Subject: OUTAGE - Sorry, it's AMAZINGLY windy here...
Posted by Kim on Tue, 31 Jul 2007 13:23:25 GMT
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going to be
slammed for the next 10 days (including this and next weekend) The dust here
will have settled by the time you get to Austin and unpack.

Have a safe trip man,

Deej

"Jeff Hoover" <

Subject: Re: OUTAGE - Sorry, it's AMAZINGLY windy here...
Posted by audioguy_editout_ on Wed, 01 Aug 2007 01:51:01 GMT
View Forum Message <> Reply to Message

mix/best mastering" kind of combination and vote on it, "Aja"
just might win.

Freakin' GREAT sounds on every single instrument in there, as
pertains to the genre being recorded, NOTHING comes across as
abrasive, yet it's not so homogenized as to not grab your
attention, either (we've all heard mixes like that, right?).

Although, I must say "My Old School" rocks almost as hot in
that same voting contest. :D

(Yes, I'm currently in a Steely Dan mode - gotta get some
breathing space from the numetal stuff i've been spending so
much time on lately! lol)

Roger Nichols gets most of the credit for all the Steely Dan
stuff, but as I understand it, Elliot Scheiner really did most
of the basic tracks recording & also the mixdowns, so maybe
he's really the unsung hero here. Not sure if this applies to
Aja, as well, but nonetheless...

....GO LISTEN TO "AJA"!!!!

NeilDo you guys think PCI audio cards will be phased out in a few years for FW
3.2GHz and USB 3.0 boxes?

http://www.macuser.com/hardware/usb_30_is_gonna_be_4_gbps_bi g_1.php

http://lowendmac.com/newsrev/mnr07/0928.html#1

http://blogs.pcworld.com/digitalworld/archives/2007/09/comin g_soonish.html"James McCloskey" <excelsm@hotmail.com> wrote:

>Oh noooooooooooooooo!!!!!!!!! : 0

Ahhh shaddup!

DCDJ,

I'm kinda glad I didn't make it as well. Having weird issues in my arms wih numbing and shooting pains. I'm not that great of a guitarist to start with, but this situation makes it really bad!!!

I'll be in touch and see how it's going for both of us in a couple weeks.

Jeff
> So you're finally out the door eh? That's good. It's also good that you
> didn't decide to come this weekend. Everything happened here just about like
> I would have predicted in my worst nightmare workwise and I'm going to be
> slammed for the next 10 days (including this and next weekend) The dust here
> will have settled by the time you get to Austin and unpack.
>
> Have a safe trip man,
>
> Deej
>
> "Jeff Hoover" <jkhoover@excite.com> wrote in message news:46fc31bc@linux...
>
>>So I'm sitting here in the van traveling down the road, not a house in
>>site in the middle of NEvada...and connected with decent enough speed to
>>browse and do newsgroup stuff without a hitch. Using Sprint's service and
>>am actually pretty pleased so far (Portland OR to Wells NV). Being a
>>independent consultant of sorts, the sixty bucks a month is well worth it
>>to keep connected to customers while on the road.
>>
>>
>>Jeff
>
>
>I would say is very possible.

Could be a while for PCIe though. SATA/eSATA should be the mainstay for storage for a while, with FW and USB for I/O and peripherals. Could make motherboards more populated with these to eliminate PCI slots (other than video cards).

Speaking of FW 3.2 - where the heck are S1600 and 1394c (twisted pair)?

Dedric

On 9/28/07 9:51 PM, in article 46fdcbaa$1@linux, "James McCloskey" <excelsm@hotmail.com> wrote:

>
> Do you guys think PCI audio cards will be phased out in a few years for FW
> 3.2GHz and USB 3.0 boxes?
>
> http://www.macuser.com/hardware/usb_30_is_gonna_be_4_gbps_bi g_1.php
>
> http://lowendmac.com/newsrev/mnr07/0928.html#1
>
> http://blogs.pcworld.com/digitalworld/archives/2007/09/comin g_soonish.html
>Ok, let me try being a little more articulate.

I came up in recording in the 1970's. Pros have a different attitude than consumers about gear. Gear is a tool. We are not loyal to brands, despite being loyal to an obsolete DAW.

Big fights about gear are for consumers, not pros, and are bad form in the extreme.

We care about Jobs and Gates exactly as much as they care for us. Zero... We care about Apple vs. PC as much as those companies care about audio pros. Next to zero.

In short, I want nothing to do with these mac vs pc fights.

They suck, they're silly, and they're boring.

Being boring is unforgiveable.

So let's give the damn platform wars a break willya?

Just a thought.

DC"James McCloskey" <excelsm@hotmail.com> wrote in message news:46fdcbaa$1@linux...
>
> Do you guys think PCI audio cards will be phased out in a few years for FW
> 3.2GHz and USB 3.0 boxes?
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> http://www.macuser.com/hardware/usb_30_is_gonna_be_4_gbps_bi g_1.php
>

> http://lowendmac.com/newsrev/mnr07/0928.html#1
>
> http://blogs.pcworld.com/digitalworld/archives/2007/09/comin g_soonish.html

I dunno.....the idea of having my audio signals shuttled around in a
packet still makes me nervous. I mean, what if one of them gets dropped and
broken?

;oP"James McCloskey" <excelsm@hotmail.com> wrote in message
news:46fdbcd9$1@linux...
>
> I know this is way off topic, but you got to check this out!
>
> http://video.yahoo.com/video/play?vid=1006571&cache=1

You mean this one?

http://www.snopes.com:80/photos/odd/haitiufos.asp

Probably did it on a Ma.....nawwww.....I won't go there.

;oD"DJ" <animix _ at _ animas _ dot _ net> wrote in message
news:46fdda93$1@linux...
>
> "James McCloskey" <excelsm@hotmail.com> wrote in message
> news:46fdbcd9$1@linux...
>>
>> I know this is way off topic, but you got to check this out!
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>> http://video.yahoo.com/video/play?vid=1006571&cache=1
>
> You mean this one?
>
> http://www.snopes.com:80/photos/odd/haitiufos.asp
>
> Probably did it on a Ma.....nawwww.....I won't go
> there.
>
> ;oD
>

Actually....he did do it on a Mac. Pretty impressive. That's probably why
you were drawn to it.

;o)I had a cassette of it years ago, then I just saw the CD in Walmart for
$6.99 recently and picked it up again. I've listened to it a bunch lately,
and yeah... it's on up in the top 5, or so, of my favorite sounding
recordings.

CLThanks Neil, I'll try that.

