Subject: Re: digital_filter

Posted by Wox on Tue, 29 Jan 2008 13:13:21 GMT

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On Mon, 28 Jan 2008 06:46:07 -0800 (PST), claire.maraldi@gmail.com
wrote:
> Hi,
>
> I have a problem of comprehension with the digital filter tool.
> I have an array "X" of n elements, the delta between two elements is
> "1unit"
> I would like to creat a pass band filter between 1unit and 60 units
> In IDL I write
> ; Filtre : 1unit<T<60unit ;
> delta=1.
> f low=1./(60.*1.)
> f high=1./(1.*1.)
> a ripple=50. & nterms=40.
> coeff=DIGITAL_FILTER(f_low*2*delta, f_high*2*delta, a_ripple, nterms)
> X_filter=convol(REFORM(X),coeff,/EDGE_TRUNCATE)
>
>
> Then when I compute FFT(X filter), I see than my signal has been cut
> between 1 unit and more than 60units (about 100 units). Are my
> definitions of f_low and f_high wrong, or is this a problem related to
> the digital filter function?
>
> Thanks
> Claire
If you plot the filter in the fequency domain, do you get what you
expect?
nfilt = N_ELEMENTS(coeff)
filter fourier = FFT(coeff)*nfilt
f filt = FINDGEN(nfilt/2+1) / (nfilt*delta)
mag = ABS(filter_fourier[0:nfilt/2])
plot,f_filt,alog(mag)
You can off course do the filtering in the fequency domain instead:
x filtered = fft(fft(x,1)*filter,-1)
```

For the filter see e.g.

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