

Algorithm for Time Domain Simulation of Analog Filters Using the Symbolic Math Toolbox

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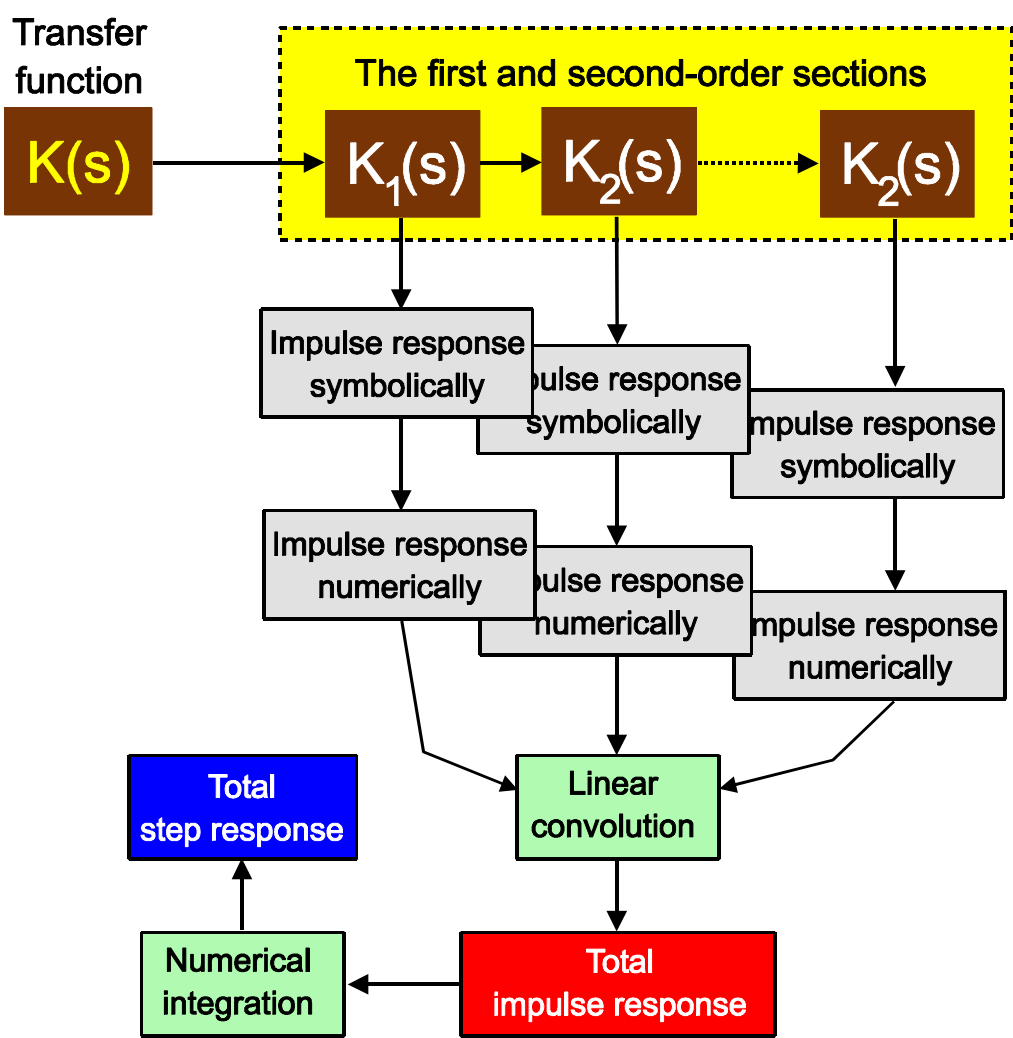
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1. Abstract

- a set of new MATLAB functions was created to extend the current possibilities of time-domain analysis of analog filters
- the transfer function of the analog filter is an input expression of the algorithm
- the transfer function of the filter is divided into the first and second-order sections
- the impulse response of each section is derived symbolically using the MATLAB Symbolic Toolbox
- the impulse response and step response of the whole filter are obtained using a linear convolution and integration
- the described technique is suitable for accurate and robust calculation of the time domain responses of arbitrary-order analog filters

