1.12 Capacitors

A capacitor (Fig. 1.27) (the old-fashioned name was condenser) is a device that has two wires sticking out of it and has the property

\[ Q = CV \]

A capacitor of \( C \) farads with \( V \) volts across its terminals has \( Q \) coulombs of stored charge on one plate, and \(-Q\) on the other.

To a first approximation, capacitors are devices that might be considered simply frequency-dependent resistors. They allow you to make frequency-dependent voltage dividers, for instance. For some applications (bypass, coupling) this is
almost all you need to know, but for other applications (filtering, energy storage, resonant circuits) a deeper understanding is needed. For example, capacitors cannot dissipate power, even though current can flow through them, because the voltage and current are 90° out of phase.

Taking the derivative of the defining equation above (see Appendix B), you get

\[ I = C \frac{dV}{dt} \]

So a capacitor is more complicated than a resistor; the current is not simply proportional to the voltage, but rather to the rate of change of voltage. If you change the voltage across a farad by 1 volt per second, you are supplying an amp. Conversely, if you supply an amp, its voltage changes by 1 volt per second. A farad is very large, and you usually deal in microfarads (\( \mu F \)) or picofarads (pF). (To make matters confusing to the uninitiated, the units are often omitted on capacitor values specified in schematic diagrams. You have to figure it out from the context.) For instance, if you supply a current of 1mA to 1\( \mu F \), the voltage will rise at 1000 volts per second. A 10ms pulse of this current will increase the voltage across the capacitor by 10 volts (Fig. 1.28).

Capacitors come in an amazing variety of shapes and sizes; with time, you will come to recognize their more common incarnations. The basic construction is simply two conductors near each other (but not touching); in fact, the simplest capacitors are just that. For greater capacitance, you need more area and closer spacing; the usual approach is to plate some conductor onto a thin insulating material (called a dielectric), for instance, aluminized Mylar film rolled up into a small cylindrical configuration. Other popular types are thin ceramic wafers (disc ceramics), metal foils with oxide insulators (electrolytics), and metallized mica. Each of these types has unique properties; for a brief rundown, see the box on capacitors. In general, ceramic and Mylar types are used for most noncritical circuit applications; tantalum capacitors are used where greater capacitance is needed, and electrolytics are used for power-supply filtering.

**Capacitors in parallel and series**

The capacitance of several capacitors in parallel is the sum of their individual capacitances. This is easy to see: Put voltage \( V \) across the parallel combination; then

\[
C_{total}V = Q_{total} = Q_1 + Q_2 + Q_3 + \ldots \\
= C_1V + C_2V + C_3V + \ldots \\
+ (C_1 + C_2 + C_3 + \ldots)V \\
\]

or

\[
C_{total} = C_1 + C_2 + C_3 + \ldots \\
\]

For capacitors in series, the formula is like that for resistors in parallel:

\[
C_{total} = \frac{1}{\frac{1}{C_1} + \frac{1}{C_2} + \frac{1}{C_3} + \ldots} \\
\]

or (two capacitors only)

\[
C_{total} = \frac{C_1C_2}{C_1 + C_2} \\
\]

![Figure 1.28](image)

Figure 1.28. The voltage across a capacitor changes when a current flows through it.
CAPACITORS

There is wide variety among the capacitor types available. This is a quickie guide to point out their major advantages and disadvantages. Our judgments should be considered somewhat subjective:

<table>
<thead>
<tr>
<th>Type</th>
<th>Capacitance range</th>
<th>Maximum voltage</th>
<th>Accuracy</th>
<th>Temperature stability</th>
<th>Leakage</th>
<th>Comments</th>
</tr>
</thead>
<tbody>
<tr>
<td>Mica</td>
<td>1pF–0.01μF</td>
<td>100–600</td>
<td>Good</td>
<td>Good</td>
<td></td>
<td>Excellent; good at RF</td>
</tr>
<tr>
<td>Tubular ceramic</td>
<td>0.5pF–100pF</td>
<td>100–600</td>
<td>Selectable</td>
<td></td>
<td></td>
<td>Several tempco (including zero)</td>
</tr>
<tr>
<td>Ceramic</td>
<td>10pF–1μF</td>
<td>50–30,000</td>
<td>Poor</td>
<td>Poor</td>
<td>Moderate</td>
<td>Small, inexpensive, very popular</td>
</tr>
<tr>
<td>Polyester (Mylar)</td>
<td>0.001μF–50μF</td>
<td>50–600</td>
<td>Good</td>
<td>Poor</td>
<td>Good</td>
<td>Inexpensive, good, popular</td>
</tr>
<tr>
<td>Polystyrene</td>
<td>10pF–2.7μF</td>
<td>100–600</td>
<td>Excellent</td>
<td>Good</td>
<td>Excellent</td>
<td>High quality, large; signal filters</td>
</tr>
<tr>
<td>Polycarbonate</td>
<td>100pF–30μF</td>
<td>50–800</td>
<td>Excellent</td>
<td>Excellent</td>
<td>Good</td>
<td>High quality, small</td>
</tr>
<tr>
<td>Polypropylene</td>
<td>100pF–50μF</td>
<td>100–800</td>
<td>Excellent</td>
<td>Good</td>
<td>Excellent</td>
<td>High quality, low dielectric absorption</td>
</tr>
<tr>
<td>Teflon</td>
<td>1000pF–2μF</td>
<td>50–200</td>
<td>Excellent</td>
<td>Best</td>
<td>Best</td>
<td>High quality, lowest dielectric absorption</td>
</tr>
<tr>
<td>Glass</td>
<td>10pF–1000pF</td>
<td>100–600</td>
<td>Good</td>
<td></td>
<td>Excellent</td>
<td>Long-term stability</td>
</tr>
<tr>
<td>Porcelain</td>
<td>100pF–0.1μF</td>
<td>50–400</td>
<td>Good</td>
<td>Good</td>
<td>Good</td>
<td>Good long-term stability</td>
</tr>
<tr>
<td>Tantalum</td>
<td>0.1μF–500μF</td>
<td>6–100</td>
<td>Poor</td>
<td>Poor</td>
<td></td>
<td>High capacitance; polarized, small; low inductance</td>
</tr>
<tr>
<td>Electrolytic</td>
<td>0.1μF–1.6F</td>
<td>3–600</td>
<td>Terrible</td>
<td>Ghastly</td>
<td>Awful</td>
<td>Power-supply filters; polarized; short life</td>
</tr>
<tr>
<td>Double layer</td>
<td>0.1F–10F</td>
<td>1.5–6</td>
<td>Poor</td>
<td>Poor</td>
<td>Good</td>
<td>Memory backup; high series resistance</td>
</tr>
<tr>
<td>Oil</td>
<td>0.1μF–20μF</td>
<td>200–10,000</td>
<td>Good</td>
<td></td>
<td></td>
<td>High-voltage filters; large, long life</td>
</tr>
<tr>
<td>Vacuum</td>
<td>1pF–5000pF</td>
<td>2000–36,000</td>
<td>Excellent</td>
<td></td>
<td></td>
<td>Transmitters</td>
</tr>
</tbody>
</table>
EXERCISE 1.12
Derive the formula for the capacitance of two capacitors in series. Hint: Because there is no external connection to the point where the two capacitors are connected together, they must have equal stored charges.

