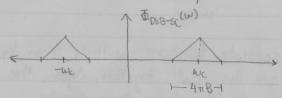
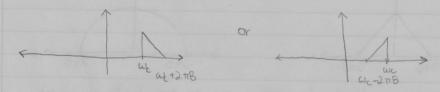
Chan and Mitran Quadrature Carrier Multiplexing

Introduction

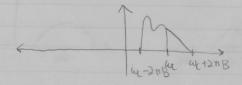
As the end of the last unit, we came up with a modulation method that is completely power-efficient. But what it lacks is spectrum efficiency.



The upper sideband and lower sidebands are simply mirror images of each other. We're sending a signal with a bandwidth of 4TB, but in order to reconstruct the baseband, we only need to send 2TB:



This way, we have 100% spectrum efficiency. Alternatively, we could send two separate signals, each one using either the upper and lower sidebands, ALSO achieving spectrum efficiency:



The first option is called Single-Sideband, which we'll cover next unit. This unit, we'll focus on the 2-in-1: quadrature carrier multiplexing.

Quadrature Carner Multiplexing

OCM looks like this:

$$\cos \omega ct$$

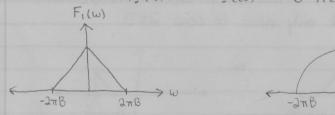
$$f_1(t) \longrightarrow \otimes \longrightarrow g(t)$$

$$f_2(t) \longrightarrow \otimes \longrightarrow g(t)$$

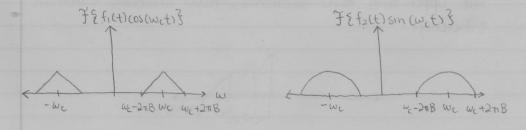
$$\sin \omega_c t$$

We take two separate functions, multiply them by sin and cos, and add the results together. Again, this is hard to visualize, so let's draw some graphs.

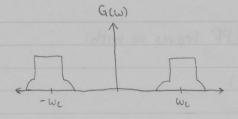
Say we have filt) <> Film, each with bandwidth fo(t) (+) Fo(w) B Hz.



Then, we multiply by either sin or cos, both of which split up the spectrum and shift them right/left by their respective frequency.



Finally, we add the results together



Since both graphs originally had bandwidths of B Hz, this result still has a band width of 2B Hz, or 47B rods.

Now, this resulting spectrum looks pretty whack. How do we recover our original signals? In engineering, we like to reuse existing technology, so let's try using the receiver from DSB-SC:

$$g(t) \longrightarrow \bigotimes \longrightarrow LPF \rightarrow d_1(t) \text{ or } d_2(t)$$

$$cos(\omega_c t) \text{ or } sin(\omega_c t)$$

$$g(t)\cos(\omega_c t) = [f_1(t)\cos(\omega_c t) + f_2(t)\sin(\omega_c t)]\cos(\omega_c t)$$

$$= f_1(t)\cos^2(\omega_c t) + f_2(t)\sin(\omega_c t)\cos(\omega_c t)$$

We can then split these terms up using trig identities.

Then, we need to pass this through a low pass filter. The first term is a low frequency and gets let through. Zwe is pretty high, so it gets rejected.

Let's do the same for sin (wct):

g(t)
$$\sin(\omega_c t) = [f_1(t)\cos(\omega_c t) + f_2(t)\sin(\omega_c t)]\sin(\omega_c t)$$

= $f_1(t)\cos(\omega_c t)\sin(\omega_c t) + f_2(t)\sin^2(\omega_c t)$
= $\frac{1}{2}f_1(t)\sin(2\omega_c t) + \frac{1}{2}f_2(t) + \frac{1}{2}f_3(t)\cos(2\omega_c t)$

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Again, passing through the LPF leaves us with:

So thankfully, we can demodulate using DSB-SC's technique, and the entire transmitter/receiver structure looks like this:

$$f_{1}(t) \longrightarrow \bigotimes \longrightarrow Cos(\omega_{c}t)$$

$$f_{2}(t) \longrightarrow \bigotimes \longrightarrow LPF \longrightarrow \frac{1}{2}f_{1}(t)$$

$$f_{2}(t) \longrightarrow \bigotimes \longrightarrow LPF \longrightarrow \frac{1}{2}f_{2}(t)$$

$$sin(\omega_{c}t)$$

$$f_{3}(t) \longrightarrow \bigotimes \longrightarrow LPF \longrightarrow \frac{1}{2}f_{2}(t)$$

Even though we've jumbled up the spectrum of fi(t) and f2(t) into g(t), we can recover our original signals EXACTLY just by multiplying by cos(wct) or sin (wct) again.

