Filter Design Results

Generated by: http://www-users.cs.york.ac.uk/~fisher/mkfilter

Summary

You specified the following parameters:

```
filtertype = Butterworth
passtype = Bandpass
ripple =
order = 1
samplerate = 8000
corner1 = 500
corner2 = 1000
adzero =
logmin =
```

Results

```
Command line: /www/usr/fisher/helpers/mkfilter -Bu -Bp -o 1 -a 6.2500000000e-02 1.2500000000e-01
raw alpha1 = 0.0625000000
raw alpha2 = 0.1250000000
warped alpha1 = 0.0633157730
warped alpha2 = 0.1318482719
gain at dc : mag = 0.0000000000e+00
gain at centre: mag = 5.962907802e+00
                                       phase = -0.0465843137 pi
gain at hf : mag = 0.0000000000e+00
S-plane zeros:
         0.0000000000 + j 0.0000000000
S-plane poles:
        -0.2153011950 + j 0.5321787262
        -0.2153011950 + j -0.5321787262
Z-plane zeros:
         1.0000000000 + j 0.0000000000
        -1.0000000000 + j 0.0000000000
Z-plane poles:
         0.7071067812 + j 0.4100958887
         0.7071067812 + j -0.4100958887
Recurrence relation:
y[n] = (-1 * x[n-2])
    + ( 0 * x[n-1])
    + (1 * x[n-0])
    + ( -0.6681786379 * y[n- 2])
    + ( 1.4142135624 * y[n-1])
```

Ansi ``C" Code

```
/* Digital filter designed by mkfilter/mkshape/gencode A.J. Fisher
  Command line: \/\www/usr/fisher/helpers/mkfilter -Bu -Bp -o 1 -a 6.2500000000e-02 1.2500000000e-01 -l */
#define NZEROS 2
#define NPOLES 2
#define GAIN 5.962907802e+00
static float xv[NZEROS+1], yv[NPOLES+1];
static void filterloop()
 { for (;;)
      \{ xv[0] = xv[1]; xv[1] = xv[2];
       xv[2] = next input value / GAIN;
       yv[0] = yv[1]; yv[1] = yv[2];
       yv[2] = (xv[2] - xv[0])
                    + (-0.6681786379 * yv[0]) + (1.4142135624 * yv[1]);
       next output value = yv[2];
     }
 }
```

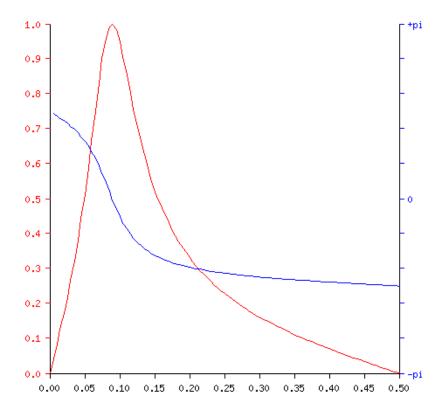
Download code and/or coefficients:

TERSE

VERBOSE

Magnitude (red) and phase (blue) vs. frequency

- *x* axis: frequency, as a fraction of the sampling rate (i.e. 0.5 represents the Nyquist frequency, which is 4000 Hz)
- y axis (red): magnitude (linear, normalized)
- y axis (blue): phase

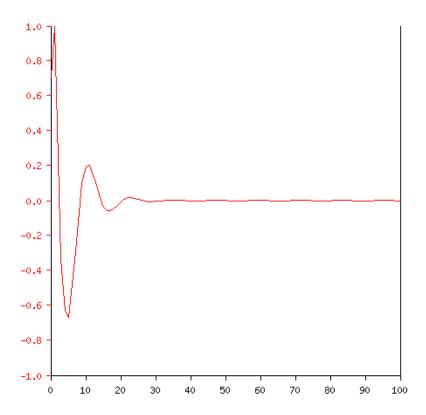


For an expanded view, enter frequency limits (as a fraction of the sampling rate) here:

Lower limit:	Upper limit:		zoom
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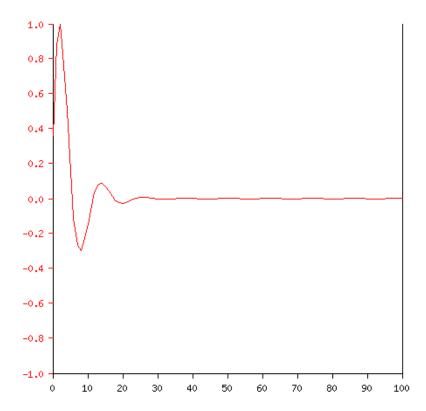
Impulse response

- *x* axis: time, in samples (i.e. 8000 represents 1 second)
- y axis (red): filter response (linear, normalized)



Step response

- *x* axis: time, in samples (i.e. 8000 represents 1 second)
- y axis (red): filter response (linear, normalized)



For a view on a different scale, enter upper time limit (integer number of samples) here:

Upper limit: zoom

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