel #3



Remote Procedure Call

/setgain (channel number = 3) (gain value = x)

Functional, reference-oriented semantics Good for highly-dynamic data structures (granular synth)

REST

REpresentational State Transfer

/channel/3/gain (x)

Emphasis on enumeration of resources Encourages stateless protocols

OOP

Object Oriented Programming

/channel/3@gain (x)

/channel/3/setgain (x)

"@" => attribute (after XPath)

RDF

Resource Description Framework

/channel,num,3/op,is,set/lvalue,is,gain (x)

Unordered set of semantic triples of {subject,predicate,object}

Programming Control Data

Section 5.3 - 5.4

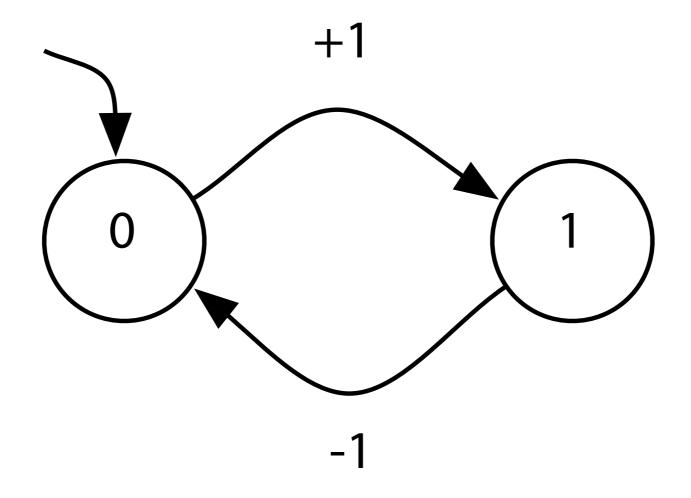
Stateless Interfaces

 A stateful protocol is one where the meaning of a message has some dependence on a previously transmitted message.

Example



Button State Machine



Stateful Encoding

/button +1

/button -1

/button +1

/button -1

Error Robustness

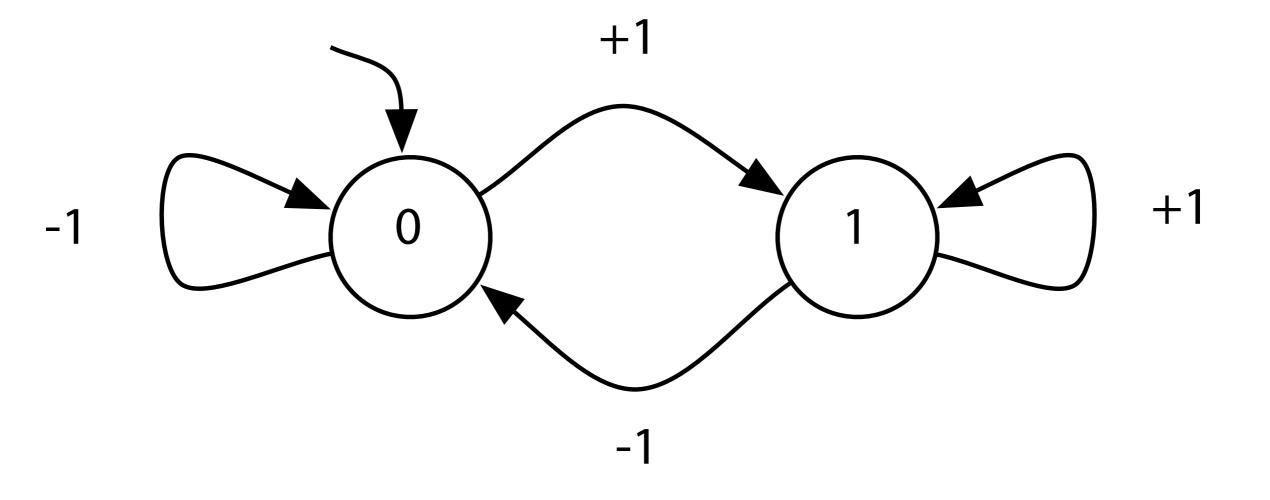
/button +1

/button +1

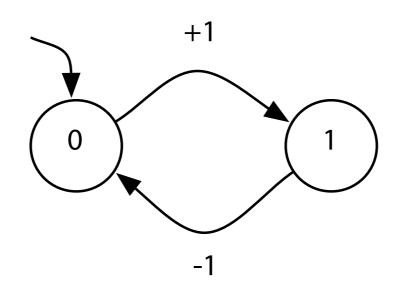
...?