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2. Basic idea of the algorithm



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3. Impulse response in a symbolic form

- the impulse response of each section is derived using the MATLAB Symbolic Toolbox (L-Laplace transform)

$$g(t) = \mathcal{L}^{-1}\{K(s)\}$$

Example 1

$$K(s) = \frac{s^2 + \omega_c^2}{s^2 + \frac{\omega_c}{Q}s + \omega_c^2} \quad \text{2nd-order band stop filter}$$

MATLAB source code

```
syms s wr Q
K = (s^2 + wr^2) / (s^2 + wr/Q*s + wr^2)
g = ilaplace(K)
```

$$g(t) = \delta(t) + 2\omega_c e^{-\omega_c t} \cos(\omega_c t) + 2 \frac{\omega_c^2}{\omega_c} e^{-\omega_c t} \sin(\omega_c t)$$
$$g(t)_{Q=0.5} = \delta(t) - (2\omega_c - 2\omega_c^2 t) e^{-\omega_c t}$$
$$x = -\frac{1}{2} \frac{\omega_c}{Q}, \quad y = \frac{1}{2} \frac{\omega_c}{Q} \sqrt{4Q^2 - 1}$$

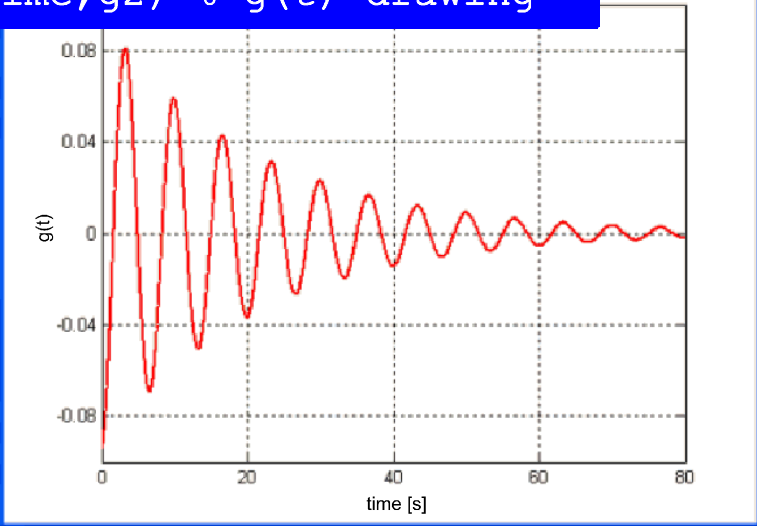
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4. Impulse response in a numeric form

