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Subject: Re: Track Counts in Native Systems?  
Posted by [IOUOI](#) on Wed, 29 Nov 2006 15:23:54 GMT  
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Mike, I'll bet you can do it - I think for the VST synth count you're looking for you may have to go ahead & RAM up to 2 gigs, but if you're going to be working at 44.1k or 48k, I'm betting you should have no problems. My Native rig is slower than the one you're going to build & I can get into the 40's in terms of track counts, and at 88.2k, to boot, and I routinely use three verbs (Usually I'll have a room verb for drums, some other kind of reverb for other instruments, and yet another - usually a Plate - for vocals, plus most of the time another group for a delay.

Now, depending on what other EFX/plugin's I'm using on individual tracks, I may have to start "Freezing" some tracks in order to be able to keep the buffer settings at such levels that I can maintain low latency (if it gets down to this, usually I'll freeze the kick, snare & bass, and sometimes the rhythm guitars - most of the time, once I get whatever compression plugin applied to those, it's pretty much "set it and forget it" for me, anyway... plus if you need to "unfreeze" them it only takes a moment, and then to re-freeze it, just a couple of mere moments more).

Regarding your VSTi's - while I don't have a ton of experience using them (I have a few, just don't use them much), I understand that if you're using a lot of sample-based VSTi's you may be better off going with a third hard drive for streaming those samples (1x system drive, 1x audio drive, and 1x drive just for the VSTi samples); so you may want to do some homework on that to see if that's indeed the most-recommended type of setup.

Neil

"Mike R." <emarenot@yahoo.com> wrote:

>

>

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Subject: Track Counts in Native Systems?  
Posted by [emarenot](#) on Wed, 29 Nov 2006 16:52:35 GMT  
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This is a multi-part message in MIME format.

-----=\_NextPart\_000\_001F\_01C71393.B6F9E860  
Content-Type: text/plain;  
charset="iso-8859-1"  
Content-Transfer-Encoding: 7bit

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Subject: Re: Track Counts in Native Systems?

Posted by [LaMont](#) on Wed, 29 Nov 2006 17:44:07 GMT

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I was going to give track counts, but after re-thinking what I was going to post, I think there things to consider.

1) Doing both, composing and recording /mixing, I feel should done on 2 machines..

2) Having said, you can do both on a fast PC/Mac setup. However, to get optimum performance, you have to make some choices.

-Compose (midi Sequence) you song first with VSTis

-Then Either freeze or render those tracks audio tracks.

-You really want to do the above first, because when you start tracking live instruments or vocals and adding fxs(plugins) depending on you track count, your PC will start to slow down..

That's why I like the 2 PC/Mac approach. 1 for Instruments, & 1 for Audio recording and mixing..

"Mike R." <emarenot@yahoo.com> wrote:

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Subject: Re: Track Counts in Native Systems?  
Posted by [Dedric Terry](#) on Wed, 29 Nov 2006 19:59:58 GMT  
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> This message is in MIME format. Since your mail reader does not understand this format, some or all of this message may not be legible.

--B\_3247649999\_923925  
Content-type: text/plain; charset="ISO-8859-1"  
Content-transfer-encoding: 8bit

at low latency all the time for commercial spots, sound design, etc. I also have a full orchestral template that will run easily on this system.

as well.

If you are building, consider an Intel core duo, dual core duo (xeon), or if you have the budget, and quad core system. The Intel core duos are slightly

seeing from DAW builders. A quad core is the fastest, but there seem to be diminishing returns on dual quad cores.

Regards,  
Dedric

On 11/29/06 9:52 AM, in article 456d8e00@linux, "Mike R."  
<emarenot@yahoo.com> wrote:

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**Subject: Re: Track Counts in Native Systems?**  
Posted by [Chris Ludwig](#) on Wed, 29 Nov 2006 22:33:14 GMT  
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Hi Dedic,  
The Core2 based system seems to be the lowest latency setup from what I've seen and tested so far.  
The Quad core that I'm using in the home system atm is letting me run projects at the lowest latency of my Fireface 800 which is 48k buffer. I haven't been able to get any demo project or project that I've done to choke at that latency. I had to redo our in house Cubase/Nuendo benchmark just to push the damn thing. The 3 UAD cards don't really like the latency that low. They seem more stable at 128 and higher lol.

There is performance hits on the Xeons at the low latencies but at least with the quad core Xeons the extra horse power makes up for it.

The Core 2 Quad runs circles around any Dual-Core Opterons I've tested.

Here is the bench mark we use currently. It's based on the same methodology of the infamous Blofelds test but with out sound card performance issues. I haven't tried this on many machine other than the Opterons, Xeons and quads so not even sure what results there will be on pre-Core 2 Duo system or a AMD X2/AM2 system. I just finished it

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Close the program then reopen the default project and repeat the test  
for each latency setting.

If you have any problem or suggestion for the test let me know so I can  
fine tune it if needed.

[http://www.adkproaudio.com/downloads/set\\_me\\_free\\_benchmark.zip](http://www.adkproaudio.com/downloads/set_me_free_benchmark.zip)

Thanks Chris

Dedric Terry wrote:

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Chris Ludwig

ADK Pro Audio  
(859) 635-5762  
www.adkproaudio.com  
chrisl@adkproaudio.com

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Subject: Re: Track Counts in Native Systems?  
Posted by [Dedric Terry](#) on Thu, 30 Nov 2006 00:38:44 GMT  
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Hi Chris,

I'll run it tomorrow and let you know the results. I'm sure my X2 will give paltry results next to the quad and probably fall well short of the core duo as well. ;-)

I'm holding out for a dual quad core at some point, but a dual core-duo is also tempting.

Regards,  
Dedric

On 11/29/06 3:33 PM, in article 456e0841@linux, "Chris Ludwig"  
<chrisl@adkproaudio.com> wrote:

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Subject: Re: Track Counts in Native Systems?

Posted by [Chris Ludwig](#) on Thu, 30 Nov 2006 01:28:54 GMT

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FYI,

It would be great if anyone with Cubase/Nuendo Apple g5 machines want to run it the test it would be great to get the numbers.

Hopefully it will open and play fine on one. I used all internal plug ins so it should be fine.:)

Chris

Chris Ludwig wrote:

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Chris Ludwig

ADK

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

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

(859) 635-5762

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Subject: Re: Track Counts in Native Systems?

Posted by [Dedric Terry](#) on Fri, 01 Dec 2006 15:19:11 GMT

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Lamont - Chris and ADK build native DAWs as a business. As far as I know they don't sell ProTools or Soundscape, but Chris can speak for himself. I for one appreciate his participation, contribution and advice in PC discussions here and don't see how it is contradictory to studio consulting advice.

If you prefer dsp based, that's great. I can't see how upgrading a PC or Mac is more expensive than upgrading a dsp DAW a la ProTools, Soundscape, etc. Then consider that you still have a PC or Mac running those systems that will have to be upgraded at some point, in some form (e.g. a new version of PT software comes out, a new conversion tool update, etc).

And if you want to run Giga, NI softsynths, etc...same thing.

And fwiw, we were advising Mike R. on what options are out there now, not talking about updating our PCs once a month.

Best of luck!

Dedric

On 12/1/06 8:49 AM, in article 457040f5\$1@linux, "LaMont" <jjdpro@ameritech.net> wrote:

>  
> Chris and others,  
>  
> This is the reason I prefer DSP based DAWs. We spend way too much time on  
> PC/Mac upgrades, chip-sets, what won't play with what..etc..

>  
> You're constantly on the DAW upgrade path which to me, is getting very old.  
> This is strange coming from a DaW builder and studio DAW consultant. But,  
> I've had enough. After my last Opteron upgrade, I said "That's it".. :)  
>  
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> Chris Ludwig <chrisl@adkproaudio.com> wrote:  
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Posted by [LaMont](#) on Fri, 01 Dec 2006 15:49:25 GMT  
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Posted by [Neil](#) on Fri, 01 Dec 2006 16:22:17 GMT  
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There are, of course, advantages, too - which I think are pretty clear to most here, but it's not all shiny happy people in Digiland, either. Just some perspective there.

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Posted by [LaMont](#) on Fri, 01 Dec 2006 16:28:41 GMT  
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Subject: Re: Track Counts in Native Systems?  
Posted by [LaMont](#) on Fri, 01 Dec 2006 16:39:04 GMT  
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Hey Neil,

No, I was not only talking about Pro-Tools. But a little perspective. There are still alot (read more) Pro-Tool Mix3 setups than HD setups. So, you don;t have to upgrade the various Digi setups if you wan to..

I was talking about PArIs, Pro Tools, Soundscape..Radar.

P.S.

I personaly would not for over the money for the Creamware stuff for summing.

The truth of the matter toimprover your ITB summing, just get a decent analog or digital mixer(Mackie..Yamaha 01x 02r,). Most of the current summing boxes are nothing more than the forementioed mixers without the faders.. Electronic Musician did a summing shoot-out this year and the results were astounding.: needless to say , that a mackie VLZ summed just as good as most of the 5k summing boxes. So that should tell you something and it should tell you that by add a better mixer to sum than the DAW is all that's needed..

I use a Soundcraft Ghost console to sum for now, and yesit even improves Paris stereo imaging. For Neundo and Pro Tools, it takes it to another level.

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Subject: Re: Track Counts in Native Systems?  
Posted by [Jamie K](#) on Fri, 01 Dec 2006 16:46:30 GMT  
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---

The native DAW upgrade path on the OSX Mac goes like this:

- 1) Upgrade your software as available, if you feel it's worthwhile.
- 2) There is no 2.

That's it. If you feel the need to tinker with hardware go work on your car. ;^)

Granted, you can upgrade the Mac hardware any time you feel you need to. I've only done that once in the last six or so years, from a dual 1GHZ

G4 PowerMac to a dual 2.5 GHZ G5 PowerMac, just to increase plugin counts without needing to freeze tracks. My current box, several years old, is still plenty fast enough to do heavy audio recording duties for years to come.

Cheers,  
-Jamie  
www.JamieKruz.com

PS. Native Instruments Komplete 4 just arrived, finally.

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Subject: Re: Track Counts in Native Systems?  
Posted by [Chris Ludwig](#) on Fri, 01 Dec 2006 19:09:30 GMT  
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Hi Lamont,  
I understand totally what you saying I have a bunch of customers that because of how they work in the studio the DSP/DAW based hybrid is the way to go. They normally have a HD rig and then native rig for VSTi/MIDI type things. Paris has been the only DSP DAW system that ever came close to what PT is doing. To have something that behaves close to a PT rig in that it has good built in hardware/latency free effects that can be run on the majority of the channels at one time with out running out of DSP. The closest thing I've seen to this would be a Nuendo system with a Euphonix or Yamaha DM-2000 console type setup. Any of the quality Digital consoles out there now will easily put in the same or higher price range as a HD/control 24 type system.

It seems to me that most people needing the mostly native setup are composers types and home and small production studios. People who do a bit of all of that will most likely have a hybrid setup. Just look at the freaks on this forum!!! :)

Chris

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Subject: Re: Track Counts in Native Systems?  
Posted by [Jamie K](#) on Fri, 01 Dec 2006 19:10:05 GMT  
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---

Hey LaMont, the NI stuff (and my other 3rd party Audio Unit plugins, Zebra, GPO, Ivory, Cameleon, etc.) already run great on my dual 2.5GHZ G5.

Tons of tracks, no prob.

When I next upgrade the studio Mac it will be for animation/graphics production reasons. The upcoming 8 core Mac Pro would be good for that.

I'm also tempted by the new core 2 duo Mac laptops, they're about as fast as my dual G5 which is amazing for a laptop. But I'll probably hold out for a laptop that can hold more than 3GB RAM.

I'm in no rush, my dual G5 works great. I'm having fun with the new NI updates.

Cheers,  
-Jamie  
www.JamieKruz.com

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Chris Ludwig

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Subject: Re: Track Counts in Native Systems?  
Posted by [LaMont](#) on Fri, 01 Dec 2006 19:27:33 GMT  
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---

"PS. Native Instruments Komplete 4 just arrived, finally."

Hey Jamie, you'll need or want that new Mac Pro core dual to run all those "sweet" Native Instruments Vstis :)

But, you are right. You should only upgrade if there is a real need.

Jamie K <Meta@Dimensional.com> wrote:

>  
>The native DAW upgrade path on the OSX Mac goes like this:  
>  
> 1) Upgrade your software as available, if you feel it's worthwhile.  
>  
> 2) There is no 2.  
>  
>That's it. If you feel the need to tinker with hardware go work on your

>car. ;^)

>

>Granted, you can upgrade the Mac hardware any time you feel you need to.

>I've only done that once in the last six or so years, from a dual 1GHZ

>G4 PowerMac to a dual 2.5 GHZ G5 PowerMac, just to increase plugin

>counts without needing to freeze tracks. My current box, several years

>old, is still plenty fast enough to do heavy audio recording duties for

>years to come.

>

>Cheers,

> -Jamie

> [www.JamieKrutz.com](http://www.JamieKrutz.com)

>

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Subject: Re: Track Counts in Native Systems?  
Posted by [Chris Ludwig](#) on Fri, 01 Dec 2006 19:32:41 GMT  
[View Forum Message](#) <> [Reply to Message](#)

---

Hi Lamont,

Lamont wrote:

> Hey Neil,  
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> are still alot (read more) Pro-Tool Mix3 setups than HD setups. So, you don;t  
> have to upgrade the various Digi setups if you wan to..

Well maybe the converters :)

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> I was talking about PAris, Pro Tools, Soundscape..Radar.  
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Paris would really be the only one that was close enough functionality  
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> P.S.  
> I personaly would not for over the money for the Creamware stuff for summing.

ditto.

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> I use a Soundcraft Ghost console to sum for now, and yesit even improves  
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Although I'd rather wish they would do the mixers in a rack mount form  
with out the automation and faders just the routing/ summing and effects  
type parts.

Actually really wish that dead-end things like the UAD and TC power  
core and various firewire DSP would stop dicking around and just write  
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giant dongles now. I'd rather have a tiny USB one.

I'd rather see these DSP people put the effects on the external converters or purely digital I/O like MADI. Then be able to add DSP chips to the boxes to be able to add more processing power to the units. But I don't see that happening see that happening soon.

>  
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Chris

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Chris Ludwig

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Subject: Re: Track Counts in Native Systems?  
Posted by [LaMont](#) on Fri, 01 Dec 2006 21:15:41 GMT  
[View Forum Message](#) <> [Reply to Message](#)

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Lol!!! :)

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Subject: Re: Track Counts in Native Systems?  
Posted by [Dedric Terry](#) on Fri, 01 Dec 2006 21:46:08 GMT  
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---

Hey Lamont - I probably did misinterpret. I don't disagree about the advantages of hybrid/dsp at all.

I think most everyone on this board knows the pros and cons both ways, and most all of us have quite a bit of experience building systems (except Mac users perhaps ;-). After all, part of the admission fee here is that you be a bit of a DAW/gear slut. ;-)

Even Mac users have to admit it gets expensive buying a new Mac every 3 years or so just to gain more power to run VSTi's and plugins that demand it (depending on the user's requirements of course). However, that is no different with dsp rigs. You end up adding \$5000 Accel cards to an HD rig to add more power, unless you start with HD3 and never need more than that. So the costs are actually relative. You don't save money by going with ProTools or Soundscape or Fairlight, you just spend it in a different way, and once you add in the cost of comparable plugins in TDM format, you do spend more, no way around it really. It's a matter of going with what works best and makes you more productive and profitable.

As Chris pointed out in another post, native is most common in the pro realm for composing and sound design (which is most of what I do, along with audio post - hence my interest in Fairlight).

I also agree with him that dedicated dsp plugin cards are getting a bit long in the tooth.

Imho, it's time for Fairlight to get out of their closed market mindset and release CC-1 as a standalone generic dsp engine for DAWs, plugins, etc (or more likely, time for other developers to get on board with it - could be in the works already from the way it sounds....). Or perhaps Clearspeed will eventually have some DAW applications.

The goal for dsp/hybrid I prefer is a more open platform so we can begin to merge our preferences in native and dsp into

a truly expandable environment - more flexibility and upgradeability with that approach - better longterm investment, imho.

Regards,  
Dedric

"LaMont" <jjdpro@ameritech.net> wrote in message news:45704a29\$1@linux...

>

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Subject: Re: Track Counts in Native Systems?  
Posted by [Jamie K](#) on Fri, 01 Dec 2006 22:37:52 GMT  
[View Forum Message](#) <> [Reply to Message](#)

---

Hey Dedic,

OK, I had to look it up:

My dual 2.5GHZ G5 cost about \$1200 additional in 2004 after selling my

previous dual 1GHZ G4. I can't say that's all that expensive for the jump in processing power, especially considering the many things the system gets used for here.