The current that flows in a capacitor during charging \((I = C\frac{dV}{dt})\) has some unusual features. Unlike resistive current, it’s not proportional to voltage, but rather to the rate of change (the “time derivative”) of voltage. Furthermore, unlike the situation in a resistor, the power \((V \times I)\) associated with capacitive current is not turned into heat, but is stored as energy in the capacitor’s internal electric field. You get all that energy back when you discharge the capacitor. We’ll see another way to look at these curious properties when we talk about reactance, beginning in Section 1.18.

1.13 RC circuits: \(V\) and \(I\) versus time

When dealing with ac circuits (or, in general, any circuits that have changing voltages and currents), there are two possible approaches. You can talk about \(V\) and \(I\) versus time, or you can talk about amplitude versus signal frequency. Both approaches have their merits, and you find yourself switching back and forth according to which description is most convenient in each situation. We will begin our study of ac circuits in the time domain. Beginning with Section 1.18, we will tackle the frequency domain.

What are some of the features of circuits with capacitors? To answer this question, let’s begin with the simple RC circuit (Fig. 1.29). Application of the capacitor rules gives
\[
C \frac{dV}{dt} = I = -\frac{V}{R}
\]

This is a differential equation, and its solution is
\[
V = Ae^{-t/RC}
\]

So a charged capacitor placed across a resistor will discharge as in Figure 1.30.

Figure 1.29

Figure 1.30. RC discharge waveform.

Time constant

The product \(RC\) is called the time constant of the circuit. For \(R\) in ohms and \(C\) in farads, the product \(RC\) is in seconds. A microfarad across 1.0k has a time constant of 1ms; if the capacitor is initially charged to 1.0 volt, the initial current is 1.0mA.

Figure 1.31

Figure 1.31 shows a slightly different circuit. At time \(t = 0\), someone connects the battery. The equation for the circuit is then
\[
I = C \frac{dV}{dt} = \frac{V_i - V}{R}
\]

with the solution
\[
V = V_i + Ae^{-t/RC}
\]
(Please don't worry if you can't follow the mathematics. What we are doing is getting some important results, which you should remember. Later we will use the results often, with no further need for the mathematics used to derive them.) The constant $A$ is determined by initial conditions (Fig. 1.32): $V = 0$ at $t = 0$; therefore, $A = -V_i$, and

$$V = V_i (1 - e^{-t/RC})$$

![Figure 1.32](image)

**Figure 1.32**

$$V(t) = \frac{1}{RC} \int_{-\infty}^{t} V_i(\tau)e^{-(t-\tau)/RC} d\tau$$

That is, the $RC$ circuit averages past history at the input with a weighting factor $e^{-\Delta t/RC}$.

In practice, you seldom ask this question. Instead, you deal in the **frequency domain** and ask how much of each frequency component present in the input gets through. We will get to this important topic soon (Section 1.18). Before we do, though, there are a few other interesting circuits we can analyze simply with this time-domain approach.

![Figure 1.33](image)

**Figure 1.33.** Output (top waveform) across a capacitor, when driven by square waves through a resistor.

**Decay to equilibrium**

Eventually (when $t \gg RC$), $V$ reaches $V_i$. (Presenting the "$5RC$ rule of thumb": a capacitor charges or decays to within 1% of its final value in 5 time constants.) If we then change $V_0$ to some other value (say, 0), $V$ will decay toward that new value with an exponential $e^{-t/RC}$. For example, a square-wave input for $V_0$ will produce the output shown in Figure 1.33.

**Exercise 1.13**

Show that the rise time (the time required to go from 10% to 90% of its final value) of this signal is $2.2RC$.

