

Audio in IP production

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Come and catch up on the Sony stand in Hall 13



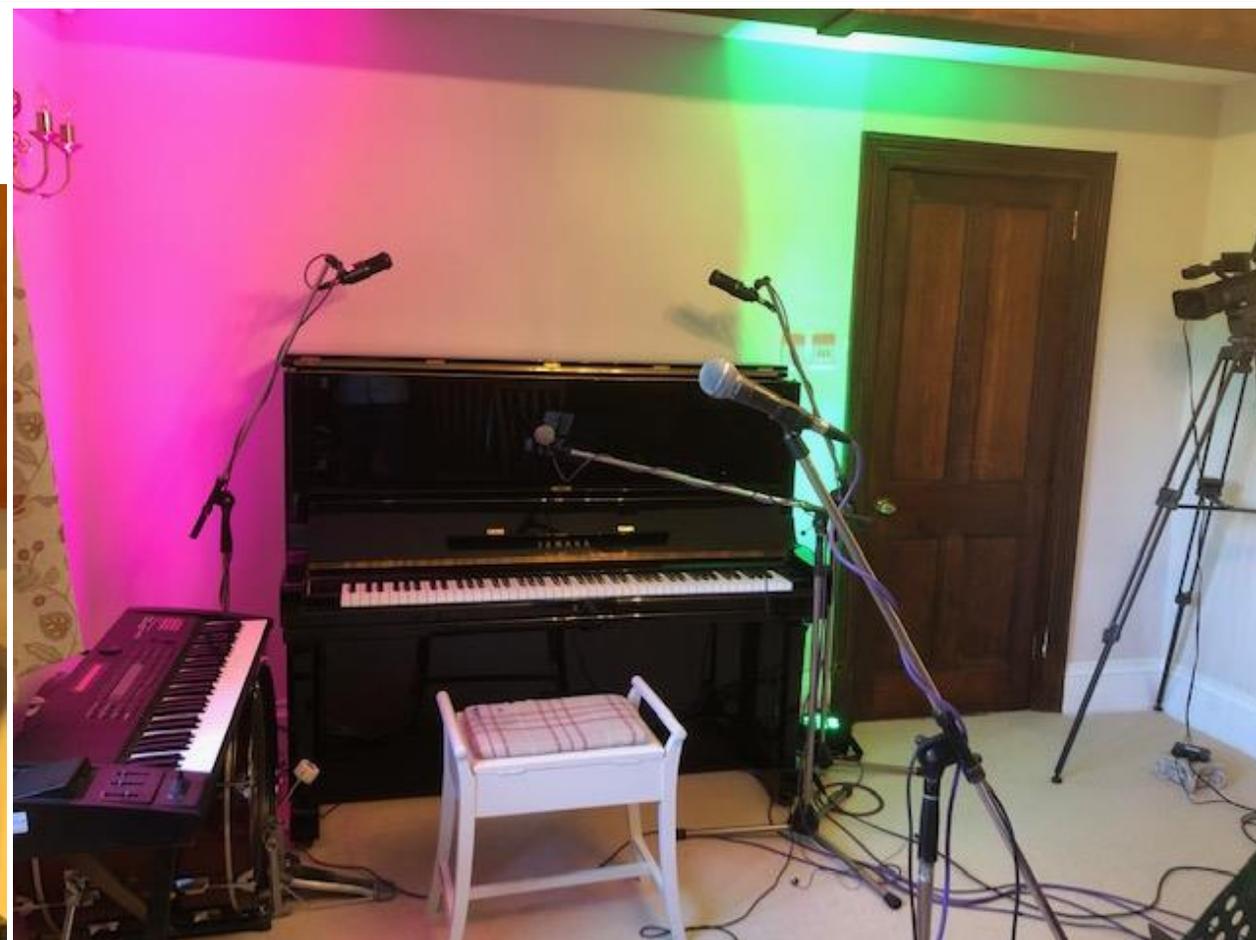
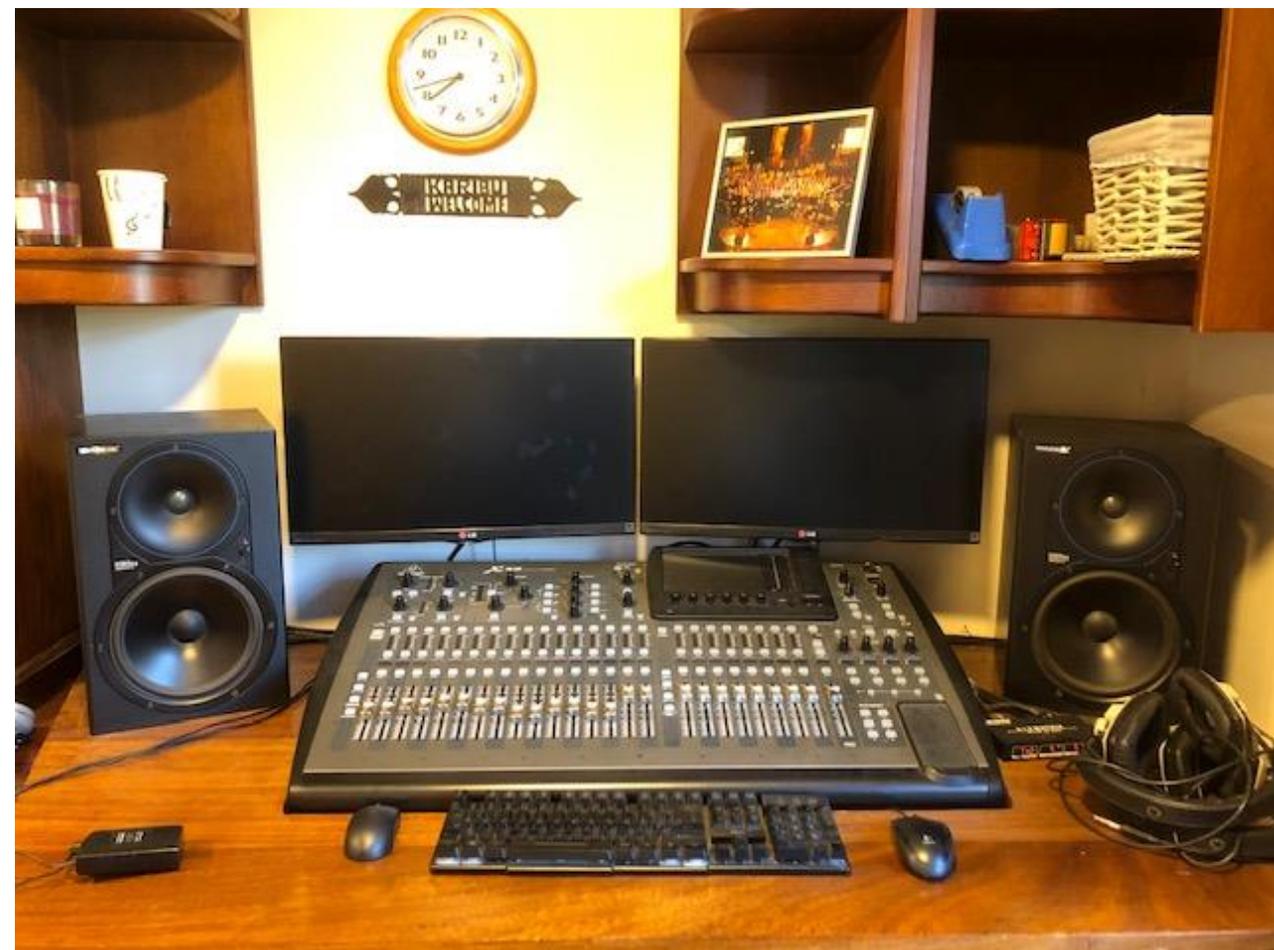
Queen Elizabeth II 1926-2022



The Queen
and I

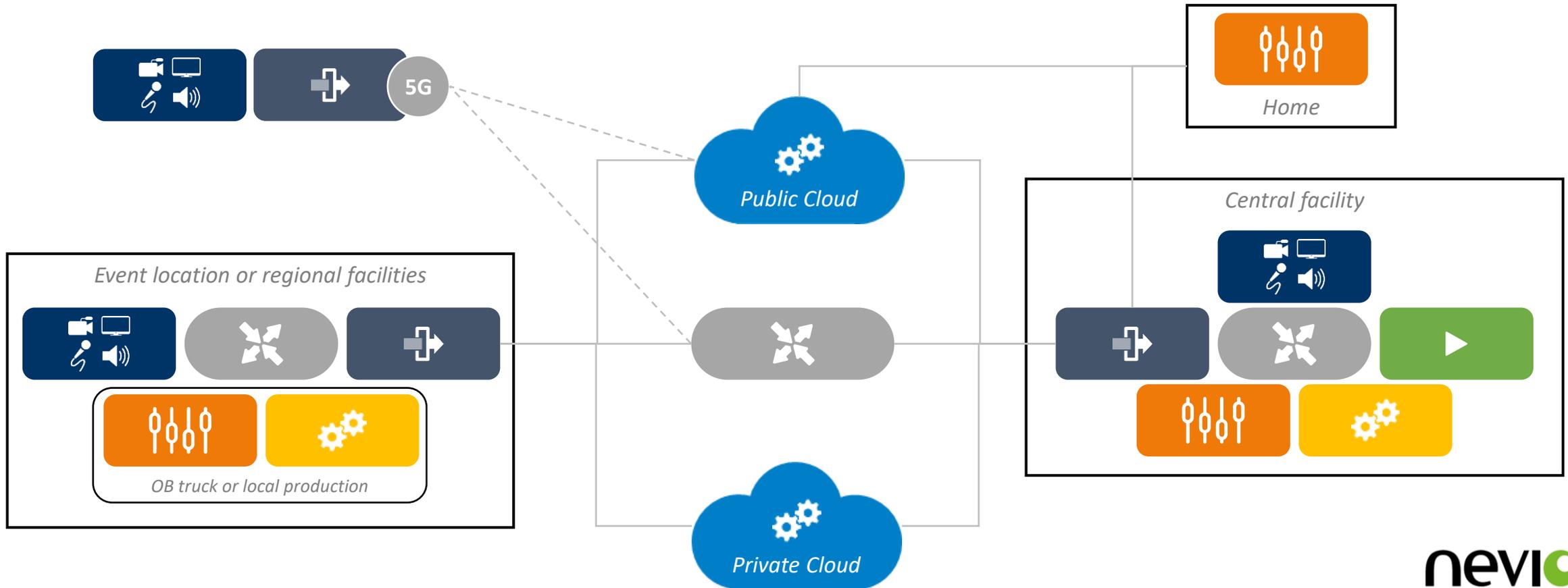


Don't forget the audio- I am an sound guy! 