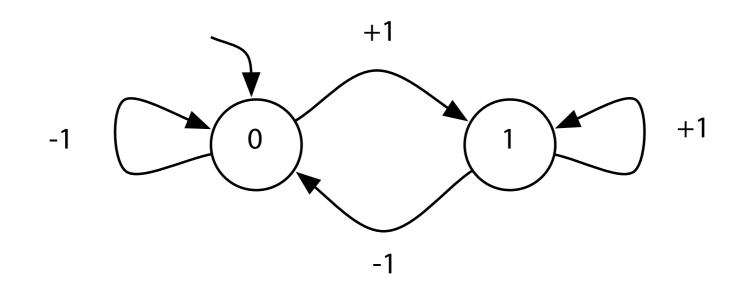
/button -1

Robust State Recovery



Parser Complexity is 2x!





Stateless Encoding (REST)

/button 0

/button 1

...?

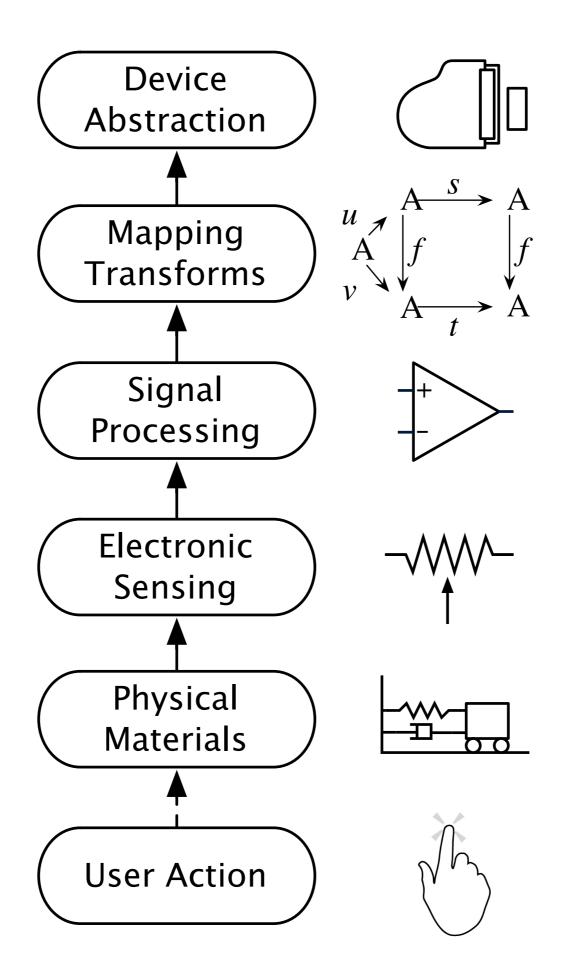
/button 0

Stateless Summary

- Stateful protocols are an optimization that reduces protocol bandwidth at the expense of protocol implementation complexity,
 - Especially when error recovery is involved
 - Otherwise, use TCP to ensure no errors (pushes complexity down to transport layer)
- Stateless interfaces can more readily support temporal constraints such as leases and expiration timestamps.

Abstraction Layering

- Effective strategy for management of complexity by encapsulation
- Can have some non-trivial complications...



