

Subject: hey John.....your Total mix tutorial???

Posted by [Deej \[4\]](#) on Fri, 26 Jan 2007 05:26:14 GMT

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mostly, probably limitations
> of
>>> the
>>>> technician, lord knows I tried for months to get a decent sound on tape.
&

Subject: Re: hey John.....your Total mix tutorial???

Posted by [rick](#) on Fri, 26 Jan 2007 10:39:10 GMT

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[illegible]

So, there's this live show, see? It's THIS Sunday, see? And I've got a piano, a cello, a soprano singer, and either a flute or a clarinet (depending upon which piece they're playing) to record and put into a PA...!

What microphones should I look at begging, borrowing, or buying to use in this situation? (All I have right now are a couple MXL 603S Mogamis and an SM57 Beta...)

Thanks for any suggestions!

Doug

<http://www.parisfaqs.com> Was experimenting with different mixes from the last concert recording I did.

Tracked straight from mixer inserts. Vocal mic was a 58 (her own mic - my preference would have been my AT AE5400, but she had a bit of a cold and didn't want to get germs on the main mics as she was the opener for Kasim Sulton).

All Paris - all EDS inserts. Put NoLimit on her vocal and just spanked the meter - voice just got fuller and ballsier the harder I pushed it. Now I know what you guys were talking about. Final mix was run through the Wavelab multiband.

Piano was a digital - the one thing you need to remember is that she called me in a panic on the Tuesday before the gig because the two guitar players she normally uses both had medical emergencies, and she was without an accompanist.

The piano player she finally found had one day to listen to the mp3's she sent, and 20 minutes before the doors opened to run through the songs with her. Had never heard them before.

Damn, it's fun working with real professionals.

Anyhow, tell me what you think. A few extraneous noises, but it's a live recording.

www.ctgonline.org/paul/open_arms.mp3

pabChuck! I've got a solution for you that will put a smile on your face! No not that, some thing else!

For around \$300.00 you could be back up and recording and the sound will put a smile on your face!

I
I
I
V

Dude, it's PARIS!!!!

C-16 \$50.00
442 \$50.00 or less
EDS 1000 \$150.00

Total \$250.00

MEC \$50.00-\$100.00

You want to record acoustic instruments and write songs, paris is great for that. Your native instruments stuff, you can use Cubase for that. You could run both on the same machine. Or run them on two different machines and use Paris as your mixer, cue system, summing box. There is no learning curve, an it's great to write and record songs with. You were happy when you used Paris in the past;) Paris is cheap, it sounds great and you will get good results quickly and easily. If you buy another Paris system, you wont lose any money, if, or when you sell it. So there is nothing to lose! Simplify your life and get another Paris system and be happy!

James

P.S. There are some great plugins for free, thanks to Chuck!

"chuck duffy" <c@c.com> wrote:

>

>Hi Carl,

>

>I play acoustic all the time, have several laying around. I got bit by the

>recording bug at age 17 or so with a 2 track then a 4 track. I love playing,

>but I really like recording and the process more. Go figure...

>

>Chuck

>"Carl Amburn" <carlamburn@hotmail.com> wrote:

>>Does an acoustic guitar work for you?

>>

>>-Carl

>>

>>"chuck duffy" <c@c.com> wrote in message news:45db6695\$1@linux...

>>>

>>> But come to think of it, I am beginning to realize that I am just a sad

>>sack

>>> about this stuff. The other day I went to play guitar and nothing worked,
>>> I left the cable plugged into the wah and the wah died. So I removed
>it
>>> from the chain and the amp didn't work. I said screw it and decided
to
>>play
>>> bass. Nothing doing. I left the cable plugged into that and the battery
>>> for the active pickup died. hahahaha
>>>
>>> "chuck duffy" <c@c.com&g

Subject: Re: hey John.....your Total mix tutorial???

Posted by [John](#) on Fri, 26 Jan 2007 13:18:45 GMT

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t; wrote:

>>> >
>>> >Well, that was a subject of another long sad post of mine. Basically
>the
>>> >room stinks, low ceilings, small, rectangular. Can't really be treated
>>because
>>> >it's the family rec room.
>>> >
>>> >The instrument I like alot, but it's a small yamaha student upright.
> It
>>> >doesn't mic well, based on surroundings mostly, probably limitations
>of
>>> the
>>> >technician, lord knows I tried for months to get a decent sound on tape.
>>> >
>>> >
>>> >That's when I got the brilliant idea to go with giga for pianos, and
>man
>>> >the sound was great but that lead to needing an 88 key weighted
>>controller,
>>> >a search that took forever. Then giga worked but it wasn't integrated
>>with
>>> >cubase and I couldn't stand that because it just got to complicated
when
>>> >all I wanted to do was take ten minutes and record a song real quick.
>>Then
>>> >I got the brilliant idea to try halion, and import my gigas, which worked
>>> >ok, but in the end I could never get acceptable, playable latencies
in
>>that
>>> >environment using the presonus.
>>> >