Distributed production

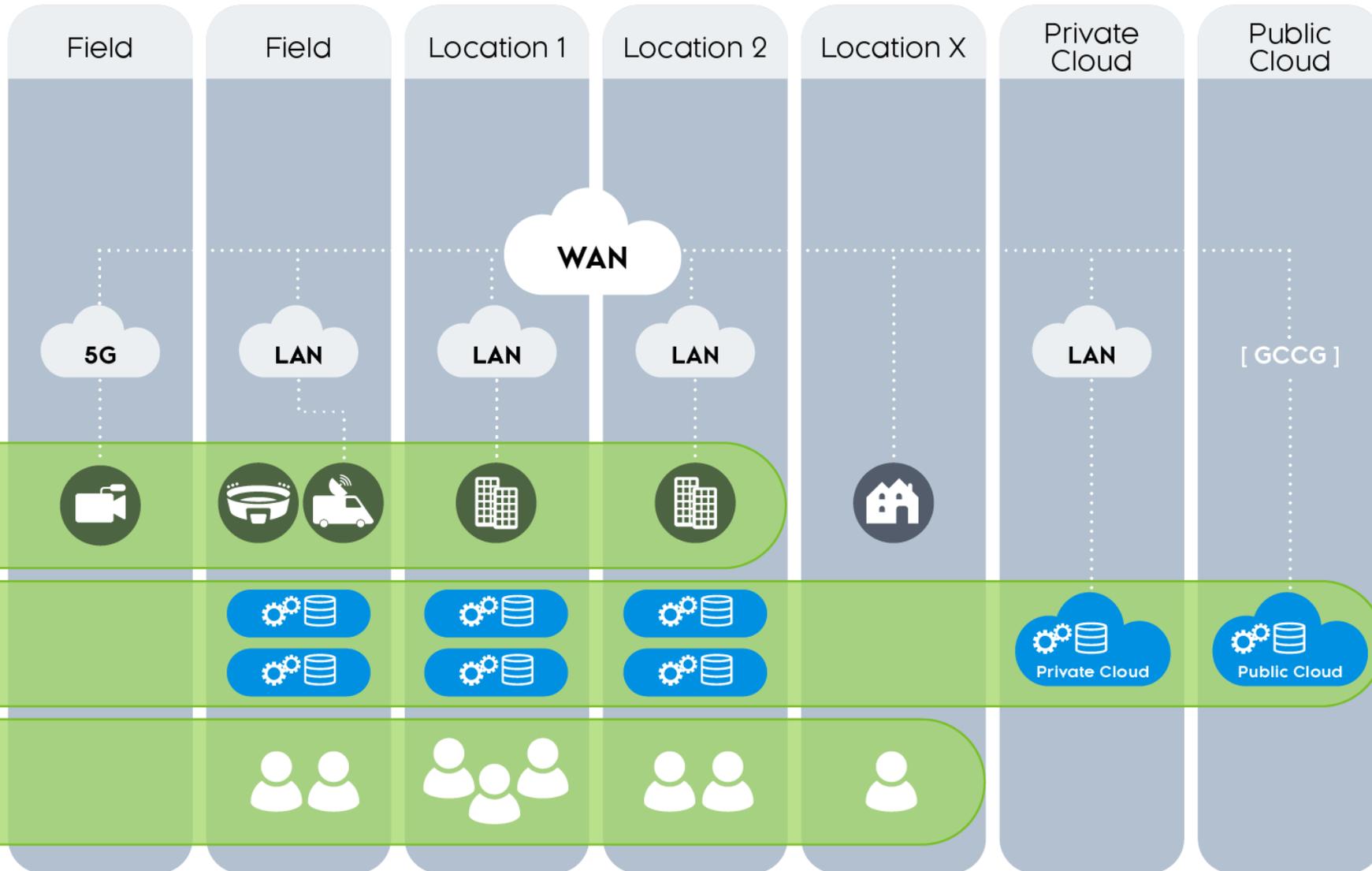
Management



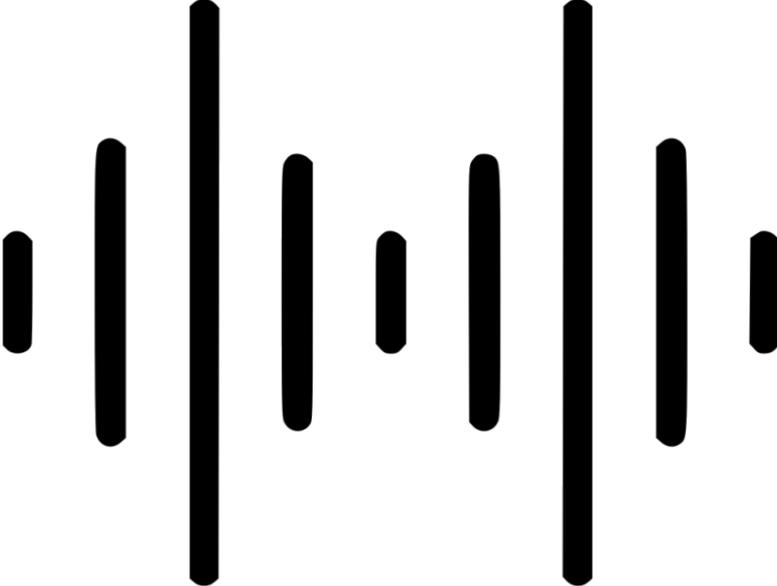
IP transformation - Places, Processing & People



The 3Ps



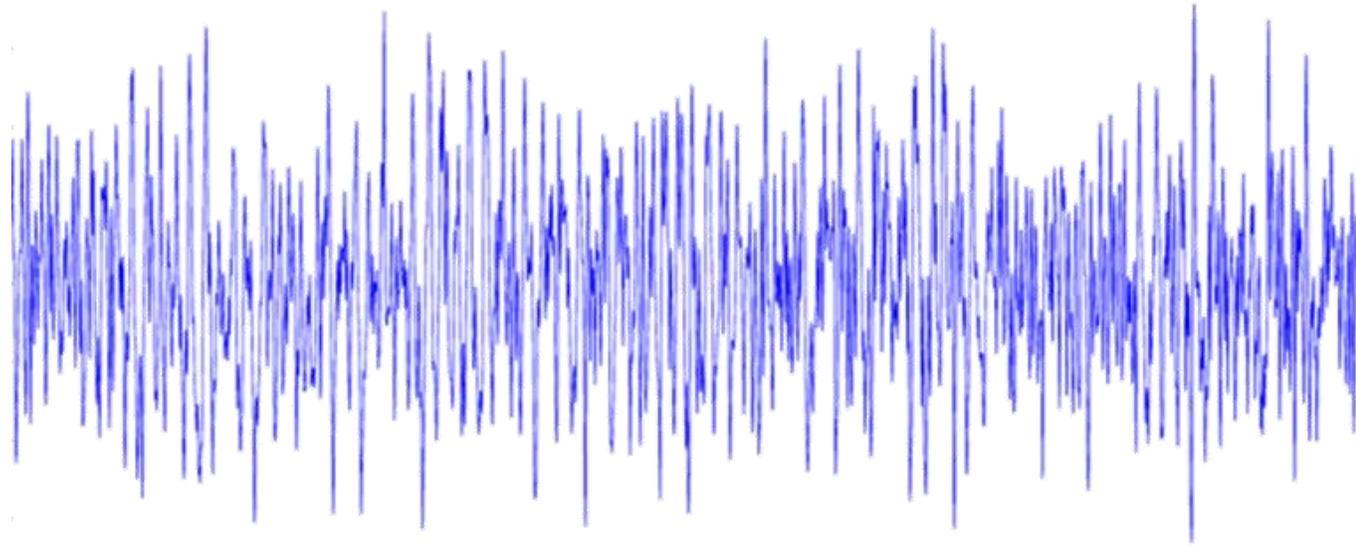
Audio requirements drive scale in facilities



The Audio folks did facility IP first!

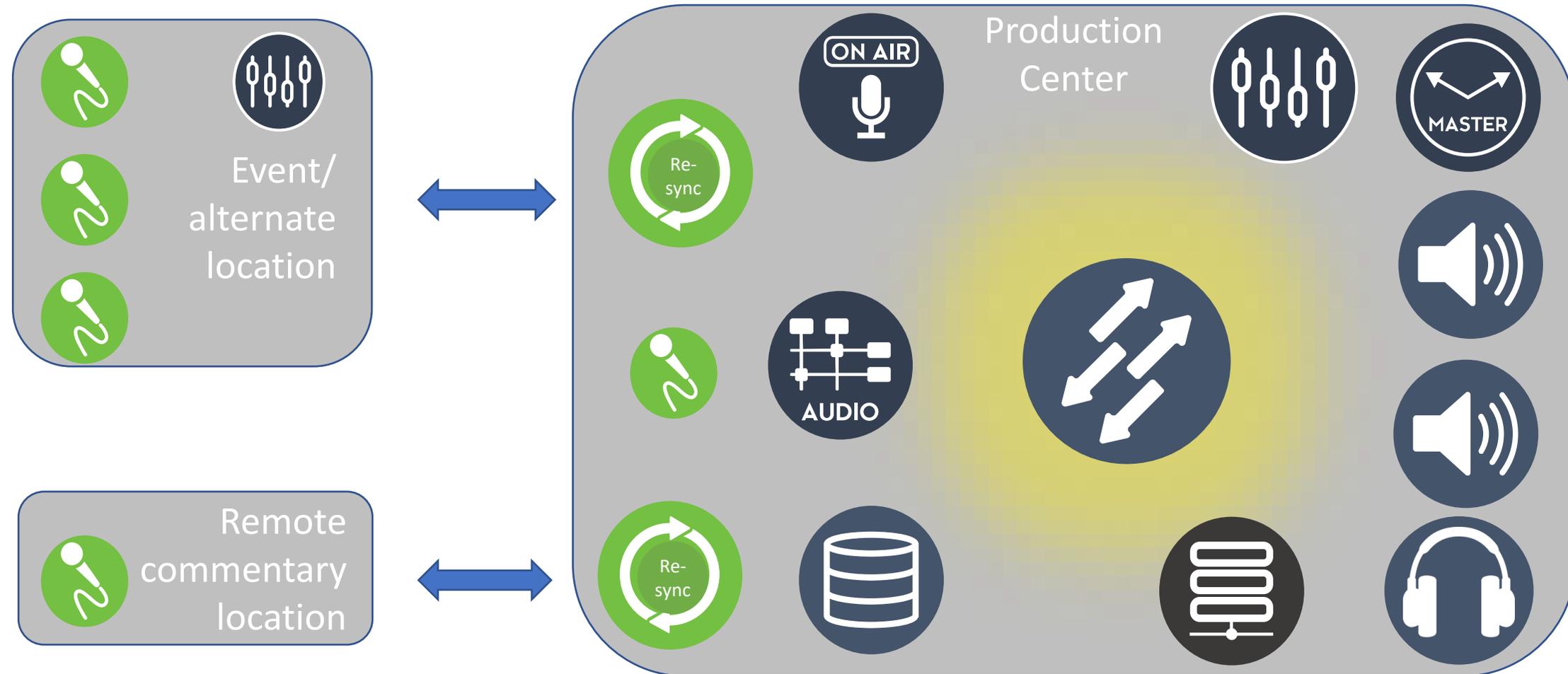


‘Most of the complexity of a production environment is the audio’

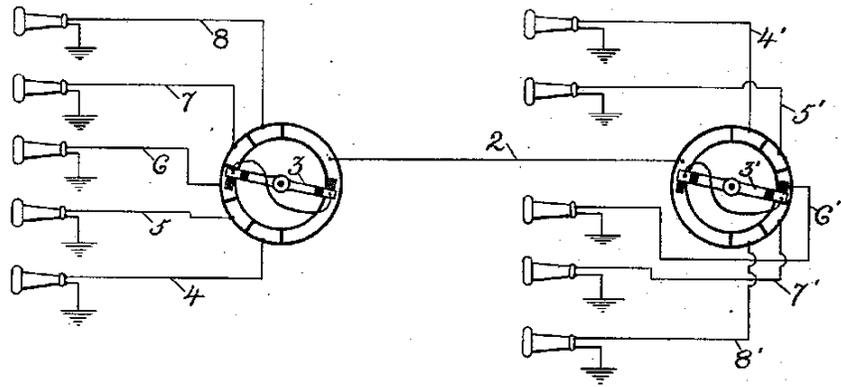


Dante 2006 Ravenna 2010

Audio facility interconnects



History of sampling - PCM is 100 years old!