Frequency and Phase Error

Once again, we're relying on the assumption that we can produce two (or four, in this case) sinusoids that are exactly in sync.

Let's explore what happens when we have a sinusoid with a frequency error Δw and a phase error Θ .

g(t)
$$\cos[(\omega_c + \Delta \omega)t + \theta] = [f_1(t) \cos(\omega_c t) + f_2(t) \sin(\omega_c t)]$$

 $\times \cos[(\omega_c + \Delta \omega)t + \theta]$

= f₁(t) cos(wct) cos [(wc+Dw)t+0] + f₂(t) sin(wct) cos [(wc+Dw)t+0] V trig identity

= $f_1(t)$ [$\frac{1}{2}$ cos[($2wc + \Delta w$) $t + \theta$] + $\frac{1}{2}$ cos ($\Delta wt + \theta$)]

+ fa(t) [1/2 sin [(2wc+Aw)t +0] + /2 sin (Awt+D)]

= 1/2 f,(t) cos[(2wc+Aw)t+B] + 1/2 f,(t) cos (Awt+B)

+ 1/2 f2(t) SIN [(2wc + DW) t + B] + 1/2 f2(t) SIN (DW + D)

Passing this through a LPF, we can reject the 2wc+ Dw and let the Dw through.

= 1/2 filt) cos (Awt+0) + 1/2 falt) sin (Awt +0)

We multiplied by cos[(wc+sw)t+0] hoping to retrieve the entirety of our fi(t) signal, but what do we actually get?

Some distortion / beats with the cos(Awt+A). We ALSO get a completely unrelated signal fa(t). As if it wasn't important enough to get our carrier sinusoids synchronized, we basically completely fuck ourselves when using QCM.

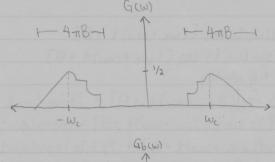
Arbitrary Spectrum

In all the modulation techniques we've looked at so far, the upper and lower sidebands are always mirror images - that is, complex conjugates.

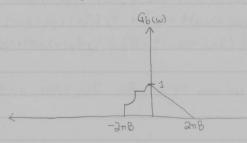
OCM allows us to create ANY arbitrary bandpass spectrum. Which really means we can make signals with any shape we want for the spectrum.*

Let's prove it.

* Not actually ANY, but any reasonable 1-to-1 shape.



Here we have some arbitrary spectrum G(w).



Next, we have Gob(w) which we get by shifting G(w) down to DC.

As such, we can define G(w) one of three ways:

This first definition is your regular multiply-by-cos shift.

In this case, G(w) is shifted to the left by we, bringing the spectrum to DC. Gb(w) is shrunk by 1/2 to accommodate for the height

Honestly, it's the same thing. It's math, guys.

Now, we can say that gb(t), the complex baseband equivalent is the time-domain function that creates Gb(w).

Okay, but so what? How does that help?

Remember that G(w) is the spectrum of some function g(t). The positive frequencies of G(w) allow us to completely reconstruct g(t) - that's what the inverse Fourier does.

So it makes sense that Gb(w) allows us to completely reconstruct gb(t).

We managed to CREATE GIO(W) by shifting things by we and modifying the amplitudes of our spectrum. In other words, by the definition

G(w) = 1/2 Gb (w-wc), w>0,

Gb(w) in combination with we COMPLETELY SPECIFY G(w). By that exact same argument, gb(t) in combination with we COMPLETELY SPECIFIES g(t).

Let's get into the math.

How do we relate $g_b(t)$ and w_c with g(t)? $g(t) \stackrel{\mathcal{F}}{\longleftrightarrow} G(w)$

as g(t) is a real-valued signal. As before, we can express G(w) in terms of Gb(w).

Since the first part is O when w is negative, and the second part is O when w is positive, we can just add them.

To get g(t) from G(w) we take the inverse Fourier.