- the numeric form of the impulse response is obtained using a substitution

Example 2

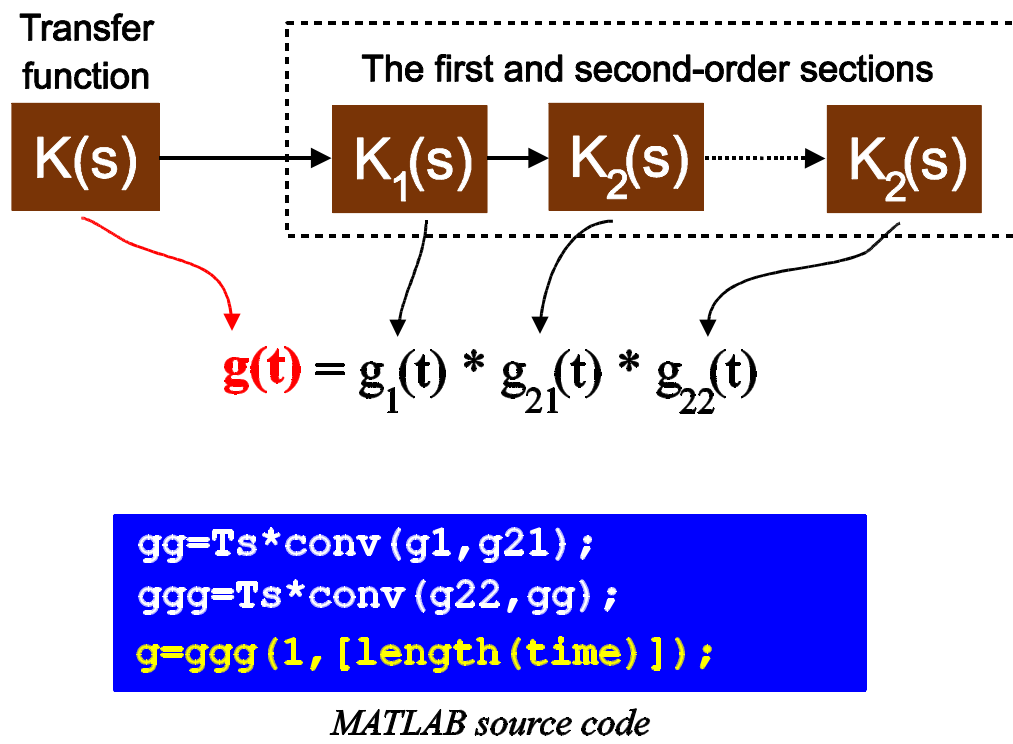
```
fr1=0.15; % resonant frequency
wr1=2*pi*fr1;
Q1=10; % Q-factor
time=0:2/100:10; % time axis
g1=subs(g, {wr,Q}, {wr1 Q1});
g2=real(subs(g1, 't', time));
plot(time,g2) % g(t) drawing
```



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5. Total impulse response

- the total impulse response of the whole filter is computed using a linear convolution



```
gg=Ts*conv(g1,g21);
ggg=Ts*conv(g22,gg);
g=ggg(1,[length(time)]);
```

MATLAB source code

Ts - sampling period

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Example 3

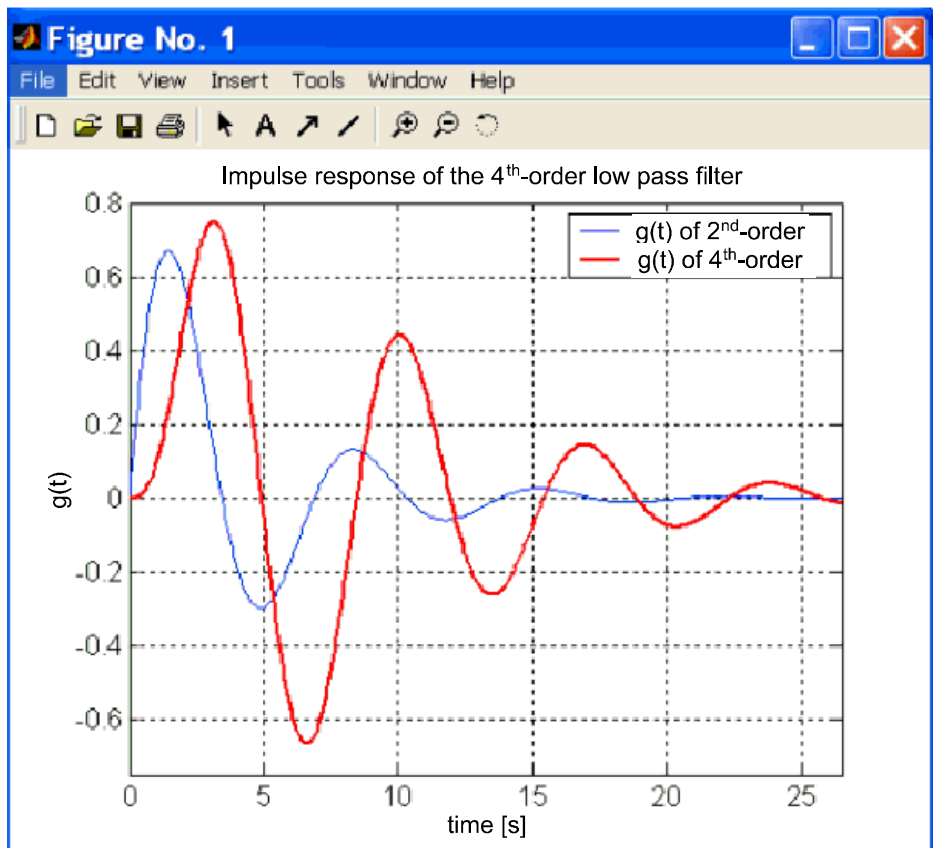
- computation of the impulse response of the 4th-order low-pass filter (two identical sections)

```
syms s wr Q
K=wr^2/(s^2+wr/Q*s+wr^2); % one section
g=ilaplace(K); % impulse response
fr1=0.15;
wr1=2*pi*fr1;
Q1=2;
maxtime=25/wr1;
numstep=500; % number of steps
time=0:maxtime/(numstep-1):maxtime;
Ts=maxtime/numstep; % sampling period
g1=subs(g, {wr,Q}, {wr1 Q1});
g2=real(subs(g1, 't', time));
gg4=Ts*conv(g2,g2); % second section
g4=gg4(1,[length(time)]);
plot(time,g2) % g(t) of 2nd-order section hold on
plot(time,g4) % g(t) of 4th-order filter
```