You might ask the obvious next question: What about $V(t)$ for arbitrary $V_i(t)$? The solution involves an inhomogeneous differential equation and can be solved by standard methods (which are, however, beyond the scope of this book). You would find

$$V(t) = \frac{1}{RC} \int_{-\infty}^{t} V_i(\tau)e^{-(t-\tau)/RC} d\tau$$

**Simplification by Thévenin equivalents**

We could go ahead and analyze more complicated circuits by similar methods, writing down the differential equations and trying to find solutions. For most purposes it simply isn't worth it. This is as complicated an $RC$ circuit as we will need. Many other circuits can be reduced to it (e.g., Fig. 1.34). By just using the Thévenin equivalent of the voltage divider formed by $R_1$ and $R_2$, you can find the output...
$V(t)$ produced by a step input for $V_0$.

**EXERCISE 1.14**

$R_1 = R_2 = 10k$, and $C = 0.1\mu F$ in the circuit shown in Figure 1.34. Find $V(t)$ and sketch it.

![Figure 1.35](image)

Figure 1.35. Producing a delayed digital waveform with the help of an $RC$.

**Example: time-delay circuit**

We have already mentioned logic levels, the voltages that digital circuits live on. Figure 1.35 shows an application of capacitors to produce a delayed pulse. The triangular symbols are "CMOS buffers." They give a HIGH output if the input is HIGH (more than one-half the dc power-supply voltage used to power them), and vice versa. The first buffer provides a replica of the input signal, but with low source resistance, and prevents input loading by the $RC$ (recall our earlier discussion of circuit loading in Section 1.05). The $RC$ output has the characteristic decays and causes the output buffer to switch $10\mu s$ after the input transitions (an $RC$ reaches $50\%$ output in $0.7RC$). In an actual application you would have to consider the effect of the buffer input threshold deviating from one-half the supply voltage, which would alter the delay and change the output pulse width. Such a circuit is sometimes used to delay a pulse so that something else can happen first. In designing circuits you try not to rely on tricks like this, but they’re occasionally handy.

**1.14 Differentiators**

Look at the circuit in Figure 1.36. The voltage across $C$ is $V_{in} - V$, so

$$I = C \frac{d}{dt}(V_{in} - V) = \frac{V}{R}$$

If we choose $R$ and $C$ small enough so that $dV/dt \ll dV_{in}/dt$, then

$$C \frac{dV_{in}}{dt} \approx \frac{V}{R}$$

or

$$V(t) = RC \frac{d}{dt} V_{in}(t)$$

That is, we get an output proportional to the rate of change of the input waveform.

![Figure 1.36](image)

Figure 1.36.

To keep $dV/dt \ll dV_{in}/dt$, we make the product $RC$ small, taking care not to "load" the input by making $R$ too small (at the transition the change in voltage across the capacitor is zero, so $R$ is the load seen by the input). We will have a better criterion for this when we look at things in the frequency domain. If you drive this circuit with a square wave, the output will be as shown in Figure 1.37.
a missing resistor termination on your signal line. If not, you must either reduce the source resistance of the signal line or find a way to reduce capacitive coupling from the offending square wave. The second case is typical of what you might see when you look at a square wave, but have a broken connection somewhere, usually at the scope probe. The very small capacitance of the broken connection combines with the scope input resistance to form a differentiator. Knowing that you've got a differentiated "something" can help you find the trouble and eliminate it.

Difficultiators are handy for detecting leading edges and trailing edges in pulse signals, and in digital circuitry you sometimes see things like those depicted in Figure 1.38. The \( RC \) differentiator generates spikes at the transitions of the input signal, and the output buffer converts the spikes to short square-topped pulses. In practice, the negative spike will be small because of a diode (a handy device discussed in Section 1.25) built into the buffer.

**Unintentional capacitive coupling**

Difficultiators sometimes crop up unexpectedly, in situations where they're not welcome. You may see signals like those shown in Figure 1.39. The first case is caused by a square wave somewhere in the circuit coupling capacitively to the signal line you're looking at; that might indicate

1.15 Integrators

Take a look at the circuit in Figure 1.40. The voltage across \( R \) is \( V_{in} - V \), so

\[
I = C \frac{dV}{dt} = \frac{V_{in} - V}{R}
\]

If we manage to keep \( V \ll V_{in} \), by keeping the product \( RC \) large, then

\[
C \frac{dV}{dt} \approx \frac{V_{in}}{R}
\]

or

\[
V(t) = \frac{1}{RC} \int_{t} V_{in}(t) \, dt + \text{constant}
\]

We have a circuit that performs the integral over time of an input signal! You can
CAPACITORS AND AC CIRCUITS

1.15 Integrators

see how the approximation works for a square-wave input: $V(t)$ is then the exponential charging curve we saw earlier (Fig. 1.41). The first part of the exponential is a ramp, the integral of a constant; as we increase the time constant $RC$, we pick off a smaller part of the exponential, i.e., a better approximation to a perfect ramp.

![Figure 1.41](image)

Note that the condition $V \ll V_{in}$ is just the same as saying that $I$ is proportional to $V_{in}$. If we had as input a current $I(t)$, rather than a voltage, we would have an exact integrator. A large voltage across a large resistance approximates a current source and, in fact, is frequently used as one.

Later, when we get to operational amplifiers and feedback, we will be able to build integrators without the restriction $V_{out} \ll V_{in}$. They will work over large frequency and voltage ranges with negligible error.

The integrator is used extensively in analog computation. It is a useful subcircuit that finds application in control systems, feedback, analog/digital conversion, and waveform generation.