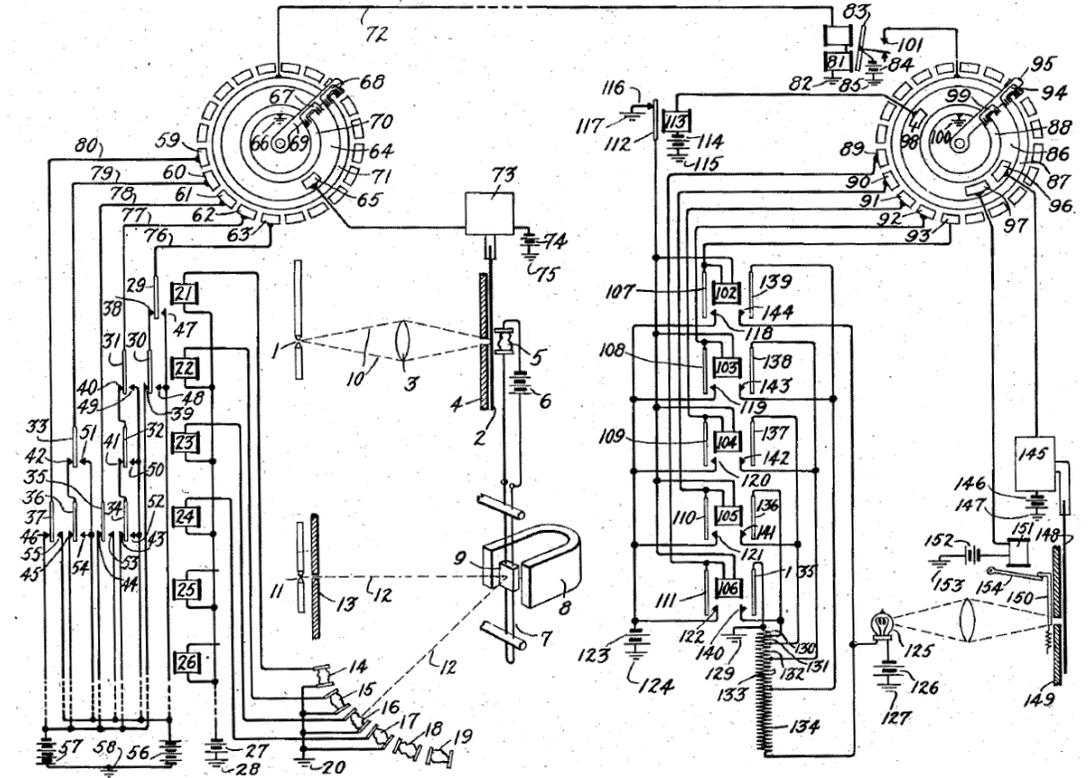


Determining sampling rate

Williard M. Miner, "Multiplex Telephony," U.S. Patent 745,734,

Filed February 26, 1903

Then Nyquist in 1924

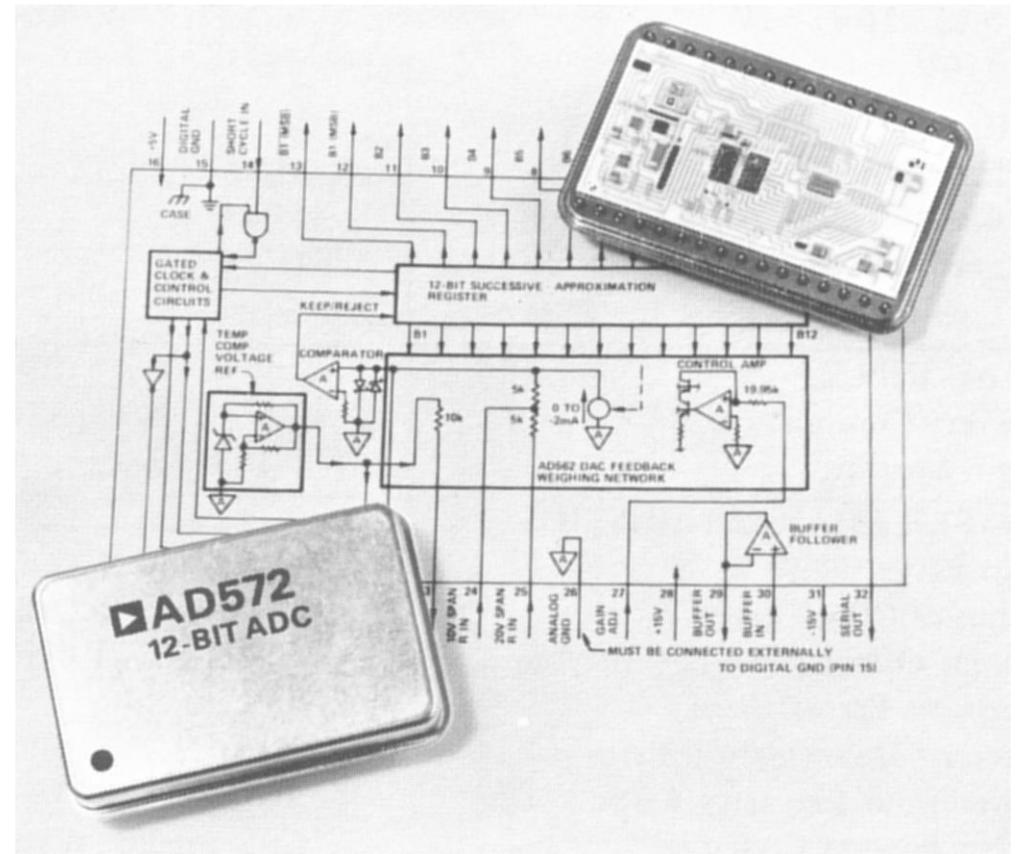


The First Disclosure of PCM: Paul M. Rainey,

"Facimile Telegraph System," U.S. Patent 1,608,527,

Filed July 20, 1921,

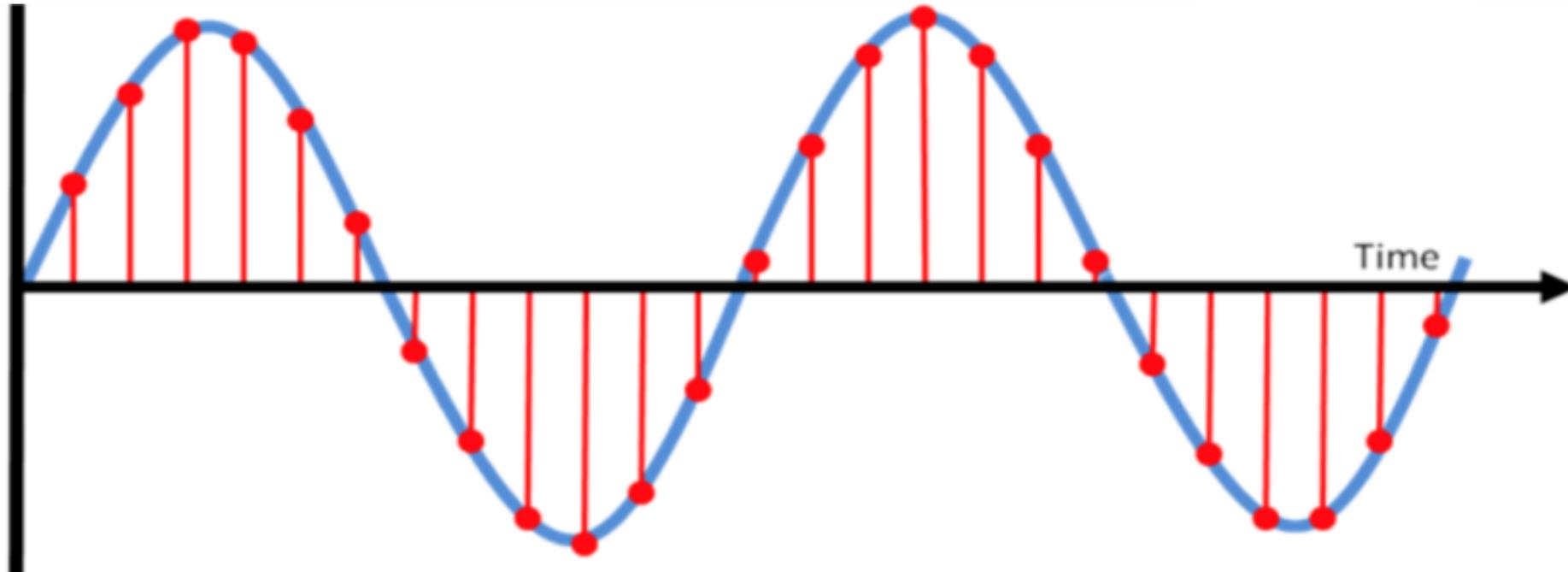
A-D conversion!



1954 "DATRAC" 11-bit, 50-kSPS Vacuum Tube ADC
Designed by Bernard M. Gordon at EPSCO
500watts, 150lbs, \$8500

AD572 12-Bit, 25- μ s Mil
Hybrid ADC, 1977

Audio acquisition



- Sample frequency
- Bit depth



Time of capture

Lots of different audio interfaces and formats!



Standardized Audio interfaces

- Analogue
- AES3
- SDI embedded
- IP 2022-6 embedded
- IP - 3326
- IP - AES67
- AES10 MADI
- AES50 AoE

Standardized IP formats

- EBU 3326
- AES67
- ST2110-30/31

Proprietary IP formats

- Livewire
- DANTE
- RAVENNA
- WheatNet
- Q-LAN

Audio production interoperability



Audio data flows – almost identical

Audio control plane – very different

Audio parameters.....lots of choices



AES3: 2 channels

Sample rate: 16kHz, 22kHz, 32kHz, 44.1kHz, 48kHz, 88.1kHz, 96kHz, 192kHz

Bit depth: 16bit, 20bit, 24bit

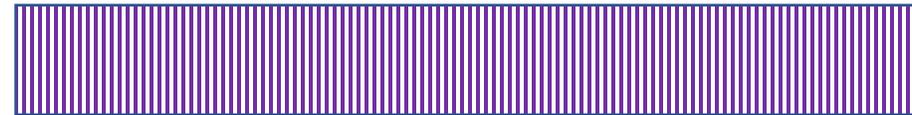
AES3 frame period: 1ms – 12ms



AES10 - MADI: 28, 56, 64 channels

Sample rate: 32kHz - 96kHz

Bit depth: 16bit, 20bit, 24bit

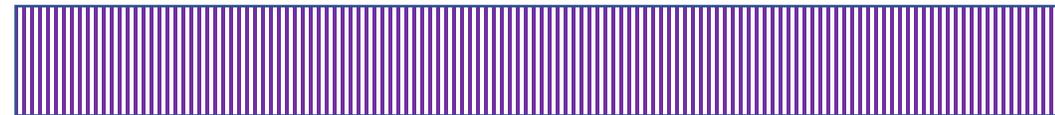


AES67: 1 - 120 channels (**1-8**)

Sample rate: 44.1kHz, **48kHz**, 96kHz

Bit depth: 16bit, 24bit

IP Packet period: 125us, 250us, 333⅓us, **1ms**, 4ms

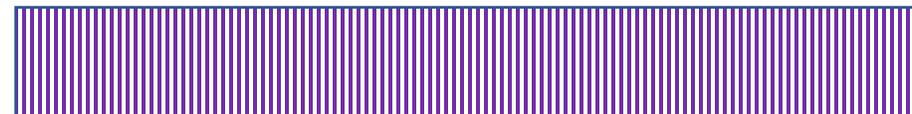


ST2110-30: **1-8** channels (1-64 channels Level C)

Sample rate: **48kHz (level A)**, 96kHz (level -x)

Bit depth: 16bit, 24bit

IP Packet period: **1ms (level A)** 125us (level B),



The ST2110 suite & NMOS



SYSTEM -10 	VIDEO -20 	AUDIO -30 	AES3-32 bit AUDIO -31 	ANCILLARY DATA -40 	TIMING -21
COMPRESSED VIDEO -22 	MULTI-PART VIDEO -23 	SD VIDEO -24 	FAST METADATA -41 	FMX -42 	2022-8 COMPOSITE
BCP-003-0x Security suite 	IS-04 Discovery and Registration	IS-05 Connection Management	IS-07 Event and Tally	IS-08 Audio Channel Mapping	IS-09 System



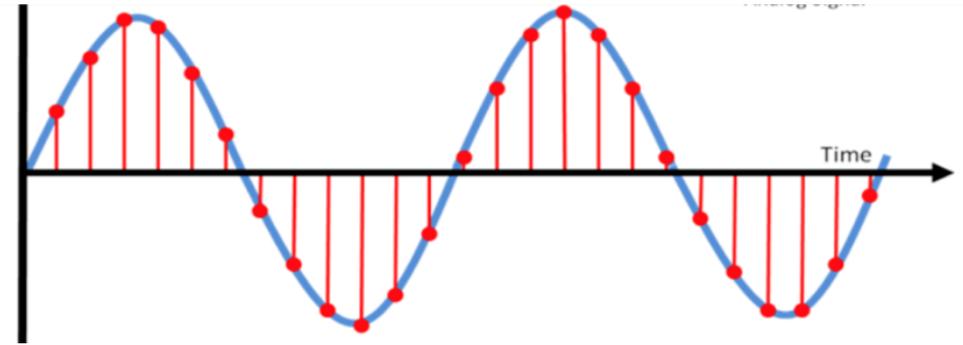
Handling 'other' audio: the ST2110-31 bits