Multi-Layer Operations

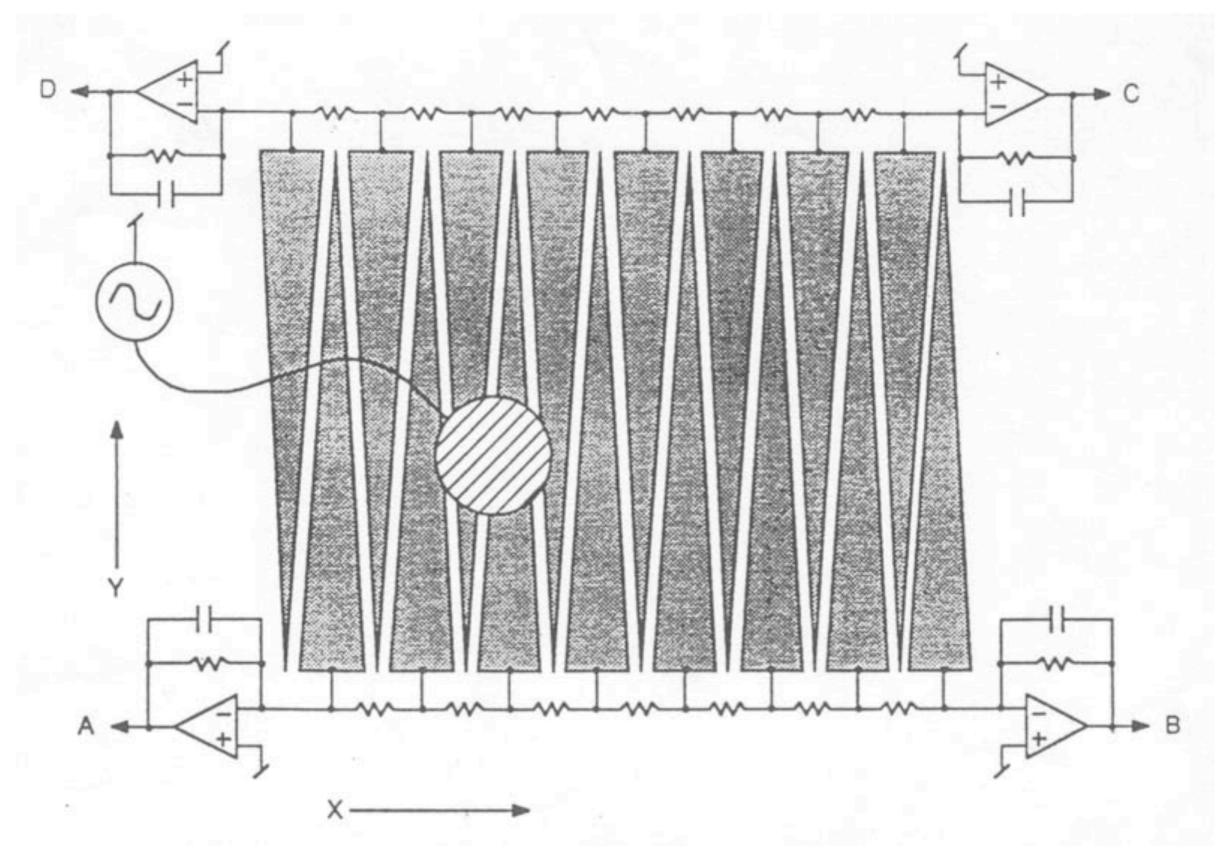
- Some transformation operations transcend the layer structure
 - Especially *mapping* transformations!

