
Subject: OT: So far little snow in NJ this winter.
Posted by [gene lennon](#) on Sat, 07 Jan 2006 23:32:10 GMT
[View Forum Message](#) <> [Reply to Message](#)

r />
> P.S. I was totally joking about the quick mix thing;)
>
>
>>
>>"James McCloskey" <excelsm@hotmail.com> wrote in message
>>news:43d559b6\$1@linux...
>>>
>>> You guys can't see the beauty in this? What a time saver! This is
>>> perfect
>>> for the client that never pay on time and wants a ruff mix. You just
> say
>>> Ok, press a button and you've got a ruff mix. Then you say, when you
> have
>>> the balance I'll give you a real mix;)
>>>
>>> James
>>>
>>>
>>> "Kim" <hiddensounds@hotmail.com> wrote:
>>>>
>>>>
>>>>LOL! Strewth! Now they've done it. ;o)
>>>>
>>>>"DJ" <animix_spam-this-ahole_@animas.net> wrote:
>>>>>Oh God.....
>>>>>
>>>>>http://www.rndigital.com/AutoMix.html
>>>>>
>>>>>;oP
>>>>>
>>>>>
>>>>>"EK Sound" <spamnot.info@eksoundNO.com> wrote in message
>>>>>news:43d52d76\$1@linux...
>>>>>> Hey,
>>>>>>
>>>>>> Found this in "the basement" at NAMM...
>>>>>>
>>>>>> http://www.rndigital.com/
>>>>>>
>>>>>> Not too much info at the moment...
>>>>>>
>>>>>> David.
>>>>>>

>>>>>

>>>>

>>

>>

>http://www.sonicfinger.com/VHSM.jpg ... gotta be my favorite out of that batch.

AA

"DJ" <animix_spam-this-ahole_@animas.net> wrote in message
news:43d5b487@linux...

> Here's another good plugin bundle that would complement yours.

>

> http://www.sonicfinger.com/

>

>

> "EK Sound" <spamnot.info@eksoundNO.com> wrote in message

> news:43d52d76\$1@linux...

>> Hey,

>>

>> Found this in "the basement" at NAMM...

>>

>> http://www.rndigital.com/

>>

>> Not too much info at the moment...

>>

>> David.

>

>This is a multi-part message in MIME format.

-----=_NextPart_000_00AE_01C62087.0A9F6590

Content-Type: text/plain;

charset="iso-8859-1"

Content-Transfer-Encoding: quoted-printable

Big Time!

Kim is it.

Tom

"Rod Lincoln" <rlincoln@nospam.kc.rr.com> wrote in message =
news:43d58bb5\$1@linux...

Yeah, no kidd'n. Thanks dude

"Rob Arsenault" <mani2@nbnet.nb.ca> wrote:

>Here...Here... I'll second that.!!

>

>Rob_A

>

>"Mike R." <emarenot@yahoo.com> wrote in message =
news:43d55d4f@linux...
>> Hey Kim,
>> I've been getting some help here lately. This wouldn't have =
happened
>without
>> your installation and upkeep of the servers. You've been a most =
cool
>> successor to Mr. Chuck..
>> Thanks for your time and effort.
>> MR
>>
>>
>
>

-----=_NextPart_000_00AE_01C62087.0A9F6590

Content-Type: text/html;

charset="iso-8859-1"

Content-Transfer-Encoding: quoted-printable

<!DOCTYPE HTML PUBLIC "-//W3C//DTD HTML 4.0 Transitional//EN">

<HTML><HEAD>

<META http-equiv=3DContent-Type content=3D"text/html; =

charset=3Diso-8859-1">

<META content=3D"MSHTML 6.00.2800.1400" name=3DGENERATOR>

<STYLE></STYLE>

</HEAD>

<BODY bgColor=3D#ffffff>

<DIV>Big Time!</DIV>

<DIV> </DIV>

<DIV>Kim is it.</DIV>

<DIV>Tom</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>"Rod Lincoln" <<A=20

=

href=3D"mailto:rlincoln@nospam.kc.rr.com">rlincoln@nospam.kc.rr.com&g=

t;=20

wrote in message <A=20

=

href=3D"news:43d58bb5\$1@linux">news:43d58bb5\$1@linux...</DIV>
Yeah=

, no=20

kidd'n. Thanks dude
"Rob Arsenault" <<A=20

href=3D"mailto:mani2@nbnet.nb.ca">mani2@nbnet.nb.ca>=20

wrote:
>Here...Here... I'll second=20

that.!!
>
>Rob_A
>
> "Mike R." <<A=20
 href=3D"mailto:emarenot@yahoo.com">emarenot@yahoo.com> wrote in =
 message=20
 news:43d55d4f@linux...
>> =
 Hey=20
 Kim,
>> I've been getting some help here lately. This =
 wouldn't have=20
 happened
>>without
>> your installation and upkeep of =
 the=20
 servers. You've been a most cool
>> successor to Mr.=20
 Chuck..
>> Thanks for your time and effort.
>>=20
 =
 MR
>>
>>
>>
>>
></BLOCKQUOTE></BODY></HTML>=

-----=_NextPart_000_00AE_01C62087.0A9F6590--Do the UAD boards have latency? I can
 remove the room latency by
 wearing headphones but I still need to remove the plugin latency for
 tracking live with plugs enabled. Does a board like a UAD have latency too?

erlilo wrote:

> ...and it's being more and more pricey to get these new late NCY filled
 > vintage effects. Roland Space Echo will cost 249\$. Their next plugins will
 > be Neve/AMS vintage signal processor products like the Neve 1073 Equalizer
 > and maybe the old, wonderful AMS reverb unit???....
 >
 > Erling
 >
 >
 > "DJ" <animix_spam-this-ahole_@animas.net> skrev i melding
 > news:43d5103f\$1@linux...
 >
 >>Yes, but it will still have a carbuerator in order to retain tht
 >>*vintage*
 >>vibe.
 >>
 >>"gene lennon" <glennon@NOSPmyrealbox.com> wrote in message
 >>news:43d50b7b\$1@linux...
 >>
 >>>"JB" <no@mail.com> wrote:
 >>>
 >>>>>NO DAW can compensate for plugin latency live on input.
 >>>>>gene
 >>>>
 >>>>Actually, I heard that the new UAD-2 card will have a built in Flux
 >>>>Capacitor that will allow recording of a .wav file before it is actually
 >>>
 >>>>played. This should allow not only 0 latency recording, but NEGATIVE

>>
>>latency
>>
>>>>recording! Unfortunately, it will only be able to play back Chuck Berry
>>>
>>>>riffs.
>>>>
>>>>Oh, the joy of technology!
>>>>
>>>
>>>My information must be incorrect. I heard the UAD-2 ran on fuel cells.
>>>g
>>
>>
>
>you guys sure have expensive taste. wow

Gary Flanigan wrote:

> "DJ" <animix_spam-this-ahole_@animas.net> wrote:

>

>>Find a cheap Avalon 737.

>

>

> I hear that the Manley VoxBox is a great front end for bass :)yeah, but that will necessitate you wearing that damn speedo...you may have to put the feelings of friends, family and clients first.

On Mon, 23 Jan 2006 14:25:52 -0500, John <no@no.com> wrote:

>I'll just heat up my room to 100C. At 212F/100C the speed is 1268 feet
>per second.

>

>TCB wrote:

>> Sound travels thorough sea level air at 1100 ft/second. There are a thousand
>> milliseconds in a second. Each foot you move away from your speakers is a
>> millisecond of latency. The latency in DSP systems is fine for tracking.
>> The compensation in native systems is great.

>>

>> Pick your poison or buy a console and tape deck.

>>

>> TCB

>>

>> John <no@no.com> wrote:

>>

>>>latency is a bummer. Is that why everyone gets a UAD, cuz pcs are still

>>

>>

>>>to slow to have no latency?

>>>Thanks
>>>
>>>TCB wrote:
>>>
>>>>Usually you have to pick your poison. DSP based DAWs (PARIS, PT, etc.)
>>
>> almost
>>
>>>>always have some very small amount of latency when playing back, but it's
>>>>exactly the same latency as the effect introduces when tracking. Native
>>
>> DAWs
>>
>>>>usually have sample accurate latency compensation at mixdown but more
>>
>> latency
>>
>>>>when tracking with an effect. I always suggest people track with cheapo
>>
>> hardware
>>
>>>>effects if they really need them when using a native DAW. But that's just
>>>>me.
>>>>
>>>>TCB
>>>>
>>>>John <no@no.com> wrote:
>>>>
>>>>
>>>>>Are there any Daws have plugin latency compensation AND allow you to
>>>>>hear the plugin while tracking.
>>>>
>>>>
>>did you try right clicking the files to see if they're marked read
only? if they are, un check them. have you tried the wav conversion
in the mac first?

On Mon, 23 Jan 2006 22:08:17 -0500, "Tom Bruhl" <arpeggio@comcast.net>
wrote:

>After every disk (we've burned over ten now) none of them
>sounded audio or showed an overview.
>I'm still trying everything in my power but I think the
>clients software is what's holding us back. He's
>using Panther/newest DP and newest Toast programs
>on a G4 laptop.
>

>Can someone sus out my Wavelab problem? I have an
>original 3.01 disk with a 3.03 patch disk too. Neither
>will allow it to start up in XP. Is this version just too old?
>When starting Wavelab it gets into the program then stops
>instantly with: "This application has failed to start because
>HypGui.dll was not found. Reinstalling the application may
>fix this problem." Needless to say it didn't.
>
>What is the solution?
>
>I haven't been frustrated in a long time. This is taking the cake.
>TIA,
>TomYou rock dude!
j

Tom Bruhl wrote:

> Big Time!
>
> Kim is it.
> Tom
>
>
> "Rod Lincoln" <rlincoln@nospam.kc.rr.com
> <mailto:rlincoln@nospam.kc.rr.com>> wrote in message
> news:43d58bb5\$1@linux...
>
> Yeah, no kidd'n. Thanks dude
> "Rob Arsenault" <mani2@nbnet.nb.ca <mailto:mani2@nbnet.nb.ca>> wrote:
> >Here...Here... I'll second that.!!
> >
> >Rob_A
> >
> >"Mike R." <emarenot@yahoo.com <mailto:emarenot@yahoo.com>> wrote
> in message news:43d55d4f@linux...
> >> Hey Kim,
> >> I've been getting some help here lately. This wouldn't have
> happened
> >without
> >> your installation and upkeep of the servers. You've been a
> most cool
> >> successor to Mr. Chuck..
> >> Thanks for your time and effort.
> >> MR
> >>
> >>
> >
> >

>The guarantee is funny, if your tune does not make the Billboard chart you get 80% back!