CLJust kidding around dude! Glad your enjoying your iPod. I hope to own an iPod one day. I own a Creative MuVo, it's a piece.

"DC" <dc@spammersincupertino.com> wrote:
>
>Ok, let me try being a little more articulate.
>
>I came up in recording in the 1970's. Pros have a different
>attitude than consumers about gear. Gear is a tool. We are
>not loyal to brands, despite being loyal to an obsolete DAW.
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>Big fights about gear are for consumers, not pros, and are
>bad form in the extreme.
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>We care about Jobs and Gates exactly as much as they care for
>us. Zero... We care about Apple vs. PC as much as those
>companies care about audio pros. Next to zero.
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>In short, I want nothing to do with these mac vs pc fights.
>
>They suck, they're silly, and they're boring.
>
>Being boring is unforgiveable.
>
>So let's give the damn platform wars a break willya?
>
>Just a thought.
>
>DC
>
>
>Used to use it to eval systems I built.

Stopped doing that. Damn thing makes almost anything sound good....

Genius stuff.

DCHey guys, I'm wondering what you guys think of this product? Do you think there is a real market for this kind of guitar? Would you buy something like this? What do you think you would be willing to pay if the quality was good?

Thanks for your opinions in advance.

http://cgi.ebay.com/ws/eBayISAPI.dll?ViewItem&item=33016

9549217&ssPageName=ADME:B:EF:US:12For the most, I'm using Adobe Audition 2, Nero 7 or PlexTools
Professional(XL) on the different machines I have. Plextools need a
Plextor burner and have plenty of things to control the burning, both
before and after. Really love it when the aftercontroll can tell me
how fair/faulty the burnings are in different ways and can print out
documentation to follow a burned CD-project.

Erling

On Fri, 28 Sep 2007 11:24:42 -0400, "Don Nafe" <dnafe@magma.ca> wrote:

>Hi All
>
>what is everyone using nowadays
>
>Don
>you're absolutely right!

On Fri, 28 Sep 2007 15:43:56 -0700, Jeff Hoover <jkhoover@excite.com>
wrote:

>rick wrote:
>
>>
>> feed 50 guys beans and you've got a pipe organ...sort of...
>>
>If you want the "pipe organ" effect, you must also serve varying sizes
>of macaroni at the same meal!!!
>
>;-)
>
>
>JeffUnfortunately the weight and size will limit your market a lot. Plus twice
as many strings to change. There is currently a digital modeling guitar
out there that does Telecaster/Strat/Les Paul with a single knob. To me
that is much more desireable.

JohnSteely Dan went stupid at a young age. Aja showed what live studio musicians
could do when they played together and had extended jams. The synergy was
amazing. It is still one of the best albums ever.

Since then they totally forgot what a simple formula it is. Now I'm luck
if I can get a boring 8 bar solo from Becker on a song they created from
a drum machine.http://www.youtube.com/watch?v=eG6X-xtVaskHey Chris

Does Wavelab let you create fades and adjust start point times etc in the
actual CD project?

D

"Chris Ludwig" <chrisl@adkproaudio.com> wrote in message
news:46fd2369$1@linux...
> wavelab
>
>
>
> Don Nafe wrote:
>> Hi All
>>
>> what is everyone using nowadays
>>
>> Don
>
> --
> Chris Ludwig
>
> ADK Pro Audio
> (859) 635-5762
> www.adkproaudio.com
> chrisl@adkproaudio.comWhy in da heel can't people learn how to make videos... Not one
mouth
matched the sound... that just drives me batty! I can' watch it like
that... They make tools to calculate the rendering and such.. I wish people
would use it...

Otherwise, very stunning news... Why don't they enforce their own rules...
instead of sitting on youtube watching it... *shakin head* Must be
Texas... lol.

"John" <no@no.com> wrote in message news:46fe1cb3$1@linux...
>
> http://www.youtube.com/watch?v=eG6X-xtVaskIt's usually one of two things. Heat or oxidation
on the daughterboard contacts.
If it's oxidation, like it was on mine, just remove the daughterboard and
get some contact cleaner and a toothbrush and scrub them really good. Regarding
heat, EDS cards get REALLY hot so you have to have serious extra fans going
o keep their life, otherwise they die a short life.

http://www.radioshack.com/search/index.jsp?kwCatId=&kw=c
ontact%20cleaner&origkw=contact%20cleaner&sr=1

JohnHI James,

PCI 32 bit slots have got about a year left on the in the consumer PC systems at best. There really isn't any need for then at all on the consumer system.
PCI-X slots will be around for a few more years on the server market till the PCI-e spec is completely independent from the PCI bridge technology.
There are still allot of PCI-x network and drive controller cards available compared to PCI-e.

Until then PCI-e devices are all over the map in performance. Something like a RME MADI or Apogee card and some of the high end HD video capture/playback cards really push the the limits of PCI-e slots right now.
But they are some of the only devices that do.

USB 3.0, Fire wire 3.2 devices will easily replace the need for PCI slots. Fire wire already replaces the need for many devices already.

What I hope this time around the requirement for stricter compliance for the chip set manufacturers.
Or just make so that only Texas Instruments can make firewire and NEC can make USB but thats not very "competitive".
PC and MACs are both victims of shitty Fire wire chip sets.

Chris

James McCloskey wrote:
> Do you guys think PCI audio cards will be phased out in a few years for FW
> 3.2GHz and USB 3.0 boxes?
>
> http://www.macuser.com/hardware/usb_30_is_gonna_be_4_gbps_bi g_1.php
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>
> http://blogs.pcworld.com/digitalworld/archives/2007/09/comin g_soonish.html
>
>

--
Chris Ludwig
ADK
chrisl@adkproaudio.com <mailto:chrisl@adkproaudio.com>
www.adkproaudio.com <http://www.adkproaudio.com/>
(859) 635-5762HI Dedric,
PCI-e isn't going any where soon. It's really just in its infancy. Once they can remove it from the PCI bridging method that they current have

to do now you will start to see it potential.
Thats gonna be a while though.

Chris

Dedric Terry wrote:
> I would say is very possible.
>
> Could be a while for PCIe though. SATA/eSATA should be the mainstay for
> storage for a while, with FW and USB for I/O and peripherals. Could make
> motherboards more populated with these to eliminate PCI slots (other than
> video cards).
>
> Speaking of FW 3.2 - where the heck are S1600 and 1394c (twisted pair)?
>
> Dedric
>
>
> On 9/28/07 9:51 PM, in article 46fdcbaa$1@linux, "James McCloskey"
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>>
>>
>
>

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Chris Ludwig
ADK
chrisl@adkproaudio.com <mailto:chrisl@adkproaudio.com>
www.adkproaudio.com <http://www.adkproaudio.com/>
(859) 635-5762HI DJ,
Yes they should be fine on the older boards.
They don't work at all on new motherboards.
I have one hear collecting dust that I think still works if you ever
plan on adding another. :)

They had a real shitty upgrade offer to the Mk2 so never upgraded.