I think Deej has me beat in having higher computer costs, and probably most everyone else here, too, who has been building machines and replacing them over and over in the last few years. And that's not even counting the time it takes to build them and debug the problems we hear about.

Meanwhile the trusty G5 still does all the tracks and AU/FX plugins I need. In fact it's gotten better with the periodic Logic updates, OSX updates and updates from plugin vendors, the system has gotten more solid and capable over time.

So I don't need to buy another computer for audio. Priceless. :^)

Any further studio computer upgrading here will be due to graphics/animation/video production. Although the G5 is doing reasonably well in those areas, too (was one reason I got it), so there's no big rush to upgrade.

I agree that it's a matter of going with what works best for whatever you do. And also, what you value in the process.

Cheers,  
-Jamie  
[www.JamieKrutz.com](http://www.JamieKrutz.com)

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>>>>>

>>>>> --

>>>>> Chris Ludwig

>>>>>

>>>>> ADK Pro Audio

>>>>> (859) 635-5762

>>>>> www.adkproaudio.com

>>>>> chrisl@adkproaudio.com

>

>

---

Subject: Re: Track Counts in Native Systems?

Posted by [Dedic Terry](#) on Sat, 02 Dec 2006 04:03:45 GMT

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Hey Jamie,

You and the other Mac users know we PC users just give you a hard time because your comps have fruit on them. ;-)

For my work, the only limit on what I could use is what I can afford (hence spaced upgrades have more to do with budget than obsolescence or problems) -

the same would be true if I were using Macs though.

I actually keep the older machines around for offline tasks and network use (effects/VSTi slaves, video playback, backup storage, etc). I just recently retired my G4 450 (former Paris Mac) and passed it along to my son for learning software, games, etc (he's 6, so it's more than enough for him).

BTW - speaking of intense animation, probably old news, but there's an HD video online from Animusic - I like some of their other work better, but fun all the same:

<http://www.apple.com/quicktime/guide/hd/animusic2dvd.html>

Regards,  
Dedric

PS: I'm starting to consider selling my HP laptop and getting a Macbook for net/email, etc, simply to be able to have both platforms handy. Just not sure how well I can integrate it into the studio at the moment to justify it though.

On 12/1/06 3:37 PM, in article 4570ac34@linux, "Jamie K" <Meta@Dimensional.com> wrote:

>  
> Hey Dedric,  
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> OK, I had to look it up:  
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> My dual 2.5GHZ G5 cost about \$1200 additional in 2004 after selling my  
> previous dual 1GHZ G4. I can't say that's all that expensive for the  
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> I agree that it's a matter of going with what works best for whatever  
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Subject: Re: Track Counts in Native Systems?  
Posted by [Neil](#) on Sat, 02 Dec 2006 04:51:10 GMT  
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Chris Ludwig <chrisl@adkproaudio.com> wrote:

>> P.S.  
>> I personally would not for over the money for the Creamware stuff for summing.  
>  
>ditto.

Why not? If you (Lamont) are willing to fork over \$2k for a digital mixer, and you can fork over \$750 instead & get a "digimixerinabox", what's the difference? ASSUMING it does, in fact, work well for summing - I'll admit I dunno yet... will find out in a week or so & you guys can determine for yourselves, since I'll post some more summing comparison clips at that time.

Neil

---

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Subject: Re: Track Counts in Native Systems?  
Posted by [DJ](#) on Sat, 02 Dec 2006 04:56:54 GMT  
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Jamie,

I'd say my computer costs were around \$1000.00, but yeah, the time it took to find the magic bullet was pretty unbelievable this time around. My problem was stability between the two DAWs that I'm now convinced was clock related (vis-a-vis syncing 3 x RME cards to ADAT sync and/or WC sync which was being generated from a standalone clock and 3 x MECs, each of which was compounding errors due to the latency between the EDS cards) and could likely have been solved by either getting a MADI based RME system or going with something like I've got here now. So far, this Pulsar system is like having Paris stability/latency on a native system and slaves flawlessly to Paris with no wierdness at all. Had I listened to Dimitrios a year ago and taken this plunge into the unknown, I would have saved myself a year of grief.

;o)

Deej

"Jamie K" <Meta@Dimensional.com> wrote in message news:4570ac34@linux...

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> -Jamie

> www.JamieKrutz.com

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>>>>> Opterons, Xeons and quads so not even sure what results there will be  
>>> on  
>>>>> pre-Core 2 Duo system or a AMD X2/AM2 system. I just finished it  
>>>>>  
>>>>> Haven't had a chance to right all the instructions out yet. Basically  
>>>>> load the project at 1024 buffer load as many magneto plug ins as you  
>>> you  
>>>>> can then save the project under a new name. Adding 1024 to the name  
>>>>> will  
>>>>> help keep track of the versions. Reload the project to confirm that  
>>>>> you  
>>>>> can run that number of plug ins. make note of the number.  
>>>>> Close the program then reopen the default project and repeat the test  
>>>>> for each latency setting.  
>>>>>  
>>>>> If you have any problem or suggestion for the test let me know so I  
>>>>> can  
>>>>> fine tune it if needed.  
>>>>>  
>>>>> [http://www.adkproaudio.com/downloads/set\\_me\\_free\\_benchmark.z ip](http://www.adkproaudio.com/downloads/set_me_free_benchmark.zip)  
>>>>>  
>>>>>  
>>>>> Thanks Chris  
>>>>>  
>>>>>  
>>>>>  
>>>>>  
>>>>> Dedric Terry wrote:  
>>>>>> I can't speak for an Athlon 64/4000, but on my X2 4400 I run a  
>>>>>> similar  
>>>>>> setup at low latency all the time for commercial spots, sound  
>>>>>> design,  
>>>>>> etc. I also have a full orchestral template that will run easily on  
>>>>>> this system.  
>>>>>>  
>>>>>> I've also run some projects of 50-100 tracks with plugins without  
>>>>>> problems as well.  
>>>>>>  
>>>>>> If you are building, consider an Intel core duo, dual core duo  
>>>>>> (xeon),  
>>>>>> or if you have the budget, and quad core system. The Intel core  
>>>>>> duos  
>>>>>> are slightly better performers than AMD chips at the moment - at  
>>>>>> least  
>>>>>> from specs I'm seeing from DAW builders. A quad core is the  
>>>>>> fastest,  
>>>>>> but there seem to be diminishing returns on dual quad cores.  
>>>>>>

>>>>>> Regards,  
>>>>>> Dedic  
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>>>>>> On 11/29/06 9:52 AM, in article 456d8e00@linux, "Mike R."  
>>>>>> <emarenot@yahoo.com> wrote:  
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>>>>>> While I'm definately going to continue to use Paris for tracking  
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>>>>>> in the box solution for writing. I'm thinking about building a  
>>>>>> computer around an Athlon 64 4000 and some sort of mobo with an  
>>> FSB  
>>>>>> around 1000mhz (I can't believe that sort of speed on the  
>>>>>> FSB -blows  
>>>>>> me away everytime I think about it..) I'd like to be able to run  
>>>>>> about five to six synths, three verbs on fx sends, and of course  
>>>>>> several comp plugs and eq's etc on the various channels. I'm  
>>>>>> wondering if anyone has a similar set up and what sort of track  
>>>>>> counts folks are getting.  
>>>>>>  
>>>>>>  
>>>>>> --  
>>>>>> Chris Ludwig  
>>>>>>  
>>>>>> ADK Pro Audio  
>>>>>> (859) 635-5762  
>>>>>> www.adkproaudio.com  
>>>>>> chrisl@adkproaudio.com  
>>

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Subject: Re: Track Counts in Native Systems?  
Posted by [chuck duffy](#) on Sat, 02 Dec 2006 14:43:14 GMT  
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"Actually really wish that dead-end things like the UAD and TC power core and various firewire DSP would stop dicking around and just write native plug-ins. I really think they are just using the hardware as giant dongles now. I'd rather have a tiny USB one."

AMEN CHRIS!

Chris Ludwig <chrisl@adkproaudio.com> wrote:  
>HI Lamont,  
>  
>  
>Lamont wrote:

>> Hey Neil,  
>> No, I was not only talking about Pro-Tools. But a little perspective.  
There  
>> are still alot (read more) Pro-Tool Mix3 setups than HD setups. So, you  
don;t  
>> have to upgrade the various Digi setups if you wan to..  
>  
>  
>Well maybe the converters :)  
>>  
>> I was talking about PAris, Pro Tools, Soundscape..Radar.  
>>  
>Paris would really be the only one that was close enough functionality  
>wise to be compared to Pro Tools. The other are really just glorified  
>hard disk recorders.  
>  
>  
>> P.S.  
>> I personally would not for over the money for the Creamware stuff for summing.  
>  
>ditto.  
>  
>>  
>> The truth of the matter toimprover your ITB summing, just get a decent  
analog  
>> or digital mixer(Mackie..Yamaha 01x 02r,). Most of the current summing  
boxes  
>> are nothing more than the forementioed mixers without the faders.. Electronic  
>> Musician did a summing shoot-out this year and the results were astounding.:  
>> needless to say , that a mackie VLZ summed just as good as most of the  
5k  
>> summing boxes. So that should tell you something and it should tell you  
that  
>> by add a better mixer to sum than the DAW is all that's needed..  
>>  
>> I use a Soundcraft Ghost console to sum for now, and yesit even improves  
>> Paris stereo imaging. For Neundo and Pro Tools, it takes it to another  
level.  
>>  
>>  
>  
>Although I'd rather wish they would do the mixers in a rack mount form  
>with out the automation and faders just the routing/ summing and effects  
  
>type parts.  
>  
>Actually really wish that dead-end things like the UAD and TC power  
>core and various firewire DSP would stop dicking around and just write

>native plug-ins. I really think they are just using the hardware as  
>giant dongles now. I'd rather have a tiny USB one.  
>  
>I'd rather see these DSP people put the effects on the external  
>converters or purely digital I/O like MADI. Then be able to add DSP  
>chips to the boxes to be able to add more processing power to the units.  
>But I don't see that happening see that happening soon.  
>  
>>  
>>  
>  
>>  
>  
>>  
>Chris  
>  
>--  
>Chris Ludwig  
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>www.adkproaudio.com  
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Subject: Re: Track Counts in Native Systems?  
Posted by [Jamie K](#) on Sat, 02 Dec 2006 15:33:37 GMT  
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Hey Deej,

\$1K in computer parts for all your variations and multiple systems since 2004, that's pretty good. I would have guessed higher. Certainly the labor and distraction is higher, but if you enjoy that then it isn't so much of a penalty. OTOH if/when you don't, then it is.

Add in your DSP FX stuff for things I'm doing native and you have me beat by far in straight investment layout. But then the question becomes: "what sounds best?" And that's subjective. There isn't an easy answer. "Listen to Dimitrios" is pretty good. Or just "listen." :^)

In Neil's test I preferred the native Cubase mix over the PARIS mix, which was interesting. I don't know if Cubase and Logic sound similar but I wouldn't be too surprised if they do. Did I prefer native because I'm used to mixing in Logic? Or is it because I've always liked a more open sound? Or is it because Neil didn't mix for the same sound in all scenarios but just left things flat for the test (a valid approach)? It's hard to say. I liked my PARIS mixes when I was using it, and I like my Logic mixes now. My guess is I can find my preferred sounds within

the toolsets of both.

I'm not in the school that sez everyone must have the same system. Diversity is great, choice is good. I enjoy hearing technical tales about struggles in the shared quest to make great music. I've walked a long path from 4 track to DAW myself.

I think the road I'm on now is simpler, more integrated and probably ultimately more affordable than the path you're on at this point, but our subjective levels of satisfaction may be comparable, now that things are finally working for both of us.

Cheers,  
-Jamie  
[www.JamieKrutz.com](http://www.JamieKrutz.com)

DJ wrote:

> Jamie,  
>  
> I'd say my computer costs were around \$1000.00, but yeah, the time it took  
> to find the magic bullet was pretty unbelievable this time around. My  
> problem was stability between the two DAWs that I'm now convinced was clock  
> related (vis-a-vis syncing 3 x RME cards to ADAT sync and/or WC sync which  
> was being generated from a standalone clock and 3 x MECs, each of which was  
> compounding errors due to the latency between the EDS cards) and could  
> likely have been solved by either getting a MAD1 based RME system or going  
> with something like I've got here now. So far, this Pulsar system is like  
> having Paris stability/latency on a native system and slaves flawlessly to  
> Paris with no wierdness at all. Had I listened to Dimitrios a year ago and  
> taken this plunge into the unknown, I would have saved myself a year of  
> grief.  
> ;o)  
> Deej  
>  
> "Jamie K" <Meta@Dimensional.com> wrote in message news:4570ac34@linux...  
>> Hey Dedic,  
>>  
>> OK, I had to look it up:  
>>  
>> My dual 2.5GHZ G5 cost about \$1200 additional in 2004 after selling my  
>> previous dual 1GHZ G4. I can't say that's all that expensive for the jump  
>> in processing power, especially considering the many things the system  
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>> most everyone else here, too, who has been building machines and replacing  
>> them over and over in the last few years. And that's not even counting the

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>> So I don't need to buy another computer for audio. Priceless. :^)  
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>> Any further studio computer upgrading here will be due to  
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>> to upgrade.  
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>> do. And also, what you value in the process.  
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>>> to merge our preferences in native and dsp into  
>>> a truly expandable environment - more flexibility and upgradeability  
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>>>  
>>> Regards,  
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>>>> pretty much quality as a PC -DAW gear slut. But, I recognized that after  
>>>> my last mega (Dual-Dual-core Opteron setup), that I was spending a lot  
>>>> of  
>>>> money for a solution that frankly, i could've purchased a HD1 or maybe  
>>>> an  
>>>> HD2..But, that's just me.. I prefer DSP based systems than Native.  
>>>>  
>>>>  
>>>> I too build DAWs-in my neck of the woods(Michgan) for studios,  
>>>>  
>>>> Detric Terry <dterry@keyofd.net> wrote:  
>>>>> Lamont - Chris and ADK build native DAWs as a business. As far as I  
>>>>> know  
>>>>> they don't sell ProTools or Soundscape, but Chris can speak for  
>>>>> himself.  
>>>>> I  
>>>>> for one appreciate his participation, contribution and advice in PC  
>>>>> discussions here and don't see how it is contradictory to studio  
>>>>> consulting

>>>> advice.  
>>>>  
>>>> If you prefer dsp based, that's great. I can't see how upgrading a PC  
>>>> or  
>>>> Mac is more expensive than upgrading a dsp DAW a la ProTools,  
>>>> Soundscape,  
>>>> etc. Then consider that you still have a PC or Mac running those  
>>>> systems  
>>>> that will have to be upgraded at some point, in some form (e.g. a new  
>>>> version of PT software comes out, a new conversion tool update, etc).  
>>>>  
>>>> And if you want to run Giga, NI softsynths, etc...same thing.  
>>>>  
>>>> And fwiw, we were advising Mike R. on what options are out there now,  
>>>> not  
>>>> talking about updating our PCs once a month.  
>>>>  
>>>> Best of luck!  
>>>>  
>>>> Detric  
>>>>  
>>>>  
>>>> On 12/1/06 8:49 AM, in article 457040f5\$1@linux, "LaMont"  
>>>> <jjdpro@ameritech.net> wrote:  
>>>>  
>>>>> Chris and others,  
>>>>>  
>>>>> This is the reason I prefer DSP based DAWS. We spend way too much time  
>>>> on  
>>>>> PC/Mac upgrades, chip-sets, what won't play with what..etc..  
>>>>>  
>>>>> You're constantly on the DAW upgrade path which to me, is getting very  
>>>> old.  
>>>>> This is strange coming from a DaW builder and studio DAW consultant.  
>>>>> But,  
>>>>> I've had enough. After my last Opteron upgrade, I said "That's it"..  
>>>>> :)  
>>>>>  
>>>>>  
>>>>>  
>>>>>  
>>>>> Chris Ludwig <chrisl@adkproaudio.com> wrote:  
>>>>>> Hi Detric,  
>>>>>> The Core2 based system seems to be the lowest latency setup from what  
>>>>>> I've seen and tested so far.  
>>>>>> The Quad core that I'm using in the home system atm is letting me run  
>>>>>> projects at the lowest latency of my Fireface 800 which is 48k  
>>>>>> buffer.