"DJ" <animix_spam-this-ahole_@animas.net> wrote:

>I was watching this one but right at the last minute, my pooper scooper
>broke and I have to budget for another one.

>

>"John Macy" <spamlessjohn@johnmacy.com> wrote in message

>news:43d5a906\$1@linux...

>>

>>

> <http://cgi.ebay.com/ws/eBayISAPI.dll?ViewItem&item=4817600091&ssPageName=ADM>

>E:B:EF:US:1

>

>And no bloated shipping charges! Nice pic too

"John Macy" <spamlessjohn@johnmacy.com> wrote in message

news:43d5a906\$1@linux...

>

>

<http://cgi.ebay.com/ws/eBayISAPI.dll?ViewItem&item=4817600091&ssPageName=ADM>

E:B:EF:US:1Hi Tom,

Do you have any third party VST plug ins?

If so move the VST plug ins directory. Usually it will be C:/Program

Files/Steinberg/VStplugins. Drag the whole directory to you desktop and

see if it starts.

Chris

Tom Bruhl wrote:

> After every disk (we've burned over ten now) none of them

> sounded audio or showed an overview.

> I'm still trying everything in my power but I think the

> clients software is what's holding us back. He's

> using Panther/newest DP and newest Toast programs

> on a G4 laptop.

>

> Can someone sus out my Wavelab problem? I have an

> original 3.01 disk with a 3.03 patch disk too. Neither

> will allow it to start up in XP. Is this version just too old?

> When starting Wavelab it gets into the program then stops

> instantly with: "This application has failed to start because

> HypGui.dll was not found. Reinstalling the application may

> fix this problem." Needless to say it didn't.

>
> What is the solution?
>
> I haven't been frustrated in a long time. This is taking the cake.
> TIA,
> Tom
>
>

--

Chris Ludwig

ADK

chrisl@adkproaudio.com <mailto:chrisl@adkproaudio.com>

www.adkproaudio.com <http://www.adkproaudio.com/>

(859) 635-5762 Well, he needs a little something for his trouble.

"cujo" <chris@nospamapplemanstudio.com> wrote in message
news:43d63fd3\$1@linux...

>
>
>
> The guarantee is funny, if your tune does not make the Billboard chart you
> get 80% back!
>
>
>
>
> "DJ" <animix_spam-this-ahole_@animas.net> wrote:
> > I was watching this one but right at the last minute, my pooper scooper
> > broke and I have to budget for another one.
> >
> > "John Macy" <spamlessjohn@johnmacy.com> wrote in message
> > news:43d5a906\$1@linux...
> >>
> >>
>
> http://cgi.ebay.com/ws/eBayISAPI.dll?ViewItem&item=48176 00091&ssPageName=AD
M
> > E:B:EF:US:1
> >
> > The only difference is that Roger Nichols was actually standing at the
booth... Hard to tell from that if it's actually a joke, or not...

David.

DJ wrote:

> Here's another good plugin bundle that would complement yours.

>
> <http://www.sonicfinger.com/>
>
>
> "EK Sound" <spamnot.info@eksoundNO.com> wrote in message
> news:43d52d76\$1@linux...
>
>>Hey,
>>
>>Found this in "the basement" at NAMM...
>>
>><http://www.rndigital.com/>
>>
>>Not too much info at the moment...
>>
>>David.
>
>
>Tom,

Have your friend try to burn a disc directly from the finder. On Panther, all he needs to do is insert a recordable cd into the internal drive. The cd will come up on his desktop. Drag the files to the cd icon (it might take a while for the files to copy over to the disc image). When done, double click the cd icon and click "Burn" on the upper right of the cd window. Maybe Toast is doing something screwy? I quit using it because of some troubles, but that was a few years ago.

Will the files from the disc he burned import back into DP? I mean are the burned files valid on his Mac? If so, and you can't get this fixed, mail me the disc and I'll see if I can batch convert them for you. Let me know if I can help.

Tony

"Tom Bruhl" <arpeggio@comcast.net> wrote in message news:43d59b80\$1@linux...
After every disk (we've burned over ten now) none of them sounded audio or showed an overview.
I'm still trying everything in my power but I think the clients software is what's holding us back. He's using Panther/newest DP and newest Toast programs on a G4 laptop.

Can someone sus out my Wavelab problem? I have an original 3.01 disk with a 3.03 patch disk too. Neither will allow it to start up in XP. Is this version just too old?

When starting Wavelab it gets into the program then stops instantly with: "This application has failed to start because HypGui.dll was not found. Reinstalling the application may fix this problem." Needless to say it didn't.

What is the solution?

I haven't been frustrated in a long time. This is taking the cake.

TIA,

TomJohn, you can finance it through PayPal for only \$4167.00 a month. What are you waiting for? ;>)

Tony

"John Macy" <spamlessjohn@johnmacy.com> wrote in message
news:43d5a906\$1@linux...

>

> <http://cgi.ebay.com/ws/eBayISAPI.dll?ViewItem&item=4817600091&ssPageName=ADME:B:EF:US:1>This is a multi-part message in MIME format.

-----=_NextPart_000_00ED_01C620DC.81AA3F70

Content-Type: text/plain;

charset="iso-8859-1"

Content-Transfer-Encoding: quoted-printable

Okay,

First off you guys are the answer men. =20

Chris' solution of moving the 3rd party VST folder was the answer on Wavelab. It used to work with those their??? There must be some rogue plugin I have to lose.

Rick,

I have been unchecking 'read only' so I don't think that's the problem. I'll double check though. He has printed at least 3 or 4 wav disks. Only once did stripwav acknowledge that it=20 was a wav and strip it. Same result though no overview or audio.

Now to figure out these Wavelab settings...

I'll try again today and tell you what happens.

Thanks,

Tom

"Chris Ludwig" <chrisl@adkproaudio.com> wrote in message =
news:43d64153@linux...

Hi Tom,

Do you have any third party VST plug ins?

If so move the VST plug ins directory. Usually it will be C:/Program=20

Files/Steinberg/VStplugins. Drag the whole directory to you desktop =
and=20
see if it starts.

Chris

Tom Bruhl wrote:

> After every disk (we've burned over ten now) none of them
> sounded audio or showed an overview.
> I'm still trying everything in my power but I think the
> clients software is what's holding us back. He's
> using Panther/newest DP and newest Toast programs
> on a G4 laptop.=20
> =20
> Can someone sus out my Wavelab problem? I have an
> original 3.01 disk with a 3.03 patch disk too. Neither
> will allow it to start up in XP. Is this version just too old?
> When starting Wavelab it gets into the program then stops
> instantly with: "This application has failed to start because
> HypGui.dll was not found. Reinstalling the application may
> fix this problem." Needless to say it didn't.
> =20
> What is the solution?
> =20
> I haven't been frustrated in a long time. This is taking the cake.
> TIA,
> Tom
> =20
> =20

--=20

Chris Ludwig

ADK

chrisl@adkproaudio.com <mailto:chrisl@adkproaudio.com>

www.adkproaudio.com <http://www.adkproaudio.com/>

(859) 635-5762

-----=_NextPart_000_00ED_01C620DC.81AA3F70

Content-Type: text/html;

charset="iso-8859-1"

Content-Transfer-Encoding: quoted-printable

<!DOCTYPE HTML PUBLIC "-//W3C//DTD HTML 4.0 Transitional//EN">

<HTML><HEAD>

<META http-equiv=3DContent-Type content=3D"text/html; =

charset=3Diso-8859-1">

<META content=3D"MSHTML 6.00.2800.1400" name=3DGENERATOR>

<STYLE></STYLE>

```

</HEAD>
<BODY bgColor=#ffffff>
<DIV><FONT face=Arial size=12>Okay,</FONT></DIV>
<DIV><FONT face=Arial size=12>First off you guys are the answer =
men.&nbsp;=20
</FONT></DIV>
<DIV><FONT face=Arial size=12>Chris' </FONT><FONT face=Arial =
size=12>solution of=20
moving the 3rd party VST folder was the answer</FONT></DIV>
<DIV><FONT face=Arial size=12>on Wavelab.&nbsp; It used to work with =
those=20
their??&nbsp; There must be</FONT></DIV>
<DIV><FONT face=Arial size=12>some rogue plugin I have to =
lose.</FONT></DIV>
<DIV><FONT face=Arial size=12></FONT>&nbsp;</DIV>
<DIV><FONT face=Arial size=12>Rick,</FONT></DIV>
<DIV><FONT face=Arial size=12>I have been unchecking 'read only' so I =
don't think=20
that's</FONT></DIV>
<DIV><FONT face=Arial size=12>the problem.&nbsp; I'll double check =
though.&nbsp;=20
He has printed at least</FONT></DIV>
<DIV><FONT face=Arial size=12>3 or 4 wav disks.&nbsp; Only once did =
stripwav=20
acknowledge that it </FONT></DIV>
<DIV><FONT face=Arial size=12>was a wav and strip it.&nbsp; Same =
result though no=20
overview or audio.</FONT></DIV>
<DIV><FONT face=Arial size=12></FONT>&nbsp;</DIV>
<DIV><FONT face=Arial size=12>Now to figure out these Wavelab=20
settings...</FONT></DIV>
<DIV><FONT face=Arial size=12></FONT>&nbsp;</DIV>
<DIV><FONT face=Arial size=12>I'll try again today and tell you what=20
happens.</FONT></DIV>
<DIV><FONT face=Arial size=12>Thanks,</FONT></DIV>
<DIV><FONT face=Arial size=12>Tom</FONT></DIV>
<BLOCKQUOTE=20
style=12"padding-right: 0px; padding-left: 5px; margin-left: 5px; =
border-left: #000000 2px solid; margin-right: 0px">
  <DIV>"Chris Ludwig" &lt;A=20
    href=12"mailto:chrisl@adkproaudio.com">chrisl@adkproaudio.com</A>&gt; =
wrote in=20
    message <A =
href=12"news:43d64153@linux">news:43d64153@linux</A>...</DIV>HI=20
    Tom,<BR>Do you have any third party VST plug ins?<BR>If so move the =
VST plug=20
    ins directory. Usually it will be C:/Program =
<BR>Files/Steinberg/VStplugins.=20

```

Drag the whole directory to you desktop and
see if it=20
 starts.