Chris

DJ wrote:
> I found one of the old PCI cards on EBay last week so I figured WTF not buy
> it and give it a spin. I actually like this system a lot. I think the plugs
> sound very good. They aren't UAD-1 plugs, but they are nicely done
> nevertheless and the Megareverb is much better sounding to my ears
> (presetwise, of course) than UA's Dreamverb. I had sort of prepared myself
> for it not to work since I've got a dual core CPU and there are some issues
> with certain AMD duallie chipsets, but it works fine with my Gigabyte GA
> K8NS Ultra 939 and Opteron 185.
>
> I'm giving it a thumbs up,.especially for $150.00. Check EBay. The old ones
> are out there cheap these days. You will need lots or room in your case
> though.this thing is longer than an EDS 1000.
>
> ;o)
>
>
>

--
Chris Ludwig
ADK
chrisl@adkproaudio.com <mailto:chrisl@adkproaudio.com>
www.adkproaudio.com <http://www.adkproaudio.com/>
(859) 635-5762lol

DJ wrote:
> .. The video guys will love it sense they
>
>> won't some wacky combination of hardware to get the same results.
>>
>>
>
> You're missing the whole point Chris. It's the journey dude.
>
> ;oD
>
>
>

--
Chris Ludwig
ADK

chrisl@adkproaudio.com <mailto:chrisl@adkproaudio.com>
www.adkproaudio.com <http://www.adkproaudio.com/>
(859) 635-5762Yellow Tools is giving away a free version of their sampler, Independence.
It is a fully functioning version with no restrictions except it can't import
WAV (or other) sound files and it has a 2Gig subset of the complete library.
Based on "Culture" and some of the Yellow Tools Giga samples I have used
it should be well worth the slow downloads.
http://www.yellowtools.us/cp21/cms/index.php?id=399&show news=527

GeneMaking a record like Aja is a lot of concentration and work. Maybe they
just got lazy.

I used to have that cassette and for about a year it was the only thing
I had in my car. It is awesome. My old school totally rocks too. Still
one of my all time favs is Haitian Divorce. It's so tasty AND funny.

John wrote:
> Steely Dan went stupid at a young age. Aja showed what live studio musicians
> could do when they played together and had extended jams. The synergy was
> amazing. It is still one of the best albums ever.
>
> Since then they totally forgot what a simple formula it is. Now I'm luck
> if I can get a boring 8 bar solo from Becker on a song they created from
> a drum machine.I would tend to agree. There's even one that will switch
tunings for you wht the flip of a modeling switch (I think it's
the Fender Strat version, IIRC). Yep, only reason to have a
doubleneck onstage these days would be for the visual thang.

Neil

"John" <no@no.com> wrote:
>
>Unfortunately the weight and size will limit your market a lot. Plus twice
>as many strings to change. There is currently a digital modeling guitar
>out there that does Telecaster/Strat/Les Paul with a single knob. To me
>that is much more desireable.
>
>JohnI'm sure some guitarists will think it's great. I don't know how big
that market might be.

I think double necked guitars are awkward and I doubt I'll ever get one.

I get a wide range of tones out of my PRS, with just one neck. Parker
Fly, some Ibanez models and Carvins, even a fat strat can cover a lot of
tonal ground without the extra weight and hassles of a double neck
instrument.

I can get an even wider range of tones with my Variax, and instant alternate tunings. Roland has their VG line that works with any guitar with a hex pickup to provide that sort of thing as well. I like the Variax better than Rolands earlier models, overall, but I haven't heard Roland's latest VG99 which is still, I think, about to ship.

Cheers,
-Jamie
www.JamieKrutz.com

James McCloskey wrote:
> Hey guys, I'm wondering what you guys think of this product? Do you think
> there is a real market for this kind of guitar? Would you buy something
> like this? What do you think you would be willing to pay if the quality
> was good?
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> Thanks for your opinions in advance.
>
> http://cgi.ebay.com/ws/eBayISAPI.dll?ViewItem&item=33016
9549217&ssPageName=ADME:B:EF:US:12
>Hi Don,

Don Nafe wrote:
> Hey Chris
>
> Does Wavelab let you create fades and adjust start point times etc in the
> actual CD project?
>
Yes and then some. To me it has the best feature set and fast work flow of any editor/Mastering program.

There are some many awesome things it can do it pisses me off that they don't have a demo download so you can see. arrrghhh

I would suggest downloading the manual and read the Audio Montage section to see the similarities to CD Architect.
ftp://ftp.steinberg.net/Download/WaveLab_6/Docs_English/Wave Lab.pdf

After looking thru the manual check out the feature comparison between the different flavors of WL to see which one will be best for you.
They offer competitive crossgrades for WL Studio 6 and the full WL that you can do sense you own CD Arch.

http://www.steinberg.net/1454_1.html

Chris

> D
>
>
> "Chris Ludwig" <chrisl@adkproaudio.com> wrote in message
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>
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>> www.adkproaudio.com
>> chrisl@adkproaudio.com
>>
>
>
>

--
Chris Ludwig
ADK
chrisl@adkproaudio.com <mailto:chrisl@adkproaudio.com>
www.adkproaudio.com <http://www.adkproaudio.com/>
(859) 635-5762In the history of guitar retail, there has never been any thing close to the market of single neck guitars. Not gonna change IMO. Double's make great pictures and shoulder injuries and that's about it.

W. Mark Wilson

"James McCloskey" <excelsm@hotmail.com> wrote in message
news:46fdf969$1@linux...
>
> Hey guys, I'm wondering what you guys think of this product? Do you think

> there is a real market for this kind of guitar? Would you buy something
> like this? What do you think you would be willing to pay if the quality
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> http://cgi.ebay.com/ws/eBayISAPI.dll?ViewItem&item=33016
9549217&ssPageName=ADME:B:EF:US:12
>"W. Mark Wilson" <xpam_mark@avidrecording> wrote:

>Double's make great pictures and shoulder injuries and that's
>about it.

LOL! Maybe the money behind this new line is a group of
chiropractors.