**Ramp generators**

At this point it is easy to understand how a ramp generator works. This nice circuit is extremely useful, for example in timing circuits, waveform and function generators, oscilloscope sweep circuits, and analog/digital conversion circuitry. The circuit uses a constant current to charge a capacitor (Fig. 1.42). From the capacitor equation $I = C(dV/dt)$, you get $V(t) = (I/C)t$. The output waveform is as shown in Figure 1.43. The ramp stops when the current source runs out of voltage,” i.e., reaches the limit of its compliance. The curve for a simple $RC$, with the resistor tied to a voltage source equal to the compliance of the current source, and with $R$ chosen so that the current at zero output voltage is the same as that of the current source, is also drawn for comparison. (Real current sources generally have output compliances limited by the power-supply volages used in making them, so the comparison is realistic.) In the next chapter, which deals with transistors, we will design some current sources, with some refinements to follow in the chapters on operational amplifiers (op-amps) and field-effect transistors (FETs). Exciting things to look forward to!

![Figure 1.42](image)

EXERCISE 1.15

A current of 1mA charges a $\mu$F capacitor. How long does it take the ramp to reach 10 volts?
INDUCTORS AND TRANSFORMERS

1.16 Inductors

If you understand capacitors, you won't have any trouble with inductors (Fig. 1.44). They're closely related to capacitors; the rate of current change in an inductor depends on the voltage applied across it, whereas the rate of voltage change in a capacitor depends on the current through it. The defining equation for an inductor is

\[ V = L \frac{dI}{dt} \]

where \( L \) is called the inductance and is measured in henrys (or mH, \( \mu \)H, etc.). Putting a voltage across an inductor causes the current to rise as a ramp (for a capacitor, supplying a constant current causes the voltage to rise as a ramp); 1 volt across 1 henry produces a current that increases at 1 amp per second.

Figure 1.44. Inductor.

As with capacitive current, inductive current is not simply proportional to voltage. Furthermore, unlike the situation in a resistor, the power associated with inductive current (\( V \) times \( I \)) is not turned into heat, but is stored as energy in the inductor's magnetic field. You get all that energy back when you interrupt the inductor's current.

The symbol for an inductor looks like a coil of wire; that's because, in its simplest form, that's all it is. Variations include coils wound on various core materials, the most popular being iron (or iron alloys, laminations, or powder) and ferrite, a black, nonconductive, brittle magnetic material. These are all ploys to multiply the inductance of a given coil by the "permeability" of the core material. The core may be in the shape of a rod, a toroid (doughnut), or even more bizarre shapes, such as a "pot core" (which has to be seen to be understood; the best description we can think of is a doughnut mold split horizontally in half, if doughnuts were made in molds).

Inductors find heavy use in radiofrequency (RF) circuits, serving as RF "chokes" and as parts of tuned circuits (see Chapter 13). A pair of closely coupled inductors forms the interesting object known as a transformer. We will talk briefly about them in the next section.

An inductor is, in a real sense, the opposite of a capacitor. You will see how that works out in the next few sections of this chapter, which deal with the important subject of impedance.

1.17 Transformers

A transformer is a device consisting of two closely coupled coils (called primary and secondary). An ac voltage applied to the primary appears across the secondary, with a voltage multiplication proportional to the turns ratio of the transformer and a current multiplication inversely proportional to the turns ratio. Power is conserved. Figure 1.45 shows the circuit symbol for a laminated-core transformer (the kind used for 60Hz ac power conversion).

Figure 1.45. Transformer.

Transformers are quite efficient (output power is very nearly equal to input power); thus, a step-up transformer gives higher voltage at lower current. Jumping ahead for a moment, a transformer of turns ratio \( n \) increases the impedance by \( n^2 \). There is very little primary current if the secondary is unloaded.

Transformers serve two important functions in electronic instruments: They
change the ac line voltage to a useful (usually lower) value that can be used by the circuit, and they “isolate” the electronic device from actual connection to the power line, because the windings of a transformer are electrically insulated from each other. Power transformers (meant for use from the 110V power line) come in an enormous variety of secondary voltages and currents: outputs as low as 1 volt or so up to several thousand volts, current ratings from a few milliamps to hundreds of amps. Typical transformers for use in electronic instruments might have secondary voltages from 10 to 50 volts, with current ratings of 0.1 to 5 amps or so.

Transformers for use at audiofrequencies and radiofrequencies are also available. At radiofrequencies you sometimes use tuned transformers, if only a narrow range of frequencies is present. There is also an interesting class of transmission-line transformer that we will discuss briefly in Section 13.10. In general, transformers for use at high frequencies must use special core materials or construction to minimize core losses, whereas low-frequency transformers (e.g., power transformers) are burdened instead by large and heavy cores. The two kinds of transformers are in general not interchangeable.

**IMPEDANCE AND REACTANCE**

Warning: This section is somewhat mathematical; you may wish to skip over the mathematics, but be sure to pay attention to the results and graphs.

Circuits with capacitors and inductors are more complicated than the resistive circuits we talked about earlier, in that their behavior depends on frequency: A “voltage divider” containing a capacitor or inductor will have a frequency-dependent division ratio. In addition, circuits containing these components (known collectively as reactive components) “corrupt” input waveforms such as square waves, as we just saw.

However, both capacitors and inductors are linear devices, meaning that the amplitude of the output waveform, whatever its shape, increases exactly in proportion to the input waveform’s amplitude. This linearity has many consequences, the most important of which is probably the following: The output of a linear circuit, driven with a sine wave at some frequency f, is itself a sine wave at the same frequency (with, at most, changed amplitude and phase).