Chris

Tom Bruhl wrote:
> After every =
 disk=20
 (we've burned over ten now) none of them
> sounded audio or =
 showed an=20
 overview.
> I'm still trying everything in my power but I think=20
 the
> clients software is what's holding us back. =
 He's
>=20
 using Panther/newest DP and newest Toast programs
> on a G4 =
 laptop.=20

>
> Can someone sus out my Wavelab problem? =
 I have=20
 an
> original 3.01 disk with a 3.03 patch disk too. =20
 Neither
> will allow it to start up in XP. Is this version =
 just=20
 too old?
> When starting Wavelab it gets into the program then=20
 stops
> instantly with: "This application has failed to start=20
 because
> HypGui.dll was not found. Reinstalling the =
 application=20
 may
> fix this problem." Needless to say it =
 didn't.
> =20

> What is the solution?
>
> I haven't been=20
 frustrated in a long time. This is taking the cake.
> =
 TIA,
>=20
 Tom
>
>

--
Chris =
 Ludwig
ADK
<A=20
 href=3D"mailto:chrisl@adkproaudio.com">chrisl@adkproaudio.com =
 <<A=20
 =
 href=3D"mailto:chrisl@adkproaudio.com">mailto:chrisl@adkproaudio.com&=
 gt;
<A=20
 href=3D"http://www.adkproaudio.com">www.adkproaudio.com <<A=20
 =
 href=3D"http://www.adkproaudio.com/">http://www.adkproaudio.com/><=
 BR>(859)=20
 635-5762</BLOCKQUOTE></BODY></HTML>

-----=_NextPart_000_00ED_01C620DC.81AA3F70--Just saw this on Gearsnitz.They say their
 going to restore the
 studios to original condition.Studios 1,2,3 and mix room will be
 for hire and the rest for Eastwest.

Seems pretty cool that they stepped up to buy it.

<http://gearsnitz.com/board/showthread.php?t=56226&page=1&pp=30>

PeteJohn <no@no.com> wrote:
 >you guys sure have expensive taste. wow

>
>Gary Flanigan wrote:
>> "DJ" <animix_spam-this-ahole_@animas.net> wrote:
>>
>>>Find a cheap Avalon 737.
>>
>>
>> I hear that the Manley VoxBox is a great front end for bass :)John <no@no.com> wrote:
>you guys sure have expensive taste. wow
What I actually use is combination of a Radial DI and a Baby Blue head.
Not that expensive.Call this guy...

Walter Woods
82054 Solano Ave. Indio, California 92201 USA

David.

Gary Flanigan wrote:
> John <no@no.com> wrote:
>
>>you guys sure have expensive taste. wow
>>
>>Gary Flanigan wrote:
>>
>>>"DJ" <animix_spam-this-ahole_@animas.net> wrote:
>>>
>>>
>>>>Find a cheap Avalon 737.
>>>
>>>
>>>I hear that the Manley VoxBox is a great front end for bass :)
>
>That live room looks magnificent.
maybe someone should start a landgrant type of thing to save these places
that no one steps in to buy.
England has The National Trust.

"Pete Ruthenburg" <ruthenburg@sbcglobal.net> wrote:
>
>Just saw this on Gearslut. They say their going to restore the
>studios to original condition. Studios 1,2,3 and mix room will be
>for hire and the rest for Eastwest.
>
> Seems pretty cool that they stepped up to buy it.
>
> <http://gearslut.com/board/showthread.php3?t=56226&page=1&pp=30>
>
>PeteThis is a multi-part message in MIME format.

-----=_NextPart_000_0019_01C620F8.DDAF9DD0
Content-Type: text/plain;
charset="iso-8859-1"
Content-Transfer-Encoding: quoted-printable

Tony,
Thanks for the Mac tip. I have forwarded them on.
Now that I'm able to use Wavelab I can try some of
those disks that didn't work with Paris. I'll give you
the run down as soon as I'm done.

Thanks for the incredible resource guys!
Tom

"Tony Benson" <tony@standinghampton.com> wrote in message =
news:43d659aa\$1@linux...
Tom,

Have your friend try to burn a disc directly from the finder. On =
Panther,=20
all he needs to do is insert a recordable cd into the internal drive. =
The cd=20
will come up on his desktop. Drag the files to the cd icon (it might =
take a=20
while for the files to copy over to the disc image). When done, double =
click=20
the cd icon and click "Burn" on the upper right of the cd window. =
Maybe=20
Toast is doing something screwy? I quit using it because of some =
troubles,=20
but that was a few years ago.

Will the files from the disc he burned import back into DP? I mean are =
the=20
burned files valid on his Mac? If so, and you can't get this fixed, =
mail me=20
the disc and I'll see if I can batch convert them for you. Let me know =
if I=20
can help.

Tony

"Tom Bruhl" <arpeggio@comcast.net> wrote in message =
news:43d59b80\$1@linux...
After every disk (we've burned over ten now) none of them

sounded audio or showed an overview.
I'm still trying everything in my power but I think the clients software is what's holding us back. He's using Panther/newest DP and newest Toast programs on a G4 laptop.

Can someone sus out my Wavelab problem? I have an original 3.01 disk with a 3.03 patch disk too. Neither will allow it to start up in XP. Is this version just too old? When starting Wavelab it gets into the program then stops instantly with: "This application has failed to start because HypGui.dll was not found. Reinstalling the application may fix this problem." Needless to say it didn't.

What is the solution?

I haven't been frustrated in a long time. This is taking the cake.
TIA,
Tom

-----=_NextPart_000_0019_01C620F8.DDAF9DD0

Content-Type: text/html;

charset="iso-8859-1"

Content-Transfer-Encoding: quoted-printable

<!DOCTYPE HTML PUBLIC "-//W3C//DTD HTML 4.0 Transitional//EN">

<HTML><HEAD>

<META http-equiv=3DContent-Type content=3D"text/html"; =
charset=3Diso-8859-1">

<META content=3D"MSHTML 6.00.2800.1400" name=3DGENERATOR>

<STYLE></STYLE>

</HEAD>

<BODY bgColor=3D#ffffff>

<DIV>Tony,</DIV>

<DIV>Thanks for the Mac tip. I have =
forwarded them=20
on.</DIV>

<DIV>Now that I'm able to use Wavelab I can =
try some=20
of</DIV>

<DIV>those disks that didn't work with =
Paris. I'll=20
give you</DIV>

<DIV>the run down as soon as I'm =
done.</DIV>

<DIV> </DIV>

<DIV>Thanks for the incredible resource=20
guys!</DIV>
<DIV>Tom</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>"Tony Benson" <<A=20
 =
 href=3D"mailto:tony@standinghampton.com">tony@standinghampton.com>=
 wrote=20
 in message <A=20
 =
 href=3D"news:43d659aa\$1@linux">news:43d659aa\$1@linux...</DIV>Tom,
=

Have=20
 your friend try to burn a disc directly from the finder. On Panther, =

all=20
 he needs to do is insert a recordable cd into the internal drive. The =
cd=20

will come up on his desktop. Drag the files to the cd icon (it =
might take=20
 a
while for the files to copy over to the disc image). When done, =
double=20
 click
the cd icon and click "Burn" on the upper right of the cd =
window.=20
 Maybe
Toast is doing something screwy? I quit using it because of =
some=20
 troubles,
but that was a few years ago.

Will the files from =
the=20
 disc he burned import back into DP? I mean are the
burned files =
valid on=20
 his Mac? If so, and you can't get this fixed, mail me
the disc and =
I'll=20
 see if I can batch convert them for you. Let me know if I
can=20
help.

Tony

"Tom Bruhl" <<A=20
 href=3D"mailto:arpeggio@comcast.net">arpeggio@comcast.net> wrote =
in message=20
 <A =
 href=3D"news:43d59b80\$1@linux">news:43d59b80\$1@linux...
After =
every=20
 disk (we've burned over ten now) none of them
sounded audio or =
showed an=20
 overview.
I'm still trying everything in my power but I think=20
 the
clients software is what's holding us back. He's
using =

 Panther/newest DP and newest Toast programs
on a G4 =
laptop.

Can=20
 someone sus out my Wavelab problem? I have an
original 3.01 =

disk with=20
a 3.03 patch disk too. Neither
will allow it to start up in =
XP. =20
Is this version just too old?
When starting Wavelab it gets into =
the=20
program then stops
instantly with: "This application has failed to =
start=20
because
HypGui.dll was not found. Reinstalling the =
application=20
may
fix this problem." Needless to say it didn't.

What =
is the=20
solution?

I haven't been frustrated in a long time. This =
is=20
taking the cake.
TIA,
Tom

</BLOCKQUOTE></BODY ></HTML>

-----=_NextPart_000_0019_01C620F8.DDAF9DD0--I am seriously close to pickin some up.
Anyone have good or bad comments?
Thanks a bunch.

Rob AHey Aaron! I haven't made a move yet. I was ready to go, but some ssssssstuff
came up, so I'm in a spending hold pattern. I've got other priorities at
the moment. I still plan to go with a 2120, and a couple of different sets
of tubes. I'm also considering the Rocktron Prophecy, but that will be later.

Thanks again for the offer, I'll take you up on it when I'm ready.

