Department of Mechatronic Engineering 2019-2020 Academic Year

Fifth Year

Second Semester Examination

Marking Scheme of McE-52066 Sensors for Mechatronic System

Date: 24.3.2022 (THU) Time: 9:00 to 12:00 noon

Ques	Solution	Marks
No.		
1.(a.)	i. Pulse count, pulse frequency	
	ii. The null position	
	iii. bandwidth	
	iv. electrically insulating material	
	v. the output signals	
1.(b.)	 i. 1. Proximity Sensor ii. 4. Both 1 and 2 are correct iii. 2. Vary unequally depending on the core position iv. iv. 4. All of the above v. 2. Decrease vi. 2. Displacement of a contact slider on a resistance vii. 3. Passive transducer viii. 1.Ferromagnetic ix. 3. seeback effect x. 3. Thermocouple xi. 1. Mutual inductance xii. 1. Encoder xiii. 3. Digital xiv. 4. Acceleration xv. 1. Shaft encoder 	

Ques No.	Solution	Marks
2.(a.)	Displacement sensor Structural system (M, K, B)	10 Marks
	Solution	
	M = 10 kg, $K = 10 N/m$, and $B = 2 N/m/s$	
	m = 5 gm, b = 0.05 N/m/s	
	The equation of free motion of the simple oscillator is given by	
	$M\ddot{\pmb{y}} + B\ddot{\pmb{y}} + Ky = 0$,(i)	5 Marks
	where y denotes the displacement of the mass from the static equilibrium	
	position. This equation is of the form	
	$\ddot{y}+2\zeta\omega_{\mathrm{n}}\dot{y}+\omega_{\mathrm{n}}^{2}y=0$,(ii)	5 marks
	where ω_n = the undamped natural frequency of the oscillator and	
	ζ = the damping ratio	
	By direct comparison of (i) and (ii),	
	$\omega_{\rm n} = \sqrt{\frac{K}{M}} {\rm and} \zeta = \frac{B}{2\sqrt{MK}}.$	
	$\omega_{d} = \sqrt{1 - \zeta^{2}} \omega_{n}$ for $0 < \zeta < 1$.	

$$\omega_{\rm d} = \sqrt{\left(1 - \frac{B^2}{4MK}\right) \frac{K}{M}}.$$

$$\omega_{\rm n} = \sqrt{\frac{K}{M}} = \sqrt{\frac{10}{10}} = 1$$

$$\omega_{\rm n} = \sqrt{\frac{K}{M}} = \sqrt{\frac{10}{10}} = 1$$

$$\widetilde{\omega}_{\rm n} = \sqrt{\frac{K}{M+m}} = \sqrt{\frac{10}{10+0.005}} = \sqrt{0.9995} = 0.99975$$

Percentage error =
$$\left[\frac{\widetilde{\omega_n} - \omega_n}{\omega_n}\right] \times 100 \% = \left[\frac{0.99975 - 1}{1}\right] \times 100 \%$$

= 0.025%

Solution 2.(b)

Digital signals (or digital representation of information) have several advantages in comparison with analog signals.

- 1. Digital signals are less susceptible to noise, disturbances, or parameter variation in instruments because data can be generated, represented, transmitted, and processed as binary words consisting of bits, which possess two identifiable states.
- 2. Complex signal processing with very high accuracy and speed is possible through digital means (hardware implementation is faster than software implementation).
- 3. High reliability in a system can be achieved by minimizing analog hardware components.
- 4. Large amounts of data can be stored using compact, high-density data storage methods.
- 5. Data can be stored or maintained for very long periods of time without any

drift or disruption by adverse environmental conditions.

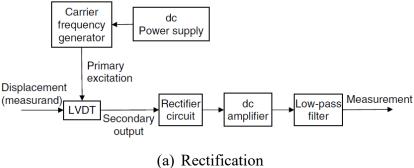
- 6. Fast data transmission is possible over long distances with no attenuation and with less dynamic delays, compared to analog signals.
- 7. Digital signals use low voltages (e.g., 0–12 V DC) and low power.
- 8. Digital devices typically have low overall cost.