"Yeah, and we'll make 'em out of BUBINGA, too!"   :D

NeilWhat he said -

Still load up the vinyl version on my VPI playing through my Beard/Counterpoint
and Maggies. Remember the first day I heard it. The sound, the playing, the

rise in intensity...

tingles

Ted

"Neil" <OIUOI@OIU.com> wrote:
>
>..is the fuckin' best recording EVER! OK, well, let's back off
>a bit & say it could POSSIBLY be the best fucking recording
>EVER. At the very least, it's definitely a candidate, yes? I'm
>not talking about the whole album (although the whole
>album is great, really, as everyone knows), I'm talking about
>the seven-minute-and-fifty-five-second title cut that was
>nailed on the 2nd take with no edits on the basic tracks.
>
>Man, I'm sitting here listening to it, and I think that if
>there was a way you could compare the best possible recordings
>of all time from ANY genre and digest them down to some kind of
>quantifiable "best sound/best playing/best execution/best
>mix/best mastering" kind of combination and vote on it, "Aja"
>just might win.
>
>Freakin' GREAT sounds on every single instrument in there, as

>pertains to the genre being recorded, NOTHING comes across as
>abrasive, yet it's not so homogenized as to not grab your
>attention, either (we've all heard mixes like that, right?).
>
>Although, I must say "My Old School" rocks almost as hot in
>that same voting contest. :D
>
>(Yes, I'm currently in a Steely Dan mode - gotta get some
>breathing space from the numetal stuff i've been spending so
>much time on lately! lol)
>
>Roger Nichols gets most of the credit for all the Steely Dan
>stuff, but as I understand it, Elliot Scheiner really did most
>of the basic tracks recording & also the mixdowns, so maybe
>he's really the unsung hero here. Not sure if this applies to
>Aja, as well, but nonetheless...
>
>...GO LISTEN TO "AJA"!!!!
>
>Neil"Neil" <OIUOIU@OIU.com> wrote:

>"Yeah, and we'll make 'em out of BUBINGA, too!" :D

I got yer Bubinga right here, bub!

Here's the only double neck I care about, and I don't want one...

http://www.juniorbrown.com/

DCLet's kill them too, while we're at it.

;o)

"John" <no@no.com> wrote in message news:46fe1cb3$1@linux...
>
> http://www.youtube.com/watch?v=eG6X-xtVaskLet'sNow.....if we can just get it to run win XP

;oD

"DC" <dc@spammersinhell.com> wrote in message news:46fdcc30$1@linux...
>
> "James McCloskey" <excelsm@hotmail.com> wrote:
>
>>Oh noooooooooooooooo!!!!!!!!! : 0
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> Ahhh shaddup!
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> http://cgi.ebay.com/ws/eBayISAPI.dll?ViewItem&item=33016
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I don't know about the guitar, but the name sounds like something that might
need to be treated with penicillin."DC" <dc@spammersinhell.com> wrote:
>
>"Neil" <OIUOIU@OIU.com> wrote:
>
>>"Yeah, and we'll make 'em out of BUBINGA, too!"   :D
>
>
>I got yer Bubinga right here, bub!
>
>Here's the only double neck I care about, and I don't want one...
>
>http://www.juniorbrown.com/
>
>DC
>

Junior Brown rocks! Well you know what I mean."DJ" <animix _ at _ animas _ dot _ net> wrote:
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>news:46fdf969$1@linux...
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>

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>
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Lol!"Don Nafe" <dnafe@magma.ca> wrote in message news:46fd51da$1@linux...
>
> "LaMont" <jjdpro@gmail.com> wrote in message news:46fd2b2e$1@linux...
>>
>> Why? CD Architect is still one the best at what it does in the business..
>
>
>
> Just looking at options.
>
>

AH HA!!!.....so you "need" it like I "need" one of these

 http://www.youtube.com/watch?v=RrOfYIvLS50

;o)"Neil" <OIUOIU@OIU.com> wrote:
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>"W. Mark Wilson" <xpam_mark@avidrecording> wrote:
>
>>Double's make great pictures and shoulder injuries and that's
>>about it.
>
>LOL! Maybe the money behind this new line is a group of
>chiropractors.
>
>"Yeah, and we'll make 'em out of BUBINGA, too!"   :D
>
>Neil

Lol!"James McCloskey" <excelsm@hotmail.com> wrote:
>
>Hey guys, I'm wondering what you guys think of this product? Do you think
>there is a real market for this kind of guitar? Would you buy something
>like this? What do you think you would be willing to pay if the quality
>was good?
>
>Thanks for your opinions in advance.
>
> http://cgi.ebay.com/ws/eBayISAPI.dll?ViewItem&item=33016
9549217&ssPageName=ADME:B:EF:US:12
>

Thanks guys for all your posts!arrgghhh!!!!But seriously, do oyu guys realize that we now have a player
(iPod Classic 160G) that can hold more than 200 CD's of
totally uncompressed audio?

There is now no reason ro not get one. None at all.

Also, there is really no reason for CD's anymore, other than
the fact that to get uncompressed audio down most of the
isp's we use would simply take too long. That will change too.

Amazing.

DCDamn, you ain't kiddin' about the slow download!! My 'puter's
been at it for almost two hours now & it says there's still an
estimated 53 minutes left!

Maybe I should get off of dial-up once & for all?

(j/k, I have DSL)

Neil

"Gene Lennon" <glennon@NOSPmyrealbox.com> wrote:
>
>Yellow Tools is giving away a free version of their sampler, Independence.
>It is a fully functioning version with no restrictions except it can't import
>WAV (or other) sound files and it has a 2Gig subset of the complete library.
>Based on "Culture" and some of the Yellow Tools Giga samples I have used
>it should be well worth the slow downloads.
> http://www.yellowtools.us/cp21/cms/index.php?id=399&show news=527
>
>Gene
>This was the first CD I bought, after wearing out SEVERAL
copies of Vinyl.

Love it.

David.