James

"Aaron Allen" <nospam@not_here.dude> wrote:
>Heh, I kinda figured that from our past conversations you were hip to it,

>but I was in play mode.
>Say, did you ever decide on a g-pre? Offer still stands man.
>AA
>
>
>"James McCloskey" <excelsm@hotmail.com> wrote in message
>news:43d5b630\$1@linux...
>>
>> "Aaron Allen" <nospam@not_here.dude> wrote:
>>>James, I actually don't man. Dont get me wrong, this sounds like a high
>> cool
>>>factor but... you're buying decision is based on clients that.. don't
pay?
>>
>>>You would spend money on the guys that won't give you any? Yikes dude.
>>>Just drop the chalupa and take 3 steps back bro.....

```

>>>
>>>AA
>>
>> Obviously you have never run a commercial Recording studio in Detroit.
>> There's always some joker asking if he can pay you the balance later.

>> "Yeah
>> man, when I get my big record deal, I'm gonna take care of you!" It made
>> me laugh, and no I never fell for it.
>>
>> James
>>
>> P.S. I was totally joking about the quick mix thing; )
>>
>>
>>>
>>>>"James McCloskey" <excelsm@hotmail.com> wrote in message
>>>>news:43d559b6$1@linux...
>>>>
>>>> You guys can't see the beauty in this? What a time saver! This is
>>>> perfect
>>>> for the client that never pay on time and wants a ruff mix. You just
>> say
>>>> Ok, press a button and you've got a ruff mix. Then you say, when you
>> have
>>>> the balance I'll give you a real mix; )
>>>>
>>>> James
>>>>
>>>>
>>>> "Kim" <hiddensounds@hotmail.com> wrote:
>>>>
>>>>
>>>>>LOL! Strewth! Now they've done it. ;o)
>>>>>
>>>>>"DJ" <animix_spam-this-ahole_@animas.net> wrote:
>>>>>>Oh God.....
>>>>>>
>>>>>>http://www.rndigital.com/AutoMix.html
>>>>>>
>>>>>>;oP
>>>>>>
>>>>>>
>>>>>>"EK Sound" <spamnot.info@eksoundNO.com> wrote in message
>>>>>>news:43d52d76$1@linux...
>>>>>>> Hey,
>>>>>>>
>>>>>>> Found this in "the basement" at NAMM...

```

>>>>>>
>>>>>> <http://www.rndigital.com/>
>>>>>>
>>>>>> Not too much info at the moment...
>>>>>>
>>>>>> David.
>>>>>>
>>>>>>
>>>>>
>>>
>>>
>>
>
>Listen closely to several different types of tunes and I'm sure you'll hear what I heard....a kacky smeary scratchy something or other somewhere in the 1.2k - 2k range

The salesmen (5 of them) thought I was until they heard it...we even switched out the demos for new speakers and sure enough it was on the new ones.

Obviously I didn't buy a pair

DOOn

"Rob Arsenault" <mani2@nbnet.nb.ca> wrote in message news:43d695bf@linux...

>I am seriously close to pickin some up.

> Anyone have good or bad comments?

> Thanks a bunch.

>

> Rob A

>

>

>

>The Chinese built HR-824's are not the same as the older versions (we have the old ones here in two rooms)... if you can get some pre-asia versions they actually sound pretty good.

Have a look at the new JBL LSR4300's as well.

David.

Don Nafe wrote:

> Listen closely to several different types of tunes and I'm sure you'll hear

> what I heard....a kacky smeary scratchy something or other somewhere in the

> 1.2k - 2k range

>

> The salesmen (5 of them) thought I was until they heard it...we even
> switched out the demos for new speakers and sure enough it was on the new
> ones.
>
> Obviously I didn't buy a pair
>
> DOn
>
>
> "Rob Arsenault" <mani2@nbnet.nb.ca> wrote in message news:43d695bf@linux...
>
>>I am seriously close to pickin some up.
>>Anyone have good or bad comments?
>>Thanks a bunch.
>>
>>Rob A
>>
>>
>>
>>
>
>
>What did you go with Don?

"Don Nafe" <dnafe@magma.ca> wrote in message news:43d69fac@linux...
> Listen closely to several different types of tunes and I'm sure you'll
hear
> what I heard....a kacky smeary scratchy something or other somewhere in
the
> 1.2k - 2k range
>
> The salesmen (5 of them) thought I was until they heard it...we even
> switched out the demos for new speakers and sure enough it was on the new
> ones.
>
> Obviously I didn't buy a pair
>
> DOn
>
>
> "Rob Arsenault" <mani2@nbnet.nb.ca> wrote in message
news:43d695bf@linux...
> >I am seriously close to pickin some up.
> > Anyone have good or bad comments?
> > Thanks a bunch.
> >
> > Rob A

> >
> >
> >
> >
>

>My vote these are for the Dynaudio acoustics (BM5A or BM6A)s.
>Very accurate sounding . You can hear "thru" your mixes.

<http://www.dynaudioacoustics.com/>
LaMont

"Rob Arsenault" <mani2@nbnet.nb.ca> wrote:

>I am seriously close to pickin some up.
>Anyone have good or bad comments?
>Thanks a bunch.

>
>Rob A

>
>
>

>Ah, I'm old enough to remember when all we cared about was pre-CBS Fender.
>Now it's pre-Asia Mackie! Careful, though, I'm also old enough to remember
>when "Made in Taiwan" meant cheap!

TCB

EK Sound <spamnot.info@eksoundNO.com> wrote:

>The Chinese built HR-824's are not the same as the older versions (we
>have the old ones here in two rooms)... if you can get some pre-asia
>versions they actually sound pretty good.

>
>Have a look at the new JBL LSR4300's as well.

>
>David.

>
>Don Nafe wrote:

>> Listen closely to several different types of tunes and I'm sure you'll
hear
>> what I heard....a kacky smeary scratchy something or other somewhere in
the
>> 1.2k - 2k range

>>
>> The salesmen (5 of them) thought I was until they heard it...we even
>> switched out the demos for new speakers and sure enough it was on the
new
>> ones.
>>

>> Obviously I didn't buy a pair
>>
>> DOn
>>
>>
>> "Rob Arsenault" <mani2@nbnet.nb.ca> wrote in message news:43d695bf@linux...
>>
>>>I am seriously close to pickin some up.
>>>Anyone have good or bad comments?
>>>Thanks a bunch.
>>>
>>>Rob A
>>>
>>>
>>>
>>>
>>
>>
>>Hey Rob

Noteperfect Alpha Pro...sight unseen...very scary buying something without hearing it first but Mike guranteed I wouldn't be disappointed...he was right!

I went hunting for two months and checked out damn near every decent brand in the 2-5K CDN range..ADAM. Dynaudio, Genelec, Quedsted, NHT, JBL, Westlake, B&W.....

To my ears the latest B&W 805's and up hit the mark so when I contacted Mike at Noteperfect he guaranteed that if his Alpha Pros didn't out perform the B&W 805's and everything else in that price range he'd refund my money less shipping. Add that to the fact I got a mismatched pair (finish-wise) for an excellent price and I couldn't go wrong. Mike even painted the veneered speaker black to match the other speaker at no charge.

The result...I am very happy and my mixes immediately began translating across systems much better than before. Great stereo imaging, tight bottom end, solid mids and smooth hi end...

They are worth at least double (if not triple) what he's selling them for when you compare them to what's on the market.

Don

"Rob Arsenault" <mani2@nbnet.nb.ca> wrote in message news:43d6a665@linux...
> What did you go with Don?
>

>
> "Don Nafe" <dnafe@magma.ca> wrote in message news:43d69fac@linux...
>> Listen closely to several different types of tunes and I'm sure you'll
> hear
>> what I heard....a kacky smeary scratchy something or other somewhere in
> the
>> 1.2k - 2k range
>>
>> The salesmen (5 of them) thought I was until they heard it...we even
>> switched out the demos for new speakers and sure enough it was on the new
>> ones.
>>
>> Obviously I didn't buy a pair
>>
>> DOn
>>
>>
>> "Rob Arsenault" <mani2@nbnet.nb.ca> wrote in message
> news:43d695bf@linux...
>> >I am seriously close to pickin some up.
>> > Anyone have good or bad comments?
>> > Thanks a bunch.
>> >
>> > Rob A
>> >
>> >
>> >
>> >
>> >
>>
>>
>
>You and I must be the same age

Don

"TCB" <nobody@ishere.com> wrote in message news:43d6b8b0\$1@linux...
>
> Ah, I'm old enough to remember when all we cared about was pre-CBS Fender.
> Now it's pre-Asia Mackie! Careful, though, I'm also old enough to remember
> when "Made in Taiwan" meant cheap!
>
> TCB
>
> EK Sound <spamnot.info@eksoundNO.com> wrote:
>>The Chinese built HR-824's are not the same as the older versions (we
>>have the old ones here in two rooms)... if you can get some pre-asia
>>versions they actually sound pretty good.

>>
>>Have a look at the new JBL LSR4300's as well.
>>
>>David.
>>
>>Don Nafe wrote:
>>> Listen closely to several different types of tunes and I'm sure you'll
> hear
>>> what I heard....a kacky smeary scratchy something or other somewhere in
> the
>>> 1.2k - 2k range
>>>
>>> The salesmen (5 of them) thought I was until they heard it...we even
>>> switched out the demos for new speakers and sure enough it was on the
> new
>>> ones.
>>>
>>> Obviously I didn't buy a pair
>>>
>>> DOn
>>>
>>>
>>> "Rob Arsenault" <mani2@nbnet.nb.ca> wrote in message
>>> news:43d695bf@linux...
>>>
>>>>I am seriously close to pickin some up.
>>>>Anyone have good or bad comments?
>>>>Thanks a bunch.
>>>>
>>>>Rob A
>>>>
>>>>
>>>>
>>>>
>>>
>>>
>>>
>Is this the same as saying mail him and giving me a phone number? Why
should I call him, is he going to donate a nice bass preamp to me?
Sweet !!! Hope it's not like a justcron gift. hehe

John

EK Sound wrote:
> Call this guy...
>
> Walter Woods
> 82054 Solano Ave. Indio, California 92201 USA

>
 > David.
 >
 > Gary Flanigan wrote:
 >
 >> John <no@no.com> wrote:
 >>
 >>> you guys sure have expensive taste. wow
 >>>
 >>> Gary Flanigan wrote:
 >>>
 >>>> "DJ" <animix_spam-this-ahole_@animas.net> wrote:
 >>>>
 >>>>
 >>>>> Find a cheap Avalon 737.
 >>>>
 >>>>
 >>>>
 >>>> I hear that the Manley VoxBox is a great front end for bass :)
 >>
 >>
 >>If you use a UAD-1 with Cubase can you monitor with effects while
 tracking with no latency (or 2ms like this article says) ?

