

Massachusetts Institute of Technology Department of Aeronautics and Astronautics Cambridge, MA 02139

Unified Engineering Spring 2005

Problem Set #13 Solutions

Spring 2005

Problem S17 Solution (Signals and Systems)

Problem Statement: Consider the quadrature modulation/demodulation system shown below. The purpose of the system is to transmit two signals, $x_1(t)$ and $x_2(t)$, over the same frequency band simultaneously. $x_1(t)$ and $x_2(t)$ are bandlimited signals, with bandwidth W. That is, their Fourier transforms $X_1(f)$ and $X_2(f)$ satisfy

$$X_1(f) = 0, \quad |f| \ge W$$
$$X_2(f) = 0, \quad |f| \ge W$$

The bandwidth is much less than the modulation frequency, f_0 . The lowpass filters shown in the diagram are ideal, with transfer function

$$L(f) = \begin{cases} 1, & |f| < W\\ 0, & |f| > W \end{cases}$$

Find the Fourier transforms of the signals $x_3(t)$, $x_4(t)$, $x_5(t)$, $x_6(t)$, $x_7(t)$, $x_8(t)$, and $x_9(t)$ in terms of $X_1(f)$ and $X_2(f)$.



Solution: Define

$$w_1(t) = \cos(2\pi f_0 t)$$

$$w_2(t) = \sin(2\pi f_0 t)$$

$$w_3(t) = 2 \cos(2\pi f_0 t)$$

$$w_4(t) = 2 \sin(2\pi f_0 t)$$

The FTs are

$$W_1(f) = \frac{1}{2}\delta(f - f_0) + \frac{1}{2}\delta(f + f_0)$$
$$W_2(f) = \frac{-j}{2}\delta(f - f_0) + \frac{j}{2}\delta(f + f_0)$$
$$W_3(f) = \delta(f - f_0) + \delta(f + f_0)$$
$$W_4(f) = -j\delta(f - f_0) + j\delta(f + f_0)$$

Therefore,

$$X_3(f) = X_1(f) * W_1(f)$$

= $\frac{1}{2}X_1(f - f_0) + \frac{1}{2}X_1(f + f_0)$

$$X_4(f) = X_2(f) * W_2(f)$$

= $\frac{-j}{2}X_2(f - f_0) + \frac{j}{2}X_2(f + f_0)$

$$X_5(f) = X_3(f) + X_4(f)$$

= $\frac{1}{2}X_1(f - f_0) + \frac{1}{2}X_1(f + f_0) + \frac{-j}{2}X_2(f - f_0) + \frac{j}{2}X_2(f + f_0)$

$$\begin{aligned} X_6(f) &= X_5(f) * W_3(f) \\ &= X_5(f - f_0) + X_5(f + f_0) \\ &= \frac{1}{2} X_1(f - 2f_0) + \frac{1}{2} X_1(f) + \frac{-j}{2} X_2(f - 2f_0) + \frac{j}{2} X_2(f) \\ &\quad + \frac{1}{2} X_1(f) + \frac{1}{2} X_1(f + 2f_0) + \frac{-j}{2} X_2(f) + \frac{j}{2} X_2(f + 2f_0) \\ &= X_1(f) + \frac{1}{2} X_1(f - 2f_0) + \frac{1}{2} X_1(f + 2f_0) \\ &\quad + \frac{-j}{2} X_2(f - 2f_0) + \frac{j}{2} X_2(f + 2f_0) \end{aligned}$$

$$\begin{aligned} X_7(f) &= X_5(f) * W_4(f) \\ &= -jX_5(f - f_0) + jX_5(f + f_0) \\ &= \frac{-j}{2}X_1(f - 2f_0) + \frac{-j}{2}X_1(f) - \frac{1}{2}X_2(f - 2f_0) + \frac{1}{2}X_2(f) \\ &\quad + \frac{j}{2}X_1(f) + \frac{j}{2}X_1(f + 2f_0) + \frac{1}{2}X_2(f) - \frac{1}{2}X_2(f + 2f_0) \\ &= X_2(f) - \frac{1}{2}X_2(f - 2f_0) - \frac{1}{2}X_2(f + 2f_0) \\ &\quad + \frac{-j}{2}X_1(f - 2f_0) + \frac{j}{2}X_1(f + 2f_0) \end{aligned}$$