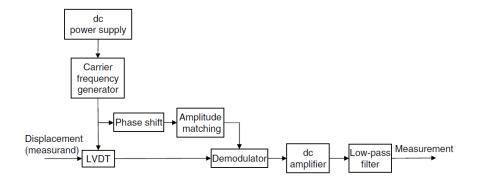
3.(a) Solution

Two methods are commonly used to interpret the crude output signal from a differential transformer: rectification and demodulation. Block diagram representations of these two procedures are given in Figure.

In the first method (rectification) the ac output from the differential transformer is rectified to obtain a dc signal. This signal is amplified and then low-pass filtered to eliminate any high-frequency noise components. The amplitude of the resulting signal provides the transducer reading. In this method, phase shift in the LVDT output has to be checked separately to determine the direction of motion.

In the second method (demodulation) the carrier frequency component is rejected from the output signal by comparing it with a phase-shifted and amplitude-adjusted version of the primary (reference) signal. Note that phase shifting is necessary because, as discussed earlier, the output signal is not in phase with the reference signal. The result is the modulating signal (proportional to x), which is subsequently amplified and filtered.

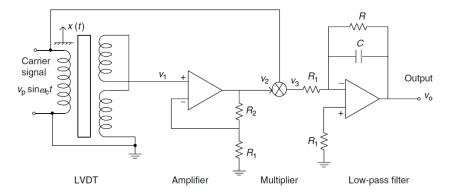




(b) Demodulation

Figure :Signal-conditioning methods for a differential transformer (a) Rectification and (b) Demodulation

3.(b) Solution



Potentials at the + and - terminals of the op-amp are nearly equal. Also, currents through these leads are nearly zero. (These are the two common assumptions used for an op-amp)Then, the current balance at node A gives,

$$\begin{aligned} \frac{V_2 - V_1}{R_2} &= \frac{V_1}{R_1} \\ V_2 &= k V_1 \\ k &= \frac{R_{1+}R_2}{R_1}, \text{ amplifier gain} \end{aligned}$$

Loss-pass filter: Since the + lead of the op-amp has approximately zero potential (ground), the voltage at point B is also approximately zero. The current balance for node B gives

$$\frac{V_3}{R_1} + \frac{V_0}{R} + c v_0^{\circ} = 0$$

$$r \frac{dV_0}{dt} + V_0 = -\frac{R}{R_1} V_3$$

Г = RC= filter time constant

The transfer function of the filter is,

$$\frac{V_0}{V_3} = -\frac{k_0}{(1+rs)}$$

the filter gain,

$$\mathbf{k}_0 = \frac{R}{R_1}$$

In the frequency domain,

$$\frac{V_0}{V_3} = -\frac{k_0}{(1+rj\omega)}$$

Finally, neglecting the phase shift in the LVDT,

$$V_1 = V_p r x(t) \sin \omega_c t$$

$$V_2 = V_p r kx(t) \sin \omega_c t$$

$$V_3 = V_p^2 r k x(t) sin^2 \omega_c t,$$

$$V_3 = \frac{V_p^2 r k}{2} [1 - \cos 2 \omega_c t]$$

The carrier signal will be filtered out by the low-pass filter with an appropriate cutoff frequency. Then,

$$V_0 = \frac{V_p^2 r k_0}{2} x(t) \# \# \#$$



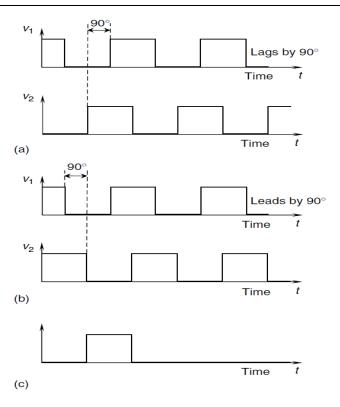


Figure: Shaped pulse signals from an incremental encoder(a) for clockwise rotation; (b) for counterclockwise rotation; (c) reference pulse signal