>>> >In the end it's all so stupid and ridiculous that I gave up, which bums
>>> me
>>> >out, because I think I am approaching writing some decent songs.
>>> >
>>> >Chuck
>>> >
>>> >Jamie K <Meta@Dimensional.com> wrote:
>>> >>
>>> >>Man, that is a sad story Chuck.
>>> >>
>>> >>Maybe you should get a stand alone recorder next to your piano.
>>> >>
>>> >>Cheers,
>>> >> -Jamie
>>> >> www.JamieKrutz.com
>>> >>
>>> >>
>>> >>chuck duffy wrote:
>>> >>> I hear you man. It kills me, because I am normally deep voodoo
>>uncommonly
>>> >>> good at solving software, hardware and driver problems.
>>> >>>
>>> >>> I'm glad all of you guys have solid rigs, but my cubase, firewire
>>experience
>>> >>> has left me never wanting to touch a computer again. I honestly
get
>>> a
>>> >pit
>>> >>> in my stomache when I think about recording. And it aint just
>>pre-sonus.
>>> >>> I have been through so much crap with this shit that it kills me.
> I
>>> >had
>>> >>> a little midiman keyboard with lots of faders and knobs and stuff
>that
>>> >was
>>> >>> perfect for B4, had a working template that mapped everything to
the
>>> draw
>>> >>> pulls and knobs on the B4.
>>> >>>
>>> >>> Loved it. Then the USB quit working. So I switched to the
>>conventional
>>> >>> MIDI port, then the keys started breaking. So I switched to an edirol
>>> >with
>>> >>> lots of knobs and faders. No mapping for B4. Spent two days learning
>>> >how
>>> >>> to create the mapping, saved the template to the device.

>>> >>>
>>> >>> Things were fine, then the template mysteriously disappeared from
>the
>>> >keyboard,
>>> >>> so I reloaded, then it stopped retaining settings completely when
>>powered
>>> >>> down. For a little while I wresled with re-loading the template
from
>>> >disk
>>> >>> to the device every time I wanted to play. I memorized the 13
>>keystrokes
>>> >>> required, but then didn't play for a while and forgot, broke out
the
>>> manual
>>> >>> and said f*** it and gave up.
>>> >>>
>>> >>> A critical hard drive died, without backup and I lost half my projects
>>> >that
>>> >>> I wanted to finish after a 10 year hiatus from actually doing any
>>serious
>>> >>> recording. 10% of the ones that I didn't loose are corrupt paris
>>projects
>>> >>> for whatever reason.
>>> >>>
>>> >>> Got a new computer, reloaded paris JUST TO EXPORT and fought with
>1/1
>>> >initialization
>>> >>> for half a day. Then got it running and faced the real world for
>a
>>couple
>>> >>> of weeks, returned to find that the demo had expired and I couldn't
>>find
>>> >>> antipace.
>>> >>>
>>> >>> Went to use GIGA on the new computer and worked through their warped
>>> re-authorization
>>> >>> process that involved me removing hacking the registry for everything
>>> >I had
>>> >>> just re-installed. Figured out how to get the GSIF drivers working
>>and
>>> >outputting
>>> >>> SPDIF from an ECHO MIA. Couldn't reliably get the pre-sonus and
MIA
>>> to
>>> >sync
>>> >>> with sp/dif so gave up and decided to go analog.
>>> >>>
>>> >>> Got it working and noticed that the piano sounded like absolute shit.

>>> > All
 >>> >>> phasey and wierd. Discovered that the pre-sonus doesn't pan in
 >>standalone
 >>> >>> mode, so it was simply layering the LR giga output together.
 >>> >>>
 >>> >>> Fire up cubase and find that the only latency that stops the
 >>stuttering
 >>> >and
 >>> >>> chattering of TWO CHANNELS OF 44.1 K won't let me play the damn thing
 >>> >in
 >>> >>> real time.
 >>> >>>
 >>> >>> This is of course all over the course of months/years. I don't get
 >>more
 >>> >>> than 10 minutes of un-interrupted time anymore, and when I do I can't
 >>> >even
 >>> >>> play a f**** midi piano.
 >>> >>>
 >>> >>> So I play the real thing, which I love, and think to myself how much
 >>> I
 >>> >would
 >>> >>> like to record the tune I'm playing. That gets me depressed so I
 >>knock
 >>> >off
 >>> >>> and head upstairs.
 >>> >>>
 >>> >>> My new tee shirt is gonna read 'i'll record when i'm dead'
 >>> >>>
 >>> >>> Chuck
 >>> >>>
 >>> >>>
 >>> >>>
 >>> >>>
 >>> >>>
 >>> >>> "Neil" <OIUOIU@OIU.com> wrote:
 >>> >>>> See? This is the kind of crap I bitched about before...
 >>> >>>> companies putting stuff out on the market that doesn't work with
 >>> >>>> half of what it's "supposed" to interface with, or putting it
 >>> >>>> out before it's ready, causing everyone who buys it to become an
 >>> >>>> unwitting beta-tester.
 >>> >>>>
 >>> >>>> Thad, you disagreed with my stance on this before, but being a
 >>> >>>> systems guy for a living, you have to agree that this is WAY
 >>> >>>> more rampant in the audio field than it is in any other... I
 >>> >>>> mean, how many times have you gotten a new piece of gear in at
 >>> >>>> work, and find it simply doesn't work as advertised? As often as
 >>> >>>> we encounter this sort of thing in the audio realm?
 >>> >>>>

>>> >>>> Neil
 >>> >>>>
 >>> >>>> Neil
 >>> >>>>
 >>> >>>>
 >>> >>>>
 >>> >>>> "chuck duffy" <c@c.com> wrote:
 >>> >>>>> Actually, two different laptops using the on-board fire-wire and
 >>with
 >>> >a
 >>> >>>> PCMCIA
 >>> >>>>> firewire interface. One desktop using on-board firewire and PCI
 >>firewire.
 >>> >>>>>
 >>> >>>>> Chuck
 >>> >>>>>
 >>> >>>>> "TCB" <nobody@ishere.com> wrote:
 >>> >>>>>> Sorry you're having so much trouble with it, Chuck. Maybe I'm
 just
 >>> >getting
 >>> >>>>>> lucky, or maybe the hardware is scared of not working around me.
 >>Are
 >>> >you
 >>> >>>>>> using it on a desktop or laptop?
 >>> >>>>>>
 >>> >>>>>> TCB
 >>> >>>>>>
 >>> >>>>>> "chuck duffy" <c@c.com> wrote:
 >>> >>>>>>> Thad,
 >>> >>>>>>>
 >>> >>>>>>>> I don't know about the general opinion, since I haven't done
 much
 >>> >research,
 >>> >>>>>>>> but my personal experience with the Firepod has been a disaster.
 >>>
 >>> >Snaps,
 >>> >>>>>>>> crackles, pops, lockups, usable only at unplayable latencies
 >>(figure
 >>> >>> that
 >>> >>>>>>>> out :-)). Two different computers, three different chipsets,
 >all
 >>> >the
 >>> >>>> driver
 >>> >>>>>>>> revs. Extremely inflexible live monitoring, lack of panning
 when
 >>> >used
 >>> >>>>> in
 >>> >>>>>>>> standalone mode. The list goes on and on, and I've basically