Low-pass filtering $x_6(t)$ and $x_7(t)$ eliminates all but the low-frequency terms, so that

$$X_8(f) = X_1(f)$$
$$X_9(f) = X_2(f)$$

Unified Engineering II

Spring 2005

Problem S18 Solution (Signals and Systems)

Do problem 8.26 from Openheim and Willksy, Signals and Systems, reprinted below:

8.26. In Section 8.2.2, we discussed the use of an envelope detector for asynchronous demodulation of an AM signal of the form $y(t) = [x(t) + A] \cos(\omega_c t + \theta_c)$. An alternative demodulation system, which also does not require phase synchronization, but does require frequency synchronization, is shown in block diagram form in Figure P8.26. The lowpass filters both have a cutoff frequency of ω_c . The signal $y(t) = [x(t) + A] \cos(\omega_c t + \theta_c)$, with θ_c constant but unknown. The signal x(t) is band limited with $X(j\omega) = 0$, $|\omega| > \omega_M$, and with $\omega_M < \omega_c$. As we required for the use of the envelope detector, x(t) + A > 0 for all t.

Show that the system in Figure P8.26 can be used to recover x(t) from y(t) without knowledge of the modulator phase θ_c .



Figure P8.26

To begin, label the signals as shown below



From the problem statement,

$$y(t) = [x(t) + A] \cos \left(2\pi f_c t + \theta_c\right)$$

Define

$$z(t) = x(t) + A$$
$$w(t) = \cos \left(2\pi f_c t + \theta_c\right)$$

The factor w(t) can be expanded as

$$w(t) = \cos\left(2\pi f_c t + \theta_c\right) = \cos\theta_c \,\cos 2\pi f_c t - \sin\theta_c \,\sin 2\pi f_c t$$

The Fourier transform of w(t) is then given by

$$W(f) = \mathcal{F}[\cos\left(2\pi f_c t + \theta_c\right)]$$

= $\frac{1}{2}\cos\theta_c \left[\delta\left(f - f_c\right) + \delta\left(f + f_c\right)\right] - \frac{1}{2}\sin\theta_c \left[-j\delta\left(f - f_c\right) + j\delta\left(f + f_c\right)\right]$
= $\frac{1}{2}\left(\cos\theta_c + j\sin\theta_c\right)\delta\left(f - f_c\right) + \frac{1}{2}\left(\cos\theta_c - j\sin\theta_c\right)\delta\left(f + f_c\right)$

The Fourier transform of z(t) = x(t) + A is given by

$$Z(f) = \mathcal{F}[z(t)] = X(f) + A\delta(f)$$

Z(f) is bandlimited, because X(f) is, and of course the impulse function is bandlimited. So the FT of y(t) is given by the convolution

$$Y(w) = Z(f) * W(f)$$

= $\frac{1}{2} [(\cos \theta_c + j \sin \theta_c) Z(f - f_c) + (\cos \theta_c - j \sin \theta_c) Z(f + f_c)]$

Next, compute the spectra of $y_1(t)$ and $y_2(t)$. To do so, we need the spectra of $w_1(t)$ and $w_2(t)$:

$$W_1(f) = \mathcal{F}[w_1(t)] = \mathcal{F}[\cos 2\pi f_c t]$$

= $\frac{1}{2} [\delta(f - f_c) + \delta(f + f_c)]$
$$W_2(f) = \mathcal{F}[w_2(t)] = \mathcal{F}[\sin 2\pi f_c t]$$

= $\frac{1}{2} [-j\delta(f - f_c) + j\delta(f + f_c)]$

Then

$$\begin{split} Y_1(f) &= W_1(f) * Y(f) \\ &= \frac{1}{2} \left[Y(f - f_c) + Y(f - f_c) \right] \\ &= \frac{1}{4} \left[(\cos \theta_c + j \sin \theta_c) \, Z(f - 2f_c) + (\cos \theta_c - j \sin \theta_c) \, Z(f) \right] \\ &\quad + \frac{1}{4} \left[(\cos \theta_c + j \sin \theta_c) \, Z(f) + (\cos \theta_c - j \sin \theta_c) \, Z(f + 2f_c) \right] \\ &= \frac{1}{2} \cos \theta_c \, Z(f) \\ &\quad + \frac{1}{4} \left[(\cos \theta_c + j \sin \theta_c) \, Z(f - 2f_c) + (\cos \theta_c - j \sin \theta_c) \, Z(f + 2f_c) \right] \end{split}$$