Example: Radio Drum



Schloss/Matthews/Boie

Radio Drum Sensor



Radio Drum Mapping

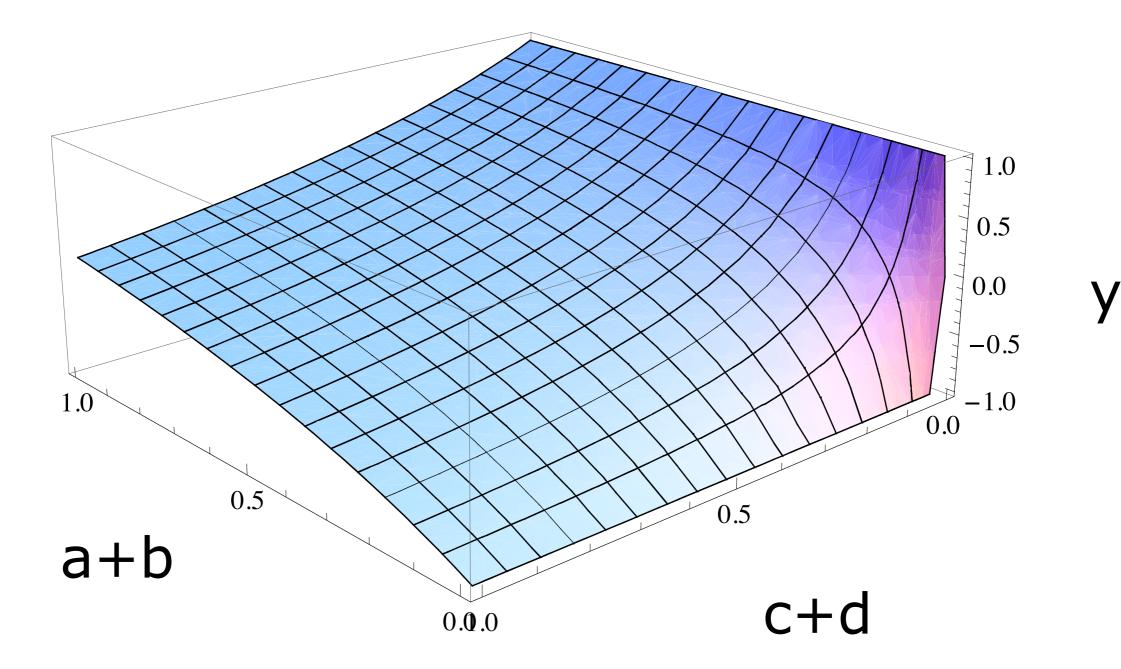
$$y = \frac{a+b-c-d}{a+b+c+d}$$

- Raw sensor to position

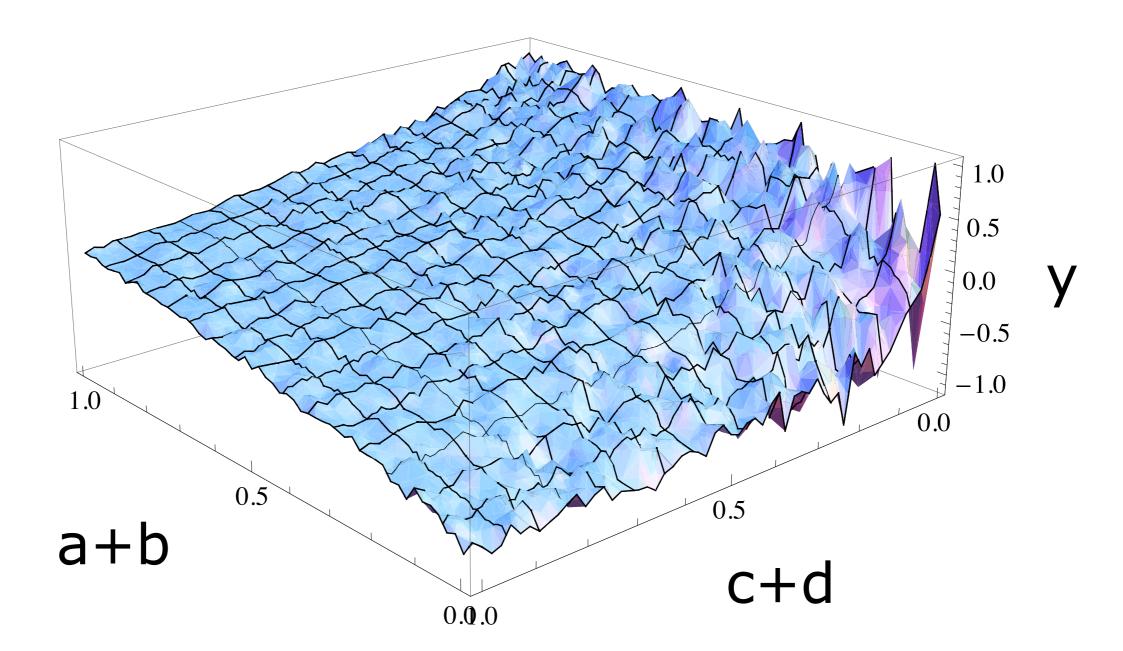
$$\|dy\| = \sqrt{\left| -\frac{a+b-c-d}{(a+b+c+d)^2} - \frac{1}{a+b+c+d} \right|^2} + \left| \frac{1}{a+b+c+d} - \frac{a+b-c-d}{(a+b+c+d)^2} \right|^2}$$