>>>>>> I haven't been able to get any demo project or project that I've done  
>>>> to  
>>>>>> choke at that latency. I had to redo our in house Cubase/Nuendo  
>>>>>> benchmark just to push the damn thing. The 3 UAD cards don't really  
>>>>>> like  
>>>>>> the latency that low. They seem more stable at 128 and higher lol.  
>>>>>>  
>>>>>> There is performance hits on the Xeons at the low latencies but at  
>>>>>> least  
>>>>>> with the quad core Xeons the extra horse power makes up for it.  
>>>>>>  
>>>>>> The Core 2 Quad runs circles around any Dual-Core Opterons I've  
>>>>>> tested.  
>>>>>>  
>>>>>>  
>>>>>> Here is the bench mark we use currently. It's based on the same  
>>>>>> methodology of the infamous Blofelds test but with out sound card  
>>>>>> performance issues. I haven't tried this on many machine other than  
>>>>>> the  
>>>>>> Opterons, Xeons and quads so not even sure what results there will be  
>>>> on  
>>>>>> pre-Core 2 Duo system or a AMD X2/AM2 system. I just finished it  
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>>>>>> Haven't had a chance to right all the instructions out yet. Basically  
>>>>>> load the project at 1024 buffer load as many magneto plug ins as you  
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>

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Subject: Re: Track Counts in Native Systems?  
Posted by [Jamie K](#) on Sat, 02 Dec 2006 16:14:20 GMT  
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Dedric Terry wrote:

> Hey Jamie,

>

> You and the other Mac users know we PC users just give you a hard time  
> because your comps have fruit on them. ;-)

Jealousy is such an ugly thing. I'll send you a banana sticker. ;^)

> For my work, the only limit on what I could use is what I can afford (hence  
> spaced upgrades have more to do with budget than obsolescence or problems) -  
> the same would be true if I were using Macs though.

Yeah, most of us have to work within some sort of budget.

Over the years Mac critics, me among them, have argued that Macs are more expensive. Too expensive. And I had other, technical, criticisms of previous generations of Macs.

However, more recently the value proposition has improved, and OSX is a huge step forward for Apple.

Although the old perceptions have inertia, the reality, from my current experience, is much better.

> I actually keep the older machines around for offline tasks and network use  
> (effects/VSTi slaves, video playback, backup storage, etc). I just recently  
> retired my G4 450 (former Paris Mac) and passed it along to my son for  
> learning software, games, etc (he's 6, so it's more than enough for him).

I have an old Pentium box lying around that cost around \$3k. It depreciated almost immediately and wasn't worth selling, so I kept it for BeOS, Linux and compatibility with old Win98 stuff just in case. It now sits with my beloved Amigas in retirement and watches the action happening on the OSX box.

My G5 is fast enough that I don't need other computers slaved for FX or instrument plugins. That makes things much simpler. The G5 can also address enough HD space that I don't need another server or, for that matter, a network. It has enough RAM (currently 2.5GB, can address up to

8GB) to do a reasonable amount of multitasking.

It's really convenient to do it all on one fast, capable box in the studio. Less bailing wire, duct tape and magic incantations needed to hold it all together. :^)

Almost no administration time needed.

> BTW - speaking of intense animation, probably old news, but there's an HD  
> video online from Animusic - I like some of their other work better, but fun  
> all the same:  
> <http://www.apple.com/quicktime/guide/hd/animusic2dvd.html>

Clever stuff. I saw another one of those a while back that also impressed me.

Cheers,  
-Jamie  
[www.JamieKrutz.com](http://www.JamieKrutz.com)

> Regards,  
> Dedic  
>  
> PS: I'm starting to consider selling my HP laptop and getting a Macbook for  
> net/email, etc, simply to be able to have both platforms handy. Just not  
> sure how well I can integrate it into the studio at the moment to justify it  
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>>>> money for a solution that frankly, I could've purchased a HD1 or maybe an  
>>>> HD2..But, that's just me.. I prefer DSP based systems than Native.  
>>>>  
>>>>  
>>>> I too build DAWs-in my neck of the woods(Michigan) for studios,  
>>>>  
>>>> Dedic Terry <dterry@keyofd.net> wrote:  
>>>>> Lamont - Chris and ADK build native DAWs as a business. As far as I know  
>>>>> they don't sell ProTools or Soundscape, but Chris can speak for himself.  
>>>>> I  
>>>>> for one appreciate his participation, contribution and advice in PC  
>>>>> discussions here and don't see how it is contradictory to studio  
>>>>> consulting  
>>>>> advice.  
>>>>>  
>>>>> If you prefer dsp based, that's great. I can't see how upgrading a PC or  
>>>>> Mac is more expensive than upgrading a dsp DAW a la ProTools, Soundscape,  
>>>>> etc. Then consider that you still have a PC or Mac running those systems  
>>>>> that will have to be upgraded at some point, in some form (e.g. a new  
>>>>> version of PT software comes out, a new conversion tool update, etc).  
>>>>>

>>>>> And if you want to run Giga, NI softsynths, etc...same thing.  
>>>>>  
>>>>> And fwiw, we were advising Mike R. on what options are out there now, not  
>>>>> talking about updating our PCs once a month.  
>>>>>  
>>>>> Best of luck!  
>>>>>  
>>>>> Detric  
>>>>>  
>>>>>  
>>>>> On 12/1/06 8:49 AM, in article 457040f5\$1@linux, "LaMont"  
>>>>> <jjdpro@ameritech.net> wrote:  
>>>>>  
>>>>>> Chris and others,  
>>>>>>  
>>>>>> This is the reason I prefer DSP based DAWS. We spend way too much time  
>>>>> on  
>>>>>> PC/MAC upgrades, chip-sets, what won't play with what..etc..  
>>>>>>  
>>>>>> You're constantly on the DAW upgrade path which to me,is getting very  
>>>>> old.  
>>>>>> This is strange coming from a DaW builder and studio DAW consultant.  
>>>>>> But,  
>>>>>> I've had enough. After my last Opteron upgrade, I said "That's it".. :)  
>>>>>>  
>>>>>>  
>>>>>>  
>>>>>>  
>>>>>> Chris Ludwig <chrisl@adkproaudio.com> wrote:  
>>>>>>> Hi Detric,  
>>>>>>> The Core2 based system seems to be the lowest latency setup from what  
>>>>>>> I've seen and tested so far.  
>>>>>>> The Quad core that I'm using in the home system atm is letting me run  
>>>>>>> projects at the lowest latency of my Fireface 800 which is 48k buffer.  
>>>>>>> I haven't been able to get any demo project or project that I've done  
>>>>> to  
>>>>>>> choke at that latency. I had to redo our in house Cubase/Nuendo  
>>>>>>> benchmark just to push the damn thing. The 3 UAD cards don't really  
>>>>>>> like  
>>>>>>> the latency that low. They seem more stable at 128 and higher lol.  
>>>>>>>  
>>>>>>> There is performance hits on the Xeons at the low latencies but at  
>>>>>>> least  
>>>>>>> with the quad core Xeons the extra horse power makes up for it.  
>>>>>>>  
>>>>>>> The Core 2 Quad runs circles around any Dual-Core Opterons I've tested.  
>>>>>>>  
>>>>>>>

>>>>>> Here is the bench mark we use currently. It's based on the same  
>>>>>> methodology of the infamous Blofelds test but with out sound card  
>>>>>> performance issues. I haven't tried this on many machine other than the  
>>>>>> Opterons, Xeons and quads so not even sure what results there will be  
>>>> on  
>>>>>> pre-Core 2 Duo system or a AMD X2/AM2 system. I just finished it  
>>>>>>  
>>>>>> Haven't had a chance to right all the instructions out yet. Basically  
>>>>>> load the project at 1024 buffer load as many magneto plug ins as you  
>>>> you  
>>>>>> can then save the project under a new name. Adding 1024 to the name  
>>>>>> will  
>>>>>> help keep track of the versions. Reload the project to confirm that you  
>>>>>> can run that number of plug ins. make note of the number.  
>>>>>> Close the program then reopen the default project and repeat the test  
>>>>>> for each latency setting.  
>>>>>>  
>>>>>> If you have any problem or suggestion for the test let me know so I can  
>>>>>> fine tune it if needed.  
>>>>>>  
>>>>>> [http://www.adkproaudio.com/downloads/set\\_me\\_free\\_benchmark.z ip](http://www.adkproaudio.com/downloads/set_me_free_benchmark.zip)  
>>>>>>  
>>>>>>  
>>>>>> Thanks Chris  
>>>>>>  
>>>>>>  
>>>>>>  
>>>>>>  
>>>>>> Dedic Terry wrote:  
>>>>>>> I can't speak for an Athlon 64/4000, but on my X2 4400 I run a similar  
>>>>>>> setup at low latency all the time for commercial spots, sound design,  
>>>>>>> etc. I also have a full orchestral template that will run easily on  
>>>>>>> this system.  
>>>>>>>  
>>>>>>> I've also run some projects of 50-100 tracks with plugins without  
>>>>>>> problems as well.  
>>>>>>>  
>>>>>>> If you are building, consider an Intel core duo, dual core duo (xeon),  
>>>>>>> or if you have the budget, and quad core system. The Intel core duos  
>>>>>>> are slightly better performers than AMD chips at the moment - at least  
>>>>>>> from specs I'm seeing from DAW builders. A quad core is the fastest,  
>>>>>>> but there seem to be diminishing returns on dual quad cores.  
>>>>>>>  
>>>>>>> Regards,  
>>>>>>> Dedic  
>>>>>>>  
>>>>>>> On 11/29/06 9:52 AM, in article 456d8e00@linux, "Mike R."  
>>>>>>> <emarenot@yahoo.com> wrote:



>>>>>>>  
>>>>>>> While I'm definately going to continue to use Paris for tracking  
>>>> and  
>>>>>>> mixing, I've been thinking more and more about getting some sort  
>>>> of  
>>>>>>> in the box solution for writing. I'm thinking about building a  
>>>>>>> computer around an Athlon 64 4000 and some sort of mobo with an  
>>>> FSB  
>>>>>>> around 1000mhz (I can't believe that sort of speed on the  
>>>>>>> FSB -blows  
>>>>>>> me away everytime I think about it..) I'd like to be able to run  
>>>>>>> about five to six synths, three verbs on fx sends, and of course  
>>>>>>> several comp plugs and eq's etc on the various channels. I'm  
>>>>>>> wondering if anyone has a similar set up and what sort of track  
>>>>>>> counts folks are getting.  
>>>>>>>  
>>>>>>>  
>>>>>>> --  
>>>>>>> Chris Ludwig  
>>>>>>>  
>>>>>>> ADK Pro Audio  
>>>>>>> (859) 635-5762  
>>>>>>> www.adkproaudio.com  
>>>>>>> chrisl@adkproaudio.com  
>>>  
>

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Subject: Re: Track Counts in Native Systems?  
Posted by [Jamie K](#) on Sat, 02 Dec 2006 16:24:35 GMT  
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---

LaMont wrote:  
> this whole vintage craze is really playing on the  
> fears of users who are so desperate to make a "Hit-Record" hat they want evey  
> out-board peice of gear that was used n a given hit-record. These folks don't  
> realize that although you may be able o purchase that old Putec, you'll never  
> be able to purchase : Talent, and song-writing /arranging skills..

Well put, LaMont.

I think the tools are at the point where it's possible to create an excellent mix on anything from Garage Band on up.

The most important piece of gear is a discerning brain connected to a good, working set of ears.

Cheers,

-Jamie  
www.JamieKrutz.com

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Subject: Re: Track Counts in Native Systems?  
Posted by [LaMont](#) on Sat, 02 Dec 2006 16:31:00 GMT  
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---

Chris,  
Great points..!!  
you wrote:

Although I'd rather wish they would do the mixers in a rack mount form with out the automation and faders just the routing/ summing and effects type parts.

-The Neve rack mixer(summing) is becoming the go unit for ITB mixing. I only used the Ghost because it was such a great deal and great performance. a true hidden gem..

You wrote:

Actually really wish that dead-end things like the UAD and TC power core and various firewire DSP would stop dicking around and just write native plug-ins. I really think they are just using the hardware as giant dongles now. I'd rather have a tiny USB one.

I totally agree! That's why I have not purchased a UAD card nor a Powercore to this day. I true rip-off dongle.(IMHO)

They recently interviewd the UAD guys and the big question was put to them: "Why not add more power to the card"?

UAD: answer was : Because they could not find a "cost-effective "DSp that would not raise the price of the card???

So, that to me sounds like: To keep their profit margins, they want to keep keep using "Cheap" "Video" cards that today our cost is around 30 bucks. UAD is so cheap they don't want add any additioinal DSP or go with a newer NVIDIA or card to give their customers more DSP juice. What a rip-off..

I can do a good mix with some Waves, SSL-Waves, URS, Sonalkis, McDSPs, and some others. Of course PARIS plugis. I don't need UAD in it's preset state.

Vendors like Vintagekings and other vinage pro0audio dealers and manufactuers are getting away with grand-larceny these days. Yes ,I do belive in free market blah blah.. But, this whole vintage craze is really playing on the fears of users who are so desperate to make a "Hit-Record" hat they want evey out-board peice of gear that was used n a given hit-record. These folks don't realize that although you may be able o purchase that old Putec, you'll never

be able to purchase : Talent, and song-writing /arranging skills..  
Stupid. Then you have all these so-called summing mixers that are no more boutique than the man on the moon. But, you know how the pro-audio hype bandwagon starts,..Now, most feel that they must have a 5k summing mixer to make great records.

And let's not get into the whold Mic-Pre/ Microphone rip-offs these days..

Chris Ludwig <chrisl@adkproaudio.com> wrote:

>HI Lamont,

>

>

>Lamont wrote:

>> Hey Neil,

>> No, I was not only talking about Pro-Tools. But a little perspective.

There

>> are still alot (read more) Pro-Tool Mix3 setups than HD setups. So, you don;t

>> have to upgrade the various Digi setups if you wan to..

>

>

>Well maybe the converters :)

>>

>> I was talking about PAris, Pro Tools, Soundscape..Radar.

>>

>Paris would really be the only one that was close enough functionality

>wise to be compared to Pro Tools. The other are really just glorified

>hard disk recorders.

>

>

>> P.S.

>> I personally would not for over the money for the Creamware stuff for summing.

>

>ditto.

>

>>

>> The truth of the matter toimprover your ITB summing, just get a decent analog

>> or digital mixer(Mackie..Yamaha 01x 02r,). Most of the current summing boxes

>> are nothing more than the forementioed mixers without the faders.. Electronic

>> Musician did a summing shoot-out this year and the results were astounding.:

>> needless to say , that a mackie VLZ summed just as good as most of the 5k

>> summing boxes. So that should tell you something and it should tell you

that

>> by add a better mixer to sum than the DAW is all that's needed..

>>

>> I use a Soundcraft Ghost console to sum for now, and yesit even improves

>> Paris stereo imaging. For Neundo and Pro Tools, it takes it to another level.

>>

>>

>

>Although I'd rather wish they would do the mixers in a rack mount form

>with out the automation and faders just the routing/ summing and effects

>type parts.

>

>Actually really wish that dead-end things like the UAD and TC power

>core and various firewire DSP would stop dicking around and just write

>native plug-ins. I really think they are just using the hardware as

>giant dongles now. I'd rather have a tiny USB one.

>

>I'd rather see these DSP people put the effects on the external

>converters or purely digital I/O like MADI. Then be able to add DSP

>chips to the boxes to be able ti add more processing power to the units.

>But I don't see that happening see that happening soon.

>

>>

>>

>

>>

>

>Chris

>

>--

>Chris Ludwig

>

>ADK Pro Audio

>(859) 635-5762

>www.adkproaudio.com

>chrisl@adkproaudio.com

---

Subject: Re: Track Counts in Native Systems?

Posted by [LaMont](#) on Sat, 02 Dec 2006 16:36:50 GMT

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Why Not you ask.?

Well, wasting time for the un-known. When, ther are proven "know' summing solutions that already rock..

Just because somebody takes a a PCI card summing approach, does not mean that a digital mixer approach is not supperior.Which it is..Plus, you go down to your loac Pro-audio storea adn actually try it out before you buy..

Mainly, I belive in PCI summing(Paris), but Most DAW users often mix the boat and have not realized thethe great benefits in a having a Digital mixer to you DAW..

"Neil" <OIUOIU@OIU.com> wrote:

>

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

>

>>> P.S.

>>> I personally would not for over the money for the Creamware stuff for summing.

>>

>>ditto.

>

>Why not? If you (Lamont) are willing to fork over \$2k for a  
>digital mixer, and you can fork over \$750 instead & get a  
>"digimixerinabox", what's the difference? ASSUMING it does, in  
>fact, work well for summing - I'll admit I dunno yet... will  
>find out in a week or so & you guys can determine for  
>yourselves, since I'll post some more summing comparison clips  
>at that time.