Timing and referencing



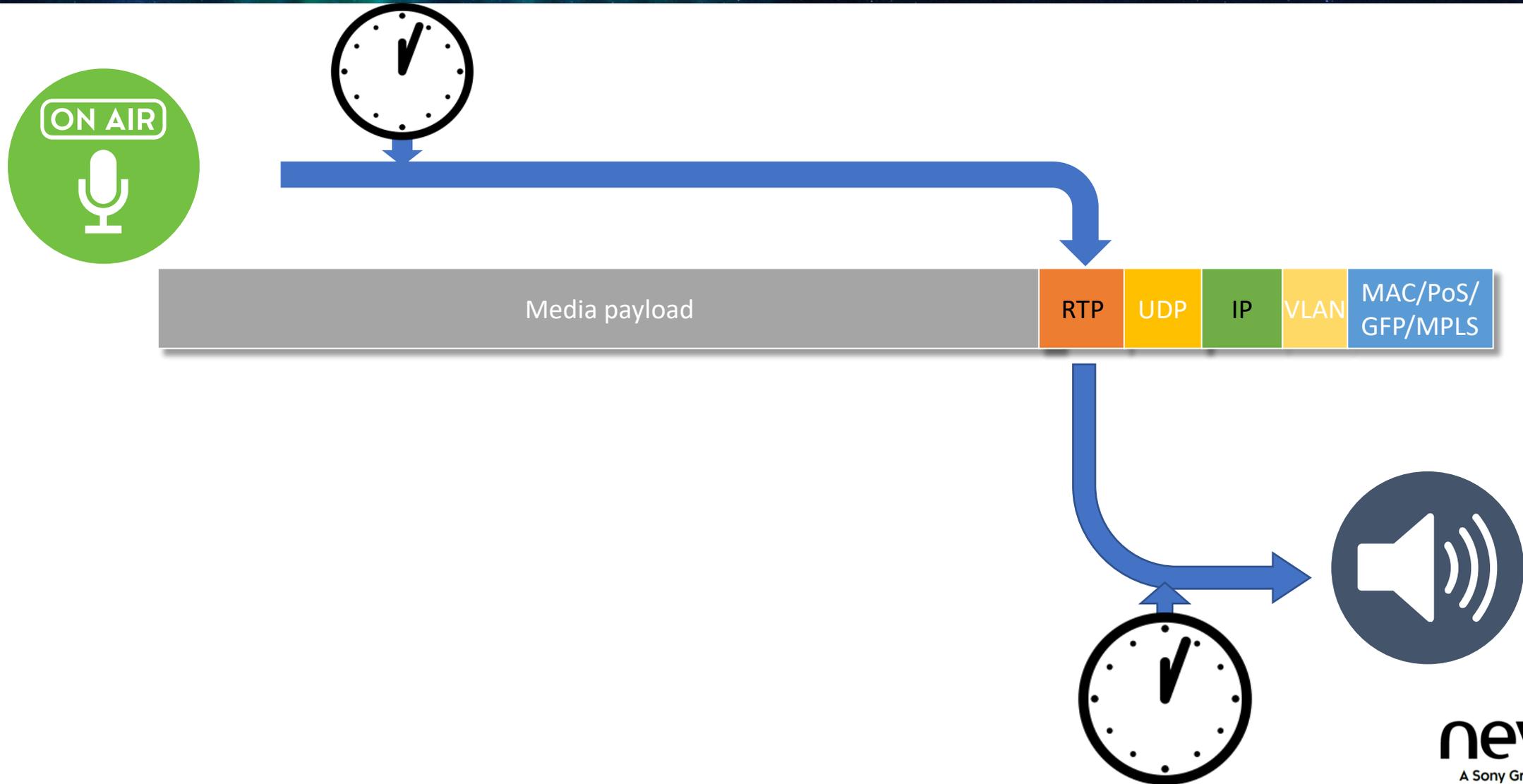
ST2059 – applying PTP IEEE1588 to media



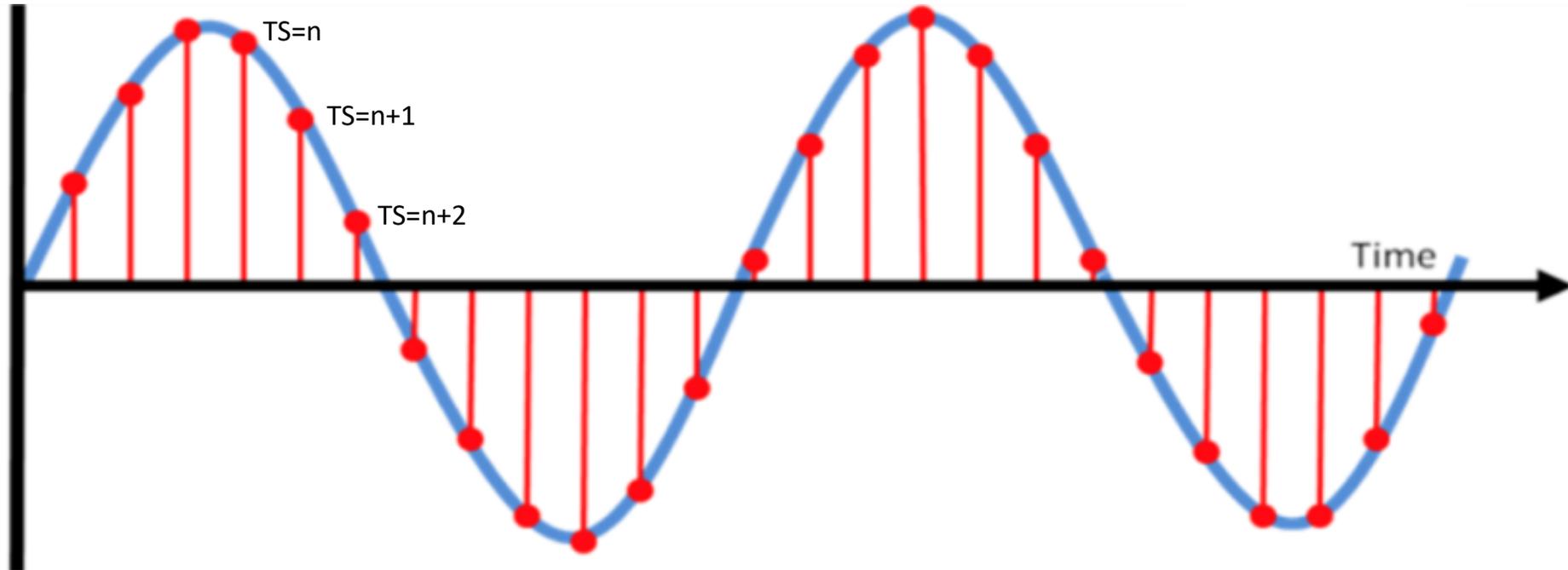
TaaS



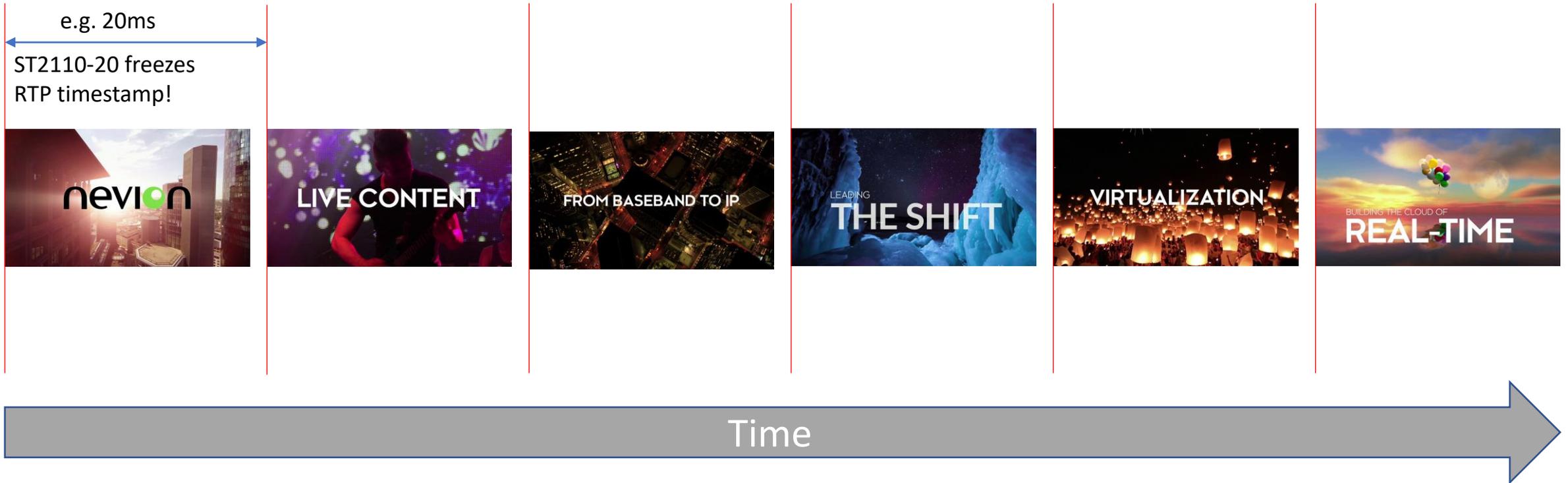
Origination time stamping in RTP



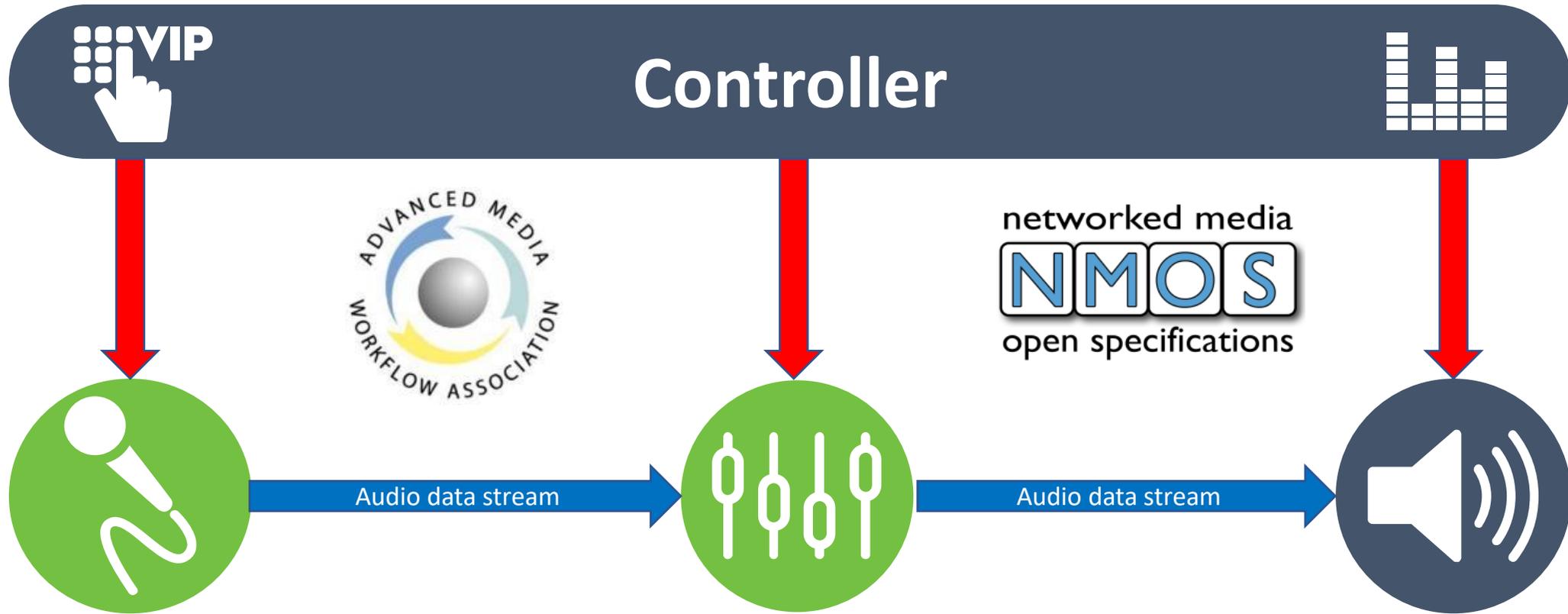
Audio per-sample timing



Video – per-frame sampling timing



What about the standardized control plane?



IS-04, IS-05, IS-08

The SDP - RFC 4566 – used in ST2110



Sender description
Video and/or audio essence
Raster size (in pixels)
Frame-rate (video)
Channel count (audio)
Sampling structure (audio/video)
Bit depth (audio/video)
Colorimetry
Source IP address and port
RTP payload ID (audio/video)
PTP grandmaster source and domain