- Norm of derivative wrt. a + b, c + d

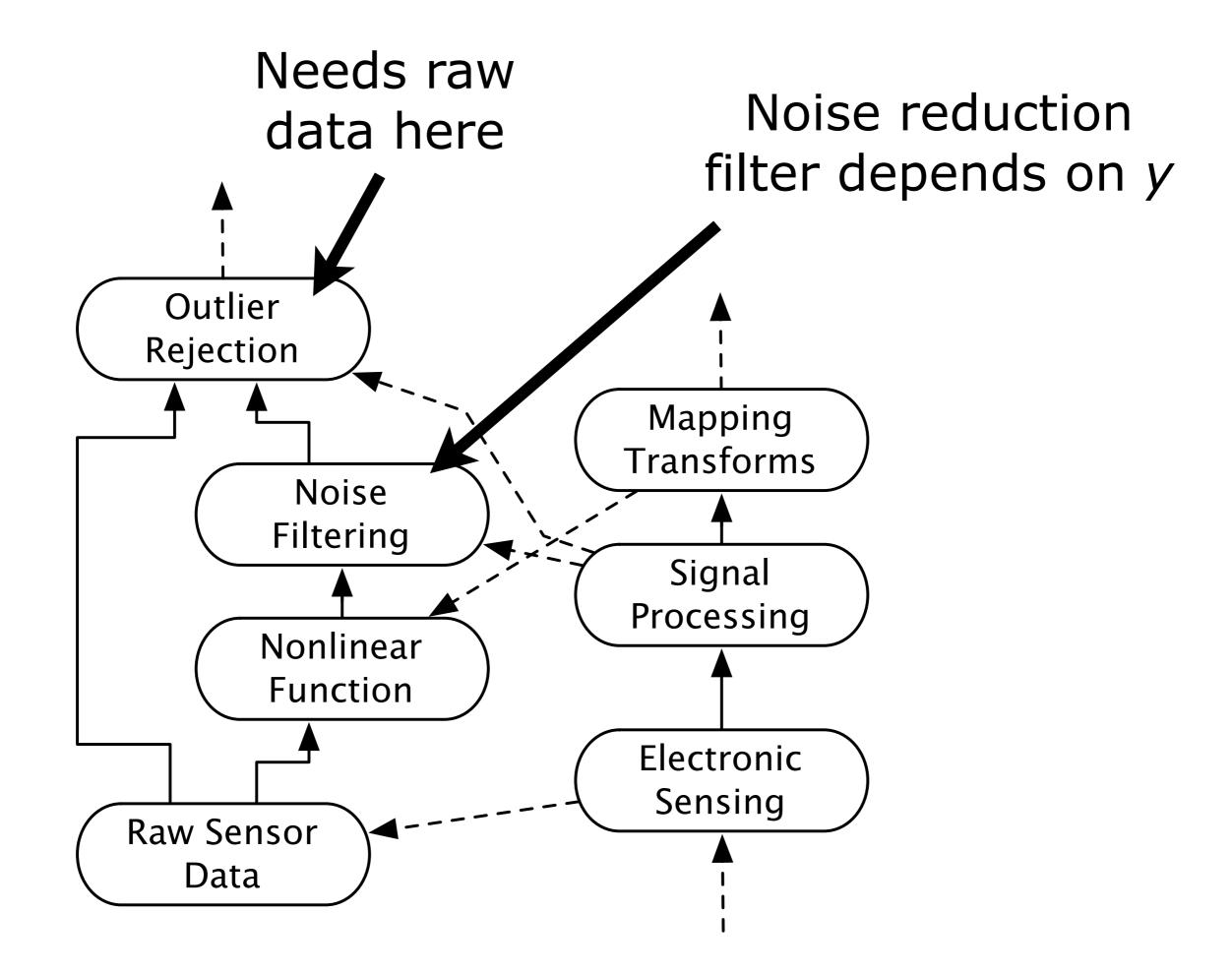
Ideal Mapping



Noise Amplification



- ||dy|| -> infinity as (a+b,c+d) -> (0,0)



Layering Summary

- Applications should maintain representations of control streams at multiple layers simultaneously when possible
 - This will support operations that need data from multiple layers.

Summary

- Clock synchronization tolerance depends on temporal control information and audio application needs (do the calculations, don't just ignore it).
- Ethernet AVB meets all the synchronization needs for audio control data, as well as bandwidth reservation. (500usec error, 2msec delay)
- There are multiple effective strategies for describing audio control interfaces (RCP, REST, OOP, RDF)
- Statefree protocols and multilayered representations can improve program reliability, enable temporal features, and increase flexibility.