MATLAB source code

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- impulse responses of the 2nd-order section and the whole 4th-order filter



Impulse responses diagram

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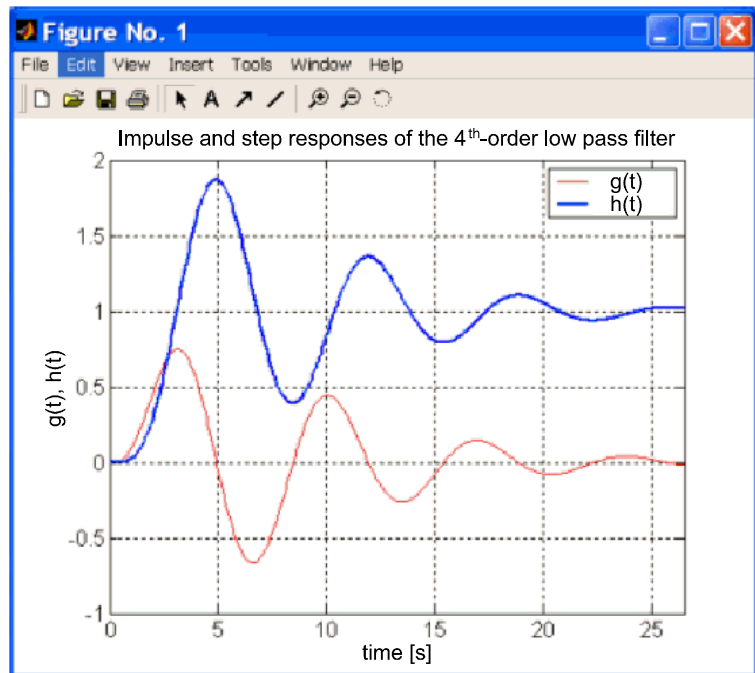
6. Total step response

- the step response of the whole filter is computed using a digital integrator (a simple IIR digital filter)

$$\text{Digital integrator } K(z) = \frac{1}{1-z^{-1}} = \frac{z}{z-1}$$

```
h=Ts*filter(1,[1 -1],g4);
h=h(1,[length(time)]);
```