Because of this remarkable property of circuits containing resistors, capacitors, and inductors (and, later, linear amplifiers), it is particularly convenient to analyze any such circuit by asking how the output voltage (amplitude and phase) depends on the input voltage, for sine-wave input at a single frequency, even though this may not be the intended use. A graph of the resulting frequency response, in which the ratio of output to input is plotted for each sine-wave frequency, is useful for thinking about many kinds of waveforms. As an example, a certain “boom-box” loudspeaker might have the frequency response shown in Figure 1.46, where the “output” in this case is of course sound pressure, not voltage. It is desirable for a speaker to have a “flat” response, meaning that the graph of sound pressure versus frequency is constant over the band of audible frequencies. In this case the speaker’s deficiencies can be corrected by introducing a passive filter with the inverse response (as shown) into the amplifiers of the radio.

As we will see, it is possible to generalize Ohm’s law, replacing the word “resistance” with “impedance,” in order to describe any circuit containing these linear passive devices (resistors, capacitors, and inductors). You could think of the subject of impedance and reactance as Ohm’s law for circuits that include capacitors and inductors. Some important terminology: Impedance is the “generalized resistance”; inductors
and capacitors have reactance (they are "reactive"); resistors have resistance (they are "resistive"). In other words, impedance = resistance + reactance (more about this later). However, you'll see statements like "the impedance of the capacitor at this frequency is ..." The reason you don't have to use the word "reactance" in such a case is that impedance covers everything. In fact, you frequently use the word "impedance" even when you know it's a resistance you're talking about; you say "the source impedance" or "the output impedance" when you mean the Thévenin equivalent resistance of some source. The same holds for "input impedance."

In all that follows, we will be talking about circuits driven by sine waves at a single frequency. Analysis of circuits driven by complicated waveforms is more elaborate, involving the methods we used earlier (differential equations) or decomposition of the waveform into sine waves (Fourier analysis). Fortunately, these methods are seldom necessary.

1.18 Frequency analysis of reactive circuits

Let's start by looking at a capacitor driven by a sine-wave voltage source (Fig. 1.47). The current is

\[ I(t) = C \frac{dV}{dt} = C \omega V_0 \cos \omega t \]

i.e., a current of amplitude \( I \), with the phase leading the input voltage by 90°. If we consider amplitudes only, and disregard phases, the current is

\[ I = \frac{V}{1/\omega C} \]

(Recall that \( \omega = 2\pi f \).) It behaves like a frequency-dependent resistance \( R = 1/\omega C \), but in addition the current is 90° out of phase with the voltage (Fig. 1.48).
elements (resistors, capacitors, inductors), the magnitudes of the currents everywhere in the circuit are still proportional to the magnitude of the driving voltage, so we might hope to find some generalization of voltage, current, and resistance in order to rescue Ohm's law. Obviously a single number won't suffice to specify the current, say, at some point in the circuit, because we must somehow have information about both the magnitude and phase shift.

Although we can imagine specifying the magnitudes and phase shifts of voltages and currents at any point in the circuit by writing them out explicitly, e.g., \( V(t) = 23.7 \sin(377t + 0.38) \), it turns out that our requirements can be met more simply by using the algebra of complex numbers to \textit{represent} voltages and currents. Then we can simply add or subtract the complex number representations, rather than laboriously having to add or subtract the actual sinusoidal functions of time themselves. Because the actual voltages and currents are real quantities that vary with time, we must develop a rule for converting from actual quantities to their representations, and vice versa. Recalling once again that we are talking about a single sine-wave frequency, \( \omega \), we agree to use the following rules:

1. Voltages and currents are \textit{represented} by the complex quantities \( V \) and \( I \). The voltage \( V_0 \cos(\omega t + \phi) \) is to be represented by the complex number \( V_0 e^{j\phi} \). Recall that \( e^{j\theta} = \cos \theta + j \sin \theta \), where \( j = \sqrt{-1} \).

2. Actual voltages and currents are obtained by multiplying their complex number representations by \( e^{j\omega t} \) and then taking the real part: \( V(t) = \Re(e^{j\omega t} V_0 e^{j\phi}) \), \( I(t) = \Re(e^{j\omega t} I_0 e^{j\phi}) \).

In other words,

\[
\begin{align*}
\text{circuit voltage versus time} & \quad \Rightarrow \quad \text{complex number representation} \\
V_0 \cos(\omega t + \phi) & \quad \Rightarrow \quad V_0 e^{j\phi} = a + jb \\
\text{multiply by } e^{j\omega t} & \quad \Rightarrow \quad V_0 e^{j\phi} e^{j\omega t} \\
& \quad \Rightarrow \quad V(t) = \Re(e^{j\omega t} V_0 e^{j\phi}) \\
\end{align*}
\]
(In electronics, the symbol $j$ is used instead of $i$ in the exponential in order to avoid confusion with the symbol $i$ meaning current.) Thus, in the general case the actual voltages and currents are given by

$$V(t) = \Re(e^{j\omega t})$$
$$I(t) = \Re(e^{j\omega t})$$

For example, a voltage whose complex representation is

$$V = 5j$$

corresponds to a (real) voltage versus time of

$$V(t) = \Re[5j \cos \omega t + 5j(j) \sin \omega t]$$
$$= -5 \sin \omega t \text{ volts}$$

**Reactance of capacitors and inductors**

With this convention we can apply complex Ohm’s law to circuits containing capacitors and inductors, just as for resistors, once we know the reactance of a capacitor or inductor. Let’s find out what these are. We have

$$V(t) = \Re(e^{j\omega t})$$

For a capacitor, using $I = C(dV/dt)$, we obtain

$$I(t) = -V_0C \omega \sin \omega t = \Re\left(\frac{V_0e^{j\omega t}}{-j/\omega C}\right)$$
$$= \Re\left(\frac{V_0e^{j\omega t}}{X_C}\right)$$

i.e., for a capacitor

$$X_C = -j/\omega C$$

$X_C$ is the reactance of a capacitor at frequency $\omega$. As an example a $1\mu F$ capacitor has a reactance of $-2653j$ ohms at 60Hz and a reactance of $-0.16j$ ohms at 1MHz. Its reactance at dc is infinite.