FAST METADATA -41

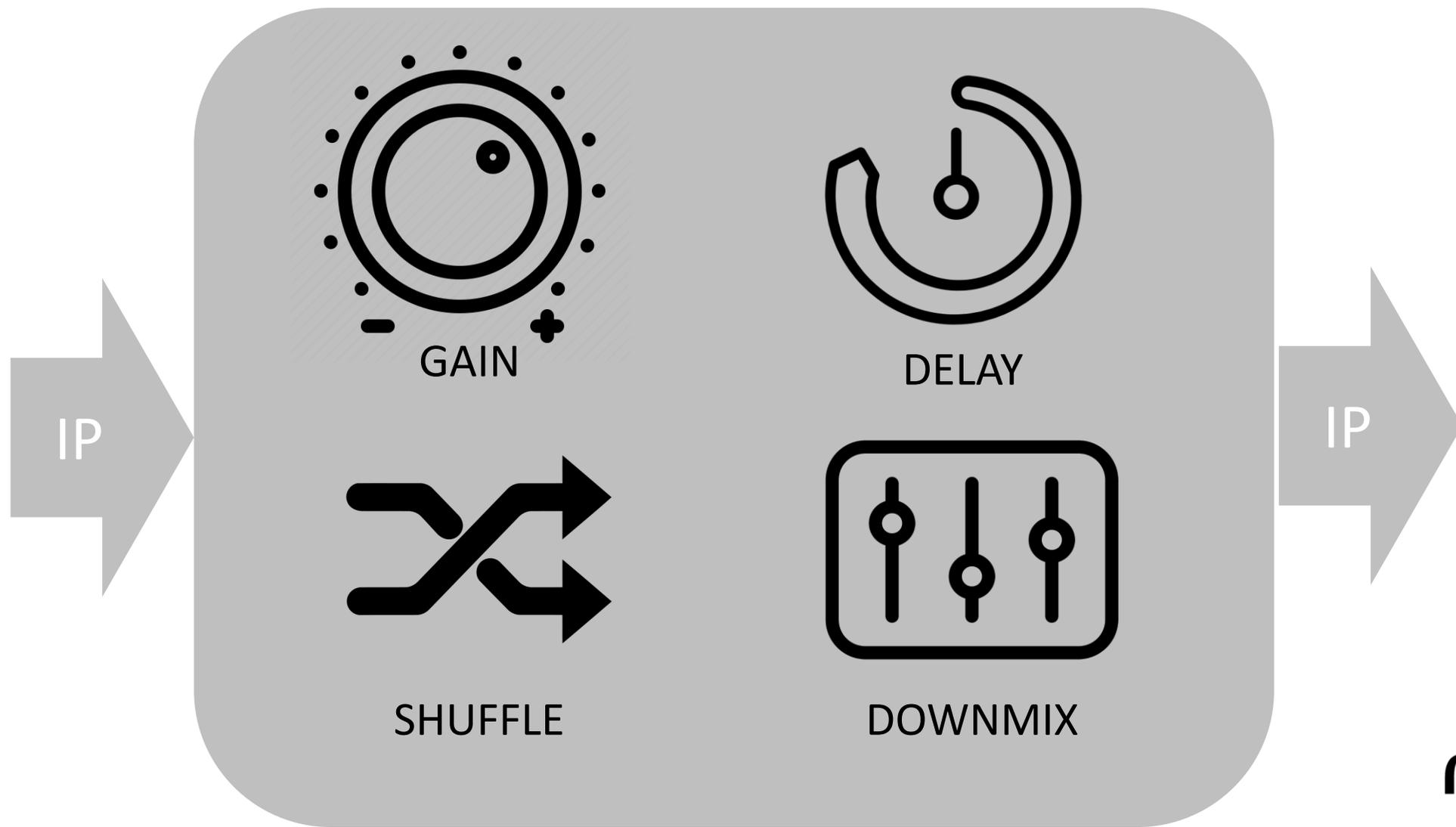


FMX -42

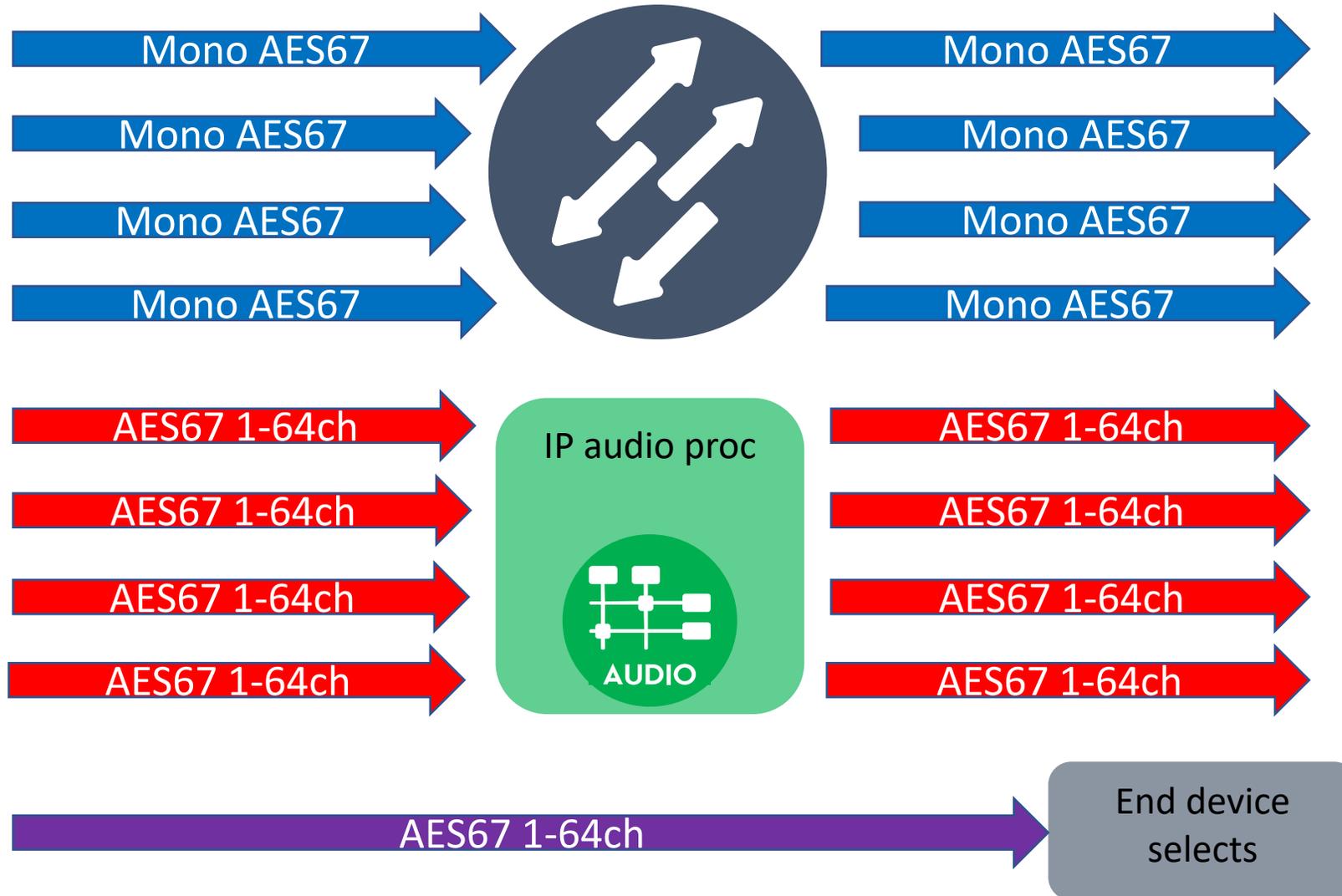


Coming soon

Audio manipulation required in IP domain



How to shuffle in the modern world



Audio templates – a few practical options



One channel - mono



Two channels - stereo

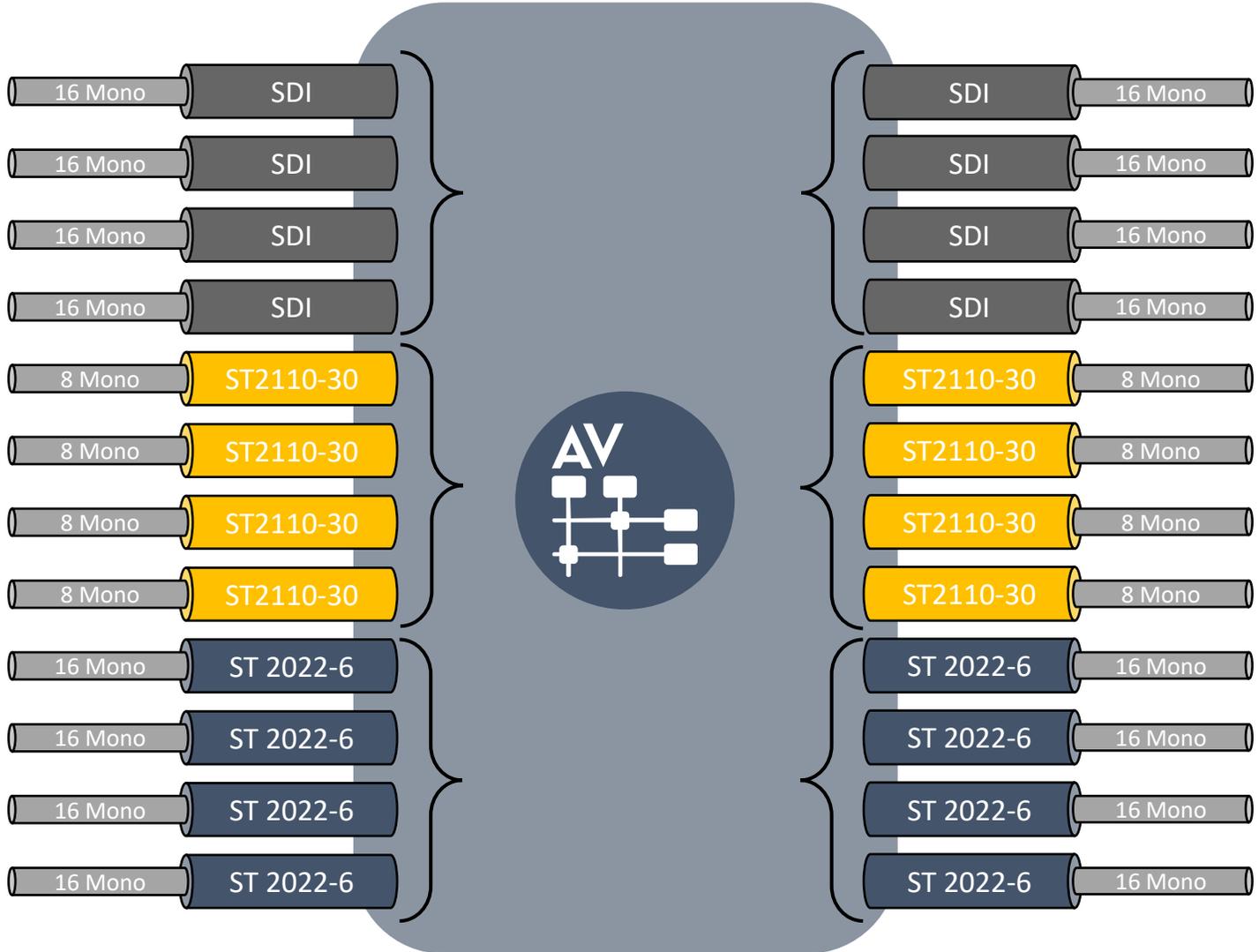


Six channels – 5.1

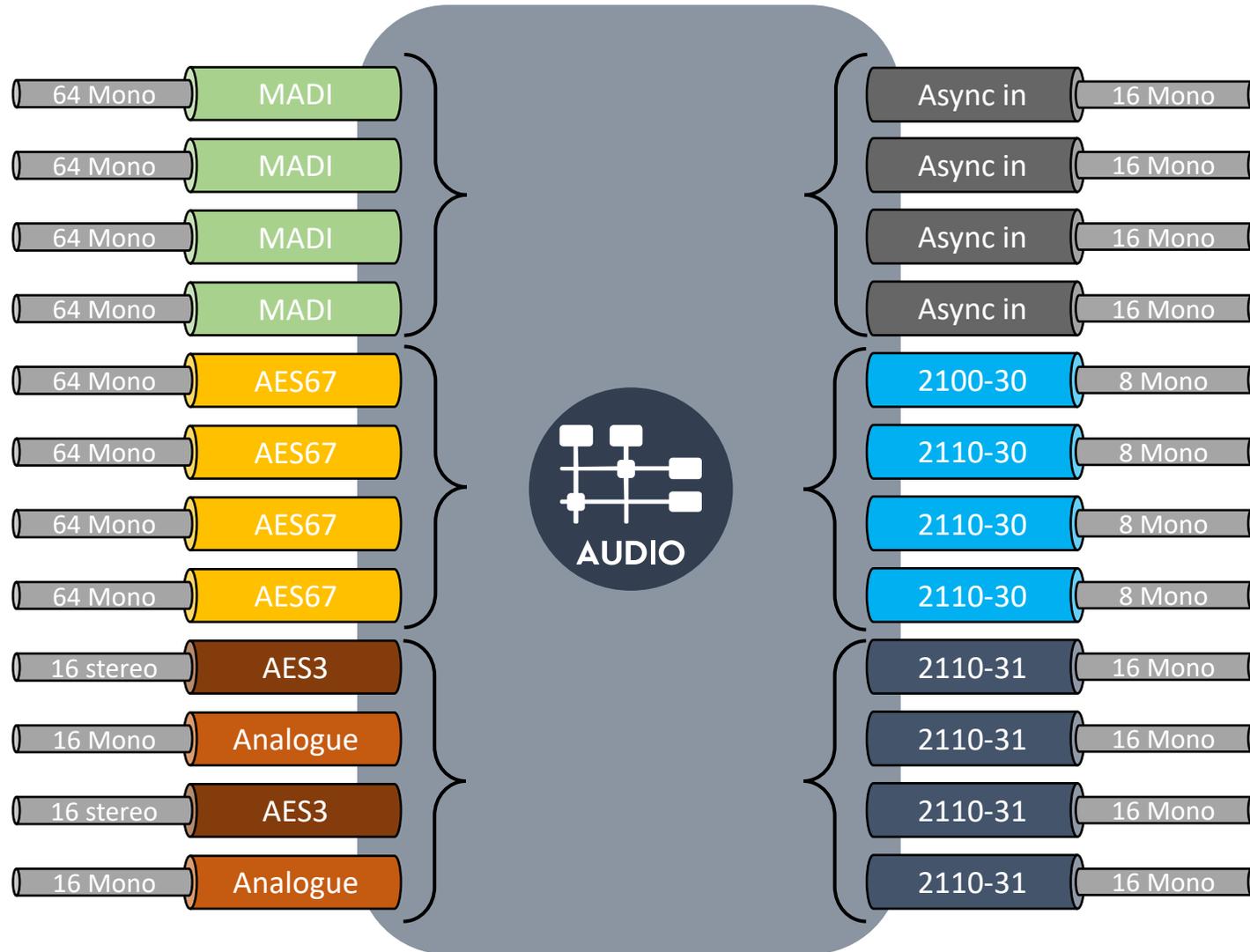


Eight channels – 4 x stereo

Video-associated audio format interfacing



Audio-only format interfacing

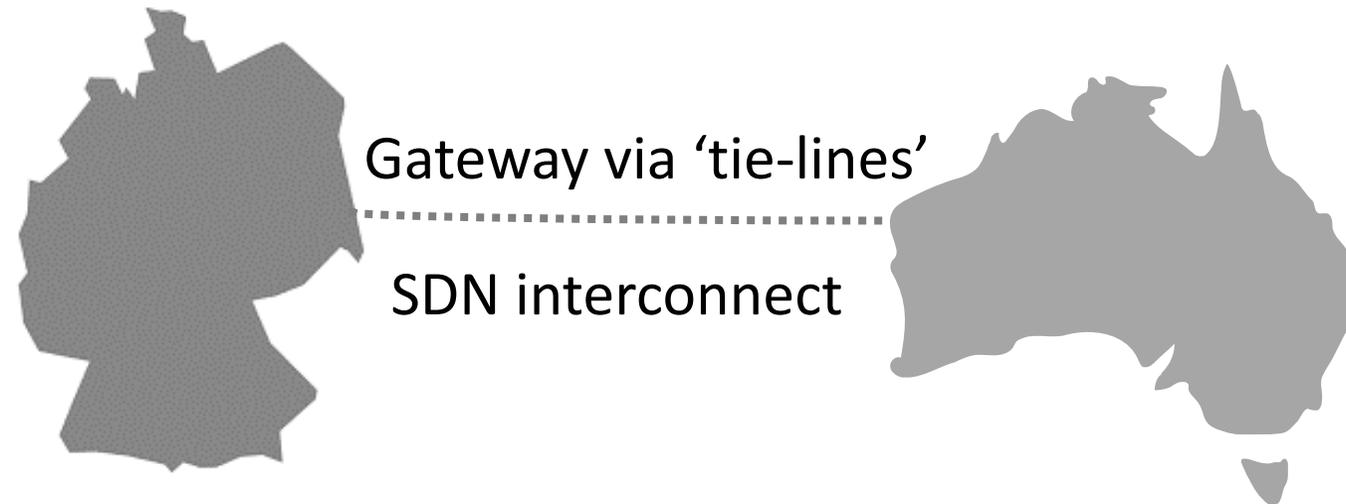


Moving outside the campus – distributed production

- WAN connectivity involved
- Longer latencies
- (Potentially) Asynchronous sources
- Layer 2 too limiting
- Layer 3 (routed) needed for larger and multi-campus networks