Neil wrote:
> ..is the fuckin' best recording EVER! OK, well, let's back off
> a bit & say it could POSSIBLY be the best fucking recording
> EVER. At the very least, it's definitely a candidate, yes? I'm
> not talking about the whole album (although the whole
> album is great, really, as everyone knows), I'm talking about
> the seven-minute-and-fifty-five-second title cut that was

> nailed on the 2nd take with no edits on the basic tracks.
>
> Man, I'm sitting here listening to it, and I think that if
> there was a way you could compare the best possible recordings
> of all time from ANY genre and digest them down to some kind of
> quantifiable "best sound/best playing/best execution/best
> mix/best mastering" kind of combination and vote on it, "Aja"
> just might win.
>
> Freakin' GREAT sounds on every single instrument in there, as
> pertains to the genre being recorded, NOTHING comes across as
> abrasive, yet it's not so homogenized as to not grab your
> attention, either (we've all heard mixes like that, right?).
>
> Although, I must say "My Old School" rocks almost as hot in
> that same voting contest. :D
>
> (Yes, I'm currently in a Steely Dan mode - gotta get some
> breathing space from the numetal stuff i've been spending so
> much time on lately! lol)
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> Roger Nichols gets most of the credit for all the Steely Dan
> stuff, but as I understand it, Elliot Scheiner really did most
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> Aja, as well, but nonetheless...
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> ...GO LISTEN TO "AJA"!!!!
>
> Neil"James McCloskey" <excelsm@hotmail.com> wrote in message
news:46fe8c7f$1@linux...
>
> "DC" <dc@spammersinhell.com> wrote:
>>
>>"Neil" <OIUOIU@OIU.com> wrote:
>>
>>>"Yeah, and we'll make 'em out of BUBINGA, too!" :D
>>
>>
>>I got yer Bubinga right here, bub!
>>
>>Here's the only double neck I care about, and I don't want one...
>>
>>http://www.juniorbrown.com/
>>
>>DC
>>
>

> Junior Brown rocks! Well you know what I mean.
>

A friend of mine (and my former engineering mentor when I was just geting started with this stuff) works with him a lot. I sat in on a couple of JB sessions a while back.
 http://www.sydec.be/About/POS/Details/ID/5f3d9117-9a37-403c- b4cf-cceeb6644d65/Thanks, I needed that...I'm going to pull it out listen to it right now.
Rod
"Neil" <OIUOI@OIU.com> wrote:
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>a bit & say it could POSSIBLY be the best fucking recording
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>he's really the unsung hero here. Not sure if this applies to
>Aja, as well, but nonetheless...
>
>...GO LISTEN TO "AJA"!!!!
>
>NeilUs guitar players are such slaves to tradition that things like this, even when brilliantly conceived and well executed, tend to fall on their butts.

Most guys who find the need for both Tele and LP sounds revel in the coolness of having two or three or more guitars onstage and (if their really cool) a guitar tech to hand the right guitar, perfectly in tune) to them at the right moment. Guys like me (journeyman hacks) tend to favor one guitar with trick wiring that can emulate both sounds. More open-minded guys find the solution in the Roland and Line 6 modeling systems. I have a PRS (w/ trick wiring, but not so much for Tele sounds) and a very nice Tele clone and I have been known to take both to gigs. I'm getting too old to carry all that gear around (I also drag a tube amp and 2-12 cabinet around a lot of the time) but I don't play often these days and I just like playing my guitars!

Another problem w/ the 2 neck idea for me: The right and left hand geometry is pretty critical for me to play my best 9such as it is) and the 2 neck idea forces the player to choose between having one right and the other too high or low or compromising on and in between height. I woudn't want to deal with that. It also looks to heavy for me...

Gantt

"James McCloskey" <excelsm@hotmail.com> wrote:
>
>Hey guys, I'm wondering what you guys think of this product? Do you think
>there is a real market for this kind of guitar? Would you buy something
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>Thanks for your opinions in advance.
>
> http://cgi.ebay.com/ws/eBayISAPI.dll?ViewItem&item=33016
9549217&ssPageName=ADME:B:EF:US:12
>Ahh Aja..I remember it well. I was fiftenn and a older friend of mine thought I should hear something that would challenge a that time musicianship..Being somewhat of a hot-shot player, Ithought bring it on.lOL!

I played that album(Still own today) everyday ftom the moment I got it(15 years old).

What amazed me was that I could not figure ou the chords. At that time, I had prided myself, on being able to comp any chord I heard. Threw me for a complete loop. My ego was quickly put into check.
But, it made me strive to become a more rounded musician. It to me 5 years to finally play the intro to Decond Blues (20years old).
Once you figure out deacon blues, the rest of the album ell into place. Next up Peg, Black Cow and so on.

Being schooled in Jazz at the time, Aja spoke to a lot us who were being mentored by the bebop cats, that You can put a new spin on Jazz. Revoice

the chords, and play some very inspired, yet very imporvised solos to boot.

I'm an offcial Aja junkie and very porud of it because it rasied the my game..

"Neil" <OIUOI@OIU.com> wrote:
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I have a question. I need to be able to switch sample rates between projects
while not having to physicaly repatch any gear. I'll be using a MADI PCI/ADI
648 but I'm sure the S/MUX would apply to an HDSP series card of any type..
When changing from a 44.1 KHz projects to a 88.2 KHz projects do the HDSP

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ADAT 1/O 2
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channels 13 & 14 combine to create channel 7
channels 15 & 16 combine to create channel 8

or is it.....

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channels 14 & 16 combine to create channel 8

or is it

ADAT I/O channels 1 & 2 combine with ADAT I/O 9 & 10 to create channel 1 & 2
ADAT I/O channels 3 & 4 combine with ADAT I/O 11 & 12 to create channel 3 & 4
ADAT I/O channels 5 & 6 combine with ADAT I/O 13 & 14 to create channel 5 & 6
ADAT I/O channels 7 & 8 combine with ADAT I/O 15 & 16 to create channel 7 & 8

I just need to know ahead of time before I jump into this. I'm going to be having to do this install without any assistance and I don't want to spend a whole day getting this wired in and then end up with a total cluster**** in my signal routing.I think it's simply that 5, 6, 7 & 8 drop off when at 88.2.

IOW, on my Multifascia, when at 88.2 and selecting an input from lightpipe, channel 2 is still channel 2, 3 is still 3, and 4 is still 4 - there's just no 5-8.

It's just a matter of each channel taking up twice the bandwidth.

Neil

"DJ" <animix _ at _ animas _ dot _ net> wrote:
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lot,
>I have a question. I need to be able to switch sample rates between projects

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would make sense. In my particular situation, that's what I need to be sure
of.

Thanks,

Deej

"Neil" <IOUIOU@OIU.com> wrote in message news:46febbab$1@linux...
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> 3 &
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> in
>>my signal routing.
>>
>>
>AFAIK, that's not how it works (combining); the protocol just
assigns a certain amount of bandwidth to each channel,
depending on the samplerate.

You won't need to repatch anything when switching from an 88.2k
project to a 44.1k project, channels 5-8 on each lightpipe just
won't exist at 88.2k

Neil

"DJ" <animix _ at _ animas _ dot _ net> wrote:
>I understand, but does this mean that 5 is combining signals with 1, 6 with

>2, 7 with 3 and 8 with 4? (a scenario that I didn't think of, BTW) That

>would make sense. In my particular situation, that's what I need to be sure

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>"Neil" <IOUIOU@OIU.com> wrote in message news:46febbab$1@linux...
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>>>
>>>
>>
>
>James,

I just stopped hurling!