>>given
 >>> >up
 >>> >>>> on
 >>> >>>>> recording because I can't afford the time or the money to research
 >>> >and
 >>> >>>> buy
 >>> >>>>> something different.
 >>> >>>>>>
 >>> >>>>>> Chuck
 >>> >>>>>> "TCB" <nobody@ishere.com> wrote:
 >>> >>>>>>> But there seems to be this general opinion that there is
 >>something
 >>> >'wrong'
 >>> >>>>>>> with the driver in all cases. Needing to have the right chipset
 >>> for
 >>> >>> an
 >>> >>>>> esoteric
 >>> >>>>>>> third party add-on is a mild pain and something that should
 be
 >>known,
 >>> >>>> but
 >>> >>>>>>> the idea that the Presonus gear doesn't work correctly or has
 >>'bad
 >>> >drivers'
 >>> >>>>>>> is just bosh. I'm not saying you were implying this, but I've
 >>started
 >>> >>>> to
 >>> >>>>>>> hear/read that in various places.
 >>> >>>>>>>
 >>> >>>>>>> TCB
 >>> >>>>>>>
 >>> >>>>>>> "TCB" <nobody@ishere.com> wrote:
 >>> >>>>>>>> I've had a Firebox for a while and have only ever had problems
 >>> with
 >>> >>>> it
 >>> >>>>>>> when
 >>> >>>>>>>> using it as what we used to call a wave device, i.e. using
 it
 >as
 >>> >a
 >>> >>> system
 >>> >>>>>>>> sound device. The ASIO driver has been rock solid for me, using
 >>> >the
 >>> >>>> on
 >>> >>>>>>> board
 >>> >>>>>>>> firewire on a Dell laptop.
 >>> >>>>>>>>
 >>> >>>>>>>> TCB

>>> >>>>>>>>>

>>> >>>>>>>>> Chris Ludwig <chrisl@adkproaudio.com> wrote:

>>> >>>>>>>>> HI DJ,

>>> >>>>>>>>> The firebox sounds and generally works well but does have some

>>> >firewire

>>> >>>>>>>>> chipset compatibility issues.

>>> >>>>>>>>> I'd recommend using a Firewire PCI card for any audio

>>interfaces

>>> >on

>>> >>>>> the

>>> >>>>>>>>> A8V.

>>> >>>>>>>>> You should also check out the Focusrite Sapphire LE or the

>TC

>>> Konnekt4D

>>> >>>>>>>>> all it he same price range.

>>> >>>>>>>>> Sonically they all sound better than the M-Audio FW410. The

>TC

>>> >Konnnect

>>> >>>>>>>>> series have a pair of excellent Mic-Pres but has issues with

>>Sonar.

>>> >>>>>>>>> Works great with Cubase, SAMP, Reaper, etc.

>>> >>>>>>>>>

>>> >>>>>>>>>

>>> >>>>>>>>>

>>> >>>>>>>>> Chris

>>> >>>>>>>>>

>>> >>>>>>>>>

>>> >>>>>>>>> DJ wrote:

>>> >>>>>>>>>

>>> >>>>>>>>> Chris,

>>> >>>>>>>>>

>>> >>>>>>>>> I've been thinking of getting one of those little Presonus

>>Firebox

>>> >>>>> gizmo's

>>> >>>>>>>>> for Amy's new PC, just so she can plug in a guitar and mic

>and

>>> >record

>>> >>>>>>>>> a

>>> >>>>>>>>> vocal/guitar track to a click when she feels a song coming

>on.

>>> >The

>>> >>>>> thing

>>> >>>>>>>>> that appeals to me about this is that it is small/simple

(at

>>> least

>>> >>>>> appears

>>> >>>>>>>>> to be), has a couple of preamps and is bundled with Cubase

>SE
>>> >so
>>> >>> our
>>> >>>>>>>>>> platforms are compatible and I can grab the project/tracks
>on
>>> >the
>>> >>>> network
>>> >>>>>>>>> >from the main DAW in the studio and import them.
>>> >>>>>>>>>> Is the Presonus driver so shabby that it would not be usable
>>> for
>>> >>> simple
>>> >>>>>>>>>> stuff like this? She will be using my old DAW rig for her
>new
>>> >graphics/music
>>> >>>>>>>>>> computer. ASUS A8V-Dlx, AMD 64 4400x2 CPU 2G RAM on XP SP2.
>I
>>> >figured
>>> >>>>>>>> the
>>> >>>>>>>>> FW
>>> >>>>>>>>>> port on this would probably handle simple two track recording.
>>> >>>>>>>>>>
>>> >>>>>>>>>> Is there another simple and inexpensive (and better) 2 track
>>> recording
>>> >>>>>>>>>> interface that is bundled with Cubase SE?
>>> >>>>>>>>>>
>>> >>>>>>>>>> Thanks,
>>> >>>>>>>>>>
>>> >>>>>>>>>> Deej
>>> >>>>>>>>>>
>>> >>>>>>>>>>
>>> >>>>>>>>>>
>>> >>>>>>>>>> "Chris Ludwig" <chrisl@adkproaudio.com> wrote in message
>>> >>>>>>>>>> news:45d88fac\$1@linux...
>>> >>>>>>>>>>
>>> >>>>>>>>>>
>>> >>>>>>>>>> HI Brandon,
>>> >>>>>>>>>> Driver -
>>> >>>>>>>>>> Not all the routing features are available in the current
>>drivers.
>>> >>>>>> You
>>> >>>>>>>>>> can
>>> >>>>>>>>>>>>>> at the moment only use the analog and the first bank of
adat
>>> >in
>>> >>> the
>>> >>>>>>>>>>>>>> internal routing control.
>>> >>>>>>>>>>>>>>
>>> >>>>>>>>>>>>>> Every unit we've gotten so far has had hardware issues.