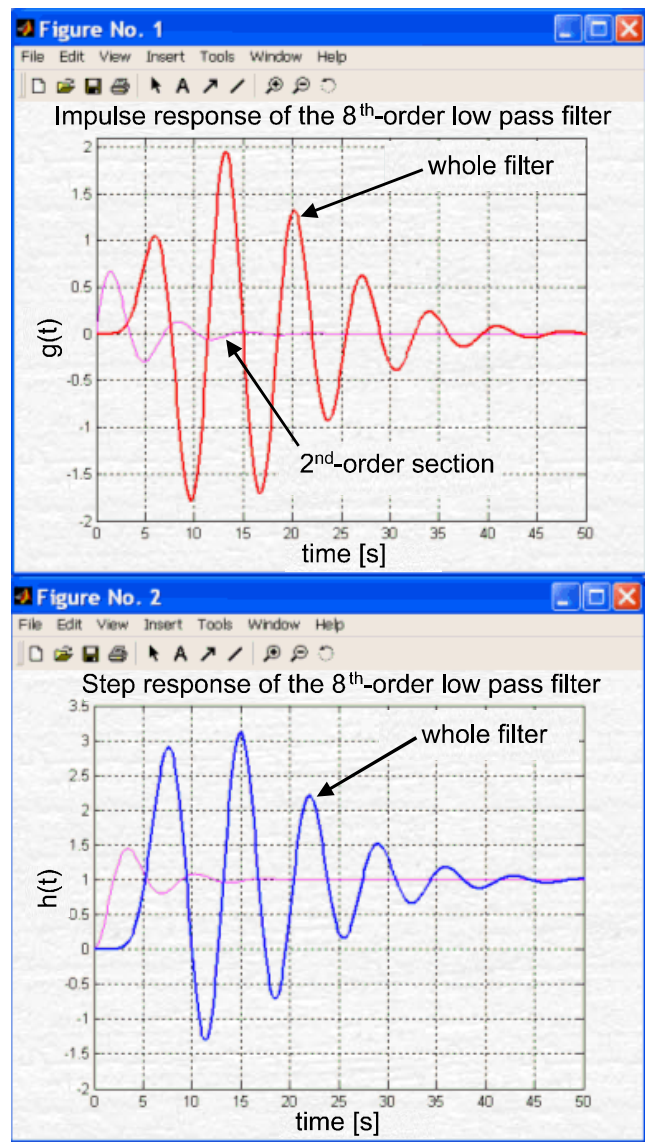
MATLAB source code



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7. Illustrational example

- time domain analysis of the 8th-order low-pass filter



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8. Generalized mathematical formalism

- the 1st and 2nd-order sections have different models according to the Q-factor ($Q=0,5[-]$ and an arbitrary value)
- nevertheless we can derive common models using a limit calculus

2nd-order high-pass filter

$$g(t)_Q = \lim_{Q \rightarrow Q} \left[\delta(t) + 2\omega_c e^{-\omega_c t} \cos(\omega_c t) + 2 \frac{\omega_c^2}{\omega_c} (1 - 2Q^2) e^{-\omega_c t} \sin(\omega_c t) \right] \quad \checkmark$$
$$g(t)_{0,5} = \lim_{Q \rightarrow 0,5} \left[\delta(t) + 2\omega_c e^{-\omega_c t} \cos(\omega_c t) + 2 \frac{\omega_c^2}{\omega_c} (1 - 2Q^2) e^{-\omega_c t} \sin(\omega_c t) \right] =$$
$$= \lim_{Q \rightarrow 0} \left[\delta(t) + 2\omega_c e^{-\omega_c t} \cos(\omega_c t) + 2 \frac{\omega_c^2}{\omega_c} (1 - 2Q^2) e^{-\omega_c t} \sin(\omega_c t) \right] =$$
$$\text{plati } x = -\frac{1}{2} \frac{\omega_c}{Q} \text{ pro } Q = 0,5 \Rightarrow x = -\omega_c$$
$$\lim_{Q \rightarrow 0} \frac{\sin(\omega_c t)}{\omega_c} = \lim_{Q \rightarrow 0} \frac{\sin(\omega_c t)}{\omega_c} = \lim_{Q \rightarrow 0} \frac{t \cos(\omega_c t)}{1} = t$$
$$= \dots \delta(t) + (\omega_c^2 t - 2\omega_c) e^{-\omega_c t} \quad \checkmark$$

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9. Conclusion

- the MATLAB system has insufficient support for time domain analysis of analog filters
- a new function was developed to extend MATLAB possibilities
- our approach is based on rigorous symbolic preprocessing; impulse responses of the 1st and 2nd-order sections are derived in a symbolic form
- the described algorithm is also suitable for higher-order filters and it is accurate and robust
- the algorithm can detect the Dirac function of some impulse responses

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