<http://www.soundonsound.com/sos/apr04/articles/pcmusician.htm>

Basically I'm wondering, if I track with headphones is there a way to
 track with effects on and not have the latency problem? Is there a
 better card or way to do this or is it just not possible?

Thanks,
 John Everything worked fine when I had ME loaded. Last week I changed my computer
 from ME to XP and all hell broke loose with my Paris setup. I Downloaded
 the XP driver and I everytime I exit Paris, My system reboots. It tells
 me Paris has encountered a serious error and reboots. I have read and followed
 the install instructions and something is still not right. Maybe I'm doing
 something wrong because I'm not really a computer person.

I am running a dell intel pent 3, 1 gig , 512 ram, 3 paris cards

Morgan tried helping me, but right after I got off the phone Paris really
 went wild. Please Someone call me at 703 597 7825 on my cell phone and walk
 me through this.

Ken Smith John <no@no.com> wrote:
 >If you use a UAD-1 with Cubase can you monitor with effects while
 >tracking with no latency (or 2ms like this article says) ?
 >

> <http://www.soundonsound.com/sos/apr04/articles/pcmusician.htm>
>
> Basically I'm wondering, if I track with headphones is there a way to
> track with effects on and not have the latency problem? Is there a
> better card or way to do this or is it just not possible?
>
> Thanks,
> John

This is from the article you pointed to:

would be about 6ms, but the additional latency due to the UAD1 insert would be 512 samples, or about 12ms, making overall latency 18ms. If you passed your audio through several insert effects, each one would add a further 12ms

The UAD-1 will always add a minimum of 12ms (For 1 plugin) to the latency of the sound card and DAW buffers.

The answer to your question remains NO.

With Paris this is not an issue, but with native DAWs, the best you can do with plugins is to use a fast card with a fast computer and zero latency plugins. The recommended method remains using outboard hardware before the converters for compression /effects and use parallel monitoring on the input. GeneCrystal is great.

Bill

"Mike R." <emarenot@yahoo.com> wrote in message news:43d3af1e@linux...

> Since I've been having such a good time playing with these free little
> softsynth beastsies, I thought I'd throw out some of my impressions for
> anybody that's interested.
>
> First off, there's ALOT of free synths out there. I think I've got about
> twenty now. I'm running them in a simple version of Orion. Many of the
> synths seem to be made with a program called "SynthEdit." This probably
> (partly) explains why many seem to sound the similar. So far I've got a
> few
> favorites: Evol, two synths by a guy named HG Fortune (STS-21, and
> Protoplasm), and another called Synth 1. I just downloaded Crystal, which
> may be pretty cool too, we'll see. I'm amazed at how responsive even my
> SBLive card is. I haven't gotten more than a couple instances yet, but
> the
> card is hangin' in there at 10-13ms. Kind of surprising for such dated
> technology.
>
> Well, just wanted to pass along some good gear vibes. I imagine most of

> you
> folks are probably already hip to all this. The first synth I messed with
> was an OB1 with four SEM modules, now I've got almost a score of synths in
> one Athlon based SEM -and their all free! Killer.
> MR
>
>Love 'em.

"Rob Arsenault" <mani2@nbnet.nb.ca> wrote in message news:43d695bf@linux...

>I am seriously close to pickin some up.
> Anyone have good or bad comments?
> Thanks a bunch.
>
> Rob A
>
>
>
>If you didn't, be sure and install the Paris Effects Subsystem.

"Ken Smith" <maddrapa@netzero.com> wrote in message news:43d6c75b\$1@linux...

>
> Everyting worked fine when I had ME loaded. Last week I changed my
computer
> from ME to XP and all hell broke loose with my Paris setup. I Downloaded
> the XP driver and I everytime I exit Paris, My system reboots. It tells
> me Paris has encountered a serious error and reboots. I have read and
followed
> the install instructions and something is still not right. Maybe Im doing
> something wrong because I not really a computer person.
>
> I am running a dell intel pent 3, 1 gig , 512 ram, 3 paris cards
>
> Morgan tried helping me, but right after I got off the phone Paris really
> went wild. Please Someone call me at 703 597 7825 on my cell phone and
walk
> me throught this.
>
> Ken SmithThis is a multi-part message in MIME format.

-----=_NextPart_000_000C_01C62147.CDBEB560

Content-Type: text/plain;
charset="iso-8859-1"

Content-Transfer-Encoding: quoted-printable

Tony,
Your technique worked great combined with
Gene's Wavelab batch processing idea. We=20
are still sussing out the Toast issues to know

just why it created such a problem.

Now we are having timing issues. My client suggests that he thinks merging files in DP is the only way to get timing correct. Does that sound right?
Tom

"Tony Benson" <tony@standinghampton.com> wrote in message = news:43d659aa\$1@linux...
Tom,

Have your friend try to burn a disc directly from the finder. On = Panther,=20
all he needs to do is insert a recordable cd into the internal drive. =
The cd=20
will come up on his desktop. Drag the files to the cd icon (it might = take a=20
while for the files to copy over to the disc image). When done, double = click=20
the cd icon and click "Burn" on the upper right of the cd window. =
Maybe=20
Toast is doing something screwy? I quit using it because of some = troubles,=20
but that was a few years ago.

Will the files from the disc he burned import back into DP? I mean are = the=20
burned files valid on his Mac? If so, and you can't get this fixed, = mail me=20
the disc and I'll see if I can batch convert them for you. Let me know = if I=20
can help.

Tony

"Tom Bruhl" <arpeggio@comcast.net> wrote in message = news:43d59b80\$1@linux...
After every disk (we've burned over ten now) none of them sounded audio or showed an overview.
I'm still trying everything in my power but I think the clients software is what's holding us back. He's using Panther/newest DP and newest Toast programs on a G4 laptop.

Can someone sus out my Wavelab problem? I have an

original 3.01 disk with a 3.03 patch disk too. Neither will allow it to start up in XP. Is this version just too old? When starting Wavelab it gets into the program then stops instantly with: "This application has failed to start because HypGui.dll was not found. Reinstalling the application may fix this problem." Needless to say it didn't.

What is the solution?

I haven't been frustrated in a long time. This is taking the cake.
TIA,
Tom

-----=_NextPart_000_000C_01C62147.CDBEB560

Content-Type: text/html;

charset="iso-8859-1"

Content-Transfer-Encoding: quoted-printable

<!DOCTYPE HTML PUBLIC "-//W3C//DTD HTML 4.0 Transitional//EN">

<HTML><HEAD>

<META http-equiv=3DContent-Type content=3D"text/html; =

charset=3Diso-8859-1">

<META content=3D"MSHTML 6.00.2800.1400" name=3DGENERATOR>

<STYLE></STYLE>

</HEAD>

<BODY bgColor=3D#ffffff>

<DIV>Tony,</DIV>

<DIV>Your technique worked great combined=20
with</DIV>

<DIV>Gene's Wavelab batch processing =
idea. We=20

</DIV>

<DIV>are still sussing out the Toast issues =
to=20

know</DIV>

<DIV>just why it created such a=20
problem.</DIV>

<DIV> </DIV>

<DIV>Now we are having timing issues. =
My=20

client</DIV>

<DIV>suggests that he thinks merging files =
in DP is the=20

</DIV>

<DIV>only way to get timing correct. =
Does that=20

</DIV>
<DIV>sound right?</DIV>
<DIV>Tom</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>"Tony Benson" <<A=20
=
href=3D"mailto:tony@standinghampton.com">tony@standinghampton.com>=
wrote=20
in message <A=20
=
href=3D"news:43d659aa\$1@linux">news:43d659aa\$1@linux...</DIV>Tom,
=

Have=20
your friend try to burn a disc directly from the finder. On Panther, =

all=20
he needs to do is insert a recordable cd into the internal drive. The =
cd=20

will come up on his desktop. Drag the files to the cd icon (it =
might take=20
a
while for the files to copy over to the disc image). When done, =
double=20
click
the cd icon and click "Burn" on the upper right of the cd =
window.=20
Maybe
Toast is doing something screwy? I quit using it because of =
some=20
troubles,
but that was a few years ago.

Will the files from =
the=20
disc he burned import back into DP? I mean are the
burned files =
valid on=20
his Mac? If so, and you can't get this fixed, mail me
the disc and =
I'll=20
see if I can batch convert them for you. Let me know if I
can=20
help.

Tony

"Tom Bruhl" <<A=20
href=3D"mailto:arpeggio@comcast.net">arpeggio@comcast.net> wrote =
in message=20
<A =
href=3D"news:43d59b80\$1@linux">news:43d59b80\$1@linux...
After =
every=20
disk (we've burned over ten now) none of them
sounded audio or =
showed an=20
overview.
I'm still trying everything in my power but I think=20
the
clients software is what's holding us back. He's
using =

Panther/newest DP and newest Toast programs
on a G4 =
laptop.

Can=20
someone sus out my Wavelab problem? I have an
original 3.01 =

disk with=20

a 3.03 patch disk too. Neither
will allow it to start up in =
XP. =20

Is this version just too old?
When starting Wavelab it gets into =
the=20

program then stops
instantly with: "This application has failed to =
start=20

because
HypGui.dll was not found. Reinstalling the =
application=20

may
fix this problem." Needless to say it didn't.

What =
is the=20

solution?

I haven't been frustrated in a long time. This =
is=20

taking the cake.
TIA,
Tom

</BLOCKQUOTE></BODY ></HTML>

-----=_NextPart_000_000C_01C62147.CDBEB560--"Edna" <edna@texomaonline.com> wrote:
>If you didn't, be sure and install the Paris Effects Subsystem.

Now only that, but when you install it, make sure the install path matches
your Paris install. If you're on version 3.0 and it's installed under C:\Program
Files\ then it should default correctly, but if it's on another drive, in
another directory, or if you're running version 2.2 or earlier then it wont
default correctly. Not only that, but when you choose the path and click
"OK" or whatever the button is, it will then change it on you and you'll
have to change it back in the text window which shows the path before continuing.
You have to watch it closely...