Seriously, some might find it interesting. Personally, I find that most of
the songs that I play are suited to either a single coil, humbucker, or P90-style
pickup. Consequently, I take a strat/tele, my PRS Soapbar II, and some type
of humbucker-equipped axe (LP, SG, ES-335, etc.) with me to every gig.

My $ .02!

"James McCloskey" <excelsm@hotmail.com> wrote:
>
>Hey guys, I'm wondering what you guys think of this product? Do you think
>there is a real market for this kind of guitar? Would you buy something
>like this? What do you think you would be willing to pay if the quality
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> http://cgi.ebay.com/ws/eBayISAPI.dll?ViewItem&item=33016
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>Hi DJ,
Neil is correct in the way he describes it.
Also it will be more flexible if you have the MADI card be master if you
are going to be switching sample rates allot.
Did you ever find out if the used ADI-648 you got is a recent enough one
to have the MIDI Remote support?
It makes configuring the front panel very easy if the unit is not in
arms/visual reach.
Also don't forget that MADI's I/O is also cut in half at 88.2/96k like
ADAT. So 32 channels instead of 64.

Chris

DJ wrote:
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> 2, 7 with 3 and 8 with 4? (a scenario that I didn't think of, BTW) That
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> Thanks,
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> Deej
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--
Chris Ludwig
ADK
chrisl@adkproaudio.com <mailto:chrisl@adkproaudio.com>
www.adkproaudio.com <http://www.adkproaudio.com/>
(859) 635-5762"Neil" <OIUOIU@OI.com> wrote:
>You won't need to repatch anything when switching from an 88.2k
>project to a 44.1k project, channels 5-8 on each lightpipe just
>won't exist at 88.2k

Oh, one thing - you won't have to repatch anything, but you
WILL probably have to refresh your insies & outsies in Cubase
in the device menu (Devices/Device Setup/VST Audiobay/VST
Inputs & VST Outputs, then select "reset" off to the bottom
right of that window for ins then outs) each time you switch
back & forth from a 44.1 project to an 88.2 one & vice-versa.

Neil"Chris Ludwig" <chrisl@adkproaudio.com> wrote in message
news:46fecf15@linux...
> Hi DJ,
> Neil is correct in the way he describes it.
> Also it will be more flexible if you have the MADI card be master if you
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> arms/visual reach.
> Also don't forget that MADI's I/O is also cut in half at 88.2/96k like
> ADAT. So 32 channels instead of 64.
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> Chris

I'm getting ready to wire up a lot of RME hardware. I'm configuring two DAWs
Master/Slave, both running Cubase 4. In my Master DAW, I've got an AES
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down/upsampled for use as send/FX in Cubase by using the ADI 8-DD units.

The other 4 x AES 32 channels will be interfacing with converters and
processors that are 88.2KHz capable.

As for the MADI/ADI 648, what I would like to accomplish is this:

ADAT 1 I/O patched to the ADI 4-DD
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.....or, does ADAT 2 I/O just disappear when switching to 88.2KHz?

I don't want to spend a whole day getting this wired in and suddenly have
signals from the left channel routed to/from one hardware device and signals
on the right channel routed to/from another hardware device."Neil" <OIUOI@OIU.com> wrote:
>At the very least, it's definitely a candidate, yes?

yes"DJ" <animix _ at _ animas _ dot _ net> wrote:
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Use 3&4 instead of 5&6... follow this pattern all the way
through the rest of the gear & you should be OK.

Chris, chime in - this is correct, yes?

Deej, if you're not cure, you can always hook up ONE unit using
1&2 and 5&6, then flip over to 88.2k & see if 5&6 now show up
on 3&4... I'm 99.9997% certain that's not the way it works,
however.

NeilI guess I could ask this another way.....

If I connected three ADI4-DD units to ADAT ports 1, 2, and 3 and I set the
clock rate to 88.2, would I get an AES stereo I/O on 1 & 2 and 5 & 6 of the
AES input/output of each of the three ADI 4-DD units?

thanks,

Deej

"DJ" <animix _ at _ animas _ dot _ net> wrote in message
news:46fee642@linux...
>
> "Chris Ludwig" <chrisl@adkproaudio.com> wrote in message
> news:46fecf15@linux...
>> Hi DJ,
>> Neil is correct in the way he describes it.
>> Also it will be more flexible if you have the MADI card be master if you
>> are going to be switching sample rates allot.
>> Did you ever find out if the used ADI-648 you got is a recent enough one
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>> It makes configuring the front panel very easy if the unit is not in

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>
> I don't want to spend a whole day getting this wired in and suddenly have
> signals from the left channel routed to/from one hardware device and
> signals on the right channel routed to/from another hardware device.
>
>That's what's going to happen tonight.

Thanks,

Deej

"Neil" <OIUOI@OIU.com> wrote in message news:46feedef$1@linux...
>
> "DJ" <animix _ at _ animas _ dot _ net> wrote:
>>As for the MADI/ADI 648, what I would like to accomplish is this:
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> on 3&4... I'm 99.9997% certain that's not the way it works,
> however.
>
> NeilThis is a multi-part message in MIME format.

------=_NextPart_000_00E0_01C802DB.DFDE99A0
Content-Type: text/plain;
 charset="iso-8859-1"
Content-Transfer-Encoding: quoted-printable

I double that yes.

  "steve the artguy" <artguy@somethingorother.net> wrote in message =
news:46feedb9$1@linux...

  "Neil" <OIUOI@OIU.com> wrote:
  >At the very least, it's definitely a candidate, yes?

  yes

I choose Polesoft Lockspam to fight spam, and you?
http://www.polesoft.com/refer.html
------=_NextPart_000_00E0_01C802DB.DFDE99A0
Content-Type: text/html;
 charset="iso-8859-1"
Content-Transfer-Encoding: quoted-printable

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charset=3Diso-8859-1">
<META content=3D"MSHTML 6.00.2800.1400" name=3DGENERATOR>