>

>Neil

---

Subject: Re: Track Counts in Native Systems?

Posted by [DJ](#) on Sat, 02 Dec 2006 16:51:12 GMT

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Hey Jamie,

"Jamie K" <Meta@Dimensional.com> wrote in message news:45719a46@linux...

>

> Hey Deej,

>

> \$1K in computer parts for all your variations and multiple systems since  
> 2004, that's pretty good. I would have guessed higher.

I was only referring to the DAW running Cubase SX. Not the audio hardware. My Paris computer has been the same since 2003 but if you lumped them together and took into account a new HD here and there, yeh.I've spent a bundle. Add the Audio hardware and it's a pretty big bundle. Not the equivalent of a HD3 system, but not cheap either. I've got 3 x RME cards sitting here in boxes right now that I may EBay pretty soon to rcover the

costs of part of this bundle

Certainly the

- > labor and distraction is higher, but if you enjoy that then it isn't so
- > much of a penalty. OTOH if/when you don't, then it is.

Well, I enjoy learning about this stuff, getting my hands dirty and my brain fried.

- >
- > Add in your DSP FX stuff for things I'm doing native and you have me beat
- > by far in straight investment layout.

O'yeah.....4 x UAD-1 cards plus the plugins for them and 3 x Magma's do add up. Still, I'm nowhere near a PT HD system.

But then the question

- > becomes: "what sounds best?" And that's subjective.

Agreed. A lot of my rig involved just seeing if I could pull it off because I was curious (or maybe obsessed ;o)

There isn't an easy

- > answer. "Listen to Dimitrios" is pretty good. Or just "listen." :^)

Well, Dimitrios has been telling me about the Pulsars for a while. I was seeking stability more than anything else after being at war with the RME system for a year and never achieving a reliable sync situation. It was workable, but during mixes, I could depend on a total lockup \*at least\* once every hour and sometimes more often. I had Cubase set to automatically backup my projects every 3 minutes so I seldom lost anything unless I was doing editing when it happened. It was very frustrating. My thinking was to try the Pulsars and possibly change my mode of working to mixing native because I didn't expect this dual DAW scenario to be any more stable so I figured I'd end up setting Paris aside and use the Pulsars. Having this clocking stability was a very happy accident actually. I didn't expect it at all. If the Pulsars didn't work, I was going to go with a MAD1 system but it really was wayyyy more than I wanted to spend and didn't really have the I/O options that I needed without spending a fortune on overkill.

- >
- > In Neil's test I preferred the native Cubase mix over the PARIS mix, which
- > was interesting. I don't know if Cubase and Logic sound similar but I
- > wouldn't be too surprised if they do. Did I prefer native because I'm used
- > to mixing in Logic? Or is it because I've always liked a more open sound?
- > Or is it because Neil didn't mix for the same sound in all scenarios but
- > just left things flat for the test (a valid approach)? It's hard to say.

I liked Neil's mixes too. All of them. The differences were subtle though I immediately picked the Paris mix as being the \*most\* pleasing to my ear. Was

it borne of familiarity with the sonic footprint rather than the superiority of the sonic footprint????...well, like you said, it's subjective and I'm sure that familiarity had something to do with it. One reason I work the way I do is because I like it. Tracking in Paris is just so effortless for me and mixing on both platforms is something I'm comfortable with. In some ways, it's a PITA, but in others it's extremely flexible and easy to use to achieve certain things.

I liked my PARIS mixes when I was using it, and I like  
> my Logic mixes now. My guess is I can find my preferred sounds within the  
> toolsets of both.

So it's all good. We get pretty overwound by the details involved in this stuff (speaking generally now as referring to engineers) but that's in our nature as anal retentive perfectionistic types and I think that whether one's primary focus in this endeavor is as a musician/producer or as an engineer, the devil is always in the details at some point. It's the thing that makes the studio world go 'round.

>  
> I'm not in the school that sez everyone must have the same system.  
> Diversity is great, choice is good. I enjoy hearing technical tales about  
> struggles in the shared quest to make great music. I've walked a long path  
> from 4 track to DAW myself.

Well, that's good, 'cause otherwise, you would have probably hired a hit man to shut me up by now. ;oD

>  
> I think the road I'm on now is simpler, more integrated and probably  
> ultimately more affordable than the path you're on at this point, but our  
> subjective levels of satisfaction may be comparable, now that things are  
> finally working for both of us.

I'm going to be exploring the possibilities of this Scope system with the thought in mind of eventually migrating to a native platform. This will involve jumping through a few hoops and mapping a controller to the mixer. I haven't decided which one yet and I'm not in any hurry. The Scope mixer and routing is what I always dreamed Paris would be. Unfortunately, VST plugins cannot be integrated into it without a third party VST application so the jury is still out as to using it as my mix platform. They do have some nice, high quality plugins and I am starting to explore them. they also have a standalone recording app called VDAT and some kind of editor too, but it's pretty hard to beat Cubase for VST integration and editing (and I'm sure Logic is just as powerful).

>  
> Cheers,  
> -Jamie  
> [www.JamieKrutz.com](http://www.JamieKrutz.com)

Take care,

Deej

>  
>  
> DJ wrote:  
>> Jamie,  
>>  
>> I'd say my computer costs were around \$1000.00, but yeah, the time it  
>> took to find the magic bullet was pretty unbelievable this time around.  
>> My problem was stability between the two DAWs that I'm now convinced was  
>> clock related (vis-a-vis syncing 3 x RME cards to ADAT sync and/or WC  
>> sync which was being generated from a standalone clock and 3 x MECs, each  
>> of which was compounding errors due to the latency between the EDS cards)  
>> and could likely have been solved by either getting a MADI based RME  
>> system or going with something like I've got here now. So far, this  
>> Pulsar system is like having Paris stability/latency on a native system  
>> and slaves flawlessly to Paris with no wierdness at all. Had I listened  
>> to Dimitrios a year ago and taken this plunge into the unknown, I would  
>> have saved myself a year of grief.  
>> ;o)  
>> Deej  
>>  
>> "Jamie K" <Meta@Dimensional.com> wrote in message news:4570ac34@linux...  
>>> Hey Dedic,  
>>>  
>>> OK, I had to look it up:  
>>>  
>>> My dual 2.5GHZ G5 cost about \$1200 additional in 2004 after selling my  
>>> previous dual 1GHZ G4. I can't say that's all that expensive for the  
>>> jump in processing power, especially considering the many things the  
>>> system gets used for here.  
>>>  
>>> I think Deej has me beat in having higher computer costs, and probably  
>>> most everyone else here, too, who has been building machines and  
>>> replacing them over and over in the last few years. And that's not even  
>>> counting the time it takes to build them and debug the problems we hear  
>>> about.  
>>>  
>>> Meanwhile the trusty G5 still does all the tracks and AU/FX plugins I  
>>> need. In fact it's gotten better with the periodic Logic updates, OSX  
>>> updates and updates from plugin vendors, the system has gotten more  
>>> solid and capable over time.  
>>>  
>>> So I don't need to buy another computer for audio. Priceless. :^)  
>>>  
>>> Any further studio computer upgrading here will be due to



>>> graphics/animation/video production. Although the G5 is doing reasonably  
>>> well in those areas, too (was one reason I got it), so there's no big  
>>> rush to upgrade.  
>>>  
>>> I agree that it's a matter of going with what works best for whatever  
>>> you do. And also, what you value in the process.  
>>>  
>>> Cheers,  
>>> -Jamie  
>>> www.JamieKruz.com  
>>>  
>>>  
>>> Detric Terry wrote:  
>>>> Hey Lamont - I probably did misinterpret. I don't disagree about the  
>>>> advantages of hybrid/dsp at all.  
>>>> I think most everyone on this board knows the pros and cons both ways,  
>>>> and most all of us have quite a bit of experience building  
>>>> systems (except Mac users perhaps ;-). After all, part of the  
>>>> admission fee here is that you be a bit of a DAW/gear slut. ;-)  
>>>>  
>>>> Even Mac users have to admit it gets expensive buying a new Mac every 3  
>>>> years or so just to gain more power to  
>>>> run VSTi's and plugins that demand it (depending on the user's  
>>>> requirements of course). However, that is no different with dsp rigs.  
>>>> You end up adding \$5000 Accel cards to an HD rig to add more power,  
>>>> unless you start with HD3 and never need more than that.  
>>>> So the costs are actually relative. You don't save money by going with  
>>>> ProTools or Soundscape or Fairlight, you just spend it in a different  
>>>> way,  
>>>> and once you add in the cost of comparable plugins in TDM format, you  
>>>> do spend more, no way around it really. It's a matter of going with  
>>>> what works best and makes you more productive and profitable.  
>>>>  
>>>> As Chris pointed out in another post, native is most common in the pro  
>>>> realm for composing and sound design (which is most  
>>>> of what I do, along with audio post - hence my interest in Fairlight).  
>>>>  
>>>> I also agree with him that dedicated dsp plugin cards are getting a bit  
>>>> long in the tooth.  
>>>> Imho, it's time for Fairlight to get out of their closed market mindset  
>>>> and release CC-1 as a standalone generic dsp engine  
>>>> for DAWs, plugins, etc (or more likely, time for other developers to  
>>>> get on board with it - could be in the  
>>>> works already from the way it sounds....). Or perhaps Clearspeed will  
>>>> eventually have some DAW applications.  
>>>>  
>>>> The goal for dsp/hybrid I prefer is a more open platform so we can  
>>>> begin to merge our preferences in native and dsp into

>>>> a truly expandable environment - more flexibility and upgradeability  
>>>> with that approach - better longterm investment, imho.  
>>>>  
>>>> Regards,  
>>>> Detric  
>>>>  
>>>> "LaMont" <jjdpro@ameritech.net> wrote in message  
>>>> news:45704a29\$1@linux...  
>>>>> Detric, I think you mis interpreted my post. Ilove what Chris and the  
>>>>> guys  
>>>>> do. My post was about just reading about various trials and  
>>>>> tribulations  
>>>>> and the cost factor of being on DAW upgrade path, which I was very much  
>>>>> apart  
>>>>> of the trend :) I was just recently(Jan 06)that I've taken this  
>>>>> pro-DSP based  
>>>>> (for me) position.  
>>>>>  
>>>>> As a person who has (1 Paris PC, 2-NativeNeundo PCs, 1 Giga PC, 1 VSTi  
>>>>> PC)I  
>>>>> pretty much quality as a PC -DAW gear slut. But, I recognized that  
>>>>> after  
>>>>> my last mega (Dual-Dual-core Opteron setup), that I was spending a lot  
>>>>> of  
>>>>> money for a solution that frankly, i could've purchased a HD1 or maybe  
>>>>> an  
>>>>> HD2..But, that's just me.. I prefer DSP based systems than Native.  
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>>>>> I too build DAWs-in my neck of the woods(Michgan) for studios,  
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>>>>> Detric Terry <dterry@keyofd.net> wrote:  
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>>>>>> know  
>>>>>> they don't sell ProTools or Soundscape, but Chris can speak for  
>>>>>> himself.  
>>>>>> I  
>>>>>> for one appreciate his participation, contribution and advice in PC  
>>>>>> discussions here and don't see how it is contradictory to studio  
>>>>>> consulting  
>>>>>> advice.  
>>>>>>  
>>>>>> If you prefer dsp based, that's great. I can't see how upgrading a  
>>>>>> PC or  
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>>>>>> etc. Then consider that you still have a PC or Mac running those  
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>>>>> that will have to be upgraded at some point, in some form (e.g. a new  
>>>>> version of PT software comes out, a new conversion tool update, etc).  
>>>>>  
>>>>> And if you want to run Giga, NI softsynths, etc...same thing.  
>>>>>  
>>>>> And fwiw, we were advising Mike R. on what options are out there now,  
>>>>> not  
>>>>> talking about updating our PCs once a month.  
>>>>>  
>>>>> Best of luck!  
>>>>>  
>>>>> Dedic  
>>>>>  
>>>>>  
>>>>> On 12/1/06 8:49 AM, in article 457040f5\$1 @linux, "LaMont"  
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>>>>>> Chris and others,  
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>>>>>> This is the reason I prefer DSP based DAWS. We spend way too much  
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>>>>>> PC/Mac upgrades, chip-sets, what won't play with what..etc..  
>>>>>>  
>>>>>> You're constantly on the DAW upgrade path which to me, is getting  
>>>>>> very  
>>>>> old.  
>>>>>> This is strange coming from a DaW builder and studio DAW consultant.  
>>>>>> But,  
>>>>>> I've had enough. After my last Opteron upgrade, I said "That's it"..  
>>>>>> :)  
>>>>>>  
>>>>>>  
>>>>>>  
>>>>>>  
>>>>>> Chris Ludwig <chrisl@adkproaudio.com> wrote:  
>>>>>>> Hi Dedic,  
>>>>>>> The Core2 based system seems to be the lowest latency setup from  
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>>>>>>> I haven't been able to get any demo project or project that I've  
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>>>>>>>  
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>>>>>>> Here is the bench mark we use currently. It's based on the same  
>>>>>>> methodology of the infamous Blofelds test but with out sound card  
>>>>>>> performance issues. I haven't tried this on many machine other than  
>>>>>>> the  
>>>>>>> Opterons, Xeons and quads so not even sure what results there will  
>>>>>>> be  
>>>>>>> on  
>>>>>>> pre-Core 2 Duo system or a AMD X2/AM2 system. I just finished it  
>>>>>>>  
>>>>>>> Haven't had a chance to right all the instructions out yet.  
>>>>>>> Basically  
>>>>>>> load the project at 1024 buffer load as many magneto plug ins as  
>>>>>>> you  
>>>>>>> you  
>>>>>>> can then save the project under a new name. Adding 1024 to the name  
>>>>>>> will  
>>>>>>> help keep track of the versions. Reload the project to confirm that  
>>>>>>> you  
>>>>>>> can run that number of plug ins. make note of the number.  
>>>>>>> Close the program then reopen the default project and repeat the  
>>>>>>> test  
>>>>>>> for each latency setting.  
>>>>>>>  
>>>>>>> If you have any problem or suggestion for the test let me know so I  
>>>>>>> can  
>>>>>>> fine tune it if needed.  
>>>>>>>  
>>>>>>> [http://www.adkproaudio.com/downloads/set\\_me\\_free\\_benchmark.z ip](http://www.adkproaudio.com/downloads/set_me_free_benchmark.z ip)  
>>>>>>>  
>>>>>>>  
>>>>>>> Thanks Chris  
>>>>>>>  
>>>>>>>  
>>>>>>>  
>>>>>>>  
>>>>>>> Dedric Terry wrote:

>>>>>>> I can't speak for an Athlon 64/4000, but on my X2 4400 I run a  
>>>>>>> similar  
>>>>>>> setup at low latency all the time for commercial spots, sound  
>>>>>>> design,  
>>>>>>> etc. I also have a full orchestral template that will run easily  
>>>>>>> on  
>>>>>>> this system.  
>>>>>>>  
>>>>>>> I've also run some projects of 50-100 tracks with plugins without  
>>>>>>> problems as well.  
>>>>>>>  
>>>>>>> If you are building, consider an Intel core duo, dual core duo  
>>>>>>> (xeon),  
>>>>>>> or if you have the budget, and quad core system. The Intel core  
>>>>>>> duos  
>>>>>>> are slightly better performers than AMD chips at the moment - at  
>>>>>>> least  
>>>>>>> from specs I'm seeing from DAW builders. A quad core is the  
>>>>>>> fastest,  
>>>>>>> but there seem to be diminishing returns on dual quad cores.  
>>>>>>>  
>>>>>>> Regards,  
>>>>>>> Dedic  
>>>>>>>  
>>>>>>> On 11/29/06 9:52 AM, in article 456d8e00@linux, "Mike R."  
>>>>>>> <emarenot@yahoo.com> wrote:  
>>>>>>>  
>>>>>>> While I'm definately going to continue to use Paris for  
>>>>>>> tracking  
>>>>>>> and  
>>>>>>> mixing, I've been thinking more and more about getting some  
>>>>>>> sort  
>>>>>>> of  
>>>>>>> in the box solution for writing. I'm thinking about building  
>>>>>>> a  
>>>>>>> computer around an Athlon 64 4000 and some sort of mobo with  
>>>>>>> an  
>>>>>>> FSB  
>>>>>>> around 1000mhz (I can't believe that sort of speed on the  
>>>>>>> FSB -blows  
>>>>>>> me away everytime I think about it..) I'd like to be able to  
>>>>>>> run  
>>>>>>> about five to six synths, three verbs on fx sends, and of  
>>>>>>> course  
>>>>>>> several comp plugs and eq's etc on the various channels. I'm  
>>>>>>> wondering if anyone has a similar set up and what sort of  
>>>>>>> track  
>>>>>>> counts folks are getting.