Similarly,

$$\begin{aligned} Y_4(f) &= W_2(f) * Y(f) \\ &= \frac{1}{2} \left[-jY(f - f_c) + jY(f - f_c) \right] \\ &= \frac{-j}{4} \left[(\cos \theta_c + j \sin \theta_c) \, Z(f - 2f_c) + (\cos \theta_c - j \sin \theta_c) \, Z(f) \right] \\ &+ \frac{j}{4} \left[(\cos \theta_c + j \sin \theta_c) \, Z(f) + (\cos \theta_c - j \sin \theta_c) \, Z(f + 2f_c) \right] \\ &= -\frac{1}{2} \sin \theta_c \, Z(f) \\ &+ \frac{1}{4} \left[(-j \cos \theta_c + \sin \theta_c) \, Z(f - 2f_c) + (j \cos \theta_c + \sin \theta_c) \, Z(f + 2f_c) \right] \end{aligned}$$

Now, when $y_1(t)$ and $y_4(t)$ are passed through the lowpass filters, the $Z(f - 2f_c)$ and $Z(f + 2f_c)$ terms are eliminated, and the Z(f) terms are passed. Therefore,

$$Y_2(f) = \frac{1}{2} \cos \theta_c Z(f)$$
$$Y_5(f) = -\frac{1}{2} \sin \theta_c Z(f)$$

and

$$y_2(t) = \frac{1}{2} \cos \theta_c \, z(t)$$
$$y_5(t) = -\frac{1}{2} \sin \theta_c \, z(t)$$

After passing these signals through the squarers, we have

$$y_3(t) = \frac{1}{4}\cos^2\theta_c \, z^2(t)$$
$$y_6(t) = \frac{1}{4}\sin^2\theta_c \, z^2(t)$$

 $y_{7}(t)\$ is the sum of these, so that

$$y_{7}(t) = y_{3}(t) + y_{7}(t)$$

= $\frac{1}{4} \left[\cos^{2} \theta_{c} z^{2}(t) + \sin^{2} \theta_{c} z^{2}(t) \right]$
= $\frac{1}{4} z^{2}(t)$

Finally, r(t) is obtained by passing taking the square root of $y_7(t)$, so that

$$r(t) = \sqrt{z^2(t)/4}$$
$$= \frac{|z(t)|}{2}$$

if the positive root is always taken. But z(t) = x(t) + A is always positive, according to the problem statement. Therefore,

$$x(t) = 2r(t) - A$$

Unified Engineering II

Problem S19 (Signals and Systems)

Do problem 8.8 from Openheim and Willksy, *Signals and Systems*, reprinted below. Note that this system implements a type of single sideband amplitude modulation. In addition to parts (a) and (b) below, do part (c).

- **8.8.** Consider the modulation system shown in Figure P8.8. The input signal x(t) has a Fourier transform $X(j\omega)$ that is zero for $|\omega| > \omega_M$. Assuming that $\omega_c > \omega_M$, answer the following questions:
 - (a) Is y(t) guaranteed to be real if x(t) is real?
 - (b) Can x(t) be recovered from y(t)?



(c) For a representative spectrum X(f), plot the resulting spectrum Y(f). You may assume that the spectrum X(f) is real, but plot both the real and imaginary parts of Y(f). Does this system in fact produce single sideband modulation?

Solution: Label the signals in the problem as below:

- **8.8.** Consider the modulation system shown in Figure P8.8. The input signal x(t) has a Fourier transform $X(j\omega)$ that is zero for $|\omega| > \omega_M$. Assuming that $\omega_c > \omega_M$, answer the following questions:
 - (a) Is y(t) guaranteed to be real if x(t) is real?
 - (b) Can x(t) be recovered from y(t)?