This logic for direction detection should be clear from Figure (.a) and Figure(b.). Another scheme can be given for direction detection. In this case, firstly detect a high level (logic high or binary 1) in signal v2 and then check whether the edge in signal v1 rises or falls during this period. As shown in Figure (a.) and Figure(b.), the following logic applies:

If rising edge in v1 when v2 is logic high) cw rotation

If falling edge in v1 when v2 is logic high) ccw rotation:

4.(b) Solution

 $\underline{Speed} = 1 \text{ rev/s}$

With 300 windows, we have 300 pulses/s

i. Pulse-counting method

Counting period =
$$\frac{1}{20}$$
 Hz = 0.05 s

Pulse count (in 0.1 s) = $300 \times 0.05 = 15$

Percentage resolution =1/15 x 100% =6.67%

ii. Pulse-timing method

At 300 pulses/s,

Pulse period = $1/300 \text{ s} = 3.33 \text{ x} 10^{-3} \text{ s}$

With a 20 MHz clock,

Clock count = $20 \times 10^6 \times 3.33 \times 10^{-3} = 66.6 \times 10^3$

Percentage resolution
$$= \frac{1}{66.6 \times 10^3} \times 100\%$$
$$= 0.0015 \%$$

Speed= 100 rev/s

N = 300 windows, pulses = 30,000 pulses/s

i. Pulse-counting method

Pulse count (in
$$0.05s$$
) = $30,000 \times 0.05 = 1500$

Percentage resolution
$$=\frac{1}{1500} \times 100\% = 0.067\%$$

ii. Pulse-timing method

At 30,000 pulses/s,

Pulse period =
$$\frac{1}{30000}$$
 s = 3.33 x 10⁻⁵

With a 10 MHz clock,

Clock count =
$$20 \times 10^6 \times 3.33 \times 10^{-5} = 666$$

Percentage resolution =
$$\frac{1}{666}$$
 x 100 % = 0.15%

Speed (rev/s)	Pulse_Counting Method(%)	Pulse_Timing Method(%)
1.0	6.67	0.0015
10	0.067	0.15

Therefore, pulse counting method is more suitable for measuring high speeds. In the pulse-timing method, the resolution degrades with speed, and hence it is more suitable for measuring low speeds.

Demodulation of the resolver

5.(a) As for differential transformers (i.e., LVDT and RVDT) transient displacement signals of a resolver can be extracted by demodulating its (modulated) outputs. As usual, this is accomplished by filtering out the carrier signal, thereby extracting the modulating signal. The two output signals v_{o1} and v_{o2} of a resolver are termed quadrature signals. Suppose that the carrier (primary) signal is $v_{ref} = v_a \sin \omega t$.

The induced quadrate signals are:

 $v_{o1} = av_a \cos \theta \sin \omega t$

 $v_{o2} = av_a \sin \theta \sin \omega t$

Multiply each quadrature signal by v_{ref} to get

$$v_{m1} = v_{o1}v_{ref} = av_a^2 \cos\theta \sin^2\omega t = \frac{1}{2}av_a^2 \cos\theta [1 - \cos2\omega t]$$

$$v_{m2} = v_{o2}v_{ref} = av_a^2 \sin\theta \sin^2\omega t = \frac{1}{2}av_a^2 \sin\theta [1 - \cos2\omega t]$$

Since the carrier frequency ω should be about 10 times the maximum frequency content of interest in the angular displacement θ , a low-pass filter can be used with a cutoff set at $\omega/10$ to remove the carrier components in v_{m1} and v_{m2} .

$$v_{f1} = \frac{1}{2}av_a^2\cos\theta.$$

This gives the demodulated outputs

$$v_{f2} = \frac{1}{2}av_{\rm a}^2\sin\theta \qquad \qquad(ii)$$

Note that Equation (i) and Equation (ii) provide both $\cos \theta$ and $\sin \theta$, and hence magnitude and sign of θ .

5.(b) Solution

Sensors can be used in a control system in several ways:

- 1. To measure the system outputs for feedback control.
- 2. To measure some types of system inputs (unknown inputs, disturbances, etc.) for feedforward control.
- 3. To measure output signals for system monitoring, diagnosis, evaluation, parameter adjustment, and supervisory control.