>>>>>>>>  
>>>>>>>>  
>>>>>>>> --  
>>>>>>>> Chris Ludwig  
>>>>>>>>  
>>>>>>>> ADK Pro Audio  
>>>>>>>> (859) 635-5762  
>>>>>>>> www.adkproaudio.com  
>>>>>>>> chrisl@adkproaudio.com  
>>

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Subject: Re: Track Counts in Native Systems?  
Posted by [Dedric Terry](#) on Sat, 02 Dec 2006 17:00:05 GMT  
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On 12/2/06 9:14 AM, in article 4571a3d1@linux, "Jamie K"  
<Meta@Dimensional.com> wrote:

> Dedric Terry wrote:  
>> Hey Jamie,  
>>  
>> You and the other Mac users know we PC users just give you a hard time  
>> because your comps have fruit on them. ;-)  
>  
> Jealousy is such an ugly thing. I'll send you a banana sticker. ;^)  
>

Excellent! Although a vegetable would be great... I think Broccoli would  
make a great computer logo. ;-)

> Over the years Mac critics, me among them, have argued that Macs are  
> more expensive. Too expensive. And I had other, technical, criticisms of  
> previous generations of Macs.  
>  
> However, more recently the value proposition has improved, and OSX is a  
> huge step forward for Apple.  
>  
> Although the old perceptions have inertia, the reality, from my current  
> experience, is much better.

You are exactly right here - price vs. performance was once a significant  
difference, but the new Intel Macs are right in line with comparable off the  
shelf PCs. Of course one can still save \$200-\$400 building a PC vs. the  
same config in a Mac, but not buying off-the-shelf. I tend to defend PCs to  
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> instrument plugins. That makes things much simpler. The G5 can also  
> address enough HD space that I don't need another server or, for that  
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> 8GB) to do a reasonable amount of multitasking.

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For me the multiple systems are needed for orchestral libraries, and PC or  
Mac, that's pretty much standard for composing (there are guys in LA I've  
talked with using even larger rigs).

With hundreds of articulations required for a score, there isn't a single  
box that I could use to run a full complement. And what I could load into a  
current PC/Mac will usually only account for about 30 seconds of scoring in  
one style/tempo.

I am guessing that with 64G of Ram in a quad quad core or dual octo core we  
might get closer to moving large scale orchestration to only one or two  
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as expected yet. But even then we'll probably just be using 24/96k sample  
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different than running outboard samplers/synths. Not quite as convenient as  
running everything in one system, but also not as bad as it sounds (actually  
quite liberating to know that the percussion section will always be there -  
no dropped timpani rolls during French horn crescendos :-)

Then if I add in other cpu/ram/disk streaming intensive VSTi's, full screen  
DV resolution video playback, the faster/more the better - I already have an  
extra PC is dedicated to full screen video (not unlike using a VCube or deck  
really).

Regards,  
Dedric

> It's really convenient to do it all on one fast, capable box in the  
> studio. Less bailing wire, duct tape and magic incantations needed to  
> hold it all together. :^)

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> Almost no administration time needed.

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>> BTW - speaking of intense animation, probably old news, but there's an HD  
>> video online from Animusic - I like some of their other work better, but fun  
>> all the same:

>> <http://www.apple.com/quicktime/guide/hd/animusic2dvd.html>

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> Clever stuff. I saw another one of those a while back that also  
> impressed me.

>  
> Cheers,

> -Jamie  
> www.JamieKruz.com  
>  
>

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Subject: Re: Track Counts in Native Systems?  
Posted by [Dedric Terry](#) on Sat, 02 Dec 2006 17:45:40 GMT  
[View Forum Message](#) <> [Reply to Message](#)

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On 12/2/06 9:31 AM, in article 45719c34\$1@linux, "LaMont"  
<jjdpro@ameritech.net> wrote:

>  
> They recently interviewd the UAD guys and the big question was put to them:  
> 'Why not add more power to the card"?  
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> UAD: answer was : Because they could not find a "cost-effective "Dsp that  
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UA is a very small company. This is the same necessity of survival that has locked Digidesign into using Motorola dsps that are far from cutting edge.

It's very expensive to change course completely with dsp or cpu based development, or start from scratch without \$5,000,000 in development budget and base a product around a higher end dsp, \*and\* expect it to pay for itself.

Joe Bryan addressed this in an interview with Sound on Sound a year or two ago. They didn't have a ton of capital at startup, so the card had to be affordable to sell enough to pay for itself, and feasible for a limited development team. Fairlight has taken a more expensive development route with the CC-1. That's probably part of the reason they are only bundling it with Dream systems, and selling different track counts/processing capability levels as licenses at different price points - to ensure they make up the heavy development cost.

I'm not trying to defend UA other than understanding their business model. But imho, too many users seem to blame companies for trying to run a profitable business when the same users are making equally biased decisions for exactly the same reason: wanting cheaper more capable products in order to spend less and make more; or buy marketing hype because xyz producer made a hit record with that gear.

I do agree the vintage gear craze is absurd beyond belief. To me there is some great character and quality in some vintage gear, but not all has the quality available today. There is way too much "follow the leader" bad



engineering going on, and companies are all to happy to appeal to whatever sells.

Dedric

---

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Subject: Re: Track Counts in Native Systems?

Posted by [DJ](#) on Sat, 02 Dec 2006 18:26:40 GMT

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---

Hi Neil,

The Cubase editor works for me because I usually end up arranging the songs during the mix. We have a bunch of drop-dead, unbelievable musicians here. One of my session players, Gary Cook has been voted national flatpicking champion and mandolin god of the universe at quite a few festival venues. There are others around here who are at that level, fiddle players, dobroists, pianists, etc. We're talking Nashville session player level. Gary is buddies with Tony Rice and a bunch of that crew in Tennessee and has hangs with/records with Charlie Daniels and his band quite a bit (no slouches in that band). Our pianist was Toby Keith's pianist for many years and has a lot of Nashville session experience. these guys can just walk in the door and create magic. What my partner and I do a lot of here is get a guitar/vocal track happening to some kind of simple rythm/click-I usually use Groove Agent. Then we bring in a bass player and drummer and nail down the rythm section. After that we bring in the rest of the players. We usually have arrangements for them and they are rehearsed so there is a general direction this should go and these guys are good so we can usually get the basics down PDQ.....but then the fun starts because these guys are mostly disgustingly egotistical virtuosos so we just turn them loose and let them go nuts all over the basic arrangement until they wear themselves out. Once all of this mayhem is finished, the comping starts. 90% of the noodling around gets shitcanned but the other 10% is golden and comes directly from the freedom we give these guys to have fun. My job is usually to integrate these snippets into the song as hooks or whatever and arrange everything so that it integrates well. I'll grab a cool lick on a guitar, a cool lick on a fiddle or mando or whatever and make a hook out of it, find a part of another instrument that counterpoints it and go to town. The Cubase editor works extremely well for this and if I need to pitch a note up or down a cent or two to achieve a nice harmonic between two instruments in a solo or whatever, then it's done quickly. Also, the fadesins/outs between tracks are so easily managed as well as basic gain changes, rendering of FX (if desired), etc. It has just spoiled me rotten. I just find it much easier/more intuitive than the Paris editor.

;o)

"Neil" <IOUOIU@OIU.com> wrote in message news:4571bbb9\$1@linux...

>  
> "DJ" <nowayjose@dude.net> wrote:  
>>but it's  
>>pretty hard to beat Cubase for VST integration and editing  
>  
> Deej, I've seen you rave about the Cubase editing features a  
> lot... what kind of editing do you find yourself doing? The  
> reason I ask is that any editing I have to do is pretty much  
> simple stuff like crossfading punches, moving the xfades around  
> so they're in the least noticeable place, the occasional snip  
> to get rid of an extraneous string noise or vocal artifact  
> ("hack!, mmmmrgh, AHEM!") lol; and that's about it, really...  
> am I missing the boat on something really cool here with regard  
> to more enhanced editing capabilities?  
>  
> Neil

---

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Subject: Re: Track Counts in Native Systems?  
Posted by [DJ](#) on Sat, 02 Dec 2006 18:29:36 GMT  
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---

I tend to defend PCs to  
some degree, but the new Macs are very nice.

Yep!!! I agree..especially because they can run Windows

;oD

"Dedric Terry" <dterry@keyofd.net> wrote in message  
news:C196FF25.5D82%dterry@keyofd.net...  
> On 12/2/06 9:14 AM, in article 4571a3d1@linux, "Jamie K"  
> <Meta@Dimensional.com> wrote:  
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>  
> Regards,  
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>  
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>> Cheers,  
>> -Jamie  
>> [www.JamieKruz.com](http://www.JamieKruz.com)  
>>  
>>  
>

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Subject: Re: Track Counts in Native Systems?  
Posted by [neil\[1\]](#) on Sat, 02 Dec 2006 18:45:29 GMT  
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"DJ" <[nowayjose@dude.net](mailto:nowayjose@dude.net)> wrote:  
>but it's  
>pretty hard to beat Cubase for VST integration and editing

Deej, I've seen you rave about the Cubase editing features a lot... what kind of editing do you find yourself doing? The reason I ask is that any editing I have to do is pretty much simple stuff like crossfading punches, moving the xfades around so they're in the least noticeable place, the occasional snip

to get rid of an extraneous string noise or vocal artifact ("hack!, mmmmrgh, AHEM!") lol; and that's about it, really... am I missing the boat on something really cool here with regard to more enhanced editing capabilities?

Neil

---

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Subject: Re: Track Counts in Native Systems?  
Posted by [DJ](#) on Sat, 02 Dec 2006 18:46:01 GMT  
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---

I talked to Tom Freeman at UA week before last. They've got something new up their sleeve but nobody's taking yet.

;o)

"Dedric Terry" <dterry@keyofd.net> wrote in message  
news:C19709D4.5D89%dterry@keyofd.net...

> On 12/2/06 9:31 AM, in article 45719c34\$1@linux, "LaMont"  
> <jjdpro@ameritech.net> wrote:

>

>>

>> They recently interviewd the UAD guys and the big question was put to  
>> them:

>> "Why not add more power to the card"?

>>

>> UAD: answer was : Because they could not find a "cost-effective "DSp that  
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>

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> I'm not trying to defend UA other than understanding their business model.  
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> Detric  
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Subject: Re: Track Counts in Native Systems?  
Posted by [Jamie K](#) on Sat, 02 Dec 2006 19:03:18 GMT  
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Detric Terry wrote:

> On 12/2/06 9:14 AM, in article 4571a3d1@linux, "Jamie K"  
> <Meta@Dimensional.com> wrote:  
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>> Detric Terry wrote:  
> You are exactly right here - price vs. performance was once a significant  
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> shelf PCs. Of course one can still save \$200-\$400 building a PC vs. the  
> same config in a Mac, but not buying off-the-shelf. I tend to defend PCs to  
> some degree, but the new Macs are very nice.

It's natural to defend a purchase. But no matter what I buy, the bottom line is we need multiple choices in the marketplace to keep the competitive pressure on. It benefits everyone who buys any particular type of computer to have at least some tiny semblance of a competitive market. It would be great to have more choices than we do.

So I actually don't want everyone to buy what I use. A market with only one solution available is not a market.

>> My G5 is fast enough that I don't need other computers slaved for FX or  
>> instrument plugins. That makes things much simpler. The G5 can also  
>> address enough HD space that I don't need another server or, for that  
>> matter, a network. It has enough RAM (currently 2.5GB, can address up to  
>> 8GB) to do a reasonable amount of multitasking.

>>

> For me the multiple systems are needed for orchestral libraries, and PC or  
> Mac, that's pretty much standard for composing (there are guys in LA I've  
> talked with using even larger rigs).

I can report unequivocally that here, with a pretty good test case,  
there is simply no need for a second machine to run instrument plugins.

I use an orchestral library, GPO, and another, the VSL subset included  
with Kontakt 2. G5 handles them no sweat. My piano main library is 35GB  
of samples. No problem on the G5. I have just about every form of  
synthesis in plugin format for sound design. I have a decent selection  
of sample libraries. The G5 handles it all. I can layer Audio Unit  
instruments until the cows come home. And I don't even have cows! :^)

If I ever exceed the CPU capabilities, I can simply freeze tracks. But I  
haven't need to do that on this machine, at least so far.

Logic does offer multi-machine support of its native plugins, so I could  
add another machine if I needed. But I haven't been pinched for  
processing power on the G5. 5GHZ on a pair of RISC chips is a lot of CPU  
cycles.

Consider: The G5 is probably 5X the speed of my last machine, which was  
maybe 8x the speed of the one before that. This box can address much  
more RAM. It has much faster HD access and multiple ways to connect more  
of them, over a TB here. Realistically, I already have the speed of  
multiple machines in one box. OSX supports multi-processors,  
multithreading, preemptive multitasking, dynamic RAM allocation, virtual  
memory, all in a user-transparent way.

Lessee, right now I have fourteen main apps up including Logic, which  
itself is running half a dozen Audio Units (B4II, Kontakt 2, Ivory,  
EVP88, Pro53, EVD6 at the moment). A light morning; I still have plenty  
of memory and processor cycles to spare.

The newer boxes are even faster than my several year old wonder (and 8  
processor cores are the next big thing), so I would say the days of  
needing to hassle with a second box are over, at least on the Mac with  
Logic and 3rd party Audio Units.

I can't speak for older systems or different designs, but if a multi-box  
system is working OK for someone, they may as well keep using it, why not?

> With hundreds of articulations required for a score, there isn't a single  
> box that I could use to run a full complement. And what I could load into a  
> current PC/Mac will usually only account for about 30 seconds of scoring in  
> one style/tempo.

Wow, really? I need to hear one of your scores. Clearly we're having different experiences. What libraries do you use?

> I am guessing that with 64G of Ram in a quad quad core or dual octo core we  
> might get closer to moving large scale orchestration to only one or two  
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> running everything in one system, but also not as bad as it sounds (actually  
> quite liberating to know that the percussion section will always be there -  
> no dropped timpani rolls during French horn crescendos :-)

Heh. My timpani and horns are very good about not dropping out, (that is, ever since that one Logic bug was finally fixed, grrrr!) and they all live happily on one box.

> Then if I add in other cpu/ram/disk streaming intensive VSTi's, full screen  
> DV resolution video playback, the faster/more the better - I already have an  
> extra PC is dedicated to full screen video (not unlike using a VCube or deck  
> really).

I used to sync via SMPTE to a BetaCamSP deck. Now I run the video in Logic, too, as Quicktime. Very slick, SMPTE offset, automatic scene detection, import/export audio from the video. I've also used Soundtrack Pro in a similar way, but much prefer to compose in Logic. Either way, it's all on one box.

Cheers,  
-Jamie  
[www.JamieKrutz.com](http://www.JamieKrutz.com)

> Regards,  
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Subject: Re: Track Counts in Native Systems?  
Posted by [Jamie K](#) on Sat, 02 Dec 2006 19:09:54 GMT  
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---

That sounds like fun! It's nice to hear about the magic making side of your experience. The highest end gear in your studio walks in on legs.

Cheers,  
-Jamie  
[www.JamieKrutz.com](http://www.JamieKrutz.com)

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> ;o)

>

>

> "Neil" <IOUOIU@OIU.com> wrote in message news:4571bbb9\$1@linux...

>> "DJ" <nowayjose@dude.net> wrote:

>>> but it's

>>> pretty hard to beat Cubase for VST integration and editing

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>> lot... what kind of editing do you find yourself doing? The

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>> so they're in the least noticeable place, the occasional snip

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>> ("hack!, mmmmrgh, AHEM!") lol; and that's about it, really...

>> am I missing the boat on something really cool here with regard

>> to more enhanced editing capabilities?

>>

>> Neil

>

>

---

Subject: Re: Track Counts in Native Systems?

Posted by [DJ](#) on Sat, 02 Dec 2006 19:25:52 GMT

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---

It's funny Jamie. I've always been able to pick up an inanimate object and figure out how to get something melodic out of it, I learned to play my grandmother's Hammond organ and piano by ear but I've never learned to read

music. In my younger days I was a fairly proficient guitarist but I really haven't played an instrument in years and I'm so rusty/incompetent now that I sorta' cringe when I pick one up. I would love to have the luxury of having time to practice and of going back to school and study music and arranging. I've always had a bit of an ear for it though I seldom have any idea at all what I'm after at the beginning. I just sorta jump into it like a kid doing a cannonball into a swimming pool and flail around until something I like spills over the rim.