The Fourier transform of x(t) is given by X(f). Then the FT of $x_1(t)$ is given by

$$X_1(f) = H(f)X(f) = \begin{cases} -jX(f), & 0 < f < f_M \\ +jX(f), & -f_M < f < 0 \\ 0, & |f| > f_M \end{cases}$$

The signal $x_2(t)$ is given by

$$x_2(t) = w_1(t)x_1(t)$$

where $w_1(t) = \cos 2\pi f_c t$. The FT of $w_1(t)$ is

$$W_1(f) = \frac{1}{2} [\delta(f - f_c) + \delta(f + f_c)]$$

The FT of $x_2(t)$ is then

$$\begin{aligned} X_2(f) &= X_1(f) * W_1(f) \\ &= \frac{1}{2} [X_1(f - f_c) + X_1(f + f_c)] \\ &= \begin{cases} -\frac{i}{2} X(f - f_c), & f_c < f < f_c + f_M \\ +\frac{i}{2} X(f - f_c), & f_c - f_M < f < f_c \\ -\frac{i}{2} X(f + f_c), & -f_c < f < -f_c + f_M \\ +\frac{i}{2} X(f + f_c), & -f_c - f_M < f < -f_c \\ 0, & \text{else} \end{cases} \end{aligned}$$

The signal $x_3(t)$ is given by

$$x_3(t) = w_2(t)x(t)$$

where $w_2(t) = \sin 2\pi f_c t$. The FT of $w_2(t)$ is

$$W_1(f) = \frac{1}{2} [-j\delta(f - f_c) + j\delta(f + f_c)]$$

The FT of $x_3(t)$ is then

$$X_{3}(f) = X(f) * W_{2}(f)$$

$$= \frac{1}{2} [-jX(f - f_{c}) + jX(f + f_{c})]$$

$$= \begin{cases} -\frac{j}{2}X(f - f_{c}), & f_{c} < f < f_{c} + f_{M} \\ -\frac{j}{2}X(f - f_{c}), & f_{c} - f_{M} < f < f_{c} \\ +\frac{j}{2}X(f + f_{c}), & -f_{c} < f < -f_{c} + f_{M} \\ +\frac{j}{2}X(f + f_{c}), & -f_{c} - f_{M} < f < -f_{c} \\ 0, & \text{else} \end{cases}$$

Finally, the FT of y(t) is given by

$$Y(f) = X_2(f) + X_3(f)$$

$$= \begin{cases} -jX(f - f_c), & f_c < f < f_c + f_M \\ 0, & f_c - f_M < f < f_c \\ 0, & -f_c < f < -f_c + f_M \\ +jX(f + f_c), & -f_c - f_M < f < -f_c \\ 0, & \text{else} \end{cases}$$

$$= \begin{cases} -jX(f - f_c), & f_c < f < f_c + f_M \\ +jX(f + f_c), & -f_c - f_M < f < -f_c \\ 0, & \text{else} \end{cases}$$

First, y(t) is guaranteed to be real if x(t) is real, because if x(t) real, X(f) has conjugate symmetry, and then Y(f) has conjugate symmetry, which implies y(t) real. Second, x(t) can be recovered from y(t) as follows. If y(t) is modulated by $2 \sin 2\pi f_c t$, the resulting signal is $z(t) = 2y(t) \sin 2\pi f_c t$, which has FT

$$Z(f) = -jY(f - f_c) + jY(f + f_c)$$

$$= \begin{cases} -X(f - 2f_c), & 2f_c < f < 2f_c + f_M \\ +X(f), & -f_M < f < 0 \\ +X(f), & 0 < f < f_M \\ -X(f + 2f_c), & -2f_c - f_M < f < -2f_c \\ 0, & \text{else} \end{cases}$$

If z(t) is then passed through a lowpass filter, with cutoff at $f = \pm f_M$, then the resulting signal is identical to x(t).





There are many ways in which to model the flow of data through a program. The short example above is one way.