If we did a similar analysis for an inductor, we would find

$$X_L = j\omega L$$

A circuit containing only capacitors and inductors always has a purely imaginary impedance, meaning that the voltage and current are always $90^\circ$ out of phase – it is purely reactive. When the circuit contains resistors, there is also a real part to the impedance. The term “reactance” in that case means the imaginary part only.

**Ohm’s law generalized**

With these conventions for representing voltages and currents, Ohm’s law takes a simple form. It reads simply

$$I = V/Z$$
$$V = IZ$$

where the voltage represented by $V$ is applied across a circuit of impedance $Z$, giving a current represented by $I$. The complex impedance of devices in series or parallel obeys the same rules as resistance:

$$Z = Z_1 + Z_2 + Z_3 + \cdots \quad \text{(series)}$$
$$Z = \frac{1}{\frac{1}{Z_1} + \frac{1}{Z_2} + \frac{1}{Z_3} + \cdots} \quad \text{(parallel)}$$

Finally, for completeness we summarize here the formulas for the impedance of resistors, capacitors, and inductors:

$$Z_R = R \quad \text{(resistor)}$$
$$Z_C = -j/\omega C = 1/j\omega C \quad \text{(capacitor)}$$
$$Z_L = j\omega L \quad \text{(inductor)}$$

With these rules we can analyze many ac circuits by the same general methods we used in handling dc circuits, i.e., application of the series and parallel formulas and Ohm’s law. Our results for circuits such as voltage dividers will look nearly the same as before. For multiply connected
networks we may have to use Kirchhoff's laws, just as with dc circuits, in this case using the complex representations for \( V \) and \( I \): The sum of the (complex) voltage drops around a closed loop is zero, and the sum of the (complex) currents into a point is zero. The latter rule implies, as with dc circuits, that the (complex) current in a series circuit is the same everywhere.

**EXERCISE 1.16**

Use the preceding rules for the impedance of devices in parallel and in series to derive the formulas (Section 1.12) for the capacitance of two capacitors (a) in parallel and (b) in series. Hint: In each case, let the individual capacitors have capacitances \( C_1 \) and \( C_2 \). Write down the impedance of the parallel or series combination; then equate it to the impedance of a capacitor with capacitance \( C \). Find \( C \).

Let's try out these techniques on the simplest circuit imaginable, an ac voltage applied across a capacitor, which we considered just previously. Then, after a brief look at power in reactive circuits (to finish laying the groundwork), we'll analyze some simple but extremely important and useful \( RC \) filter circuits.

Imagine putting a 1\( \mu \)F capacitor across a 110 volt (rms) 60Hz power line. What current flows? Using complex Ohm's law, we have

\[
Z = -j/\omega C
\]

Therefore, the current is given by

\[
I = V/Z
\]

The phase of the voltage is arbitrary, so let us choose \( V = A \), i.e. \( V(t) = A \cos \omega t \), where the amplitude \( A = 110\sqrt{2} \approx 156 \) volts. Then

\[
I = j\omega CA \approx 0.059 \sin \omega t
\]

The resulting current has an amplitude of 59mA (41.5mA rms) and leads the voltage by 90\(^\circ\). This agrees with our previous calculation. Note that if we just wanted to know the magnitude of the current, and didn't care what the relative phase was, we could have avoided doing any complex algebra: If

\[
A = B/C
\]

then

\[
A = B/C
\]

where \( A \), \( B \), and \( C \) are the magnitudes of the respective complex numbers; this holds for multiplication, also (see Exercise 1.17). Thus, in this case,

\[
I = V/Z = \omega CV
\]

This trick is often useful.

Surprisingly, there is no power dissipated by the capacitor in this example. Such activity won't increase your electric bill; you'll see why in the next section. Then we will go on to look at circuits containing resistors and capacitors with our complex Ohm's law.

**EXERCISE 1.17**

Show that if \( A = BC \), then \( A = BC \), where \( A \), \( B \), and \( C \) are magnitudes. Hint: Represent each complex number in polar form, i.e., \( A = Ae^{i\theta} \).
Rather than worrying about this result in general, let’s look at some simple, but very important, examples.

![Diagram](image)

Figure 1.51. Generalized voltage divider: a pair of arbitrary impedances.

### 1.19 RC filters

By combining resistors with capacitors it is possible to make frequency-dependent voltage dividers, owing to the frequency dependence of a capacitor’s impedance \( Z_C = -j/\omega C \). Such circuits can have the desirable property of passing signal frequencies of interest while rejecting undesired signal frequencies. In this section you will see examples of the simplest such RC filters, which we will be using frequently throughout the book. Chapter 5 and Appendix H describe filters of greater sophistication.