>>> >>>>>>>>>> dead or abnormally noisy inputs or outputs.
>>> >>>>>>>>>> Weird zippering sounds when using the mic gain and also
the
>>> headphone
>>> >>>>>>>>>> channels.
>>> >>>>>>>>>> Grounding noise and buzzing units.
>>> >>>>>>>>>>
>>> >>>>>>>>>> If they work out some of these mechanical and or drivers
>>issues
>>> >>> then
>>> >>>>>>>> the
>>> >>>>>>>>>> unit will be a great device but so far I've had to replace
>>almost
>>> >>>> every
>>> >>>>>>>>>> unit with some other device.
>>> >>>>>>>>>>
>>> >>>>>>>>>> The Focusrite Sapphire Pro is the only other in that price
>>range
>>> >>>> with
>>> >>>>>>>>>> similar features. It doesn't have the routing features of
>the
>>> >Presonus
>>> >>>>>>>>>> but
>>> >>>>>>>>>> has the same number of I/O.
>>> >>>>>>>>>>
>>> >>>>>>>>>> In the long run I think you really better off going with
>the
>>> >MOTU
>>> >>>>>>>>>> and
>>> >>>>>>>>>> RME
>>> >>>>>>>>>> products.
>>> >>>>>>>>>>
>>> >>>>>>>>>> If the the b

Subject: Re: hey John.....your Total mix tutorial???
Posted by [John \[1\]](#) on Fri, 26 Jan 2007 13:20:21 GMT
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udget is an issues then I would recommend.
>>> >>>>>>>>

Subject: Re: hey John.....your Total mix tutorial???
Posted by [Deej \[4\]](#) on Fri, 26 Jan 2007 16:19:24 GMT
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rives are formatted with 4kb Cluster size.

> This is good for the OS correct? BUT not for the Audio drive CORrect?

> The Audio drive should be something larger like 128k or something correct?

> How can I change this without removing all the data from it?

> thx

> b

--

Chris Ludwig

ADK Pro Audio

(859) 635-5762

www.adkproaudio.com

chrisl@adkproaudio.com HI Brandon,

Right click on "My Computer" go to properties/advanced/performance settings/advanced/virtual memory. press the change button

First select the C drive then select custom size set the min max to the same amount. For 1 gig of ram use 3072 as min./max, for 2 gigs use 4096.

Second select each of the other drives listed and choose 'No paging File" for each drive other than the C drive.

Press OK then apply and then reboot and you should be good.

Chris

Brandon wrote:

> I set the swap file on its own drive before. Now I want to set it back where

> it goes normally. How do I do that?

> thx

> b

--

Chris Ludwig

ADK Pro Audio

(859) 635-5762

www.adkproaudio.com

chrisl@adkproaudio.com no i used the XP installation disk

--

Thanks,

Brandon

"Chris Ludwig" <chrisl@adkproaudio.com> wrote in message news:45dc76ab\$1@linux...

> Hi Brandon,

>

> How did you format the drives when you installed windows?

> Did you use a 3rd party program to do it?

>

>

> Chris

>

> Brandon wrote:

>> According to my defragmenter results report both my

>> OS and AUDIO drives are formatted with 4kb Cluster size.

>> This is good for the OS correct? BUT not for the Audio drive CORrect?

>> The Audio drive should be something larger like 128k or something

>> correct?

>> How can I change this without removing all the data from it?

>> thx

>> b

>

> --

> Chris Ludwig

>

> ADK Pro Audio

> (859) 635-5762

> www.adkproaudio.com

> chrisl@adkproaudio.com My feeling is that the ear pieces are the most important consideration. Unfortunately, like you say, there's no way to "try out" many different ear pieces. I would think any decent transmitter and receiver should be able to cover getting the audio to the ears. Beyond that, EQ and compression will be a consideration. I would recommend though researching and buying the best ear pieces you can afford.

Tony

"Ed" <AskMe@email.com> wrote in message news:45db67b2\$1@linux...

> Thanks Tony...

>

> Yeah, I am still a bit puzzled as to which brand/model to start with...