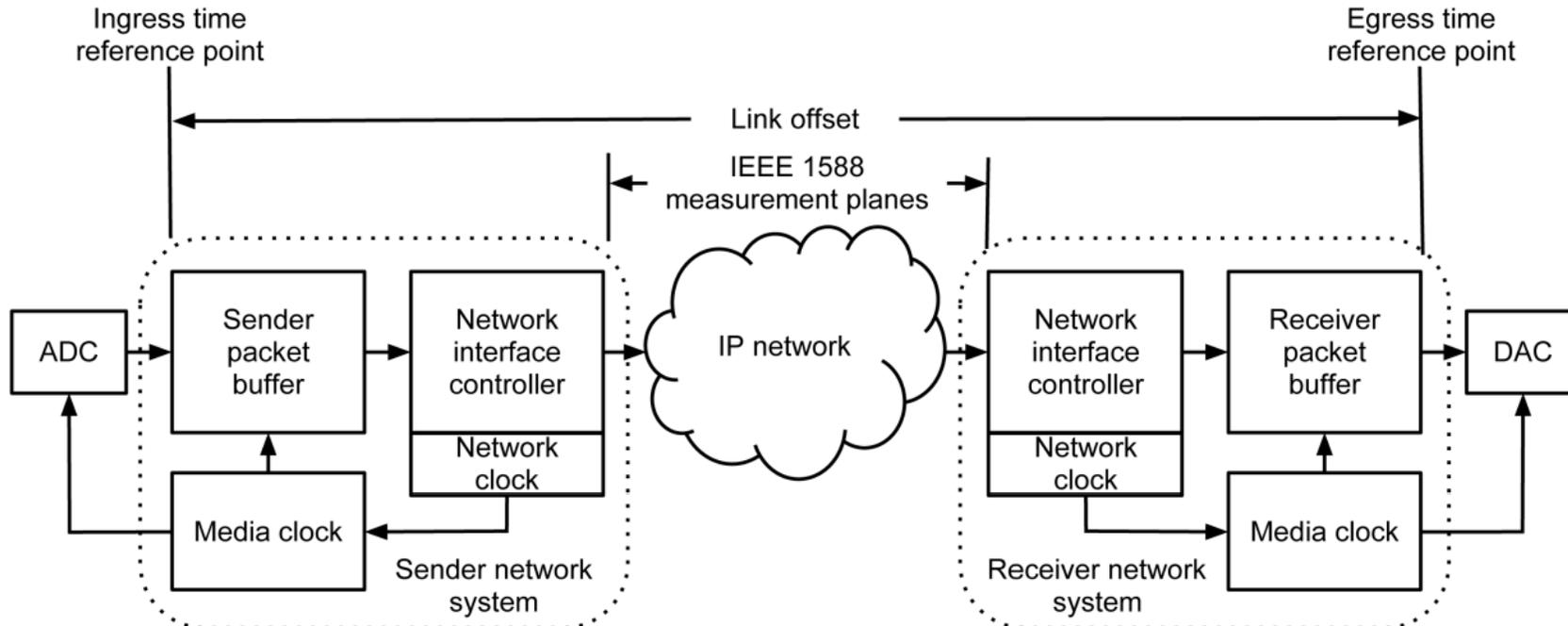
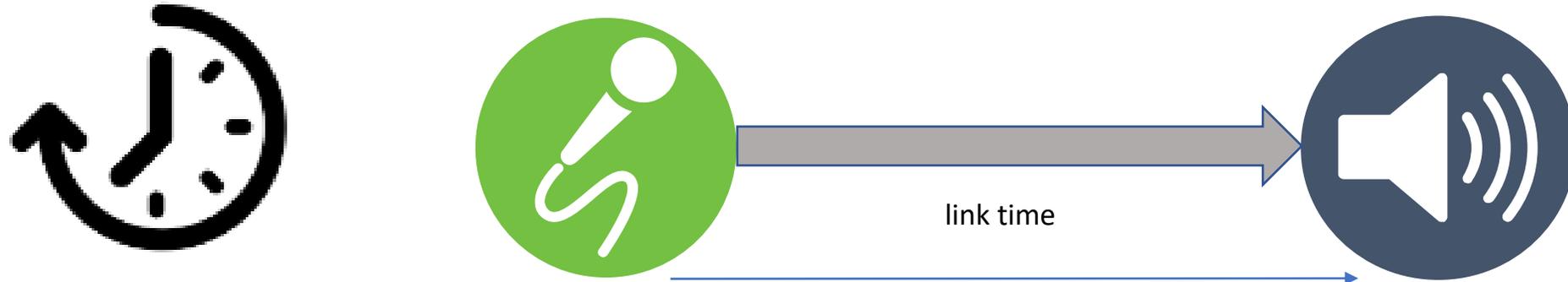


Interconnecting proprietary audio Islands

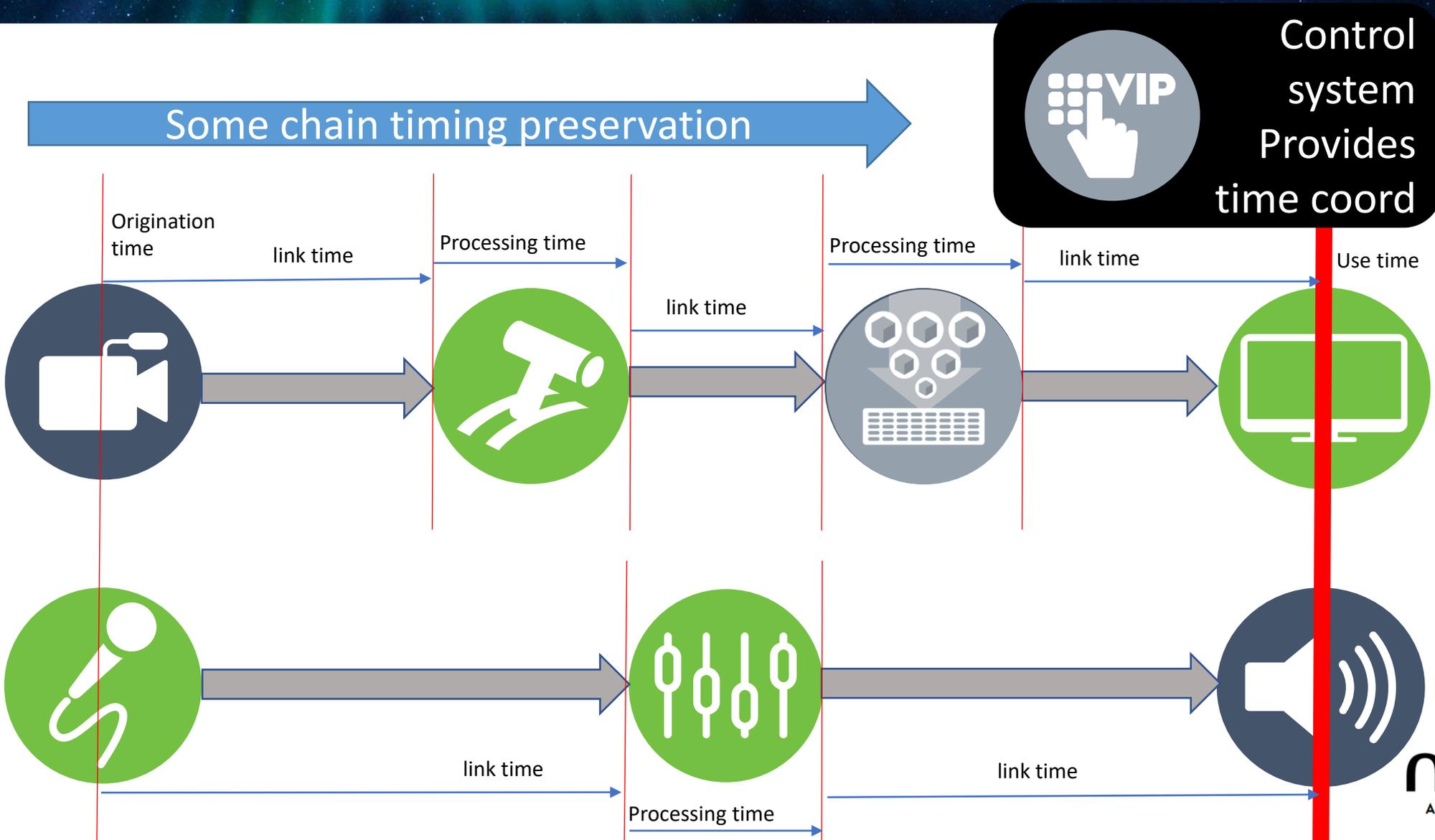


Plug and play 'toolkit' needed to transition to standardised formats

AES67 – defines link offset



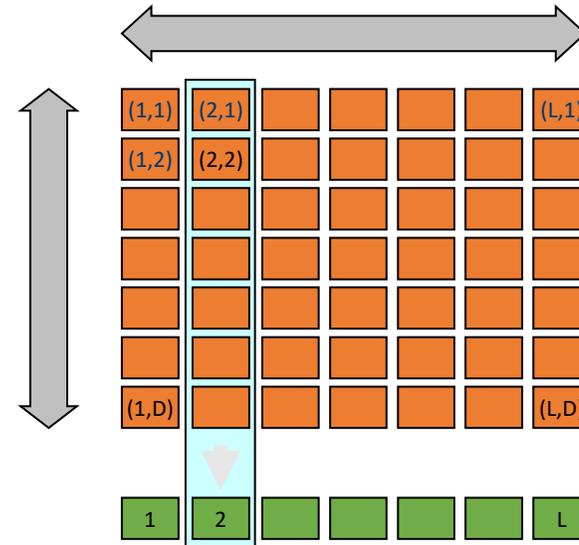
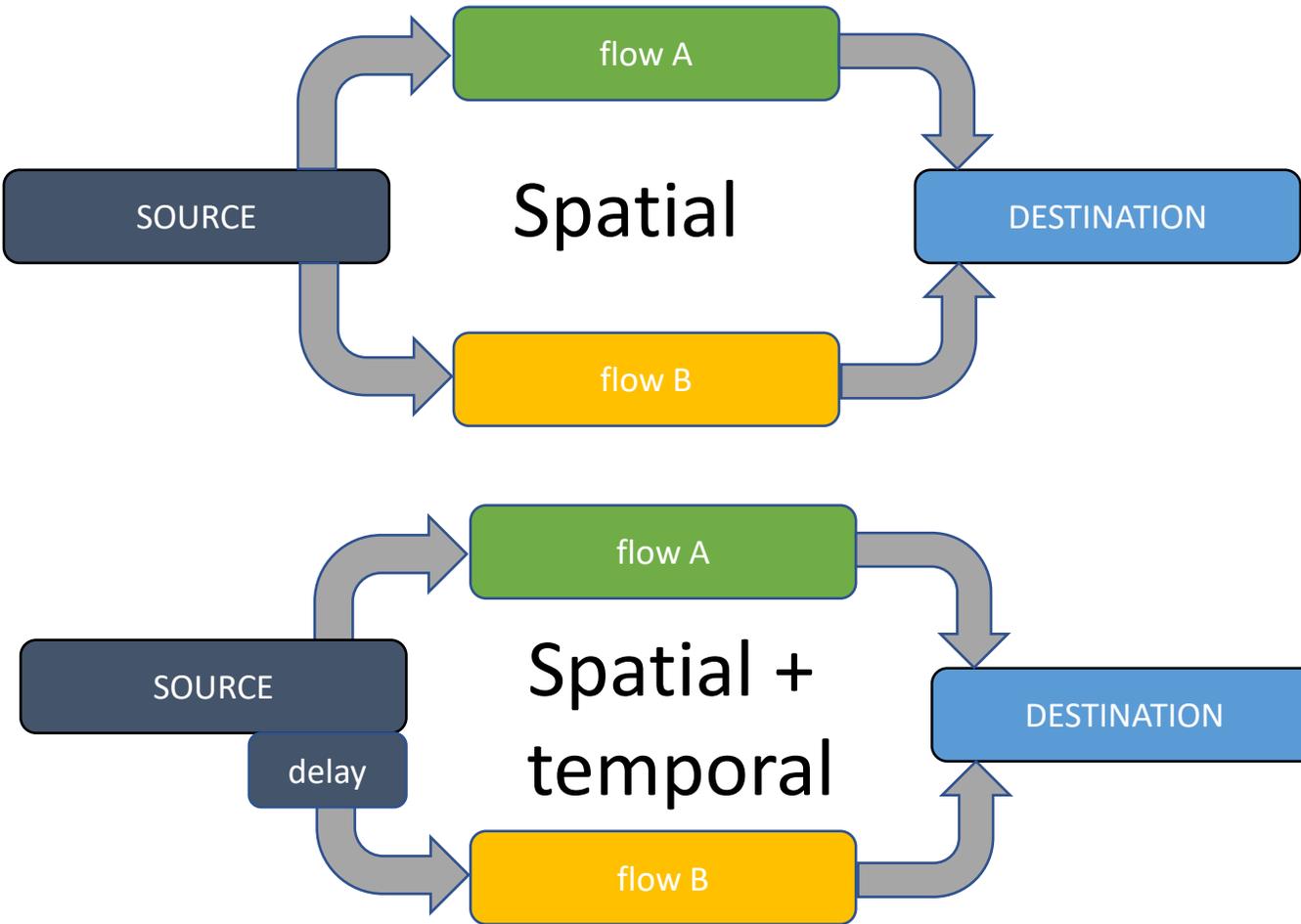
Hybrid timing reconciliation



PTP holdover is capable of being very long – let's make it so!