The important thing to understand is that even though code may be in hundreds of different packages and procedures, these are just ways of dividing the code into smaller understandable segments. If you took all the files from one program, it would be possible to make one large data flow diagram. package My_Test_Package is
 (i) My_Integer : Integer := 1;

procedure My_Integer_Increment;

procedure My_Integer_Double_Increment;

procedure Manipulate_Integer(
current_integer_value : out integer);

end My_Test_Package;

package body My_Test_Package is

procedure My_Integer_Increment is
begin
(A) My_Integer := My_Integer + 1;
end My_Integer_Increment;

procedure My_Integer_Double_Increment is begin

(B) My_Integer := My_Integer + 2; end My_Integer_Double_Increment;

procedure Manipulate_Integer(Current_Integer_Value : out Integer) is begin

- (C) if My_Integer mod 2 = 0 then
 - (D) Current_Integer_Value := My_Integer/2; else
- (E) Current_Integer_Value := My_Integer+1; end if; end Manipulate_Integer;

end My_Test_Package;

1. with My_Test_Package; 2. 3. with Ada.Text_lo; 4. 5. procedure Demo_Data_Flow is 6. Counter : Integer :=1; My_Test_Integer : Integer; 7. 8. 9. begin 10. for I in 1..5 loop 11. if $I \mod 2 = 0$ then 12. My_Test_Package.My_Integer_Increment; 13. else 14. My_Test_Package.My_Integer_Double_Increment; 15. end if: end loop; 16. 17. 18. loop 19. exit when Counter > 10; 20. My_test_Package.Manipulate_Integer(My_Test_Integer) if My_Test_Integer mod 2 = 0 then 21. 22. Counter := Counter +2; 23. My_Test_Package.My_Integer_Increment; 24. else 25. Counter := Counter +1; My_Test_Package.My_Integer_Double_Increment; 26. 27. end if: Ada.Text_Io.Put("My Test Integer Value is"); 28. 29. Ada.Text Io.Put(Integer'Image(My Test Integer)); 30. Ada.Text_IO.New_Line; 31. Ada.Text_lo.Put(Integer'Image(Counter)); 32. Ada.Text_lo.New_Line; 33. end loop: 34. Ada.Text_Io.Put_Line("Exited Loop"); 35. end Demo_Data_Flow;

- Defs on the following lines • • 1. def My_Integer 6. def Counter 12, (A). def My_Integer 14, (B). def My_Integer 20 (D or E). def My_Test_Integer 20,29 22. def Counter 23, (A). def My_Integer 25. def Counter 26, (B). def My_Integer
 - All defs, some use 1,20 12,20 14,20 23,20 26,20 6,19 25,19
 - 22,19

- Uses on the following lines
 - 12, (A). C-use My_Integer
 - 14, (B). C-use My_Integer
 - 19. P-use Counter
 - 20. P-use My_Integer
 - 21. P-use My_Test_Integer and C-use My_Test_Integer (D or E)
 - 22. C-use Counter
 - 23, (A). C-Use My_Integer
 - 25. C-use of Counter
 - 26, (B). C-use My_Integer
 - 29. C-use My_Test_Integer
 - 31. C-use Counter

C18) $\alpha) (A + \overline{B}C) \cdot (\overline{A}C + B\overline{C}).$ ABC A+BC AC+BE X.Y 000 0 0 O 0 O 00 C 0 0 0 ()A B There are several different solutions ...

C19 a) $((A+BC) \cdot (AC+BC)) + ((AB+C) \cdot (AB+BC))$ Given AAC + ABC + BAC + ABAB + ABC + Expanded: CAB + CBC XX removed: ABC+BAC+ABC +CAB (ABO+ABC (ABC)+ ABC rewrite: BC + BC simplify IN) X 7 A+BC AC+BC AB+C AB+BC 5) WX YZ ABC \bigcirc 0 0 () 0 G 0 0 0 t 0 1 1 ()0 1 0 1 1 O 0 0 0 O $(\ni$ 0 O 0 0 0 0 O O 0 1 ()C 0 Use this for K-Mgp

C 0 A/B 0 0 0 0 10 0 BC+BC If you wonted to be very Interval in this translation: c)B otpi for your onderstonding C BC B BC for year ->| BC + BC