;o)

"Jamie K" <Meta@Dimensional.com> wrote in message news:4571ccf4@linux...

>

> That sounds like fun! It's nice to hear about the magic making side of  
> your experience. The highest end gear in your studio walks in on legs.

>

> Cheers,

> -Jamie

> www.JamieKruz.com

>

>

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---

Subject: Re: Track Counts in Native Systems?  
Posted by [LaMont](#) on Sat, 02 Dec 2006 19:40:36 GMT  
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---

Good Post Decric ..

There was this debate (last year) on gearslutz about the UAD card issue and it was an overwhelming fact that most of the users would gladly pay "more" money for a "faster" DSP card..

And to be honest, I think I would galdly pay more for the UAD if it had more horsepower than he current model(s).

Soemthing not quite right in UAD land with concernng the UAD powered plugins business model. Factor in this :

-Their plugins have never been hacked, nor iare being sold on the black market.

So, unlike other Plugin manufacturers, their seeing a real profit and not loses on plugins.  
So what gives..??

c Terry <dterry@keyofd.net> wrote:

>On 12/2/06 9:31 AM, in article 45719c34\$1@linux, "LaMont"

><jjdpro@ameritech.net> wrote:

>

>>

>> They recently interviewd the UAD guys and the big question was put to them:

>> "Why not add more power to the card"?

>>

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>

---

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Subject: Re: Track Counts in Native Systems?  
Posted by [Jamie K](#) on Sat, 02 Dec 2006 19:44:29 GMT  
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I like that image - the reflexes of swimming are similar to the reflexes of making music in a group, diving in, taking risks, splashing around and seeing what everyone does with it, having fun.

There are some great books and web sites on arranging if you do get some time to spare. Maybe now that your system is coalescing you can spend less time on that and free up some time for books and arranging exploration.

I just got together with another drummer who is giving me tips on kit techniques to help me to my next level on drums in exchange for music theory lessons to help him to his next level on both drums and keyboards. The first lesson went great and I'll be practicing ghost notes on snare this week...

Cheers,  
-Jamie  
[www.JamieKruz.com](http://www.JamieKruz.com)

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>

---

Subject: Re: Track Counts in Native Systems?  
Posted by [LaMont](#) on Sat, 02 Dec 2006 19:45:19 GMT  
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Neil if I may  
I use Neundo, which Cuabse is based off of:  
I too have ben raving about neundo/SX editing since it's inception. Here's  
why:

Highlight a given track, right-mouse click..Boom.. It's all there for the  
chosing.  
-Process a track -normalize, volume/down, add convolution, cut,copy,posy,  
Silence..

-Right-mouse click..Process with either Dirrectx or vst plugins..

I don;t knwo about you, but the right mouse click, is what set's the Steinbergs  
apart from all other DAWs.. That's speed and power.

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>Neil

---

---

Subject: Re: Track Counts in Native Systems?  
Posted by [DJ](#) on Sat, 02 Dec 2006 20:04:58 GMT  
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---

They are making a profit and having a life with it?

;o)

"LaMont" <jjdpro@ameritech.net> wrote in message news:4571c8a4\$1@linux...

>  
> Good Post Decric ..  
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> There was this debate (last year) on gearslutz about the UAD card issue  
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>>  
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>>  
>

---

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Subject: Re: Track Counts in Native Systems?

Posted by [Dedric Terry](#) on Sat, 02 Dec 2006 20:51:38 GMT

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"Jamie K" <Meta@Dimensional.com> wrote in message news:4571cb6a@linux...

>> With hundreds of articulations required for a score, there isn't a single  
>> box that I could use to run a full complement. And what I could load into  
>> a

>> current PC/Mac will usually only account for about 30 seconds of scoring  
>> in

>> one style/tempo.

>

> Wow, really? I need to hear one of your scores. Clearly we're having  
> different experiences. What libraries do you use?

I'm using East West's Quantum Leap Symphonic XP library at the moment: 38G  
total. For some things

I then add Symphonic Choirs at 35G (just one section (e.g. sopranos) uses  
most of my free Ram).

Then factor in Stormdrum, Kontakt 2, Absynth, and numerous other VSTi's for  
more varied, or modern/cinematic uses  
and it adds up really fast. EW actually recommends up to 8 PCs for their  
platinum library (24-bit).

The composers I've chatted with a few times in LA use 300 track templates  
for composing, multiple

PCs and a large number of outboard samplers/synths - usually 50-100 inputs.

Although I'm not anywhere

near that scale of outboard gear, I can see, and feel the need to greatly  
expand my rig in my work more and more now.

I'm getting ready to put together a rather involved and dynamic piece for a  
new demo - I'll email you a link when it's done.

> Heh. My timpani and horns are very good about not dropping out, (that is,  
> ever since that one Logic bug was finally fixed, grrrr!) and they all live  
> happily on one box.

>

That really depends on the percussionist though. Some guys are more  
reliable than others, esp. if they are former  
rock drummers. ;-) Hey, the trumpet player fell asleep during my  
wedding....didn't even take a timpani roll to  
get him to drop out. He did come back for the recession though. :-)

With a larger templates I use up most of my Ram, and since one of the  
crescendo Fr.Horn instruments I use is actually a mod-wheel dynamic  
crossfade of three sample sets,  
as are the timpani modwheel crossfade instruments, they tend to suck down  
any remaining ram and cpu power very fast  
when a lot of other instruments are loaded and running.

> I used to sync via SMPTE to a BetaCamSP deck. Now I run the video in  
> Logic, too, as Quicktime. Very slick, SMPTE offset, automatic scene  
> detection, import/export audio from the video. I've also used Soundtrack  
> Pro in a similar way, but much prefer to compose in Logic. Either way,  
> it's all on one box.

Apple has video down cold - H.264 is a superb codec, but sadly doesn't work  
with Nuendo(PC- too bad).

AVI works fine but I've never been happy with compression and size vs.  
quality tradeoffs.

Inline video in Nuendo actually works very well - Quicktime MJpeg works well  
here though better when encoded with Vegas

than Quicktime Pro oddly. (I usually run a compressed 320x240 window in  
Nuendo locally just for sync and spotting).

I also have a PC sync'd via System Link running full screen DV (720x480)  
MJpeg (Quicktime playback in Cubase 4) -

looks quite good for client previews (not as clear as HD or external  
monitored DV, but it works well). I'm planning to add a system  
running the new Decklink HDMI PCIe card for full res HD playback to an HD  
LCD TV fairly soon - for eye candy mainly.

It's pretty stunning what we can accomplish with a single computer now  
though. More power and flexibility is necessary for some  
things, but there is a ton of great music that can be produced with even  
just a laptop.

Regards,  
Dedric

>  
> Cheers,  
> -Jamie  
> [www.JamieKrutz.com](http://www.JamieKrutz.com)

>  
>  
>  
>> Regards,  
>> Dedric  
>>

>>> It's really convenient to do it all on one fast, capable box in the  
>>> studio. Less bailing wire, duct tape and magic incantations needed to  
>>> hold it all together. :^)

>>>  
>>> Almost no administration time needed.

>>>  
>>>

>>>> BTW - speaking of intense animation, probably old news, but there's an

>>>> HD  
>>>> video online from Animusic - I like some of their other work better,  
>>>> but fun  
>>>> all the same:  
>>>> <http://www.apple.com/quicktime/guide/hd/animusic2dvd.html>  
>>> Clever stuff. I saw another one of those a while back that also  
>>> impressed me.  
>>>  
>>> Cheers,  
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>>> [www.JamieKrutz.com](http://www.JamieKrutz.com)  
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---

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Subject: Re: Track Counts in Native Systems?  
Posted by [Dedric Terry](#) on Sat, 02 Dec 2006 21:06:02 GMT  
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I don't think Powercore has been hacked either, but it's a stock Motorola 56k chipset.

It's hard to hack low level code on a proprietary processing card (i.e though it's just a graphics card the processing code isn't run of the mill OS code even if it's C++ or whatever).

It's way easier to hack an OS that has SDKs galore floating around and so more holes in it than applications.

I really think their business model was "we ain't got a mint to fund this puppy so what works for \$500k max, marketing and payroll included?"

As DJ said, there may be something significant on the horizon from them anyway. Having built hardware in a slightly larger tech company, I can attest to the approach they are taking as making complete sense for their size and market. We ran had 50 employees and only \$1.2 mil in annual revenue (over 10 years ago) to fund everything we did - there were no shoot for the moon high end projects or redesigns just because customers wanted bigger, better and faster. We had to make what we had work, sell and pay for itself.

UA has made a huge impact on that market without even being the most well-heeled of the dsp developers. Imho, kudos to them and I hope the next product is even more successful than the UAD-1 has been.

Wouldn't it be a hoot if they signed a deal with Fairlight to use the CC-1 platform, or Clearspeed cards? Being the supposed equivalent of 8 HD Accel cards, the CC-1 could be about 16-20 times the power of a UAD-1 (rough guess - I ran the numbers last week, but don't remember exactly). I have no idea how the Clearspeed would translate. Both are probably left field musings though.

Regards,  
Dedric

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>>ago. They didn't have a ton of capital at startup, so the card had to be  
>>affordable to sell enough to pay for itself, and feasible for a limited  
>>development team. Fairlight has taken a more expensive development route  
>>with the CC-1. That's probably part of the reason they are only bundling  
> it  
>>with Dream systems, and selling different track counts/processing  
>>capability  
>>levels as licenses at different price points - to ensure they make up the  
>>heavy development cost.  
>>  
>>I'm not trying to defend UA other than understanding their business model.  
>>But imho, too many users seem to blame companies for trying to run a  
>>profitable business when the same users are making equally biased  
>>decisions  
>>for exactly the same reason: wanting cheaper more capable products in  
>>order  
>>to spend less and make more; or buy marketing hype because xyz producer  
> made  
>>a hit record with that gear.  
>>  
>>I do agree the vintage gear craze is absurd beyond belief. To me there  
> is  
>>some great character and quality in some vintage gear, but not all has the  
>>quality available today. There is way too much "follow the leader" bad  
>>engineering going on, and companies are all too happy to appeal to whatever  
>>sells.  
>>  
>>Dedric  
>>  
>

---

Subject: Re: Track Counts in Native Systems?  
Posted by [LaMont](#) on Sat, 02 Dec 2006 21:26:08 GMT  
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Yes of course Dj, but why have are not giving the user's (customers) what

they want.

The user(customers) don;t want more plugings from UAD, but just a faster DSP card to run more of whthe plugings they have..

But, UAD just releases another plugin to seemingly apease the customer base. However, it's not working..

"DJ" <nowayjose@dude.net> wrote:

>They are making a profit and having a life with it?

>

>;o)

>

>"LaMont" <jjdpro@ameritech.net> wrote in message news:4571c8a4\$1@linux...

>>

>> Good Post Decric ..

>>

>> There was this debate (last year) on gearslutz about the UAD card issue

>> and

>> it was an overwhelming fact that most of the users would gladly pay "more" money for a "faster" DSP card..

>>

>> And to be honest, I think I would galdly pay more for the UAD if it had

>> more

>> horsepower than he current model(s).

>>

>> Soemthing not quite right in UAD land with concernng the UAD powered plugins

>> business model. Factor in this :

>>

>> -Their plugins have never been hacked, nor iare being sold on the black

>> market.

>>

>> So, unlike other Plugin manufactuers, their seeing a real profit and not loses on plugins.

>> So what gives..??

>>

>>

>> c Terry <dterry@keyofd.net> wrote:

>>>On 12/2/06 9:31 AM, in article 45719c34\$1@linux, "LaMont"

>>><jjdpro@ameritech.net> wrote:

>>>

>>>>

>>>> They recently interviewd the UAD guys and the big question was put to them:



>>>> 'Why not add more power to the card'?

>>>>

>>>> UAD: answer was : Because they could not find a "cost-effective "DSp

>>>> that

>>>> would not raise the price of the card???

>>>

>>>UA is a very small company. This is the same necessity of survival that

>>>has locked Digidesign into using Motorola dsps that are far from cutting

>>>edge.

>>>

>>>It's very expensive to change course completely with dsp or cpu based

>>>development, or start from scratch without \$5,000,000 in development

>>>budget

>>>and base a product around a higher end dsp, \*and\* expect it to pay for

>>>itself.

>>>

>>>Joe Bryan addressed this in an interview with Sound on Sound a year or

two

>>>ago. They didn't have a ton of capital at startup, so the card had to

be

>>>affordable to sell enough to pay for itself, and feasible for a limited

>>>development team. Fairlight has taken a more expensive development route

>>>with the CC-1. That's probably part of the reason they are only bundling

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>>>for exactly the same reason: wanting cheaper more capable products in

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>>>to spend less and make more; or buy marketing hype because xyz producer

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>>>  
>>>Dedric  
>>>  
>>  
>  
>

---

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Subject: Re: Track Counts in Native Systems?  
Posted by [Martin Harrington](#) on Sat, 02 Dec 2006 23:18:29 GMT  
[View Forum Message](#) <> [Reply to Message](#)

---

Hey Dedric,  
How do you use a separate PC for full screen video, and how are they sync'd,  
(sync)?  
Interesting

--  
Martin Harrington  
[www.lendaneer-sound.com](http://www.lendaneer-sound.com)

"Dedric Terry" <[dterry@keyofd.net](mailto:dterry@keyofd.net)> wrote in message  
news:C196FF25.5D82%dterry@keyofd.net...  
> On 12/2/06 9:14 AM, in article 4571a3d1@linux, "Jamie K"  
> <[Meta@Dimensional.com](mailto:Meta@Dimensional.com)> wrote:  
>  
>> Dedric Terry wrote:  
>>> Hey Jamie,  
>>>  
>>> You and the other Mac users know we PC users just give you a hard time  
>>> because your comps have fruit on them. ;-)  
>>  
>> Jealousy is such an ugly thing. I'll send you a banana sticker. ;^)  
>>  
>  
> Excellent! Although a vegetable would be great... I think Broccoli would  
> make a great computer logo. ;-)  
>  
>> Over the years Mac critics, me among them, have argued that Macs are  
>> more expensive. Too expensive. And I had other, technical, criticisms of  
>> previous generations of Macs.  
>>  
>> However, more recently the value proposition has improved, and OSX is a  
>> huge step forward for Apple.  
>>  
>> Although the old perceptions have inertia, the reality, from my current  
>> experience, is much better.  
>

> You are exactly right here - price vs. performance was once a significant  
> difference, but the new Intel Macs are right in line with comparable off  
> the  
> shelf PCs. Of course one can still save \$200-\$400 building a PC vs. the  
> same config in a Mac, but not buying off-the-shelf. I tend to defend PCs  
> to  
> some degree, but the new Macs are very nice.  
>>  
>> My G5 is fast enough that I don't need other computers slaved for FX or  
>> instrument plugins. That makes things much simpler. The G5 can also  
>> address enough HD space that I don't need another server or, for that  
>> matter, a network. It has enough RAM (currently 2.5GB, can address up to  
>> 8GB) to do a reasonable amount of multitasking.  
>>  
> For me the multiple systems are needed for orchestral libraries, and PC or  
> Mac, that's pretty much standard for composing (there are guys in LA I've  
> talked with using even larger rigs).  
>  
> With hundreds of articulations required for a score, there isn't a single  
> box that I could use to run a full complement. And what I could load into  
> a  
> current PC/Mac will usually only account for about 30 seconds of scoring  
> in  
> one style/tempo.  
>  
> I am guessing that with 64G of Ram in a quad quad core or dual octo core  
> we  
> might get closer to moving large scale orchestration to only one or two  
> systems, but the low latency capabilities in the dual quads aren't scaling  
> as expected yet. But even then we'll probably just be using 24/96k sample  
> libraries to continue the networking requirements. It really isn't any  
> different than running outboard samplers/synths. Not quite as convenient  
> as  
> running everything in one system, but also not as bad as it sounds  
> (actually  
> quite liberating to know that the percussion section will always be  
> there -  
> no dropped timpani rolls during French horn crescendos :-)  
>  
> Then if I add in other cpu/ram/disk streaming intensive VSTi's, full  
> screen  
> DV resolution video playback, the faster/more the better - I already have  
> an  
> extra PC is dedicated to full screen video (not unlike using a VCube or  
> deck  
> really).  
>  
> Regards,

> Dedic  
>  
>> It's really convenient to do it all on one fast, capable box in the  
>> studio. Less bailing wire, duct tape and magic incantations needed to  
>> hold it all together. :^)  
>>  
>> Almost no administration time needed.  
>>  
>>  
>>> BTW - speaking of intense animation, probably old news, but there's an  
>>> HD  
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>>> fun  
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>>> <http://www.apple.com/quicktime/guide/hd/animusic2dvd.html>  
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>> Clever stuff. I saw another one of those a while back that also  
>> impressed me.  
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>> Cheers,  
>> -Jamie  
>> [www.JamieKruz.com](http://www.JamieKruz.com)  
>>  
>>  
>

---

Subject: Re: Track Counts in Native Systems?  
Posted by [Dedic Terry](#) on Sun, 03 Dec 2006 04:03:06 GMT  
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---

Hi Martin,

I'm running Nuendo on the main system and Cubase 4 on the second. I have them linked via VST System Link (SPDIF). SL is more accurate than MTC, locks faster, and chases jog/scrub better. The video/Cubase 4 PC runs a full screen MJpeg window, in Cubase 4, using Quicktime (17" Samsung LCD). I convert the original Final Cut file in Vegas as MJpeg .mov, DV 720x480, progressive, best quality, etc.