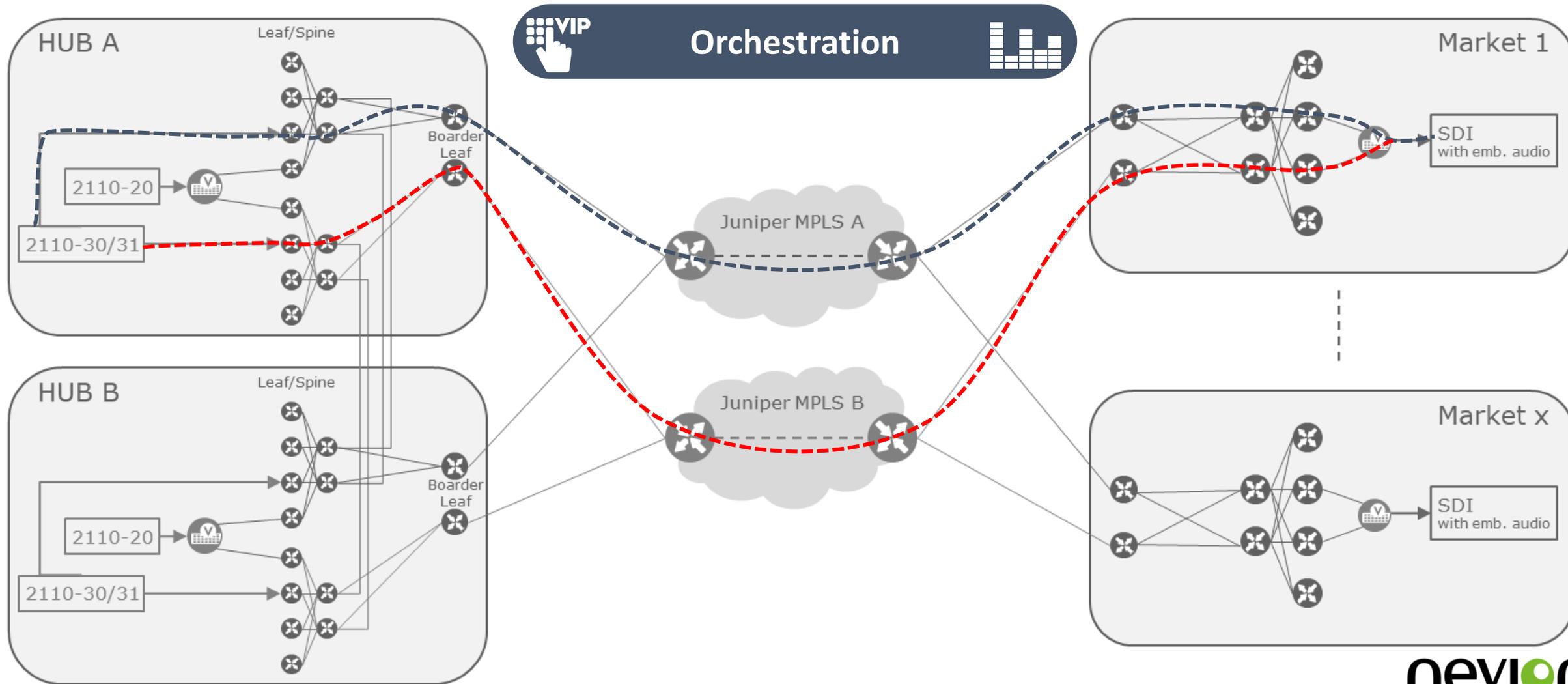
Protection – on and off campus – video & audio



FEC

n	n+1	n+2	n+3
n+4	n+5	n+6	n+7
n+8	n+9	n+10	n+11
n+12	n+13	n+14	n+15
n+16	n+17	n+18	n+19
n+20	n+21	n+22	n+23
n+24	n+25	n+26	n+27
n+28	n+29	n+30	n+31
n+32	n+33	n+34	n+35
n+36	n+37	n+38	n+39
n+40	n+41	n+42	n+43