A simpler representation can be found using De Margon's laws to convert the expression: AB -> A+B A+B > AB Given: BC + BC Invert twice: BE + BC opply ReMargon: BC BC

C20 a) Given: $(A+\overline{B}) \cdot (\overline{A}+B) \cdot (\overline{B}+\overline{C}) \cdot (\overline{A}+C)$ expond = $(\overline{A}\overline{A} + AB + \overline{B}\overline{A} + B\overline{B}) \cdot (\overline{B}\overline{A} + \overline{B}C + \overline{C}\overline{A} + \overline{C}C)$ Export + AABA + AABC + AACA + AACC + ABBA + ABBC + ABCA + ABCC + BABA + BABC + BACA + BACO + BBBA BBBC + BBCA + BBCA + XX because (true and false) = false always? remore b) AABA + AABC + AACA + AACC + ABBA + ABBC + ABEA + ABEC + BABA + BABC + BACA + BACC + BABA + BABC + BACA + BREC result: AB + ABC + ABC simpley: AB + AB AB

c) $(A+\overline{B}) \cdot (\overline{A}+B) \cdot (\overline{B}+\overline{c}) \cdot (\overline{A}+c)$ BCA+BA+BB+EA+CWXYZ A I 0 0 6 1 0 0 1 1 K-Mgg: AB C O B

d) Many students understood ABCD as not (ABCD) many other students: ABCD as A.B.C.D D Since this was inclear both will be accepted as a solution. The answers below will represent O then D Answer for O 0000 Since we are taking the or 001 of all of these terms OL 01 Hebet that ABCD is the 10 for so many of the coves means we only med to verify the 1 last case -O 0 100 1 110 0

Using a K-Mgp for this case is trivial: C/D/00011110 A/B 00 10 This is how you would circle of you wonted SOP for m ... (but we don't wont that) Method 1:1 but for pas form... (D) 00 01 1P 10 $\frac{A/B}{OO} \left(1 \right) \left$ 10 when using a Kingp for POS form things are alittle different: you invert each unable and add them your $\begin{array}{c} AB & CD \\ \hline 1 & 1 \\ \hline \end{array} \xrightarrow{AB} \overline{CD} \xrightarrow{AB} \overline{CD} \xrightarrow{A+B+C+D} \\ \hline \hline 0 & 0 \\ \hline \end{array} \xrightarrow{O+O+O+O+O} \end{array}$

Nethod 2: This is the way Professor Lundquist tag ht you, this is the mothed you are responsible to know. 1 Whe the truth table ABCD cortier turth table for OO Intermediate steps O O Ó Sethe Ó T 1 1 @ Invert the autput result 3 Create the K-Mgp besed on F AB Ø

D Write the expression: Note F -> F = ABCD @ Convert w/ DeMargers Low F=ABCD F=A+B+C+D

Answer for 2 ABCD ABC ABCD ABCD ABCD ABCD ABCD ABCD ABC \bigcirc C I O \bigcirc -1 O I Cot tired of writing zeroes ... K-Meg: C/D A/B t D ł O

expression-BD+ABC+ABD+BCD [Mathod 1] -- converting this guy is a pain & Lets do a POS K-Mgp: 40 00 01 11 10 A/R 6 0 0 $\left(\overline{B}+\overline{D}\right)\cdot\left(\overline{A}+\overline{B}+\overline{C}\right)\cdot\left(\overline{B}+\overline{C}+D\right)\cdot\left(\overline{A}+\overline{B}+D\right)$

Method 2 This is the method dow one responsible for lenving & O write the toth table Ē ABCD L È O O See the edict much tabl O Sast Ø ((O Invert the adjust result -O Kreate K-map bood on F do Á/B

(D write the expression: Note F -> F = BD + ABC + BCD + ABD (5) Convert W De Margon's Lows F = BD + ABC+ BCD + ABD F= BD ABC + BCD + ABD $F = (\overline{B} + \overline{D}) \cdot (\overline{A} + \overline{B} + \overline{C}) \cdot (\overline{B} + \overline{C} + \overline{D}) \cdot (\overline{A} + \overline{B} + \overline{D})$ F= (B+D) • (A+B+D) • (B+E+D) • (A+B+D)