Cheers,

Kim.Only a lawyer could have dreamt this one up.

Rich

"John Macy" <spamlessjohn@johnmacy.com> wrote in message
news:43d5a906\$1@linux...

>

>

<http://cgi.ebay.com/ws/eBayISAPI.dll?ViewItem&item=4817600091&ssPageName=ADME:B:EF:US:1>So what does this part of the article mean?

However, one possible solution would be to try running a dedicated DSP
card such as the PowerCore or UAD1 alongside a DSP-assisted soundcard
such as the Mixtreme or original Pulsar. This would allow 'zero latency'
monitoring with DSP effects on your live input signals, with the option
of further high-quality delay-compensated plug-in insert and send

gene lennon wrote:

> John <no@no.com> wrote:

>

>>If you use a UAD-1 with Cubase can you monitor with effects while
>>tracking with no latency (or 2ms like this article says) ?

>>

>> <http://www.soundonsound.com/sos/apr04/articles/pcmusician.htm>

>>

>>Basically I'm wondering, if I track with headphones is there a way to
>>track with effects on and not have the latency problem? Is there a
>>better card or way to do this or is it just not possible?

>>

>>Thanks,

>>John

>

>

> This is from the article you pointed to:

>

> would be about 6ms, but the additional latency due to the UAD1 insert would
> be 512 samples, or about 12ms, making overall latency 18ms. If you passed
> your audio through several insert effects, each one would add a further 12ms

>

> The UAD-1 will always add a minimum of 12ms (For 1 plugin) to the latency
> of the sound card and DAW buffers.

>

> The answer to your question remains NO.

>

> With Paris this is not an issue, but with native DAWs, the best you can do
> with plugins is to use a fast card with a fast computer and zero latency
> plugins. The recommended method remains using outboard hardware before the
> converters for compression /effects and use parallel monitoring on the input.

> Gene

>Thanks for the heads up on Dyns LaMont, I had heard about em but never
looked deeper. After a little googlin, I found that these have a really good
rep online, some say they surpass HRs in many aspects. Gonna get some
pricing on these. Its tough buying monitors out here, nobody stocks anything
so "try before you buy" is not really an option.

Thanks

Rob

"LaMont" <jjdpro@ameritech.net> wrote in message news:43d6ab66\$1@linux...

>

> My vote these are for the Dynaudio acoustics (BM5A or BM6A)s.

> Very accurate sounding . You can hear "thru" your mixes.

>

> <http://www.dynaudioacoustics.com/>

> LaMont
>
>
> "Rob Arsenault" <mani2@nbnet.nb.ca> wrote:
>>I am seriously close to pickin some up.
>>Anyone have good or bad comments?
>>Thanks a bunch.
>>
>>Rob A
>>
>>
>>
>>
>I'll second Lamont's opinion on the Dynaudios... I have a pair
of the BM5A's and I think they're great.

Neil

"Rob Arsenault" <info@studiomanitou.com> wrote:
>Thanks for the heads up on Dyns LaMont, I had heard about em but never
>looked deeper. After a little googlin, I found that these have a really
good
>rep online, some say they surpass HRs in many aspects. Gonna get some
>pricing on these. Its tough buying monitors out here, nobody stocks anything

>so "try before you buy" is not really an option.
>
>Thanks
>Rob
>
>"LaMont" <jjdpro@ameritech.net> wrote in message news:43d6ab66\$1@linux...
>>
>> My vote these are for the Dynaudio acoustics (BM5A or BM6A)s.
>> Very acurate sounding . You can hear "thru" your mixes.
>>
>> <http://www.dynaudioacoustics.com/>
>> LaMont
>>
>>
>> "Rob Arsenault" <mani2@nbnet.nb.ca> wrote:
>>>I am seriously close to pickin some up.
>>>Anyone have good or bad comments?
>>>Thanks a bunch.
>>>
>>>Rob A
>>>
>>>

>>>
>>>
>>
>
>Deej

I have been external out on 4 Aux's for my headphone mix into a Furman HDS 16. Why can't I send 8 trks out my aux external from one EDS and 8 out the aux external on the other EDS card.

I'll know if will work in a few weeks. Just think PARIS tracks mixed down on an AMEK 2500.....Next to an Ampex MM1200 What a world..

You got to love it

DJ" <animix_spam-this-ahole_@animas.net> wrote:

>Hmmmmm.....guys???.....major brain fart here, likely caused by
>.....well.....nevermind what, I simply fucked up. This is actually
>not possible to my knowlefdge without routing the track from the Paris
>insert sends into a summing device. Using the auxes is just a dead end
>because you would still have to patch the insert sends to L & R Paris auxes
>and since you can't sum patch points on the aux module, what's the use in
>that?

>

>Now back to the CR. Big mix to do, lots of acoustic instruments, with track
>counts in low 20's for the most part.....14 songs and needs to be
>finished in two weeks. This wouldn't be a problem except I have a day job
>that is very busy these days and some evenings I'm just too fried to feel
>like firing up the beast and going to work all night (though that's what's
>happening right now).

>

>If I think of a workaround for this, I'll post it up here.

>Sorry.....false hope sucks.....I know.

>

>Deej

>

>

>

>tonehouse" <zmcleod@comcast.net> wrote in message news:43c83675\$1@linux...

>> Hi DJ...please tell us all how to do this....

>> "> >

>> > >You could use the Paris auxes to create stem mixes if you didn't want
>to

>> > fly

>> > >16 individual tracks to an external mixer..

>> > >

>> > >;o)

>> > >

>> > > "Dominic" <bertstudio@aol.com> wrote in message
>news:43c7bbe7\$1@linux...
>> > >
>> > > Looking for help mixing with PARIS outside the box.
>> > > How do you have it set up? What I was originally thinking was
>> > > to "external out" each channel but that would only get me 8 outs.
>> > > I am running 2 EDS cards so what I was thinking was to 8 out on
>> > > one card and 8 out on the other, but I would think there would
>> > > be an easier way.
>> > >
>> > > Your advice would be greatly appreciated.
>> > > Thanx Dominic
>> > >
>> > >
>> >
>>
>>
>
>Gotta agree with both Lamont and Neil...nice monitors, especially the
6's....good value and good sound, just not my cup of tea but they were high
on the list.

Let me get you the name of the distributor in Montreal...they have an
excellent return policy for trying out different brands that they carry

Don

"Neil" <IUOI@OIU.com> wrote in message news:43d784ff\$1@linux...
>
> I'll second Lamont's opinion on the Dynaudios... I have a pair
> of the BM5A's and I think they're great.
>
> Neil
>
>
> "Rob Arsenault" <info@studiomanitou.com> wrote:
>>Thanks for the heads up on Dyns LaMont, I had heard about em but never
>>looked deeper. After a little googlin, I found that these have a really
> good
>>rep online, some say they surpass HRs in many aspects. Gonna get some
>>pricing on these. Its tough buying monitors out here, nobody stocks
>>anything
>
>>so "try before you buy" is not really an option.
>>
>>Thanks

>>Rob
>>
>>"LaMont" <jjdpro@ameritech.net> wrote in message news:43d6ab66\$1@linux...
>>>
>>> My vote these are for the Dynaudio acoustics (BM5A or BM6A)s.
>>> Very accurate sounding . You can hear "thru" your mixes.
>>>
>>> <http://www.dynaudioacoustics.com/>
>>> LaMont
>>>
>>>
>>> "Rob Arsenault" <mani2@nbnet.nb.ca> wrote:
>>>>I am seriously close to pickin some up.
>>>>Anyone have good or bad comments?
>>>>Thanks a bunch.
>>>>
>>>>Rob A
>>>>
>>>>
>>>>
>>>>
>>>>
>>>
>>
>>
>John <no@no.com> wrote:
>So what does this part of the article mean?
>
>
>However, one possible solution would be to try running a dedicated DSP
>card such as the PowerCore or UAD1 alongside a DSP-assisted soundcard
>such as the Mixtreme or original Pulsar. This would allow 'zero latency'

>monitoring with DSP effects on your live input signals, with the option

>of further high-quality delay-compensated plug-in insert and send

>

It means if you run a DSP-assisted soundcard such as the Mixtreme or original Pulsar, you can use the effects on the card during live recording to monitor through, and then use other plugins like the UAD-1 later for mixing with. Unfortunately, it never sounds the same, and this kind of software based dual split signal can be very complex to set up in some systems. For less money you could get an RNC.I have the "USA" 824's " they are very good for the money,and they work well in a small space..I use them with a sub-woofer....Of course there are better speakers for way more money....You people were throwing out comparisons with Dynaudios..not fair....There are several "better" small

nearfields these days if you have the cash...Genelecs,JBL,Tannoy,etc,etc...
"Bill Lorentzen" <bill@lorentzen.ws> wrote in message
news:43d6f5b2\$1@linux...
> Love 'em.
>
> "Rob Arsenault" <mani2@nbnet.nb.ca> wrote in message
news:43d695bf@linux...
> >I am seriously close to pickin some up.
> > Anyone have good or bad comments?
> > Thanks a bunch.
> >
> > Rob A
> >
> >
> >
> >
> >
>
>This is a multi-part message in MIME format.

-----=_NextPart_000_0095_01C6219F.73A92760
Content-Type: text/plain;
charset="iso-8859-1"
Content-Transfer-Encoding: quoted-printable

Spin Auidio VST-DX Wrapper..Putt .dll in win/sys32/ensoniq/plugins
"Tom Bruhl" <arpeggio@comcast.net> wrote in message =
news:43d59779@linux...
Paul,
Wrap that bad boy.
Tom
"Paul" <paulbeach@adelphia.net> wrote in message =
news:43d57a48\$1@linux...

Just purchased Drumagog [4.03] at NAMM. Runs fine in my WIN 2000 =
audio apps.

But... can't seem to get Paris to see it. I searched the forum, and =
tried
a couple of things there, but no luck. I apologize if I missed the =
answer.

I'm using win 98 SE for PARIS [dual boot machine]. None of my VST =
..dll files
appear in the designated VST folder. So, is this a Win 98 issue?