MJpeg seems to work best for native PC playback in general as I can cut down to 75-80% of the original's size and still have acceptable quality (slightly less clarity than the original, but no blocking or color degradation issues as with some other codecs).

I don't have anything else running on this system when syncing video, mainly for client playback, otherwise it's a waste of Cubase 4 :-). This machine is slightly slow for this (Athlon XP 1900 with ATA100 drives), so it skips a

frame every now and then, but maintains sync (though I'm still looking to eliminate the skipping completely in other ways).

Sys Link with Cubase 4 and Nuendo 3 really works quite well. It's a temporary solution for video, but a faster PC with SATA II Drives would probably be perfectly fine.

Regards,  
Dedric

On 12/2/06 4:18 PM, in article 45720768\$1 @linux, "Martin Harrington" <lendan@bigpond.net.au> wrote:

> Hey Dedric,  
> How do you use a separate PC for full screen video, and how are they sync'd,  
> (sync)?  
> Interesting

---

Subject: Re: Track Counts in Native Systems?  
Posted by [Jamie K](#) on Sun, 03 Dec 2006 05:05:35 GMT  
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Hey Dedric, what is your cpu configuration (dual or single) and speed?  
How much RAM do you have?

It sounds like your setup is more RAM starved for sample space, and shouldn't necessarily be CPU starved for processing. Can you set the disk streaming for more efficient RAM usage?

Large libraries shouldn't need to be in RAM all at once. I run the NI stuff you listed and a bunch of other Audio Unit plugins, including some pretty large sample sets, and it's no prob here. Granted I don't run the EW stuff and if you're right about EW, I'm glad I don't.

I have 2.5 GB RAM at present and it's been enough so far for some fairly large arrangements. Other than the sample-based plugs, some AUs are synths that depend more on CPU power than RAM and they run fine in large projects as well, with the G5 muscle. Plus live instrument tracks and copious FX plugins, no sweat, one box.

What's with 8 PCs recommended to run the EW plugs? If it's just samples, that's way overkill for CPU power, assuming they're talking about current boxes.

If it's RAM limitations, then A) sample libraries shouldn't have to run completely in RAM, and B) current machines have a lot of RAM support available to buffer samples - for example 16GB on current MacPro boxes,

which should be plenty o' RAM for sample buffering.

If it's an HD i/o speed bottleneck for streaming samples, how about adding more SATA cards/drives?

I also wonder about the use of 300 tracks. A real orchestra doesn't have 300 sections. It doesn't even have 300 individual players.

Are you using 300 tracks on your stuff? I'm not. Shoot me if I do, the music would get lost in the overproduction.

Maybe the folks you talked to are just trying to save time preloading a lot of plugins they won't necessarily use, that's a choice I guess. But it'd be quicker and waste less resources to work out the arrangement with fewer plugins (but still plenty). And then add what's needed to refine, if anything. I'm not going to fault them for using outboard samplers, that just shows they've been around a while, have a lot invested in that setup, and it works for them. But anyone starting now wouldn't likely go that route.

BTW, GPO switches articulations/sample sets on the fly and runs very efficiently. Big ensembles, small footprint.

Sorry to hear about the trumpet player at your wedding, are you sure it wasn't an EW plugin? ;^) I'm really looking forward to hearing your demo piece!

Cheers,  
-Jamie  
[www.JamieKruz.com](http://www.JamieKruz.com)

PS. You're right about laptops, next laptop I get I'll evaluate for using the plugins at live shows. It should be as fast or faster than my G5, although with less RAM expansion and HD flexibility.

Dedric Terry wrote:

```
> "Jamie K" <Meta@Dimensional.com> wrote in message news:4571cb6a@linux...
>>> With hundreds of articulations required for a score, there isn't a single
>>> box that I could use to run a full complement. And what I could load into
>>> a
>>> current PC/Mac will usually only account for about 30 seconds of scoring
>>> in
>>> one style/tempo.
>> Wow, really? I need to hear one of your scores. Clearly we're having
>> different experiences. What libraries do you use?
>
> I'm using East West's Quantum Leap Symphonic XP library at the moment: 38G
```

> total. For some things  
> I then add Symphonic Choirs at 35G (just one section (e.g. sopranos) uses  
> most of my free Ram).  
> Then factor in Stormdrum, Kontakt 2, Absynth, and numerous other VSTi's for  
> more varied, or modern/cinematic uses  
> and it adds up really fast. EW actually recommends up to 8 PCs for their  
> platinum library (24-bit).  
> The composers I've chatted with a few times in LA use 300 track templates  
> for composing, multiple  
> PCs and a large number of outboard samplers/synths - usually 50-100 inputs.  
> Although I'm not anywhere  
> near that scale of outboard gear, I can see, and feel the need to greatly  
> expand my rig in my work more and more now.  
>  
> I'm getting ready to put together a rather involved and dynamic piece for a  
> new demo - I'll email you a link when it's done.  
>  
>> Heh. My timpani and horns are very good about not dropping out, (that is,  
>> ever since that one Logic bug was finally fixed, grrrr!) and they all live  
>> happily on one box.  
>>  
> That really depends on the percussionist though. Some guys are more  
> reliable than others, esp. if they are former  
> rock drummers. ;-) Hey, the trumpet player fell asleep during my  
> wedding....didn't even take a timpani roll to  
> get him to drop out. He did come back for the recession though. :-)  
>  
> With a larger templates I use up most of my Ram, and since one of the  
> crescendo Fr.Horn instruments I use is actually a mod-wheel dynamic  
> crossfade of three sample sets,  
> as are the timpani modwheel crossfade instruments, they tend to suck down  
> any remaining ram and cpu power very fast  
> when a lot of other instruments are loaded and running.  
>  
>> I used to sync via SMPTE to a BetaCamSP deck. Now I run the video in  
>> Logic, too, as Quicktime. Very slick, SMPTE offset, automatic scene  
>> detection, import/export audio from the video. I've also used Soundtrack  
>> Pro in a similar way, but much prefer to compose in Logic. Either way,  
>> it's all on one box.  
>  
> Apple has video down cold - H.264 is a superb codec, but sadly doesn't work  
> with Nuendo(PC- too bad).  
> AVI works fine but I've never been happy with compression and size vs.  
> quality tradeoffs.  
> Inline video in Nuendo actually works very well - Quicktime MJpeg works well  
> here though better when encoded with Vegas  
> than Quicktime Pro oddly. (I usually run a compressed 320x240 window in  
> Nuendo locally just for sync and spotting).

> I also have a PC sync'd via System Link running full screen DV (720x480)  
> MJpeg (Quicktime playback in Cubase 4) -  
> looks quite good for client previews (not as clear as HD or external  
> monitored DV, but it works well). I'm planning to add a system  
> running the new Decklink HDMI PCIe card for full res HD playback to an HD  
> LCD TV fairly soon - for eye candy mainly.

>  
> It's pretty stunning what we can accomplish with a single computer now  
> though. More power and flexibility is necessary for some  
> things, but there is a ton of great music that can be produced with even  
> just a laptop.

>  
> Regards,  
> Detric

>  
>> Cheers,  
>> -Jamie  
>> www.JamieKruz.com

>>  
>>  
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>>> Regards,  
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>>>> It's really convenient to do it all on one fast, capable box in the  
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>>>>  
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Subject: Re: Track Counts in Native Systems?

Posted by [Martin Harrington](#) on Sun, 03 Dec 2006 05:06:11 GMT

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Thanks Dedic,

I may give it a try although I get full screen playback in Nuendo now with no problems, but I lose one of my screens, (2 x 19inch LCD's).

Since I built my new comp,(Intel core 2 duo E1600), everything's been running like a dream.

--

Martin Harrington

[www.lendaneer-sound.com](http://www.lendaneer-sound.com)

"Dedic Terry" <[dterry@keyofd.net](mailto:dterry@keyofd.net)> wrote in message  
news:C1979A8A.5DBB%[dterry@keyofd.net](mailto:dterry@keyofd.net)...

> Hi Martin,

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> Regards,

> Dedic

>

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> <lendan@bigpond.net.au> wrote:  
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>> Interesting  
>

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---

Subject: Re: Track Counts in Native Systems?  
Posted by [Dedic Terry](#) on Sun, 03 Dec 2006 05:20:43 GMT  
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Hi Martin - That's certainly easier.

I usually run a heavy VSTi load while scoring that I need to cut down what I run for video on the main system, and maintain both screens for editing with post too. The separate video PC is really just eye candy. Not necessary for scoring or spotting effects.

I've heard great things about the core 2 duos, and quad cores. Great to hear yours is running well.

Regards,  
Dedic

On 12/2/06 10:06 PM, in article 457258dd\$1@linux, "Martin Harrington" <lendan@bigpond.net.au> wrote:

> Thanks Dedic,  
> I may give it a try although I get full screen playback in Nuendo now with  
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---

---

Subject: Re: Track Counts in Native Systems?  
Posted by [Dedic Terry](#) on Sun, 03 Dec 2006 07:00:43 GMT  
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---

Jamie,

My system is an AMD X2 (dual core) with 2G of Ram and am planning to bump to 4G soon. EWQLSO uses NI's Kompakt player and I have the system pretty well tweaked for this.

More ram will buy me a little more space to load instruments, and a few more voices, but latency, disk streaming and cpu power also becomes an issue with these libraries at a point.

GPO is a great set - I used it for quite a while before moving to EWQLSO, and it would run fully loaded on my system I am sure (haven't used it since getting the X2). I also have Kirk Hunter's Emerald, though I only use it infrequently for a more unique sound (and where I don't need the depth of EWQL). There is a significant difference between the amount of processing (cpu, ram and disk) required between GPO and EWQLSO. Where GPO might have a single Marcato sample for Violins 1, EWQLSO will have 8 variations depending on what phrasing is required.

Regarding 300 track templates, no I don't use that many yet, but the need is there - I just haven't taken, or had the time to start setting it up. That is really borne out of the necessity of having quick access to what you might use most to minimize loading and setup times, esp. for guys scoring for weekly TV. Even with the 60-100 mid tracks I might use on a short piece, setup really gets tedious and time consuming.

Running the numbers for example:

Just for orchestra with EWQLSO, each subsection may have 30 articulations or more, then you have roughly 10-12 subsections/instruments (Vlns 1 & 2, Violas, Cellos, Basses, Trumpets, Trombones, Fr. Horns, Clarinets, Oboes, Flutes, percussion, solo instruments, etc) - right there are 300+ midi tracks to access each individually. Less common articulations may be loaded only as needed (one doesn't often need a Psycho minor 3rd half step run up ;-).

Then add in several VSTi's on a slave PC, outboard synths/samplers, etc - all that you want accessible simply by enabling a track and selecting a preset/patch, and you have a very large template.

I also frequently run a config similar to what you are describing for many projects with no problem. I can load up, for example, Kontakt 2, Stormdrum, Intakt, 4-5 instances of Rapture (imho, one of the best soft synths on the market), 2 instances of Absynth, plugins, audio tracks, and never push the system.

....but just load up the full orchestra and she starts whinin' about union regulations and demandin' double scale. ;-)

East West's 8 PC recommendation is probably based on each being a 3GHz Pentium or comparable system with 2G ram each minimum, given the timeframe of the manual, and that is for the 24-bit Platinum edition that has 3 mic positions for each instrument, including 5.1 surround samples - cutting that

to 4 core duos is possible. Spreading out any such larger library (even 16 bit stereo samples with one mic position) into just four sections makes it easier and faster to manage loading since each PC can be loading its' section at the same time when starting the session, and each can run at lower latency because it isn't pushed to the limit.

Regards,  
Dedric

PS: the trumpet player was a Mac user... lol ;-)

On 12/2/06 10:05 PM, in article 45725893@linux, "Jamie K" <Meta@Dimensional.com> wrote:

- >
- > Hey Dedric, what is your cpu configuration (dual or single) and speed?
- > How much RAM do you have?
- >
- > It sounds like your setup is more RAM starved for sample space, and
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>  
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>> "Jamie K" <Meta@Dimensional.com> wrote in message news:4571cb6a@linux...  
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>>>> a  
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>>>> one style/tempo.  
>>> Wow, really? I need to hear one of your scores. Clearly we're having  
>>> different experiences. What libraries do you use?  
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>> I then add Symphonic Choirs at 35G (just one section (e.g. sopranos) uses  
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>> Then factor in Stormdrum, Kontakt 2, Absynth, and numerous other VSTi's for  
>> more varied, or modern/cinematic uses  
>> and it adds up really fast. EW actually recommends up to 8 PCs for their  
>> platinum library (24-bit).  
>> The composers I've chatted with a few times in LA use 300 track templates

>> for composing, multiple  
>> PCs and a large number of outboard samplers/synths - usually 50-100 inputs.  
>> Although I'm not anywhere  
>> near that scale of outboard gear, I can see, and feel the need to greatly  
>> expand my rig in my work more and more now.  
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>> new demo - I'll email you a link when it's done.  
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>>> happily on one box.  
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>> That really depends on the percussionist though. Some guys are more  
>> reliable than others, esp. if they are former  
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>> wedding....didn't even take a timpani roll to  
>> get him to drop out. He did come back for the recession though. :-)  
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>> With a larger templates I use up most of my Ram, and since one of the  
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>> looks quite good for client previews (not as clear as HD or external  
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>> running the new Decklink HDMI PCIe card for full res HD playback to an HD  
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>> It's pretty stunning what we can accomplish with a single computer now

>> though. More power and flexibility is necessary for some  
>> things, but there is a ton of great music that can be produced with even  
>> just a laptop.  
>>  
>> Regards,  
>> Dedic  
>>  
>>> Cheers,  
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>>>  
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Subject: Re: Track Counts in Native Systems?  
Posted by [Jamie K](#) on Sun, 03 Dec 2006 08:43:05 GMT  
[View Forum Message](#) <> [Reply to Message](#)

Thanks for the info, Dedic. So the difference between running on one box or multiple boxes comes down to not only the speed/RAM/HD access of the box but the plugin set you choose. My plugin set is happy on my G5,

your set is not happy on your AMD.

Over time that distinction will melt away, as computers continue gaining power.

It's also interesting to see plugin designers coming up with algorithmic emulations, which have reduced RAM requirements, as opposed to straight sample-based approaches. We'll see more of this down the road.

I'll have to check out your setup sometime. Is EWQLSO a tedious way to orchestrate? I find GPO very performance oriented and reasonably quick to use, in addition to its efficient use of RAM/CPU.