The End

Your comments / feedback:

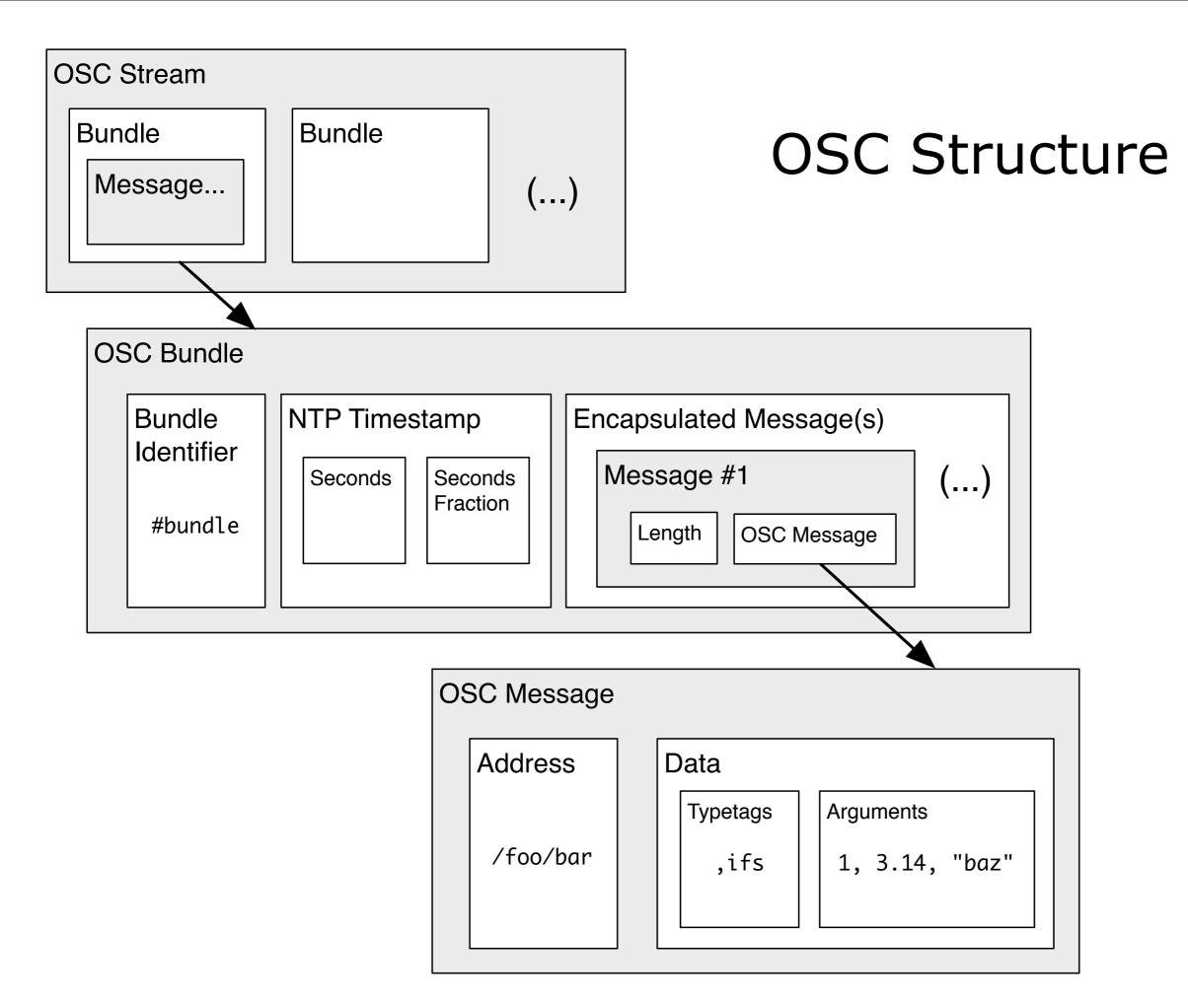
andy@cnmat.berkeley.edu

Appendix

(omitted slides)

Synchronization

- Suppose there is a set of changes to be applied all at once or not at all.
- Suppose there is a set of changes that should be committed by time T, after which the request is considered to be expired.
- In OSC we use Bundles to express frames of temporally-synchronized data.
- The quality of distributed synchronization is limited by the clock distribution error, which is a network layer or frame layer service.



