

Exam in TSTE14 Analog Filters

Exam code: TEN1

Date: 2016-03-21 **Time:** 14–18

Place: TER3

Examiner: Håkan Johansson

Department: ISY

Allowed aids: Pocket calculator
Wanhammar: Tables and formulas for analog and digital filters
Söderkvist: Formler och Tabeller
Ingelstam, Rönngren, Sjöberg: Tefyma
Ekbohm: Tabeller och Formler NT
Nordling: Physics Handbook for Science and Engineering
Strid: Formler och Lexikon
Mathematical tables

Number of tasks: 7

Grading: Maximum 70 points, 30 points required to pass the exam.

Note that a **motivation/solution** is required to get the maximal number of points for a problem.

Note that 10, 8, 6, 4, or 2 points obtained at the **seminars** means that you do not have to solve Problem 1, 1(a)-(d), 1(a)-(c), 1(a)-(b), or 1(a), respectively.

Solutions: Will be published no later than three days after the exam at <http://www.commsys.isy.liu.se/en/student/kurser/TSTE14/>

Result: Available 2016-04-04

- 1
 - a. Explain the meaning of a linear and time-invariant system (filter). (2 p)
 - b. What is the difference between second-order lowpass, highpass, and bandpass sections, regarding their zero locations? (2 p)
 - c. What are the potential drawbacks of using frequency transformations when designing bandpass and bandstop filters? (2 p)
 - d. Why is it important to use low-sensitivity filter realizations? (2 p)
 - e. Explain what the gain-bandwidth product of an OP amplifier is, and why it is desired to have a large gain-bandwidth product. (2 p)

- 2 Synthesize a Causer filter that meets the following specification: $\omega_c = 2\pi$ Mrad/s, $\omega_s = 2\pi \times 2$ Mrad/s, $A_{\max} = 0.1$ dB ($\rho = 15\%$), and $A_{\min} = 50$ dB. Determine the poles and zeros, and indicate their locations in the s -plane. The filter order should not be higher than necessary. (10 p)

- 3 Realize a highpass Chebyshev-I filter that meets the following specification: $\omega_c = 2\pi \times 5$ Mrad/s, $\omega_s = 2\pi$ Mrad/s, $A_{\max} = 0.5$ dB, and $A_{\min} = 40$ dB. Use a π -type ladder structure. Determine the capacitor and inductor values assuming source and load resistors with the same value according to $R_S = R_L = 2$ k Ω . (10 p)

- 4 The transfer function of a second-order section can generally be expressed as

$$H(s) = \frac{as^2 + bs + c}{s^2 + ds + e}$$

- a. Determine a , b and c (zero or nonzero constants) so that $H(s)$ corresponds to a bandpass filter and a bandstop (notch) filter, respectively. (4 p)
- b. Assume that $a = b = 0$ and $c = e$ which makes $H(s)$ correspond to a lowpass filter with a DC gain of $|H(j0)| = 1$. When $e > d^2/2$, $|H(j\omega)|$ has a maximum value for $\omega = \omega_{peak}$, $0 < \omega_{peak} < \infty$. Determine ω_{peak} and $|H(j\omega_{peak})|$ when $e = 10$ and $d = 1$. (6 p)

- 5 a. Derive the transfer function $H(s) = V_{\text{out}}(s)/V_{\text{in}}(s)$ for the active filter below, and show that it can be written as $H(s) = H_1(s)H_2(s)$ where

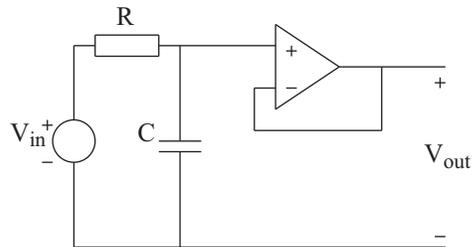
$$H_1(s) = \frac{1}{1 + sRC}, \quad H_2(s) = \frac{A(s)}{1 + A(s)}$$

with $A(s)$ being the transfer characteristics of the OP amplifier (that is, a non-ideal OP amplifier is assumed). (5 p)

- b. Determine the 3dB angular frequency for $H_1(s)$, $H_2(s)$, and $H(s)$. Use reasonable approximations if necessary. For a lowpass filter $H(s)$, the 3dB angular frequency, say ω_{3dB} , is defined by $|H(j\omega_{3dB})| = G/\sqrt{2}$ when $|H(j0)| = G$. Assume that $R = 10 \text{ k}\Omega$, $C = 100 \text{ nF}$, and

$$A(s) = \frac{A_0}{1 + \frac{s}{\omega_0}}$$

with $A_0 = 10^5$ and $\omega_0 = 2\pi \times 10 \text{ rad/s}$. (5 p)



- 6 For a lowpass-to-bandpass frequency transformation

$$S = s + \frac{\omega_I^2}{s}$$

where $\omega_I^2 = \omega_{c1}\omega_{c2} = \omega_{s1}\omega_{s2}$, show that:

- a. The frequency responses are related as $H_{BP}(j\omega) = H_{LP}(j\Omega)$, where $\Omega = \omega - \omega_I^2/\omega$. (2 p)
- b. The passband region of the lowpass filter, $\Omega \in [-\Omega_c, \Omega_c]$ is mapped to the two passband regions of the bandpass filter, $\omega \in [\omega_{c1}, \omega_{c2}]$ and $\omega \in [-\omega_{c2}, -\omega_{c1}]$ where $\Omega_c = \omega_{c2} - \omega_{c1}$. Note that the passband regions given above include the negative frequencies. (8 p)

- 7 For a doubly resistively terminated LC -filter, seen in the figure below, derive the transfer function, defined by $H(s) = V_{\text{out}}(s)/V_{\text{in}}(s)$, expressed in terms of R_S , R_L , $A(s)$, $B(s)$, $C(s)$, and $D(s)$, where $A(s)$, $B(s)$, $C(s)$, and $D(s)$ are the entries in the chain matrix of the two-port between R_S and R_L . (10 p)