### High-pass filters

Figure 1.52 shows a voltage divider made from a capacitor and a resistor. Complex Ohm’s law gives

\[
\frac{V_{\text{in}}}{Z_{\text{total}}} = \frac{V_{\text{in}}}{R - (j/\omega C)}
\]

\[
= \frac{V_{\text{in}}[R + (j/\omega C)]}{R^2 + 1/\omega^2 C^2}
\]

(For the last step, multiply top and bottom by the complex conjugate of the denominator.) So the voltage across \( R \) is just

\[
V_{\text{out}} = Z_R = R = \frac{V_{\text{in}}[R + (j/\omega C)]R}{R^2 + (1/\omega^2 C^2)}
\]
Most often we don't care about the phase of $V_{out}$, just its amplitude:

$$V_{out} = (V_{out}V_{out}^*)^{1/2}$$

$$= \frac{R}{[R^2 + (1/\omega^2 C^2)]^{1/2}} V_{in}$$

Note the analogy to a resistive divider, where

$$V_{out} = \frac{R_1}{R_1 + R_2} V_{in}$$

Here the impedance of the series $RC$ combination (Fig. 1.53) is as shown in Figure 1.54. So the "response" of this circuit, ignoring phase shifts by taking magnitudes of the complex amplitudes, is given by

$$V_{out} = \frac{R}{[R^2 + (1/\omega^2 C^2)]^{1/2}} V_{in}$$

$$= \frac{2\pi f RC}{[1 + (2\pi f RC)^2]^{1/2}} V_{in}$$

and looks as shown in Figure 1.55. We could have gotten this result immediately by taking the ratio of the magnitudes of impedances, as in Exercise 1.17 and the example immediately preceding it; the numerator is the magnitude of the impedance of the lower leg of the divider ($R$), and the denominator is the magnitude of the impedance of the series combination of $R$ and $C$.

![Figure 1.52. High-pass filter.](image)

![Figure 1.53](image)

You can see that the output is approximately equal to the input at high frequencies (how high? $\approx 1/RC$) and goes to zero at low frequencies. This is a very important result. Such a circuit is called a high-pass filter, for obvious reasons. It is very common. For instance, the input to the oscilloscope (Appendix A) can be switched to ac coupling. That's just an $RC$ high-pass filter with the bend at about 10Hz (you would use ac coupling if you wanted to look at a small signal riding on a large dc voltage). Engineers like to refer to the $-3\text{dB}$ "breakpoint" of a filter (or of any circuit that behaves like a filter). In the case of the simple $RC$ high-pass filter, the $-3\text{dB}$ breakpoint is given by

$$f_{3\text{dB}} = 1/2\pi RC$$

Note that the capacitor lets no steady current through ($f = 0$). This use as a dc blocking capacitor is one of its most frequent applications. Whenever you need to couple a signal from one amplifier to another, you almost invariably use a capacitor. For instance, every hi-fi audio...
amplifier has all its inputs capacitively coupled, because it doesn't know what dc level its input signals might be riding on. In such a coupling application you always pick $R$ and $C$ so that all frequencies of interest (in this case, 20Hz–20kHz) are passed without loss (attenuation).

![Graph of reactance vs frequency]  

Figure 1.56. A. Reactance of inductors and capacitors versus frequency; all decades are identical, except for scale. B. A single decade from part A expanded, with standard 20% component values shown.

You often need to know the impedance of a capacitor at a given frequency (e.g., for design of filters). Figure 1.56 provides a very useful graph covering large ranges of capacitance and frequency, giving the value of $|Z| = 1/2\pi fC$.

![Diagram of RC filter]  

Figure 1.57

As an example, consider the filter shown in Figure 1.57. It is a high-pass filter with the 3dB point at 15.9kHz. The impedance of a load driven by it should be much larger than 1.0k in order to prevent circuit loading effects on the filter's output, and the driving source should be able to drive a 1.0k load without significant attenuation (loss of signal amplitude) in order to prevent circuit loading effects by the filter on the signal source.

![Diagram of low-pass filter]  

Figure 1.58. Low-pass filter.

**Low-pass filters**

You can get the opposite frequency behavior in a filter by interchanging $R$ and $C$ (Fig. 1.58). You will find

$$V_{\text{out}} = \frac{1}{(1 + \omega^2 R^2 C^2)^{1/2}} V_{\text{in}}$$

as seen in Figure 1.59. This is called a low-pass filter. The 3dB point is again at a frequency

$$f = 1/2\pi RC$$

Low-pass filters are quite handy in real life. For instance, a low-pass filter can be used to eliminate interference from nearby radio and television stations (550kHz–800MHz), a problem that plagues audio
amplifiers and other sensitive electronic equipment.

Figure 1.59. Frequency response of low-pass filter.

EXERCISE 1.21
Show that the preceding expression for the response of an RC low-pass filter is correct.

The low-pass filter's output can be viewed as a signal source in its own right. When driven by a perfect ac voltage (zero source impedance), the filter's output looks like $R$ at low frequencies (the perfect signal source can be replaced by a short, i.e., by its small-signal source impedance, for the purpose of impedance calculations). It drops to zero impedance at high frequencies, where the capacitor dominates the output impedance. The signal driving the filter sees a load of $R$ plus the load resistance at low frequencies, dropping to $R$ at high frequencies.