I have a project coming up soon, where I know Drumagog will really =
help.
Thanks in advance for any help.

-----=_NextPart_000_0095_01C6219F.73A92760

Content-Type: text/html;
charset="iso-8859-1"

Content-Transfer-Encoding: quoted-printable

<!DOCTYPE HTML PUBLIC "-//W3C//DTD HTML 4.0 Transitional//EN">

<HTML><HEAD>

<META http-equiv=3DContent-Type content=3D"text/html; =
charset=3Diso-8859-1">

<META content=3D"MSHTML 6.00.2800.1522" name=3DGENERATOR>

<STYLE></STYLE>

</HEAD>

<BODY bgColor=3D#ffffff>

<DIV>Spin Auidio VST-DX Wrapper..Putt .dll =
in=20

win/sys32/ensoniq/plugins</DIV>

<BLOCKQUOTE dir=3Dltr=20

style=3D"PADDING-RIGHT: 0px; PADDING-LEFT: 5px; MARGIN-LEFT: 5px; =
BORDER-LEFT: #000000 2px solid; MARGIN-RIGHT: 0px">

<DIV>"Tom Bruhl" <<A=20

href=3D"mailto:arpeggio@comcast.net">arpeggio@comcast.net> wrote =
in message=20

news:43d59779@linux...</DIV>

<DIV>Paul,</DIV>

<DIV>Wrap that bad boy.</DIV>

<DIV>Tom</DIV>

<BLOCKQUOTE=20

style=3D"PADDING-RIGHT: 0px; PADDING-LEFT: 5px; MARGIN-LEFT: 5px; =
BORDER-LEFT: #000000 2px solid; MARGIN-RIGHT: 0px">

<DIV>"Paul" <<A=20

=

href=3D"mailto:paulbeach@adelphia.net">paulbeach@adelphia.net> =
wrote in=20

message <A=20

=

href=3D"news:43d57a48\$1@linux">news:43d57a48\$1@linux...</DIV>
Just=
=20

purchased Drumagog [4.03] at NAMM. Runs fine in my WIN 2000 audio=20
apps.
But... can't seem to get Paris to see it. I searched the =

forum, and=20

tried
a couple of things there, but no luck. I apologize if I =
missed the=20

answer.

I'm using win 98 SE for PARIS [dual boot =
machine]. None=20

of my VST .dll files
appear in the designated VST folder. =
So, is=20

this a Win 98 issue?

I have a project coming up soon, where I =
know=20

Drumagog will really help.
Thanks in advance for any=20
help.</BLOCKQUOTE></BLOCKQUOTE></BODY></HTML>

-----=_NextPart_000_0095_01C6219F.73A92760--bummer, thanks for the explanation. Sounds like pcs have a long way to go to get quality live effects.

gene lennon wrote:

> John <no@no.com> wrote:

>

>>So what does this part of the article mean?

>>

>>

>>However, one possible solution would be to try running a dedicated DSP
>>card such as the PowerCore or UAD1 alongside a DSP-assisted soundcard
>>such as the Mixtreme or original Pulsar. This would allow 'zero latency'

>

>

>>monitoring with DSP effects on your live input signals, with the option

>

>

>>of further high-quality delay-compensated plug-in insert and send

>>

>

>

> It means if you run a DSP-assisted soundcard such as the Mixtreme or original
> Pulsar, you can use the effects on the card during live recording to monitor
> through, and then use other plugins like the UAD-1 later for mixing with.
> Unfortunately, it never sounds the same, and this kind of software based
> dual split signal can be very complex to set up in some systems. For less
> money you could get an RNC.

>> This is from the article you pointed to:

>

> "So if you were running with a 256-sample buffer at 44.1kHz, playback
> latency

> would be about 6ms, but the additional latency due to the UAD1 insert
> would

> be 512 samples, or about 12ms, making overall latency 18ms. If you passed
> your audio through several insert effects, each one would add a further
> 12ms

> to the overall latency."

>

> The UAD-1 will always add a minimum of 12ms (For 1 plugin) to the latency
> of the sound card and DAW buffers.

>

> The answer to your question remains NO.

I actually do this. The scenario outlined above is using a 256-sample buffer. An RME card can use a 64-sample buffer, using a fast computer (I use an Athlon 3200). This gives 1.5ms latency per pass, or 6ms total latency with a UAD-1. If you were to use 96k, you could cut the latency in half, but you would need a very fast computer.

Now getting my PC to do this was no easy task. I had to turn off most XP services, and I have to disable my network card. I get no audible glitches, but who knows if I'm dropping samples, so I only do this when I'm messing around. For serious recording, I jack the buffer up to 1024 samples and use Zero Latency Monitoring in the RME card.

So for me, at this time, this is not a solution for serious recording, but it can be done, and I think with the right system, with the proper tweaks, it can be done reliably. I had a talk with a Steinberg product specialist at NAMM about this. My question specifically was "why can't I access *send* effects while using direct monitoring". He said they are waiting for CPU's to get faster before implimenting this. I made the point that send effects would usually be used for delay based effects (delay, chorus, reverb), so the added latency shouldn't be too much of an issue. Using a bit of native verb to wet a vocal take for instance... usually the verb effect itself has predelay in it already, so what's a few more milliseconds there. He agreed that it made sense.

As for tracking live through software comps etc... Nuendo can do this now (via inserts on input buses), but it will never get to "0" for latency. There will always be *some* processing delay no matter how fast machines get.

David.

John wrote:

> bummer, thanks for the explanation. Sounds like pcs have a long way to
> go to get quality live effects.

>

> gene lennon wrote:

>

>> John <no@no.com> wrote:

>>

>>> So what does this part of the article mean?

>>>

>>>

>>> However, one possible solution would be to try running a dedicated
>>> DSP card such as the PowerCore or UAD1 alongside a DSP-assisted
>>> soundcard such as the Mixtreme or original Pulsar. This would allow
>>> 'zero latency'

>>

>>

>>
>>> monitoring with DSP effects on your live input signals, with the option
>>
>>
>>
>>> of further high-quality delay-compensated plug-in insert and send

>>>
>>
>>
>> It means if you run a DSP-assisted soundcard such as the Mixtreme or
>> original
>> Pulsar, you can use the effects on the card during live recording to
>> monitor
>> through, and then use other plugins like the UAD-1 later for mixing with.
>> Unfortunately, it never sounds the same, and this kind of software based
>> dual split signal can be very complex to set up in some systems. For less
>> money you could get an RNC.
>>I think it's great that Roger got so involved in this that it seems legit on
first look. Now I think he needs to volunteer to be one of the studio
visitors in Sonicfinger...maybe a Roger Nichols and Bea Arthur combined visit
from hell.

;o)

"EK Sound" <spamnot.info@eksoundNO.com> wrote in message
news:43d64eee@linux...
> The only difference is that Roger Nichols was actually standing at the
> booth... Hard to tell from that if it's actually a joke, or not...
>
> David.
>
> DJ wrote:
>
> > Here's another good plugin bundle that would complement yours.
> >
> > <http://www.sonicfinger.com/>
> >
> >
> > "EK Sound" <spamnot.info@eksoundNO.com> wrote in message
> > news:43d52d76\$1@linux...
> >
> >>Hey,
> >>
> >>Found this in "the basement" at NAMM...
> >>
> >><http://www.rndigital.com/>
> >>

> >>Not too much info at the moment...
> >>
> >>David.
> >
> >
> >>I actually do this. The scenario outlined above is using a 256-sample
>buffer. An RME card can use a 64-sample buffer, using a fast computer (I
use
>an Athlon 3200). This gives 1.5ms latency per pass, or 6ms total latency

>with a UAD-1. If you were to use 96k, you could cut the latency in half,

>but you would need a very fast computer.
>
>Now getting my PC to do this was no easy task. I had to turn off most XP

>services, and I have to disable my network card. I get no audible glitches,

>but who knows if I'm dropping samples, so I only do this when I'm messing

>around. For serious recording, I jack the buffer up to 1024 samples and
use
>Zero Latency Monitoring in the RME card.
>
>So for me, at this time, this is not a solution for serious recording, but

>it can be done, and I think with the right system, with the proper tweaks,

>it can be done reliably.
>
>

get the UAD-1 down to 6ms, but it is incredibly taxing on the system. Perhaps

still sounds better and the ultra high-end computer systems cost as much
or more than good audio hardware.

GeneYou can do this if you set the analog I/O appropriately in the Paris
patchbay, then set the respective aux to External and stereo, enable the
respective aux on the tracks you want to hear on the respective input of the
HDS-16, physically connect the output from the respective auxes to the
inputs of the HDS16 and crank the volume pot on the HRM-16 mixer. If you
want to hear the entire mix on the HRM mixer, route your main mix (not
monitor mix) to stereo D of the HDS-16. Now that you have that working, set
you mixer to post fader and
go buy yourself an Alesis Wedge on EBay and hook it up to the FX I/O on the

HRM mixer and you'll never have to worry about jacking around with FX levels from the board. The HDS/HRM is a great system, IMO, plenty of gain available for 600 ohm cans and more versatile I/O situation (at least for studio applications) than the (somewhat similar) Mytek Private Q system which, when similarly configured, costs around \$1000.00 more and exponentially more and more as it is expanded. Something about running line level audio down those big old SCSI cables sounds much *phatter* than the LAN based HDS-6 system, to my ears.

Welcome to the club.

;o)

Deej

"Dominic" <BERTSTUDIO@aol.com> wrote in message news:43d78e2e\$1@linux...

>

> Deej

>

> I have been external out on 4 Aux's for my headphone mix into a
> Furman HDS 16. Why can't I send 8 trks out my aux external from
> one EDS and 8 out the aux external on the other EDS card.

>

>

> I'll know if will work in a few weeks. Just think PARIS tracks
> mixed down on an AMEK 2500.....Next to an Ampex MM1200
> What a world..

> You got to love it

> DJ" <animix_spam-this-ahole_@animas.net> wrote:

> >Hmmmmmm.....guys???.....major brain fart here, likely caused by

> >.....well.....nevermind what, I simply fucked up. This is actually

> >not possible to my knowlefdge without routing the track from the Paris

> >insert sends into a summing device. Using the auxes is just a dead end

> >because you would still have to patch the insert sends to L & R Paris
auxes

> >and since you can't sum patch points on the aux module, what's the use in

> >that?

