Subject: Re: bug in IDL's hanning() window-generating function Posted by bennetsc on Sat, 04 Aug 2001 07:18:03 GMT

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In article <3B6B1D98.1665BFFE@rsinc.com>, Harold Cline <harold@rsinc.com> wrote: > Just an FYI for everyone on this topic. ;-) > > Thanks. > Harold Cline > IDL Product Manager

- > RELEASE_NOTE for IDL5.5:
- > HANNING:
- > The formula to compute the Hanning window has been changed
- > from 0.5 * (1-Cos(2*!Pi*k/(N-1))) to 0.5 * (1-Cos(2*!Pi*k/N)).
- > Because of the change from 1/(N-1) to 1/N, the Hanning window
- > will no longer
- > appear symmetric in the time domain, but will give improved results for
- > spectral estimation.

That's very good to know.

- > > Additional notes:
- > When filters such as the Hanning are to be used as
- > finite-impulse response
- > (FIR) filtering in the time domain, it is useful to have the filter be
- > symmetric, and go to zero at both ends. However, if you are

First off, data filtering and data windowing (a.k.a. data tapering) are two different things. One can design an FIR data filter by modifying a spectral window to pass or stop certain frequency/wavenumber ranges as desired. So a Hann spectral window (i.e. FFT(HANNING(length of desired filter))) could be modified and then used as a time-domain filter.

- > filtering your
- > time series in preparation for doing spectral analysis, the
- > filter should
- > not be symmetric, and should not go to zero at one of the
- > endpoints. Thus,

That is correct. Note, however, that one often wishes to calculate a spectrum without filtering. In my work, for example, I have not filtered the data before applying the window and taking the transform because I want to see all of the resolvable

spectrum. A data window is still necessary to prevent generation of spurious spectral power that would result from the break in "continuity" at the data segment boundaries.

IMHO, IDL's DIGITAL_FILTER() provides a much better filter than it is possible to contrive from a Hann spectral window, though it can take a bit of experimentation to find good values to specify for "A" (maximum level Gibbs ringing.) With the right value for "A", DIGITAL_FILTER() can even outperform a filter based upon the sinc function (FT of a boxcar window) at the same filter order. For those who want to try this, code a fairly wide filter size and code 43.93125 for the A argument to DIGITAL FILTER(). Then generate a sinc-based filter with the same number of coefficients and compare the frequency responses of the two filters. IIRC, the sinc-based filter has a maximum Gibbs ringing amplitude in the range of only 2%-3% and the maximum ringing amplitude from the filter you get from DIGITAL FILTER() is about half that, while the transition ramps of the two filters are practically indistinguishable. (A look at the code for DIGITAL FILTER() looks like it might be a since filter that's been tightened up a bit further by multiplying it by a Kaiser-Bessel data window. Pretty neat.)

A caveat here: certain configurations of arguments to DIGITAL_FILTER() yield filters that should preserve the mean of the data, but don't. This can easily be fixed by normalizing all the coefficients before convolving them with the data. For example.

```
filter = digital_filter(0., .00777778, 43.93125, 1125)
filter = filter / total(filter)
> the new behavior of the HANNING function assumes it will be used for
> spectral analysis.
> Reference: Oppenheim & Schafer (Discrete-Time Signal Processing,
> 1999), p.
> 468-469.
   Thanks for fixing it.
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* "Lay then the axe to the root, and teach governments humanity. * It is their sanguinary punishments which corrupt mankind." * -- _The_Rights_of_Man_ by Tom Paine (1791.)