> and it isn't a product you can just try... I am a little surprised that

> the Samson is getting decent reviews, being so cheap. In fact some

> reviews I read, the Samson sounds just as good as some of the higher

> priced units. Of course everyone recommends replacing the plugs with

> better ones... thanks again... take care... ~ Ed

>

>

>

> "Tony Benson" <tony@standinghampton.com> wrote in message
> news:45d9fa4f\$1@linux...
>> Ed,
>>
>> My wife and I play in a classic rock band. We play around 70 gigs a year.
>> She's the lead singer. I drum, sing back-up, and often run sound from the
>> stage. We both use Shure PSM600 systems with Sensaphonics (custom molded)
>> dual driver ear phones. I haven't tried the PSM400's but have heard
>> several reviews from people who moved from the 400's to the 600's and all
>> have said the 600's sounded much better. Of course, when people spend
>> more money, they sometimes "think" things sound better, so take that info
>> with a grain of salt. I do know that the 600 systems are used by a lot of
>> the "big boys" and they are proven, road ready systems. Anyway, my wife
>> started with the Shure E3's ear pieces. The problem with in-ears is that
>> in order to get the best sound at the lowest volumes, the ear-pieces have
>> to seal very well. Some people may be able to get a good seal with
>> standard ear pieces. My wife likes her custom molds much better, but they
>> are also dual driver, so I'm sure that makes a difference also. I've only
>> ever used the custom molded ear pieces, and they sound very good. One
>> thing to remember is that it will take some time to adjust to the "cut
>> off" feeling you get with in ears. If you rely on a lot of communication
>> with the audience or other players, you might have some problems.
>> Bleeding some stage mics in can help, but you can get to point where it
>> starts to defeat the purpose of using the in ears. Sensaphonics is making
>> a new model that has built-in ambience mics, but I'm sure they aren't
>> cheap. I don't mean to put you off the idea, just letting you know that
>> there is some acclimation time involved. Also, when I'm running sound
>> from the stage, I can't use the in-ears. I rely too much on how the sound
>> from the mains comes back to me to run things effectively.
>>
>> Tony
>>
>>
>>
>> "Ed" <AskMe@email.com> wrote in message news:45d8eecb\$1@linux...
>>> Hi All.. Hopefully I can find some experts on the Shure in-ear personal
>>> monitor systems. I am planning to start investing in a good system and
>>> I am having a hard time finding the info I need on these. Practical
>>> info.. not technical...
>>>
>>> First, I read that the PSM200 is mono... so that is out. I am
>>> considering the PSM400, but I don't understand why there is such a price
>>> climb when looking at the PSM600 and the PSM700. The PSM400 seems to
>>> have everything I need. 4 channels with the E3 plugs... Something that
>>> I can't seem to find out is... The PSM400 has the P4T (transmitter) and
>>> the P4R (receiver)... Since I can't afford the whole package at once, I
>>> plan to buy the pieces... but my question is... if I have a P4T and I
>>> find a deal on a P6R... would they work together?

>>>
>>> Is there really a sonic difference in the E3 versus the E4 or E5?
>>>
>>> My ultimate plans are to have a good system that is expandable... and if
>>> other band members want to jump and purchase their own receiver and
>>> plugs... it will be compatible...
>>>
>>> Any help? Also, if there are other systems that is better recommended..
>>> please write...
>>>
>>
>>
>
>Click Start, then right-click My Computer and select Properties.

Click the Advanced tab, then click the Settings button in the Performance pane.

Click the Advanced tab, then click Change in the Virtual Memory pane.

You can set your swap file location/size here.

Pretty simple, eh?

Thanks Bill G for the wonderful UI!

Mark

"brandon" <a@a.com> wrote:

>
>I set the swap file on its own drive before. Now I want to set it back where
>it goes normally. How do I do that?
>thx
>bl tried a reinstall to address issues. No EDS effects show up.
Repeated the process same thing. VST's are showing up. Installing to
c:\ParisPro.

Where to next?

HoovDo these cards happily coexist? Even in a Magma? Also, does anyone know how long they are? I can't find that info on the TC site.

ThanksAt last I have managed to do it ...
Only way I could configure it is the following.
I used a normal usb midi interface and midioverlan software midi.
I also used console DX/VST wrapper which is the only one that can configure
midi ports in and out from within as a vst plugin !
So by letting Paris be master via midioverlan to Cubase on a second computer

, although on same computer with pipes (midioverlan) could be done too...
Then Opening console on a paris audio track and putting an empty 16bit or 24 bot wavefile you open console and configure the usbmidi input as midi0 for console, then open the VSTI you have installed in your Paris rig and connect them with wires (like with Paris patchbay) and voila your cuabse send midi sequence that is read by your VSTI inside Paris and audio output stays in Paris.

Sounds complicated but it works.

It has a big delay (meaning you cannot play live but you can play it inside Cubase and then let just cubase follow Paris MTC and compensate for delay by moving ahead of time the cubase midi tracks or offest mtc...

Regards,

DimitriosPlease note that youer Cubase computer must have a hardware midi to which will connect to usbmidi hardware of Paris .

So to summarize you need two hardware midi interfaces one for each computer.

Maybe you can also make them work mtc connected with only these two midi interfaces but I am using midioverlan and I am extremely happy with it...

Regards,

Dimitrios

"Dimitrios" <musurgio@otenet.gr> wrote:

>

>At last I have managed to do it ...

>Only way I could configure it is the following.

>I used a normal usb midi interface and midioverlan software midi.

>I also used console DX/VST wrapper which is the only one that can configure

>midi ports in and out from within as a vst plugin !

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>for console, then open the VSTI you have installed in your Paris rig and

>connect them with wires (like with Paris patchbay) and voila your cuabse

>send midi sequence that is read by your VSTI inside Paris and audio output

>stays in Paris.

>Sounds complicated but it works.

>It has a big delay (meaning you cannot play live but you can play it inside

>Cubase and then let just cubase follow Paris MTC and compensate for delay

>by moving ahead of time the cubase midi tracks or offest mtc...

>Regards,

>DimitriosDavid,

I think you should seriously look into Metric Halo's Moblie I/Os. The Firewire interface seamlessly connects with Macbook Pros! I use mine in the studio, but it is equally at home on the road.