International use case - Audio signal flows



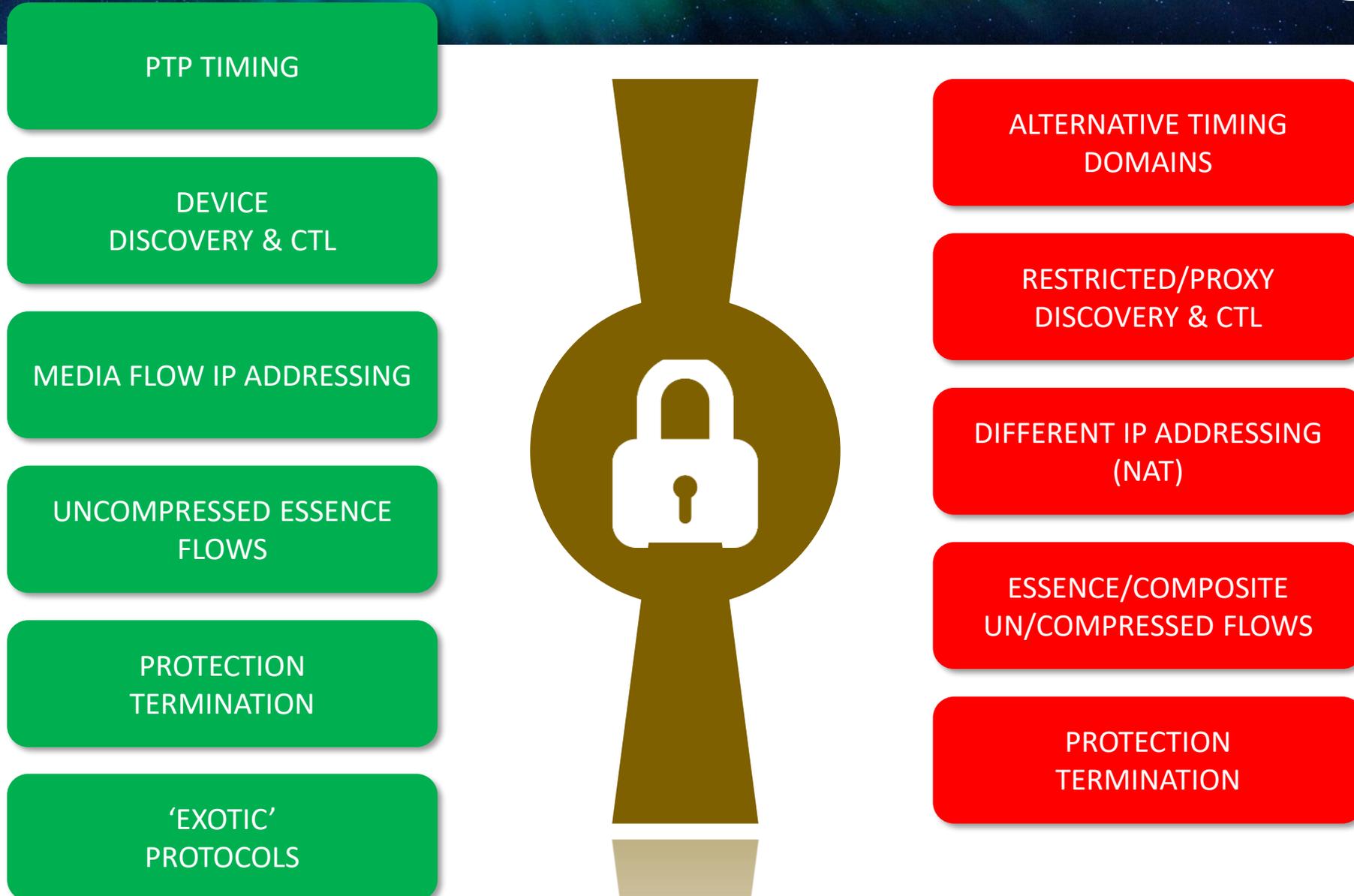
Secure networks need managed networks



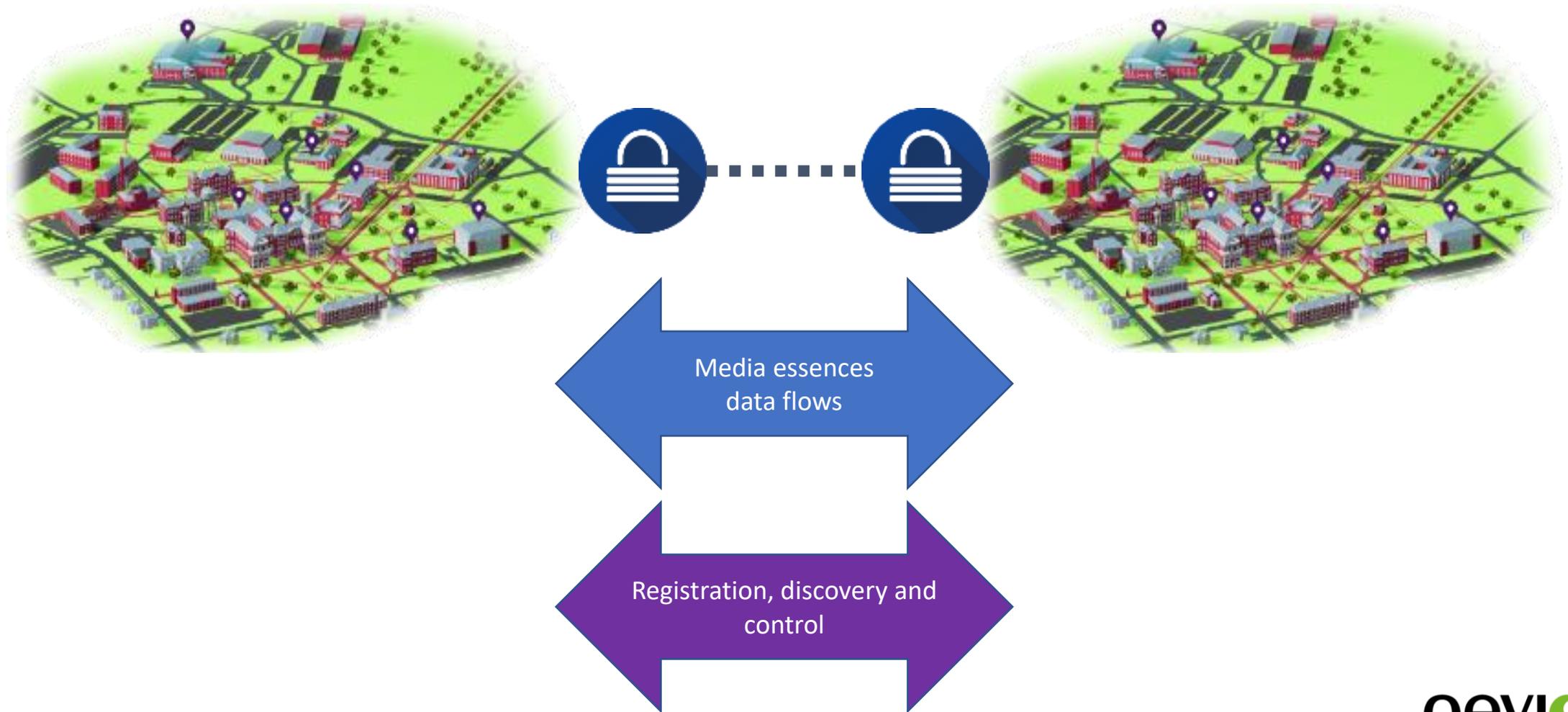
SDN control



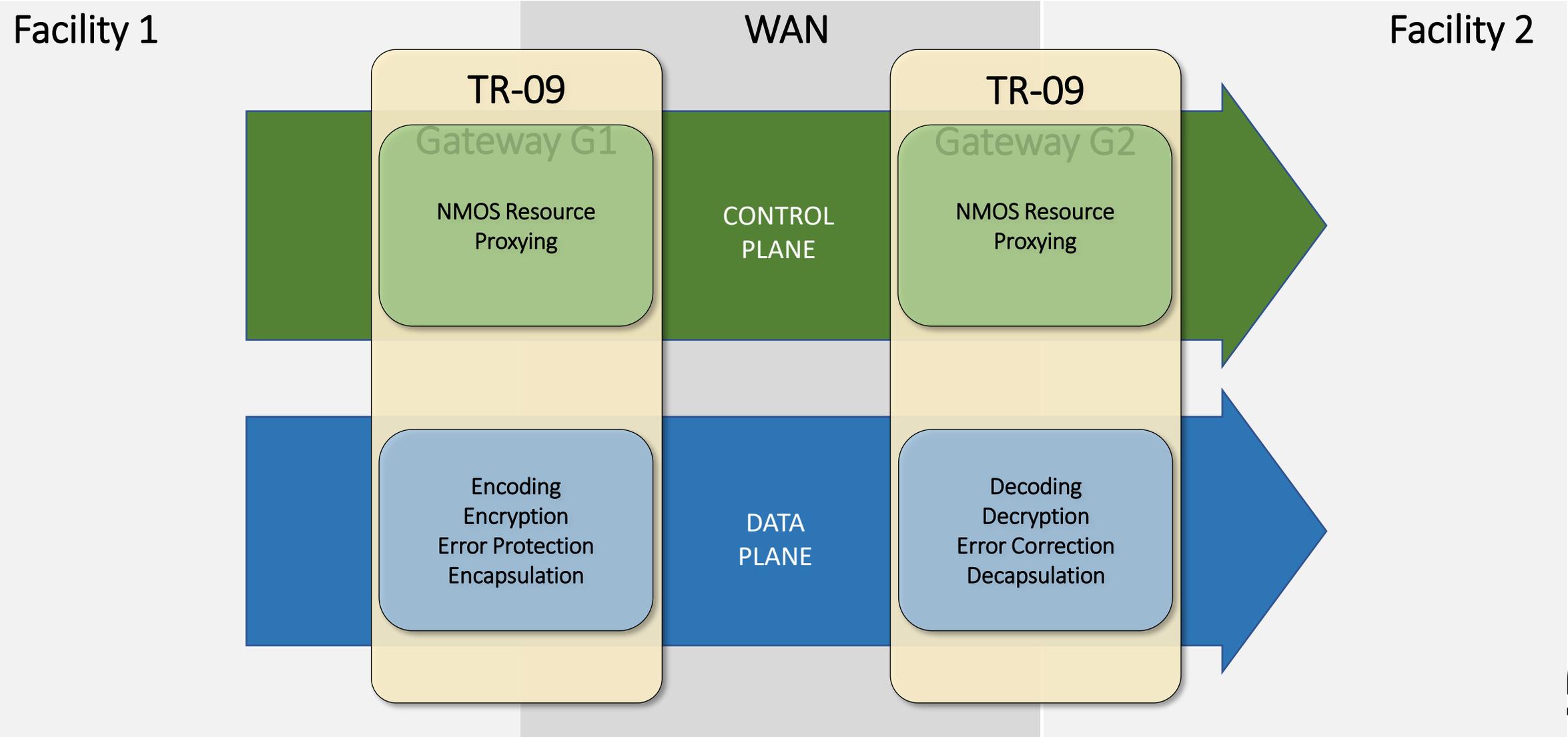
Going off-campus – the IP facility media edge



ST2110-WAN AG - two layers of focus – data and control



ST2110-WAN

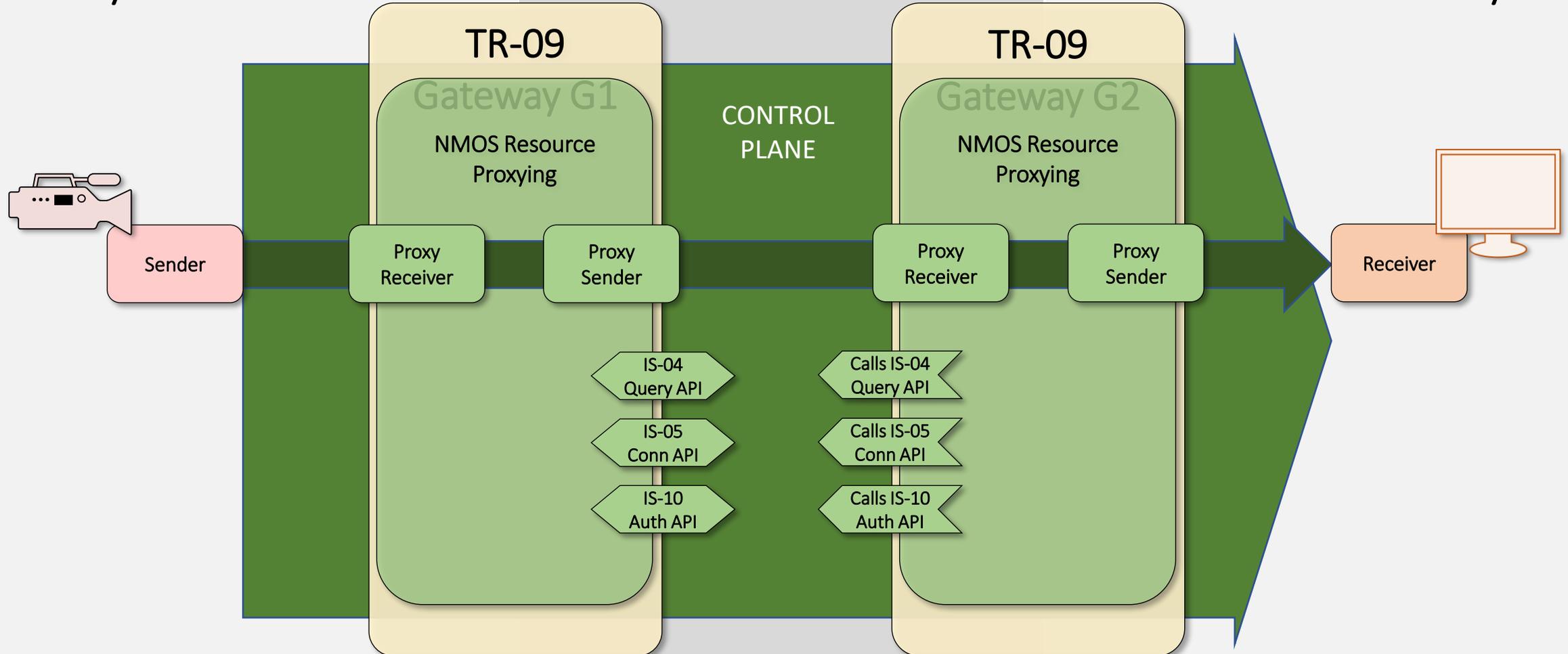


2110-WAN control plane

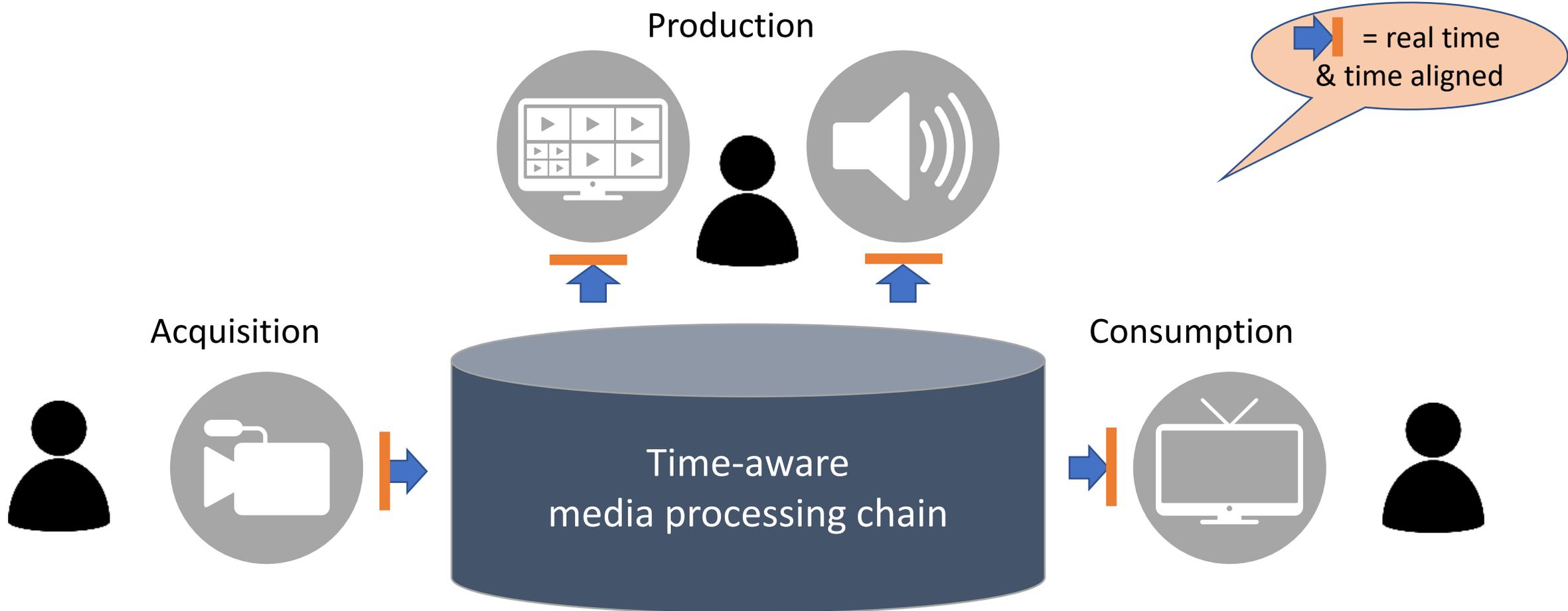
Facility 1

WAN

Facility 2



The broadcast end game



Deterministic data transfer

Linear and non-linear



RTP/UDP

2022-7

FEC

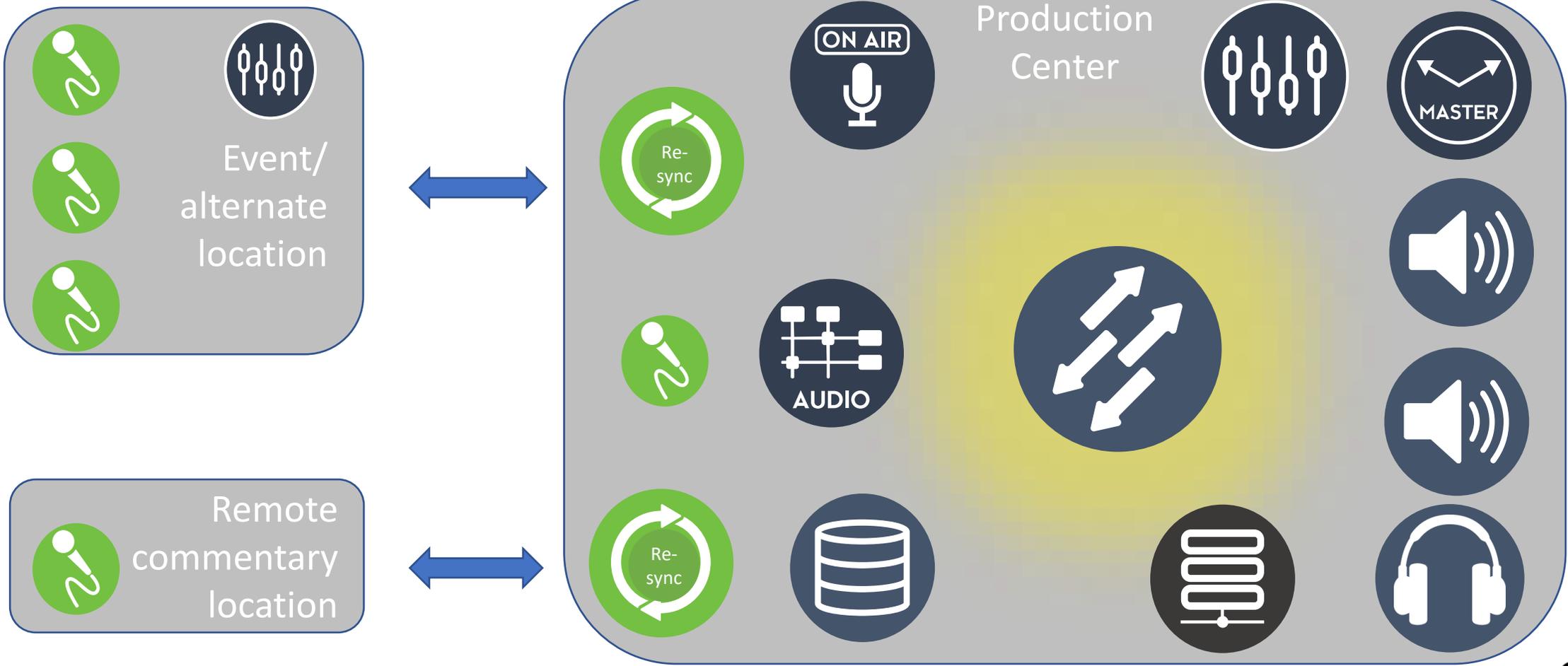
ARQ

TCP

Infiniband

SRD

In conclusion



Thank you!



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Check out our other presentations on-line

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Any Questions?

IP SHOWCASE™

