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Subject: Re: Using IDL to make a signal filter  
Posted by [Wox](#) on Mon, 14 Jul 2008 17:33:06 GMT  
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On Mon, 14 Jul 2008 04:08:15 -0700 (PDT), ICBM0926  
<ICBM0926@gmail.com> wrote:

> I have an 1D vector data in IDL from an analytical laser formula which  
> contains 2 laser frequencies. I wrote a program trying to filter my 1D  
> vector and get the waveform of one of the frequencies. I used 1D FFT  
> and a mask  
> function(step function). I applied the mask function to the frequency  
> domain data. I've covered both positive and negative frequencies. I  
> did inverse 1d FFT to retrieve the signal. I found that  
> the amplitude of the signal is only half as it should be. Could  
> anybody tell me what went wrong?

In the example below, you see first a sum of three sinus waves, all  
with amplitude 1. After filtering two of them out, 1 sinus is  
remaining with amplitude 1, not 0.5 as in your case. Does this help?

pro test

```
; Time domain
freq1=2.
freq2=3.
freq3=4.
dtime=0.05
ntime=200

time=dtime*findgen(ntime)
signal=sin(2!*pi*freq1*time)+sin(2!*pi*freq2*time)+sin(2!*pi *freq3*time)

; Frequency domain
nfreq=ntime/2+1
freq=findgen(nfreq)/(dtime*ntime)
freq=[freq,reverse(-freq[1:nfreq-1-(~(ntime mod 2))])]
fsignal=fft(signal,-1)

; Frequency domain filter
f_low = 0
f_high = 2.5
steep=20.
freqfilter= 1./(1.+(freq/f_high)^steep)
fsignalfilt=fsignal*freqfilter
```

```
; Back to time domain
signalfilt=fft(fsignalfilt,1)

; Plot
window,0
!P.MULTI=[0,2,2]
  plot,freq[0:nfreq-1],abs(fsignal[0:nfreq-1])^2,xtitle='frequency',ytitle='spectrum',title='Original
Spectrum'
  plot,freq[0:nfreq-1],freqfilter[0:nfreq-1],xtitle='frequency ',ytitle='filter',title='Filter'
  plot,freq[0:nfreq-1],abs(fsignalfilt[0:nfreq-1])^2,xtitle='frequency',ytitle='filtered
spectrum',title='Filtered spectrum'

window,1
!P.MULTI=[0,1,2]
plot,time,signal,xtitle='time',ytitle='signal'
plot,time,signalfilt,xtitle='time',ytitle='filtered signal'
end
```

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