Both Meric halo's ULN and 2882's have great pres and coverters. It depends on how many channels you want to record at once. The ULN has two channels and the 2882 has 8. Of course one can connect several together for more simultaineus

tracks Many MH users are using them to record exceptionally high quality live recordings. Metric Halo MIO Control software also has a sweet recording interface so no other DAW is required.

Check out www.mhlab.com and their excellent user group for more info.

Good Luck!
Steve

"David Evan" <davidevanmusic@yahoo.com> wrote:

>hey everybody,
>it's been a while since i've been here. still using paris but about to sell
>everything (i may keep paris so that in ten years when suddenly, everybody's
>drooling for that great, "vintage paris sound" i'll sell it and retire).
i
>have mics, pres, racks, etc. i'm selling it because i'm about to move out
>of my home and live a life of touring on the road. so, i need to get a
>laptop and a mobile recording setup for recording simple demos on the road.
>i'm keeping one at4050 for vocals, and am planning to get a rode nt4 for
>stereo micing my ac. guit. any suggestions about programs and firewire
>micpre interfaces?
>
>thanks!
>-david evan
>www.myspace.com/davidevanmusic
>
>or he's blonde...

On Wed, 21 Feb 2007 07:54:35 -0600, "Aaron Allen"
<know-spam@not_here.dude> wrote:

>That's just age man :>
>
>AA
>
>
>"Mark McCurdy" <mark@mccurdy.net> wrote in message news:45dc49ef@linux...
>>I think the sad thing is it took me a sec.. :\n/>>>
>>
>>
>>
>> "Neil" <IOOIU@OIU.com> wrote in message news:45dbe056\$1@linux...
>>>
>>>

>>> Blonde #1: That backpack you're carrying sure looks heavy! What
>>> have you got in there?
>>>
>>> Blonde #2: Watermelons!
>>>
>>> Blonde #1: If I can guess how many watermelons are in your
>>> backpack, would you let me have one of them?
>>>
>>> Blonde #2: If you can guess how many watermelons are in my
>>> backpack, I'll let you have both of them!
>>>
>>> Blonde #1: Are there three?
>>
>>
>Yes, but I've got a U5 also. Actually, everything here has a di on it, so I could track a bass sextet,
no problem!

I think I'll keep the 737, get the Babyface mod and use it for mixing to run key mono channels
back through for extra warmth and eq.

I'd still love to check out the Great River though, I think that would cover all my preamp needs for
years to come.

Cheers,

TC

DJ wrote:

>> I'm just not overly excited about the preamp on the 737 compared to my
>> Vipre and API's. I just never use it.
>
> makes a great bass DI
>
> ;o)
>
> "TC" <tc@spammetodeathyoubastards.org> wrote in message
> news:45db763f@linux...
>> Thanks guys for all the info.
>>
>> Tyrone,
>>
>> The Babyface mod is a mod for the compressor, giving it a faster attack.
>>
>> Basically it will make the compressor more usable.
>> <http://www3.telus.net/thewitt/>
>>
>> I'm just not overly excited about the preamp on the 737 compared to my

>> Vipre and API's. I just never use it.
>>
>> Cheers,
>>
>> TC
>>
>>
>> Tyrone Corbett wrote:
>>> TC, what's the "Babyface" mod for the 737SP? I am considering getting rid
>>> of mine as well for the Millinea pre....heard it at Genes
>>> place...actually
>>> heard it before in another studio and liked it alot.
>>>
>>> Tyrone
>

<http://blogs.zdnet.com/Apple/?p=441&tag=nl.e622Crappy!>

I really am starting to hate dvd/cd anyway as a format.

I'm hoping usb flash drives will keep coming down in price, and maybe non-writable flash drives will eventually replace dvds and cds for movies and audio.

Most of the movies I rent are glitchy. Probably the dvd player, but still..

Cheers,

TC

Mark McDermott wrote:

>
> <http://blogs.zdnet.com/Apple/?p=441&tag=nl.e622>This contains the oft lauded Macopener software. A total cluster**** which included the complete hosing of a brand new hard drive and a few scary moments as I scrambled to recover my boot sector on my OS drive which suddenly failed to be recognized after I ran the quick format utility on a completely different drive (the aforementioned "hosed" drive)...and spent a couple of hours on the phone with Dataviz tech support who were very nice and polite but finally threw in the towel themselves and referred me to another company.

If you want to open Mac files on a PC, use Macdrive (www.mediafour.com).Howdy.

I was wondering about user forums and hit counts and hardware and electricity. How many users can Kim's setup handle, for instance? You measure that sort of thing in "hits", right? What is Kim's setup, anyway?

How much power does it use per month? How much hardware and power is necessary for a site that attracts, say, 3,000 people? 10,000? 50,000?

I was wondering these things because a beloved user forum, "Chowhound", has gone completely to shit since being sold by the founder to CNet. They've absolutely wrecked the thing, and many of the old-school Hounds have left the building, so to speak. The founder claims adamantly that he couldn't afford to keep the site up and running, that it was going to go away entirely if he didn't get deep pockets in on the action, but it somehow doesn't add up for me. I was wondering if he's telling the truth or lying to hide a greedy streak.

JimmyI sold my original "white" Great River MP-2MH, and I regret it. Awesome preamp, and they've stopped making it for a while. They claim they'll reintroduce it with the NV-style controls, but we'll see.

I have an NV now, and it kills on everything. Great River makes awesome gear. The EQ is also excellent.

Jimmy

"TC" <tc@spammetodeathyoubastards.org> wrote in message news:45db3060\$1@linux...

>

> Is there any difference between the Great River 1NV that Mercenary sells, and the standard one? I actually can't find one that isn't an "ME", even Atlas sells them.