> >

> >Now back to the CR. Big mix to do, lots of acoustic instruments, with
track

> >counts in low 20's for the most part.....14 songs and needs to be

> >finished in two weeks. This wouldn't be a problem except I have a day job

> >that is very busy these days and some evenings I'm just too fried to feel

> >like firing up the beast and going to work all night (though that's
what's

> >happening right now).

> >

> >If I think of a workaround for this, I'll post it up here.

> >Sorry.....false hope sucks.....I know.

> >

> >Deej

> >

> >

> >

> >tonehouse" <zmcleod@comcast.net> wrote in message
news:43c83675\$1@linux...

> >> Hi DJ...please tell us all how to do this....

> >> "> >

> >> > >You could use the Paris auxes to create stem mixes if you didn't
want

> >to

> >> > fly

> >> > >16 individual tracks to an external mixer..

> >> > >

> >> > >;o)

> >> > >

> >> > >"Dominic" <bertstudio@aol.com> wrote in message
> >news:43c7bbe7\$1@linux...

> >> > >>

> >> > >> Looking for help mixing with PARIS outside the box.

> >> > >> How do you have it set up? What I was originally thinking was

> >> > >> to "external out" each channel but that would only get me 8 outs.

> >> > >> I an running 2 EDS cards so what is was thinking was to 8 out on

> >> > >> one card and 8 out on the other, but I would think there would

> >> > >> be an easier way.

> >> > >>

> >> > >> Your advice would be greatly appreciated.

> >> > >> Thanx Dominic

> >> > >

> >> > >

> >> >

> >>

> >>

> >

> >

> >

>I'm trying out various permutations of gain staging on a mix. This is a mix
with a lot of acoustic instruments. I have tried lots of options. The one
that is working best for this kind of mix is to set the submix faders so tht
the overall submix is barely clipping the submix bus, then set the global
fader to -2.2dB, set NoLimit on the Global fader with the thresjhild at -1.8
and output at -0.6. The mix sounds very smooth and phat, but very detiled
and non-aggressive and the kicker is that subsequent application of UAD-1 PL
on the stereo mix allows a very transparent gain boost without any
significant perceptual skewing of the relative levels. Since there is no
mastering budget for this project I'm mixing here, I'm having to really do a
lot of experimenting with all phases of the process and it's very

interesting....and stressful.

The verdict, IMO, is to save yourself lots of wailing, gnashing of teeth, rending of garments and tearing of hair and make sure your client has a mastering budget.

;o)I don't see the point. Hardware still sounds better and the ultra high-end computer systems cost as much or more than good audio hardware.

I agree with you. I built a fairly *mooselike* DAW recently to see if I could get this happening. My dual core is running at almost 5000MHz (if you take both cores into account). Still, it's not ready for this kind of thing in any kind of critical scenario.

Another thing I've found is that even with 4 x UAD-1 cards, if I get more than 17 UAD-1 plugins happening (with the UAD meter showing only around 50%) I start getting crackling in the audio. I called UA about this yesterday and they said to drastically increase the buffer settings on my RME cards. WTF??? I built this system so I could mix at low latencies (I have my reasons for wanting to mix at low latencies) and now if I use even half the horsepower available to me with the UAD-1 cards, the whole scenario is in the toilet.

Quad dual cores might be the ticket. When I can justify that kind of expense for convenience, I'll probably just buy a Neve Capricorn instead.

;o)

"gene lennon" <glennon@NOSPmyrealbox.com> wrote in message news:43d7ac51\$1@linux...

>

>

>

> >I actually do this. The scenario outlined above is using a 256-sample
> >buffer. An RME card can use a 64-sample buffer, using a fast computer (I
> use

> >an Athlon 3200). This gives 1.5ms latency per pass, or 6ms total latency
>

> >with a UAD-1. If you were to use 96k, you could cut the latency in
half,

>

> >but you would need a very fast computer.

> >

> >Now getting my PC to do this was no easy task. I had to turn off most XP
>

> >services, and I have to disable my network card. I get no audible
glitches,

>
> >but who knows if I'm dropping samples, so I only do this when I'm messing
>
> >around. For serious recording, I jack the buffer up to 1024 samples and
> use
> >Zero Latency Monitoring in the RME card.
> >
> >So for me, at this time, this is not a solution for serious recording,
but
>
> >it can be done, and I think with the right system, with the proper
tweaks,
>
> >it can be done reliably.
> >
> >
>
> I've done this as well. Using my G5 at 96, I can get down to 32 buffers
and
> get the UAD-1 down to 6ms, but it is incredibly taxing on the system.
Perhaps
> on a Quad, but other than for academic reasons, I don't see the point.
Hardware
> still sounds better and the ultra high-end computer systems cost as much
> or more than good audio hardware.
>
> As to Quads..report to follow.
> Gene
> If I could track 32 channels with all the effects going at once that
would be more than anyone could buy hardware for.

DJ wrote:

> I don't see the point. Hardware
> still sounds better and the ultra high-end computer systems cost as much
> or more than good audio hardware.
>
> I agree with you. I built a fairly *mooselike* DAW recently to see if I
> could get this happening. My dual core is running at almost 5000MHz (if you
> take both cores into account). Still, it's not ready for this kind of thing
> in any kind of critical scenario.
>
> Another thing I've found is that even with 4 x UAD-1 cards, if I get more
> than 17 UAD-1 plugins happening (with the UAD meter showing only around 50%) I
> start getting crackling in the audio. I called UA about this yesterday and
> they said to drastically increase the buffer settings on my RME cards.
> WTF??? I built this system so I could mix at low latencies (I have my
> reasons for wanting to mix at low latencies) and now if I use even half the
> horsepower available to me with the UAD-1 cards, the whole scenario is in

> the toilet.
>
> Quad dual cores might be the ticket. When I can justify tht kind of expense
> for convenience, I'll probably just buy a Neve Capricorn instead.
>
> ;o)
>
> "gene lennon" <glennon@NOSPmyrealbox.com> wrote in message
> news:43d7ac51\$1@linux...
>
>>
>>
>>>I actually do this. The scenario outlined above is using a 256-sample
>>>buffer. An RME card can use a 64-sample buffer, using a fast computer (I
>>
>>use
>>
>>>an Athlon 3200). This gives 1.5ms latency per pass, or 6ms total latency
>>
>>>with a UAD-1. I If you were to use 96k, you could cut the latency in
>
> half,
>
>>>but you would need a very fast computer.
>>>
>>>Now getting my PC to do this was no easy task. I had to turn off most XP
>>
>>>services, and I have to disable my network card. I get no audible
>
> glitches,
>
>>>but who knows if I'm dropping samples, so I only do this when I'm messing
>>
>>>around. For serious recording, I jack the buffer up to 1024 samples and
>>
>>use
>>
>>>Zero Latency Monitoring in the RME card.
>>>
>>>So for me, at this time, this is not a solution for serious recording,
>
> but
>
>>>it can be done, and I think with the right system, with the proper
>
> tweaks,
>
>>>it can be done reliably.

>>>
>>>
>>
>>I've done this as well. Using my G5 at 96, I can get down to 32 buffers
>
> and
>
>>get the UAD-1 down to 6ms, but it is incredibly taxing on the system.
>
> Perhaps
>
>>on a Quad, but other than for academic reasons, I don't see the point.
>
> Hardware
>
>>still sounds better and the ultra high-end computer systems cost as much
>>or more than good audio hardware.
>>
>>As to Quads..report to follow.
>>Gene
>>
>
>
>> If I could track 32 channels with all the effects going at once that
> would be more than anyone could buy hardware for.

Well hmmm...you can pretty much do this in Paris, ..depending on how many
FX you're planning to use you would need a multi card system The Paris FX
work well for this, IMHO. that's what the whole dedicated DSP thing is all
about.

Deej

"John" <no@no.com> wrote in message news:43d7be34\$1@linux...
> If I could track 32 channels with all the effects going at once that
> would be more than anyone could buy hardware for.
>
> DJ wrote:
> > I don't see the point. Hardware
> > still sounds better and the ultra high-end computer systems cost as much
> > or more than good audio hardware.
> >
> > I agree with you. I built a fairly *mooselike* DAW recently to see if I
> > could get this happening. My dual core is running at almost 5000MHz (if
you
> > take both cores into account). Still, it's not ready for this kind of
thing
> > in any kind of critical scenario.

> >
> > Another thing I've found is that even with 4 x UAD-1 cards, if I get more
> > than 17 UAD-1 plugins happening (with the UAD meter showing only around 50%) I
> > start getting crackling in the audio. I called UA about this yesterday and
> > they said to drastically increase the buffer settings on my RME cards.
> > WTF??? I built this system so I could mix at low latencies (I have my
> > reasons for wanting to mix at low latencies) and now if I use even half the
> > horsepower available to me with the UAD-1 cards, the whole scenario is in
> > the toilet.
> >
> > Quad dual cores might be the ticket. When I can justify that kind of expense
> > for convenience, I'll probably just buy a Neve Capricorn instead.
> >
> > ;o)
> >
> > "gene lennon" <glennon@NOSPmyrealbox.com> wrote in message
> > news:43d7ac51\$1@linux...
> >
> >>
> >>
> >>> I actually do this. The scenario outlined above is using a 256-sample
> >>> buffer. An RME card can use a 64-sample buffer, using a fast computer (I
> >>
> >>> use
> >>
> >>> an Athlon 3200). This gives 1.5ms latency per pass, or 6ms total latency
&g

Subject: Re: OT: So far little snow in NJ this winter.

Posted by [rick](#) on Sun, 08 Jan 2006 12:31:20 GMT

[View Forum Message](#) <> [Reply to Message](#)

/>

>3) This Product: Is it just a dedicated DSP Mixer, thus letting the Mac's
cpu(s)

> handle the recording and plugins?

>

>4)* With all the "whining" I have been doing about the lack of any DAW manufacturer

> "Willing to " to head to toe with Digi. Could this be the Part Native solution/DS