We know \mathcal{F}^{-1} \mathcal{E} $\mathcal{G}_b(\omega-\omega)$ \mathcal{E}_b is $\mathcal{G}_b(t)$ $e^{j\omega_b t}$ from the transform properties in the first couple of units. But what happens to \mathcal{F}^{-1} \mathcal{E} \mathcal{G}_b $(-\omega-\omega)^*$ \mathcal{F}^{-2} We'll have to do this the old-fashioned way: by definition.

We'll have to integrate by substitution:

Let
$$u = -w - wc$$
 $w = -u - wc$ $du = -du$

We also need to change the bounds of integration.

$$\omega = -\infty \rightarrow -\infty = -u - wc$$

$$-\infty + wc = -u$$

$$u = \infty$$

$$W = \infty \rightarrow \infty = -u - wc$$

$$0 + wc = -u$$

$$u = -\infty$$

$$\mathcal{F}^{-1}\xi G_{b}(-w-w_{c})^{*}\mathcal{F} = \frac{1}{2\pi}\int_{-\infty}^{+\infty}G_{b}(u)^{*}e^{j(-u-w_{c})t}(-du)$$

The -du is gross, so we'll get rid of the - to flip the bounds of integration again.

$$= \frac{1}{2\pi} \int_{-\infty}^{\infty} G_b(u)^* e^{j(-u-w_c)t} du$$

$$= \frac{1}{2\pi} \int_{-\infty}^{\infty} G_b(u)^* e^{-jut} e^{-jw_ct} du$$

$$= \frac{e^{-jw_ct}}{2\pi} \int_{-\infty}^{\infty} G_b(u)^* e^{-jut} du$$

$$= \frac{e^{-jw_ct}}{2\pi} \int_{-\infty}^{\infty} G_b(u)^* e^{-jut} du$$

We'll pull out the complex conjugate

$$= \frac{-j\omega_c t}{2\pi} \left(\int_{-\infty}^{\infty} G_b(u) e^{-jut} du \right) *$$

literally just $2\pi \times$ the inverse

Fourier fransform

= $e^{-j\omega_c t} \left[2\pi g_b(t) \right]^*$ = $q_b(t)^* e^{-j\omega_c t}$

So overall,

Already, this might look vaguely familiar. Let's extract the complex conjugate.

Remember that anything + its complex conjugate is 2 × Re & thing 3.

Alright, we're about to get FUNKY. Check these two expansions.

$$g_{blt}$$
 = Re & g_{blt} 3 + $j Im & g_{blt} 3
$$e^{jw_{c}t} = cos(w_{c}t) + j sin(w_{c}t)$$$

Everything is a sum of its real and imaginary parts. The exponent has been expanded to its sum of sinusoids.

Let's cross-multiply.

The purely real terms are the ones at the top and bottom, so we keep them and toss the rest.

If this doesn't look familiar, you haven't been paying enough attention

This is the EXACT definition for QCM. We've proved that if we want to generate some arbitrary passbond signal Gb(w), we just need to choose filt) and folt):

$$f_1(t) = Re \{ g_b(t) \}$$

 $f_2(t) = -I_m \{ g_b(t) \}$

f.(t) is known as the "in phase" component and fa(t) is known as the "quadrature" component.

Let's summarize

Summary

- 1) g(t) is some real-valued signal that occupies the frequency band [ωc 2π B, ωc + 2π B], a bandwidth of 4πB.
- 2) Gbcw) is the positive shape of the spectrum we want to create. It is shifted down to DC and as such, occupies F2πB, 2πB]. This bandwidth is 2πB (positive only).
- 3) gb(t) is the complex baseband equivalent model (or complex envelope) of Gb(w). Obviously, as they're the same signal, gb(t)'s bandwidth is 2718.
- 4) We select filt) as Re & gb(t)3. We select falt) as -Im & gb(t)3.

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5) We can then create Gb(w) using QCM by generating g(t) = f,(t) cos (wct) + fo(t) sin (wct).

EVERY real-valved passband signal can be generated using QCM. It can do the same thing as every technique out there - the only question left is whether it's the simplest method for our problem.

For example - what if we don't have two messages to send, and we don't need to construct some arbitrary spectrum?

Coming up next - single side band.