>

> <http://www.mercenaryeditions.com/me-1nv.shtml>

>

> I've been wanting to check out one of these, and I'm wondering how it would complement my current preamps:

>

> 1 UA LA610

> 2 x API 512c

> GT Vipre

> Avalon 737sp

>

> Primary application: vocals, acoustic guitar, electric, percussion.

>

> I'm seriously considering ditching the Avalon 737, as it gets almost no use these days. Either that or getting the Babyface mod to it.

>

> Any thoughts on the great river? I'm interested in how it would differ from the API's?

>

> I'm also considering getting something like an Avalon 747 for 2 buss coloring and eq. I tried one years ago and liked the extra dimension that it

seemed to give the mix.

>

> Any opinions are appreciated!

>

> Cheers,

>

> TCMy experiences with Dataviz were not very good either.

"DJ" <www.aarrrrggghhh!!!.com> wrote:

>This contains the oft lauded Macopener software. A total cluster**** which

>included the complete hosing of a brand new hard drive and a few scary

>moments as I scrambled to recover my boot sector on my OS drive which

>suddenly failed to be recognized after I ran the quick format utility on

a

>completely different drive (the aforementioned "hosed" drive)...and spent

a

>couple of hours on the phone with Dataviz tech support who were very nice

>and polite but finally threw in the towel themselves and referred me to

>another company.

>

>If you want to open Mac files on a PC, use Macdrive (www.mediafour.com).

>

>Wow! That's amazing that Apple would replace the DVD drive for free. The
damaged was caused by somebody else's defective product. So if my MacBook
was run over by a defective cement truck, would Apple replace it? ;)

James

"Mark McDermott" <nospam@stateofwail.com> wrote:

>

>

><http://blogs.zdnet.com/Apple/?p=441&tag=nl.e622>I'm sure that Mr. Jobs would use his private jet
to personally replace it!

;)

"James McCloskey" <excelsm@hotmail.com> wrote:

>

>Wow! That's amazing that Apple would replace the DVD drive for free. The

>damaged was caused by somebody else's defective product. So if my MacBook

>was run over by a defective cement truck, wo

Subject: Re: hey John.....your Total mix tutorial???

Posted by [rick](#) on Sat, 27 Jan 2007 10:39:30 GMT

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

>>
>>>>drives, both 10K and 7.5K RPM and also IDE RAID arrays with drives of
>>>>different cache amounts...you definitely have to compare 8MB cache to
> 16MB
>>>
>>>>cache on every brand that offers this. We need to know the fastest
>>>>possible
>>>
>>>>configuration that will work while weighing the risks of opting for
>>>>performance over security and any other stuff that might come up during
>>> your
>>>>odyssey.
>>>>
>>>>Don't let these guys pull the wool over your eyes. You need to know all
>>> of
>>>>these things.
>>>>
>>>>
>>>>;o)
>>>>
>>>>
>>>>"Chris Ludwig" <chrisl@adkproaudio.com> wrote in message
>>>>news:45dd1ccc@linux...
>>>>> exactly!!!!
>>>>>
>>>>>
>>>>>
>>>>> TCB wrote:
>>>>>
>>>>>>Main response--it doesn't matter anymore. Disk performance is about
>
>>>>>>100th
>>>>>>on the list of issues for DAWs.
>>>>>>Secondary response. The outside of the disk will have the highest
>>>>>>throughput,
>>>>>>i.e. number of MB/s. The lowest _latency_ will be when files are
>>>>>>contiguous,
>>>>>>regardless of which part of the disk they are on. Latency is a
>>>>>>function
>>>
>>>>>>how
>>>>>>much time the head of the drive spends seeking, i.e. not doing
>>>>>>anything
>>>
>>>>>>while

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>>>
>>>>>spins
>>>>>marginally faster. A 10k drive has more throughput than a 7.2k drive.
>>>>>None of which matters, at all, these days. Just format it and start
>>>>>recording.
>>>>>
>>>>>
>>>>>TCB
>>>>>
>>>>>"Brandon" <a@a.com> wrote:
>>>>>
>>>>>>Ok so if you partition a hard drive into two partitions which one
>>>>>>will
>>> be
>>>>>>on the outside of the disk and which will be on the inside of the
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>>>>>>One for recording and tracking (on the outside of the platter)
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>>>>>>I will be using Partition Magic 8.
>>>>>>thx
>>>>>>b
>>>>>>
>>>>>>
>>>>>>
>>>>>>
>>>>>
>>>>> --
>>>>> Chris Ludwig
>>>>> ADK
>>>>> chrisl@adkproaudio.com <mailto:chrisl@adkproaudio.com>
>>>>> www.adkproaudio.com <http://www.adkproaudio.com/>
>>>>> (859) 635-5762
>>>>>
>>>>>
>>>>>
>>>
>>
>>
>Also addressed the distortion in NOLimit...can you say monitor pot?

DOH!!!!

that one got me a time or two as well.....

;o)

"Jeff Hoover" <jkhoover@excite.com> wrote in message news:45dd4dc4@linux...

> We're Baaaack!

>

> Finally got it after cleaning out (including a couple pointers in the
> registry) and reinstalling all to default directories. For some reason,
> the effects subsystem wouldn't go anywhere else for me.

>

> Also addressed the distortion in NOLimit...can you say monitor pot?

>

> DOH!!!!

>

> Hoov

>> 2 places during that install as I recall - one in your choosen directory
>> installed path (in this case, that'd be C:\ParisPro) and the system
>> directory is the windows\system32\ensoniq\plugins folder (from memory,
>> not anywhere near a paris rig at the moment).....

>>

>> AA

>>

>> "Jeff Hoover" <jkhoover@excite.com> wrote in message

>> news:45dcea9b\$1@linux...