<STYLE></STYLE>
</HEAD>
<BODY bgColor=3D#ffffff>
<DIV><FONT face=3DArial size=3D2>I double that yes.</FONT></DIV>
<DIV><FONT face=3DArial size=3D2></FONT> </DIV>
<DIV> </DIV>
<BLOCKQUOTE=20
style=3D"PADDING-RIGHT: 0px; PADDING-LEFT: 5px; MARGIN-LEFT: 5px; =
BORDER-LEFT: #000000 2px solid; MARGIN-RIGHT: 0px">
  <DIV>"steve the artguy" <A=20
  =
href=3D"mailto:artguy@somethingorother.net">artguy@somethingorother.net</=
A>>=20
  wrote in message <A=20
  =
href=3D"news:46feedb9$1@linux">news:46feedb9$1@linux</A>...</DIV>
"Nei=
l"=20
  <<A href=3D"mailto:OIUOI@OIU.com">OIUOI@OIU.com</A>> =
wrote:
>At the=20
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yes</BLOCKQUOTE>
<DIV><FONT size=3D2>

I choose Polesoft Lockspam to fight spam, =
and=20
you?
<A=20
href=3D"http://www.polesoft.com/refer.html">http://www.polesoft.c

---

Subject: Re: OUTAGE - Sorry, it's AMAZINGLY windy here...
Posted by Kim on Wed, 01 Aug 2007 01:53:50 GMT
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could POSSIBLY be the best fucking recording
EVER. At the very least, it's definitely a candidate, yes? I'm
not talking about the whole album (although the whole
album is great, really, as everyone knows), I'm talking about
the seven-minute-and-fifty-five-second title cut that was
nailed on the 2nd take with no edits on the basic tracks.

Man, I'm sitting here listening to it, and I think that if
there was a way you could compare the best possible recordings
of all time from ANY genre and digest them down to some kind of
quantifiable "best sound/best playing/best execution/best

Subject: Re: OUTAGE - Sorry, it's AMAZINGLY windy here...
Posted by Kim on Wed, 01 Aug 2007 03:22:44 GMT
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om/refer=
..html</A>   </FONT></DIV></BODY ></HTML>

------=_NextPart_000_00E0_01C802DB.DFDE99A0--I'm thinking this place (Meson Sevilla - http://www.mesonsevilla.com/media/websitemesonsevilla.html) might be nice. Tapas & sangria for the Paris faithful?

Let's meet as the show ends about 6pm just outside the main hall entrance, then we can hoof or cab it up to the restaurant. If you want to just meet at the restaurant, plan to get there about 6:30. Reservation will be in the name of Paris!

Right now, I am counting:

Me
Ron (my guest)
Sakis
Chris
Morgan
Gene
Brian
Kerry
Tyrone

Am I missing anyone? If you are planning to attend and bring a guest(s), please let me know so I can make a reservation.

- Paul Artola
  Ellicott City, MarylandIf I'm not mistaken, I believe Nichols mastered that record directly through an Ibanez SDR1000 straight to master reel.

WMW

"Rod Lincoln" <rlincoln@nospam.kc.rr.com> wrote in message news:46fe9ab8$1@linux...
>
> Thanks, I needed that...I'm going to pull it out listen to it right now.
> Rod
> "Neil" <OIUOI@OIU.com> wrote:
>>
>>..is the fuckin' best recording EVER! OK, well, let's back off
>>a bit & say it could POSSIBLY be the best fucking recording
>>EVER. At the very least, it's definitely

Subject: Re: OUTAGE - Sorry, it's AMAZINGLY windy here...
Posted by JeffH on Wed, 01 Aug 2007 04:42:40 GMT

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>>nailed on the 2nd take with no edits on the basic tracks.
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>>there was a way you could compare the best possible recordings
>>of all time from ANY genre and digest them down to some kind of
>>quantifiable "best sound/best playing/best execution/best
>>mix/best mastering" kind of combination and vote on it, "Aja"
>>just might win.
>>
>>Freakin' GREAT sounds on every single instrument in there, as
>>pertains to the genre being recorded, NOTHING comes across as
>>abrasive, yet it's not so homogenized as to not grab your
>>attention, either (we've all heard mixes like that, right?).
>>
>>Although, I must say "My Old School" rocks almost as hot in
>>that same voting contest. :D
>>
>>(Yes, I'm currently in a Steely Dan mode - gotta get some
>>breathing space from the numetal stuff i've been spending so
>>much time on lately! lol)
>>
>>Roger Nichols gets most of the credit for all the Steely Dan
>>stuff, but as I understand it, Elliot Scheiner really did most
>>of the basic tracks recording & also the mixdowns, so maybe
>>he's really the unsung hero here. Not sure if this applies to
>>Aja, as well, but nonetheless...
>>
>>...GO LISTEN TO "AJA"!!!!
>>
>>Neil
>HI Paul,
That sounds awesome to me.

Chris

Paul Artola wrote:
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> http://www.mesonsevilla.com/media/websitemesonsevilla.html) might be
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Subject: Re: OUTAGE - Sorry, it's AMAZINGLY windy here...
Posted by audioguy_editout_ on Wed, 01 Aug 2007 04:45:45 GMT
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br />
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>
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>
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> guest(s), please let me know so I can make a reservation.
>
> - Paul Artola
> Ellicott City, Maryland
>
>
>
>

--
Chris Ludwig
ADK
chrisl@adkproaudio.com <mailto:chrisl@adkproaudio.com>
www.adkproaudio.com <http://www.adkproaudio.com/>
(859) 635-5762DJ wrote:
> "Chris Ludwig" <chrisl@adkproaudio.com> wrote in message
> news:46fecf15@linux...
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>>Did you ever find out if the used ADI-648 you got is a recent enough one
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Subject: Re: OUTAGE - Sorry, it's AMAZINGLY windy here...
Posted by Kim on Wed, 01 Aug 2007 05:23:46 GMT
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Deej,

What's the purpose of the second DAW? If you're passing audio back and
forth for processing, this might be a great scenario for Wormmhole.
Pass the audio ether

Subject: Re: OUTAGE - Sorry, it's AMAZINGLY windy here...
Posted by Kim on Wed, 01 Aug 2007 05:25:28 GMT
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net and use the ADAT sync on the multiface for
synching the two.

Just a thought,

Jeff> Deej,
>
> What's the purpose of the second DAW? If you're passing audio back and
> forth for processing, this might be a great scenario for Wormmhole. Pass
> the audio ethernet and use the ADAT sync on the multiface for synching
> the two.

>
> Just a thought,
>
> Jeff

Scratch that...appears Wormhole went the way of Paris...ok now you have
to have it or you'll.....

;-)

JeffPlus.....ADAT sync is only one way on HDSP cards..input. I could use a
BRC I suppose. The Steiny Systemline works nicely on two comps anyway.

I've got everything sync'ed up to external WC and I can cycle through
everything from 44.1 thru 96k and everything syncs up fine but I'm hearing
no stereo output signal in cubase. I'm sure this is due to the ADI 648
routing matrix being scrambled. Time to RTFM to get my head around this.
Lots to learn about this beast. I'm pretty stoked about it though. I think
that I may have finally found an audio system that will make my bizarre and
twisted dreams come true.