I also keep a violin handy when I need complete control of articulation. :^)

Cheers,  
-Jamie  
[www.JamieKrutz.com](http://www.JamieKrutz.com)

PS. The groom was also a Mac user, although not exclusively. :^)

Dedric Terry wrote:

> Jamie,  
>  
> My system is an AMD X2 (dual core) with 2G of Ram and am planning to bump to  
> 4G soon. EWQLSO uses NI's Kompakt player and I have the system pretty well  
> tweaked for this.  
>  
> More ram will buy me a little more space to load instruments, and a few more  
> voices, but latency, disk streaming and cpu power also becomes an issue with  
> these libraries at a point.  
>  
> GPO is a great set - I used it for quite a while before moving to EWQLSO,  
> and it would run fully loaded on my system I am sure (haven't used it since  
> getting the X2). I also have Kirk Hunter's Emerald, though I only use it  
> infrequently for a more unique sound (and where I don't need the depth of  
> EWQL). There is a significant difference between the amount of processing  
> (cpu, ram and disk) required between GPO and EWQSOL. Where GPO might have a  
> single Marcato sample for Violins 1, EWQLSO will have 8 variations depending  
> on what phrasing is required.  
>  
> Regarding 300 track templates, no I don't use that many yet, but the need is  
> there - I just haven't taken, or had the time to start setting it up. That  
> is really borne out of the necessity of having quick access to what you  
> might use most to minimize loading and setup times, esp. for guys scoring



> for weekly TV. Even with the 60-100 mid tracks I might use on a short  
> piece, setup really gets tedious and time consuming.  
>  
>  
> Running the numbers for example:  
>  
> Just for orchestra with EWQLSO, each subsection may have 30 articulations or  
> more, then you have roughly 10-12 subsections/instruments (Vlns 1 & 2,  
> Violas, Cellos, Basses, Trumpets, Trombones, Fr. Horns, Clarinets, Oboes,  
> Flutes, percussion, solo instruments, etc) - right there are 300+ midi  
> tracks to access each individually. Less common articulations may be loaded  
> only as needed (one doesn't often need a Psycho minor 3rd half step run up  
> ;-).  
>  
> Then add in several VSTi's on a slave PC, outboard synths/samplers, etc -  
> all that you want accessible simply by enabling a track and selecting a  
> preset/patch, and you have a very large template.  
>  
> I also frequently run a config similar to what you are describing for many  
> projects with no problem. I can load up, for example, Kontakt 2, Stormdrum,  
> Intakt, 4-5 instances of Rapture (imho, one of the best soft synths on the  
> market), 2 instances of Absynth, plugins, audio tracks, and never push the  
> system.  
>  
> ...but just load up the full orchestra and she starts whinin' about union  
> regulations and demandin' double scale. ;-)  
>  
> East West's 8 PC recommendation is probably based on each being a 3GHz  
> Pentium or comparable system with 2G ram each minimum, given the timeframe  
> of the manual, and that is for the 24-bit Platinum edition that has 3 mic  
> positions for each instrument, including 5.1 surround samples - cutting that  
> to 4 core duos is possible. Spreading out any such larger library (even 16  
> bit stereo samples with one mic position) into just four sections makes it  
> easier and faster to manage loading since each PC can be loading its'  
> section at the same time when starting the session, and each can run at  
> lower latency because it isn't pushed to the limit.  
>  
> Regards,  
> Dedic  
>  
> PS: the trumpet player was a Mac user... lol ;-)  
>  
> On 12/2/06 10:05 PM, in article 45725893@linux, "Jamie K"  
> <Meta@Dimensional.com> wrote:  
>  
>> Hey Dedic, what is your cpu configuration (dual or single) and speed?  
>> How much RAM do you have?  
>>

>> It sounds like your setup is more RAM starved for sample space, and  
>> shouldn't necessarily be CPU starved for processing. Can you set the  
>> disk streaming for more efficient RAM usage?  
>>  
>> Large libraries shouldn't need to be in RAM all at once. I run the NI  
>> stuff you listed and a bunch of other Audio Unit plugins, including some  
>> pretty large sample sets, and it's no prob here. Granted I don't run the  
>> EW stuff and if you're right about EW, I'm glad I don't.  
>>  
>> I have 2.5 GB RAM at present and it's been enough so far for some fairly  
>> large arrangements. Other than the sample-based plugs, some AUs are  
>> synths that depend more on CPU power than RAM and they run fine in large  
>> projects as well, with the G5 muscle. Plus live instrument tracks and  
>> copious FX plugins, no sweat, one box.  
>>  
>> What's with 8 PCs recommended to run the EW plugs? If it's just samples,  
>> that's way overkill for CPU power, assuming they're talking about  
>> current boxes.  
>>  
>> If it's RAM limitations, then A) sample libraries shouldn't have to run  
>> completely in RAM, and B) current machines have a lot of RAM support  
>> available to buffer samples - for example 16GB on current MacPro boxes,  
>> which should be plenty o' RAM for sample buffering.  
>>  
>> If it's an HD i/o speed bottleneck for streaming samples, how about  
>> adding more SATA cards/drives?  
>>  
>> I also wonder about the use of 300 tracks. A real orchestra doesn't have  
>> 300 sections. It doesn't even have 300 individual players.  
>>  
>> Are you using 300 tracks on your stuff? I'm not. Shoot me if I do, the  
>> music would get lost in the overproduction.  
>>  
>> Maybe the folks you talked to are just trying to save time preloading a  
>> lot of plugins they won't necessarily use, that's a choice I guess. But  
>> it'd be quicker and waste less resources to work out the arrangement  
>> with fewer plugins (but still plenty). And then add what's needed to  
>> refine, if anything. I'm not going to fault them for using outboard  
>> samplers, that just shows they've been around a while, have a lot  
>> invested in that setup, and it works for them. But anyone starting now  
>> wouldn't likely go that route.  
>>  
>> BTW, GPO switches articulations/sample sets on the fly and runs very  
>> efficiently. Big ensembles, small footprint.  
>>  
>> Sorry to hear about the trumpet player at your wedding, are you sure it  
>> wasn't an EW plugin? ;^) I'm really looking forward to hearing your demo  
>> piece!

>>  
>> Cheers,  
>> -Jamie  
>> www.JamieKruz.com  
>>  
>> PS. You're right about laptops, next laptop I get I'll evaluate for  
>> using the plugins at live shows. It should be as fast or faster than my  
>> G5, although with less RAM expansion and HD flexibility.  
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Subject: Re: Track Counts in Native Systems?  
Posted by [Dedric Terry](#) on Sun, 03 Dec 2006 13:37:36 GMT  
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On 12/3/06 1:43 AM, in article 45728b8a@linux, "Jamie K"  
<Meta@Dimensional.com> wrote:

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> box or multiple boxes comes down to not only the speed/RAM/HD access of  
> the box but the plugin set you choose. My plugin set is happy on my G5,  
> your set is not happy on your AMD.

It's a size issue. GPO (2G) is significantly smaller than EWQLSO (38G), so  
it really isn't taxing your G5 as much as EWQL would.

> I'll have to check out your setup sometime. Is EWQLSO a tedious way to  
> orchestrate? I find GPO very performance oriented and reasonably quick  
> to use, in addition to its efficient use of RAM/CPU.

Certainly - feel free to pop down for a visit sometime. The advantage of  
EWQLSO, VSL or Sonic Implants libraries is that they provide a much wider  
range of realism, and creating that realism in a score is time consuming,  
though most have quite a few performance shortcuts to simplify the process -  
it's easy enough to be as quick or as accurate and detailed as desired.

Regards,

Dedric

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> I also keep a violin handy when I need complete control of articulation. :^)  
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> PS. The groom was also a Mac user, although not exclusively. :^)  
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Subject: Re: Track Counts in Native Systems?  
Posted by [Jamie K](#) on Sun, 03 Dec 2006 18:06:44 GMT  
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Dedric Terry wrote:

> On 12/3/06 1:43 AM, in article 45728b8a@linux, "Jamie K"  
> <Meta@Dimensional.com> wrote:  
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>> your set is not happy on your AMD.  
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> It's a size issue. GPO (2G) is significantly smaller than EWQLSO (38G), so  
> it really isn't taxing your G5 as much as EWQL would.

But since EWQL runs in Kontakt, and Kontakt can do disk streaming, there must be a happy medium where you could have much less than 38GB of RAM and still access the entire library.

If current machines can't yet do this in one box, that day will come. Of course for anyone willing to freeze tracks as they go, one box should handle it already. Continued development of algorithmic emulations will also likely lead to less RAM heavy solutions for orchestrations, some pieces of that puzzle are already available.

Since I'm not using EWQL, and even though I am using several other large sample libraries, being able to run on a single box without freezing is a great way to work. Compared to every system I've used in the past, I really appreciate this setup.

But it's also great that working on multiple boxes is getting the job done for you with EWQL. For anyone who really likes and needs EWQL the

extra hardware would be worth it.

>> I'll have to check out your setup sometime. Is EWQLSO a tedious way to  
>> orchestrate? I find GPO very performance oriented and reasonably quick  
>> to use, in addition to its efficient use of RAM/CPU.

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> Certainly - feel free to pop down for a visit sometime. The advantage of  
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> range of realism, and creating that realism in a score is time consuming,  
> though most have quite a few performance shortcuts to simplify the process -  
> it's easy enough to be as quick or as accurate and detailed as desired.

Next time I'm heading your way I'll stop in. I can see advantages and disadvantages to the EWQLSO approach and it will be nice to hear how you're using it.

I have a lot of respect for the GPO approach to make a library that's efficient and easily performable in real time, with clever control functions that allow enough variation and oft-used articulations to create credible orchestrations. I also like that it's a dry library that can be mixed into any space. The EWQLSO approach of samples recorded with room space included is also valid and useful for many applications, and it's an impressively deep collection.

Cheers,  
-Jamie  
[www.JamieKrutz.com](http://www.JamieKrutz.com)

PS. Just saw a report of the emergency radio system test in the Springs. Did the air force jam your garage door? "When a secretive Air Force installation in Colorado Springs began testing the radio signal, it knocked out remote control garage door openers around the area. The communication system, intended to reach first responders, uses the same frequency as an estimated 50 million garage door openers." -AP

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>> PS. The groom was also a Mac user, although not exclusively. :^)

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Subject: Re: Track Counts in Native Systems?  
Posted by [Dedric Terry](#) on Sun, 03 Dec 2006 22:13:02 GMT  
[View Forum Message](#) <> [Reply to Message](#)

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On 12/3/06 11:06 AM, in article 45730fa6@linux, "Jamie K"  
<Meta@Dimensional.com> wrote:

> But since EWQL runs in Kontakt, and Kontakt can do disk streaming, there  
> must be a happy medium where you could have much less than 38GB of RAM  
> and still access the entire library.

DSD can only do so much streaming and swapping out with that amount of data  
and number of instruments and orchestral library requires. With multiple  
instruments DSD is handling multiple separate processes, not just one as  
with a grand piano. (That may not be quite accurate, but it makes sense in  
light of how system requirements vary when I load a large piano sample vs.  
multiple orchestral instruments).

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> If current machines can't yet do this in one box, that day will come. Of  
> course for anyone willing to freeze tracks as they go, one box should  
> handle it already. Continued development of algorithmic emulations will  
> also likely lead to less RAM heavy solutions for orchestrations, some  
> pieces of that puzzle are already available.

But emulations come at the expense of higher cpu usage, and realism. It's  
certainly an interesting approach and does work fairly well in the VSTi's  
I've tried, for what they are, but it still doesn't approach the realism of  
a large sample library.... which itself doesn't completely replace the  
realism of the real thing, though it can be quite convincing.

> Next time I'm heading your way I'll stop in. I can see advantages and  
> disadvantages to the EWQLSO approach and it will be nice to hear how  
> you're using it.

Please do! Seeing and hearing is the easiest way to get a feel for why many  
composers go this route. But don't anticipate benching GPO for EWQLSO  
though - the whole goal is to maximize creativity, and when you have a setup  
you are excited about working with and are productive with, I think it's  
best to stick with it.

Regards,  
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> PS. Just saw a report of the emergency radio system test in the Springs.  
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> installation in Colorado Springs began testing the radio signal, it  
> knocked out remote control garage door openers around the area. The  
> communication system, intended to reach first responders, uses the same  
> frequency as an estimated 50 million garage door openers." -AP

PS: Our garage door opener is working - maybe it's in the Navy radio band.  
:-)

>  
>  
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Subject: Re: Track Counts in Native Systems?  
Posted by [AlexPlasko](#) on Sun, 03 Dec 2006 23:32:20 GMT  
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dedric  
how many instances are you able to run together? do you run EWQLSO and  
EWQLSC at the same time? I have choirs, and am trying to hold out for a  
group buy offering on orchestra.  
word builder is amazing isnt it?  
"Detric Terry" <dterry@keyofd.net> wrote in message  
news:C19899FE.5E01%dterry@keyofd.net...  
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>> Did the air force jam your garage door? "When a secretive Air Force  
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>>> Regards,  
>>> Dedic  
>>>  
>>>> I also keep a violin handy when I need complete control of  
>>>> articulation. :^)  
>>>>  
>>>> Cheers,  
>>>> -Jamie  
>>>> www.JamieKruz.com  
>>>>  
>>>> PS. The groom was also a Mac user, although not exclusively. :^)  
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Subject: Re: Track Counts in Native Systems?  
Posted by [Dedic Terry](#) on Mon, 04 Dec 2006 00:30:25 GMT  
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Hi Alex,

I run about 8 instances of EWQL's Kompakt player. I can't run Choirs simultaneously - too memory intensive, for the switched parts at least. All the East West stuff is on sale for about 50% off right now. That's what I paid for Choirs in a group buy.

Regards,  
Dedic

On 12/3/06 4:32 PM, in article 45735c0c@linux, "alex plasko" <alex.plasko@snet.net> wrote:

> dedric  
> how many instances are you able to run together? do you run EWQLSO and  
> EWQLSC at the same time? I have choirs, and am trying to hold out for a  
> group buy offering on orchestra.  
> word builder is amazing isnt it?  
> "Dedic Terry" <dterry@keyofd.net> wrote in message  
> news:C19899FE.5E01%dterry@keyofd.net...  
>> On 12/3/06 11:06 AM, in article 45730fa6@linux, "Jamie K"  
>> <Meta@Dimensional.com> wrote:

>>  
>>> But since EWQL runs in Kontakt, and Kontakt can do disk streaming, there  
>>> must be a happy medium where you could have much less than 38GB of RAM  
>>> and still access the entire library.  
>>  
>> DSD can only do so much streaming and swapping out with that amount of  
>> data  
>> and number of instruments and orchestral library requires. With multiple  
>> instruments DSD is handling multiple separate processes, not just one as  
>> with a grand piano. (That may not be quite accurate, but it makes sense  
>> in  
>> light of how system requirements vary when I load a large piano sample vs.  
>> multiple orchestral instruments).  
>>>  
>>> If current machines can't yet do this in one box, that day will come. Of  
>>> course for anyone willing to freeze tracks as they go, one box should  
>>> handle it already. Continued development of algorithmic emulations will  
>>> also likely lead to less RAM heavy solutions for orchestrations, some  
>>> pieces of that puzzle are already available.  
>>  
>> But emulations come at the expense of higher cpu usage, and realism. It's  
>> certainly an interesting approach and does work fairly well in the VSTi's  
>> I've tried, for what they are, but it still doesn't approach the realism  
>> of  
>> a large sample library.... which itself doesn't completely replace the  
>> realism of the real thing, though it can be quite convincing.  
>>  
>>> Next time I'm heading your way I'll stop in. I can see advantages and  
>>> disadvantages to the EWQLSO approach and it will be nice to hear how  
>>> you're using it.  
>>  
>> Please do! Seeing and hearing is the easiest way to get a feel for why  
>> many  
>> composers go this route. But don't anticipate benching GPO for EWQLSO  
>> though - the whole goal is to maximize creativity, and when you have a  
>> setup  
>> you are excited about working with and are productive with, I think it's  
>> best to stick with it.  
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>> Regards,  
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Subject: Re: Track Counts in Native Systems?  
Posted by [Jamie K](#) on Mon, 04 Dec 2006 18:09:03 GMT  
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Detric Terry wrote:

> On 12/3/06 11:06 AM, in article 45730fa6@linux, "Jamie K"  
> <Meta@Dimensional.com> wrote:  
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> I've tried, for what they are, but it still doesn't approach the realism of  
> a large sample library.... which itself doesn't completely replace the  
> realism of the real thing, though it can be quite convincing.

We've gone from pipe organs taking up the space of a small house to software plugins inside laptops. An amazing progression of ensemble emulation tools, and the progression will continue.

I suspect algorithmic emulations will take over eventually. CPU power is less and less a problem. Potential advantages in performability, low storage/bandwidth requirements and ever improving accuracy/quality will tilt the balance at some point.

Meanwhile, speaking of realism, you're right. A great use of these libraries is as mockups in preparation for directing a real orchestra. I've also found that combining real instrument tracks with sampled tracks aids greatly in producing a convincing piece.

>> Next time I'm heading your way I'll stop in. I can see advantages and  
>> disadvantages to the EWQLSO approach and it will be nice to hear how  
>> you're using it.

>

> Please do! Seeing and hearing is the easiest way to get a feel for why many  
> composers go this route. But don't anticipate benching GPO for EWQLSO  
> though - the whole goal is to maximize creativity, and when you have a setup  
> you are excited about working with and are productive with, I think it's  
> best to stick with it.

Good advice. Whenever I add a new approach I usually keep the existing ones around and integrate the new approach gradually, and only where it makes sense.

Cheers,  
-Jamie  
[www.JamieKruz.com](http://www.JamieKruz.com)

> Regards,  
> Dedic

>

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