>>

>>>I assume the right path is where I installed parispro or is it to the
>>>ensoniq directory in the windows folder?

>>>

>>>JH

>>>

>>>Aaron Allen wrote:

>>>

>>>>Sounds to me like the subsystem FX patch didn't have the right install
>>>>path. Be 'very' observant of where it's installing that patch.

>>>>

>>>>Uninstall the patch completely.

>>>>Restart PC.

>>>>Install patch - watch that path! - install tries to add a line to it.

>>>>Restart PC.

>>>>Give it a try.

>>>>

>>>>AA

>>>>

>>>>"Jeff Hoover" <jkhoover@excite.com> wrote in message

>>>>news:45dce294\$1@linux...

>>>>

>>>>

>>>>>Aaron Allen wrote:
>>>>>
>>>>>
>>>>>>XP?
>>>>>>
>>>>>>"Jeff Hoover" <jkhoover@excite.com> wrote in message
>>>>>>news:45dc82ba\$1@linux...
>>>>>>
>>>>>>
>>>>>>
>>>>>>>I tried a reinstall to address issues. No EDS effects show up.
>>>>>>>Repeated the process same thing. VST's are showing up. Installing
>>>>>>>to c:\ParisPro.
>>>>>>>
>>>>>>>Where to next?
>>>>>>>
>>>>>>>Hoov
>>>>>>>
>>>>>>>
>>>>>>>
>>>>>>>
>>>>>>>yes...
>>>>>>>
>>>>>>>Hoov
>>>>>>>
>>>>>>>
>>I cant start recording because my MOTU 8 pre hasnt arrived yet. That is why
I have time to
think of this kind of stuff.

--

Thanks,

Brandon

"DJ" <www.aarrrrggghhh!!!.com> wrote in message news:45dda7b1@linux...
> ...or at least more dynamic.....
>
> ;o)
>
> "TCB" <nobody@ishere.com> wrote in message news:45dda3a7\$1@linux...
>>
>> I think when you record to a drive done with MaxBlast, though, everything
>> sound warmer.
>>
>> TCB

```

>>
>> "DJ" <www.aarrrrggghhh!!!.com> wrote:
>>> Damn right!!!.....though I always heard that if you use Partition Magic,
>> at
>>> some point, the *magic* is bound to happen.
>>>
>>>;o)
>>>
>>> "TCB" <nobody@ishere.com> wrote in message news:45dd23e8$1@linux...
>>>>
>>>> Don't forget to use that wonderful MaxBlast application Maxtor includes
>>
>>>> for
>>>> formatting!
>>>>
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>>>> "DJ" <www.aarrrrggghhh!!!.com> wrote:
>>>>> (gasp!!!!.....sputter!!!!!!).....Brandon!!!!..... .don't
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>>>>> to
>>>>> these guys!!!!.....spend at least two weeks testing different drives
>>>>> and
>>>>> permutations of drive arrays, like RAID 0, or 0-1 with both SCSI and
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>>>>>
>>>>> drives, both 10K and 7.5K RPM and also IDE RAID arrays with drives of
>>>>> different cache amounts...you definitely have to compare 8MB cache to
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```

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 >>>>>
 >>>>>
 >>>>
 >>>
 >>>
 >>
 >
 >http://www.youtube.com/watch?v=wX_ukHFNP7YWhile you're at it, you should partition a 40G
 5400 RPM notebook drive so
 that the 1st partition is only 2G and then compare the speed of this
 partition to the speed of a 10k RPM SATA RAID0 array that is 90% full. that
 should keep you busy until the MOTU shows up.

 ;o)

 "Brandon" <a@a.com> wrote in message news:45ddb785\$1@linux...
 >I cant start recording because my MOTU 8 pre hasnt arrived yet. That is why
 >I have time to
 > think of this kind of stuff.
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 >
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 > Thanks,
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 > Brandon
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>>>>>>> Chris Ludwig  
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>>>>>>> www.adkproaudio.com <http://www.adkproaudio.com/>  
>>>>>>> (859) 635-5762  
>>>>>>>  
>>>>>>>  
>>>>>>>  
>>>>>>>  
>>>>>>>  
>>>>>>>
```

let the phone calls begin. That was very cool. Always loved that song. Stevie just rocks. Thanks for posting that link DJ.

just getting in ?

Thanks

Dimitrios

"DJ" <www.aarrrrggghhh!!!.com> wrote:

>Dimitrios,

>

>If the spdif out of the A/D converter is fed to the Pulsar spdif input,
then

>the Pulsar would be receiving clock signal from the AD. To my knowledge,
a

>device can't be clocked from it's own digital output, only from a signal

>arriving at it's digital input if it is set to slave so the AD would not
be

>clocked from the Mytek.unless it has a BNC clock input that can be fed a
WC

>signal.

>

>As for the DAC-1, I use this and it will basically lock to the incoming

>clock signal from Paris/Mytek and then upsample this signal before

>converting it back to analog. The newer units upsample to 192k. Mine

>upsamples to 96k.

>

>That's pretty cool that you got VSTi to work with Paris.

>

>;o)

>

>

>

>"Dimitrios" <musurgio@otenet.gr> wrote in message news:45dda357\$1@linux...

>>

>> Hi,

>> I would like to ask:

>> My pulsar card is wordclocked from Mytek so the spdif input of the card

>> ,probably,

>> is driven from this clock.

>> FIRST QUESTION: I have a prototype AD converter (very nice) that has spdif

>> output. Now connecting this output to the Pulsar spdif input would the

>> ACTUAL

>> ad converter sound be CLOCKED ?

>> Up till now I was using Behringer ultramath converter which was accepting

>> the ad's spdif out and reclocked with mytek's wordclock using the bnc