We'll know more next week. The RME BOB that I need to interface with the AES
32 card hasn't arrived here yet.

Deej

"Jeff Hoover" <jkhoover@excite.com> wrote in message news:46ff33e3@linux...
>
>> Deej,
>>
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>> forth for processing, this might be a great scenario for Wormmhole. Pass
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>> Just a thought,
>>
>> Jeff
>
> Scratch that...appears Wormhole went the way of Paris...ok now you have to
> have it or you'll.....
>
>
>
>

> ;-)
>
> Jeffwhere are you now BTW?

"Jeff Hoover" <jkhoover@excite.com> wrote in message news:46ff3218@linux...
> DJ wrote:
>> "Chris Ludwig" <chrisl@adkproaudio.com> wrote in message
>> news:46fecf15@linux...
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Subject: Re: OUTAGE - Sorry, it's AMAZINGLY windy here...
Posted by Martin Harrington on Wed, 01 Aug 2007 05:29:31 GMT
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>> & 2 of the ADI 4-DD when used at 88.2 sample rates-correct?
>>
>> ADI 4-DD I/O 5 & 6 patched into a Mytek Stereo AD/DA units
>>
>> ADI 4-DD I/O 7 & 8 not used because they will be combined with Channels 5
>> & 6 of the ADI 4-DD when used at 88.2 sample rates-correct?
>>
>> ADAT 2 I/O patched to the ADAT I/O of the Multiface Slave, which when
>> running both DAWs at 88.2KHz.
>>
>> I just want to be sure that, when configured in this manner and using
>> 88.2 sample rates on the master and slave DAWS, the various devices would
>> show up in Totalmix and Cubase VST I/O as follows:
>>
>> HDSP MADI 1 I/O would be left channel of ADI 4-DD channel 1 & 2 which are
>> interfacing with the Slave DAW output L and the Benchmark DAC-1 input L
>> HDSP MADI 2 I/O would be right channel of ADI 4-DD channel 1 & 2 which
>> are interfacing with the Slave DAW output R and the Benchmark DAC-1 input
>> R
>>
>> HDSP MADI 3 I/O would be left channel of ADI 4-DD channel 5 & 6 which are
>> interfacing with the Mytek Stereo 96 AD/DA L
>> HDSP MADI 4 I/O would be right channel of ADI 4-DD channel 5 & 6 which
>> are interfacing with the Mytek Stereo 96 AD/DA R
>>
>> HDSP MADI 5 I/O would be left channel signal from Multiface ADAT I/O 1 &
>> 2 on slave DAW
>> HDSP MADI 6 I/O would be right channel signal from Multiface ADAT I/O 1 &
>> 2 on slave DAW
>>
>> HDSP MADI 7 I/O would be left channel signal from Multiface ADAT I/O 3 &
>> 4 on slave DAW
>> HDSP MADI 8 I/O would be right channel signal from Multiface ADAT I/O 3 &
>> 4 on slave DAW
>>
>>
>> .....or, does ADAT 2 I/O just disappear when switching to 88.2KHz?
>>
>> I don't want to spend a whole day getting this wired in and suddenly have
>> signals from the left channel routed to/from one hardware device and
>> signals on the right channel routed to/from another hardware device.
>>
>>

> Deej,
>
> What's the purpose of the

Subject: Re: OUTAGE - Sorry, it's AMAZINGLY windy here...
Posted by JeffH on Wed, 01 Aug 2007 08:08:43 GMT
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second DAW? If you're passing audio back and
> forth for processing, this might be a great scenario for Wormmhole. Pass
> the audio ethernet and use the ADAT sync on the multiface for synching the
> two.
>
> Just a thought,
>
> JeffMan.....I wish I could come. I'd like to dive naked into a giant cauldron
of paella ;

;o)

"Chris Ludwig" <chrisl@adkproaudio.com> wrote in message
news:46ff30eb@linux...
> HI Paul,
> That sounds awesome to me.
>
> Chris
>
> Paul Artola wrote:
>> I'm thinking this place (Meson Sevilla -
>> http://www.mesonsevilla.com/media/websitemesonsevilla.html) might be
>> nice. Tapas & sangria for the Paris faithful?
>>
>> Let's meet as the show ends about 6pm just outside the main hall
>> entrance, then we can hoof or cab it up to the restaurant. If you want
>> to just meet at the restaurant, plan to get there about 6:30.
>> Reservation will be in the name of Paris!
>>
>> Right now, I am counting:
>>
>> Me
>> Ron (my guest)
>> Sakis
>> Chris
>> Morgan
>> Gene
>> Brian
>> Kerry

>> Tyrone
>>
>> Am I missing anyone? If you are planning to attend and bring a
>> guest(s), please let me know so I can make a reservation.
>>
>> - Paul Artola
>> Ellicott City, Maryland
>>
>>
>>
>>
>
> --
> Chris Ludwig
> ADK
> chrisl@adkproaudio.com <mailto:chrisl@adkproaudio.com>
> www.adkproaudio.com <http://www.adkproaudio.com/>
> (859) 635-5762Today I received the second comment ever on my blog where I post my old tunes,
and the very few new tunes I am working on. I went booking over, and logged
in to read the comment. It was from a guy looking for a manual for a tascam
488 :-)

ChuckBoy am I glad I ripped my entire CD collection to uncompressed .wav files
instead of some other format.

I can't imagine ripping 400 CDs again!

Chuck
"DC" <dc@spammersinhell.com> wrote:
>
>But seriously, do oyu guys realize that we now have a player
>(iPod Classic 160G) that can hold more than 200 CD's of
>totally uncompressed audio?
>
>There is now no reason ro not get one. None at all.
>
>Also, there is really no reason for CD's anymore, other than
>the fact that to get uncompressed audio down most of the
>isp's we use would simply take too long. That will change too.
>
>Amazing.
>
>DC
>Good Morning Chuck,
Where is your Blog ?

Morgan :)

"chuck duffy" <c@c.com> wrote in message news:46ff8f68$1@linux...
>
> Today I received the second comment ever on my blog where I post my old
> tunes,
> and the very few new tunes I am working on. I went booking over, and
> logged
> in to read the comment. It was from a guy looking for a manual for a
> tascam
> 488 :-)
>
> ChuckI recently recorded and mixed the Lennon & McCartney gem, In My Life. I
would appreciate mix input from my amigos. Anything too loud, too soft,
EQ