In Figure 1.60, we've plotted the same low-pass filter response with logarithmic axes, which is a more usual way of doing it. You can think of the vertical axis as decibels, and the horizontal axis as octaves (or decades). On such a plot, equal distances correspond to equal ratios. We've also plotted the phase shift, using a linear

Figure 1.60. Frequency response (phase and amplitude) of low-pass filter, plotted on logarithmic axes. Note that the phase shift is $45^\circ$ at the 3dB point and is within $6^\circ$ of its asymptotic value for a decade of frequency change.
vertical axis (degrees) and the same logarithmic frequency axis. This sort of plot is
good for seeing the detailed response even when it is greatly attenuated (as at right);
we'll see a number of such plots in Chapter 5, when we treat active filters. Note
that the filter curve plotted here becomes a straight line at large attenuations, with
a slope of −20dB/decade (engineers prefer to say “−6dB/octave”). Note also that the
phase shift goes smoothly from 0° (at frequencies well below the breakpoint) to 90°
(well above it), with a value of 45° at the −3dB point. A rule of thumb for single-
section RC filters is that the phase shift is ≈ 6° from its asymptotic value at 0.1f_{3dB}
and 10f_{3dB}.

EXERCISE 1.22
Prove the last assertion.

An interesting question is the following: Is it possible to make a filter with some
arbitrary specified amplitude response and some other specified phase response? Sur-
prisingly, the answer is no: The demands of causality (i.e., that response must follow
cause, not precede it) force a relationship between phase and amplitude response of
realizable analog filters (known officially as the Kramers-Kronig relation).

**RC differentiators and integrators in the frequency domain**

The RC differentiator that we saw in Section 1.14 is exactly the same circuit as the
high-pass filter in this section. In fact, it can be considered as either, depending on
whether you’re thinking of waveforms in the time domain or response in the fre-
quency domain. We can restate the earlier time-domain condition for its proper
operation (V_{out} ≪ V_{in}) in terms of the frequency response: For the output to be
small compared with the input, the signal frequency (or frequencies) must be well
below the 3dB point. This is easy to check.

Suppose we have the input signal

\[ V_{in} = \sin \omega t \]

Then, using the equation we obtained earlier for the differentiator output,

\[ V_{out} = RC \frac{d}{dt} \sin \omega t = \omega RC \cos \omega t \]

and so \( V_{out} \ll V_{in} \) if \( \omega RC \ll 1 \), i.e., \( RC \ll 1/\omega \). If the input signal contains a
range of frequencies, this must hold for the highest frequencies present in the input.

The RC integrator (Section 1.15) is the same circuit as the low-pass filter; by
similar reasoning, the criterion for a good integrator is that the lowest signal frequen-
cies must be well above the 3dB point.

**Inductors versus capacitors**

Inductors could be used, instead of capacitors, in combination with resistors to make
low-pass (or high-pass) filters. In practice, however, you rarely see RL low- or
high-pass filters. The reason is that inductors tend to be more bulky and expensive
and perform less well (i.e., they depart further from the ideal) than capacitors. If
you have a choice, use a capacitor. One exception to this general statement is the
use of ferrite beads and chokes in high-frequency circuits. You just string a few
beads here and there in the circuit; they make the wire interconnections slightly in-
ductive, raising the impedance at very high frequencies and preventing “oscillations,”
without the added resistance you would get with an RC filter. An RF “choke” is an
inductor, usually a few turns of wire wound on a ferrite core, used for the same purpose
in RF circuits.

| 1.20 Phasor diagrams |

There’s a nice graphic method that can be very helpful when trying to understand
reactive circuits. Let’s take an example, namely the fact that an RC filter attenu-
ates 3dB at a frequency \( f = 1/2\pi RC \),
which we derived in Section 1.19. This is true for both high-pass and low-pass filters. It is easy to get a bit confused here, because at that frequency the reactance of the capacitor equals the resistance of the resistor; so you might at first expect 6dB attenuation. That is what you would get, for example, if you were to replace the capacitor by a resistor of the same impedance (recall that 6dB means half voltage). The confusion arises because the capacitor is reactive, but the matter is clarified by a phasor diagram (Fig. 1.61). The axes are the real (resistive) and imaginary (reactive) components of the impedance. In a series circuit like this, the axes also represent the (complex) voltage, because the current is the same everywhere. So for this circuit (think of it as an $R-C$ voltage divider) the input voltage (applied across the series $R-C$ pair) is proportional to the length of the hypotenuse, and the output voltage (across $R$ only) is proportional to the length of the $R$ leg of the triangle. The diagram represents the situation at the frequency where the magnitude of the capacitor’s reactance equals $R$, i.e., $f = 1/2\pi RC$, and shows that the ratio of output voltage to input voltage is $1/\sqrt{2}$, i.e., -3dB.

![Phasor diagram](image.png)

Figure 1.61

The angle between the vectors gives the phase shift from input to output. At the 3dB point, for instance, the output amplitude equals the input amplitude divided by the square root of 2, and it leads by $45^\circ$ in phase. This graphic method makes it easy to read off amplitude and phase relationships in $RLC$ circuits. For example, you can use it to get the response of the high-pass filter that we previously derived algebraically.

**EXERCISE 1.23**

Use a phasor diagram to derive the response of an $RC$ high-pass filter:

$$V_{\text{out}} = \frac{R}{[R^2 + \left(1/\omega^2 C^2\right)]^{1/2}} V_{\text{in}}$$

**EXERCISE 1.24**

At what frequency does an $RC$ low-pass filter attenuate by 6dB (output voltage equal to half the input voltage)? What is the phase shift at that frequency?

**EXERCISE 1.25**

Use a phasor diagram to obtain the low-pass filter response previously derived algebraically.