Framing

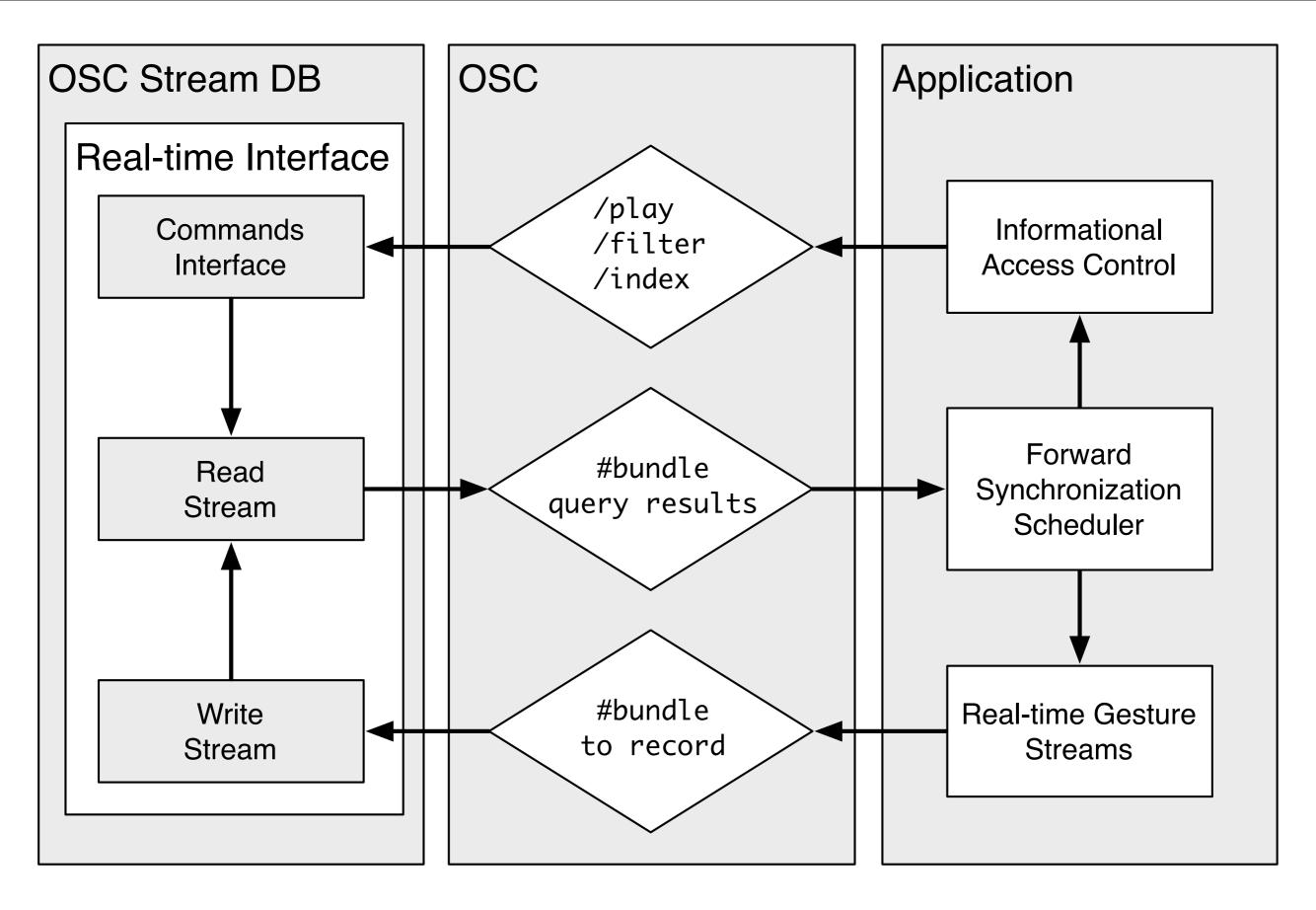
- OSC needs a transport that includes a framing structure (such as datagram messaging, UDP)
- Any serial transport can be adapted to support framing with a frame encoding:
 - SLIP RFC1055: a byte-quoted encoding that is robust to interruption (with doubleended variant)
 - int32 length preamble: requires an assured serial transport (TCP) (see OSC 1.0)

Routing

- Implementations should expose as much detail as possible from the network routing layer, so that applications can make full use of routing capabilities of the transport.
 - e.g., bidirectional UDP
 - Reverse NAT traversal (OSCgroups)

Bulk Transports

- File pointers and databases can be treated as classes of serial transport having a bulk-IO delay distribution model (OSC Stream DB).
 - On serial transports we use a SLIP RFC1055 encoding to provide a datagram framing around OSC



OSC Stream DB