Chapter 5

PERFORMANCE OF 802.11B/G IN THE
INTERFERENCE LIMITED REGIME

One of the key advantages of mesh networks over cellular-based networks is that mesh nodes are relatively inexpensive. Hence, it is economically feasible to spread mesh nodes at a high enough density to provide high data rates to a dense user population. Specifically, the high node density results in a short distance between receivers and transmitters, and hence high SNR channels and high bit-rates are possible. However, an important issue of high density networks is that transmissions are significantly impacted by interference. Thus, in low density networks, transmission errors are due to low SNR, while in high density networks, transmission errors are due to low SIR. The first scenario is referred to as the noise limited regime and the second scenario is referred to as the interference limited regime. The behavior of 802.11 is well understood when transmissions are noise limited. However, performance in the interference limited case is considerably less well understood. This work explores 802.11b/g in the interference limited regime.

The findings presented here are the result of a large number of laboratory experiments. Thus, the channels between transmitters and receivers were controlled and all transmissions from external sources were eliminated with EMF shielding. The experiments lead to two key findings. First, in the interference limited regime, the behavior of 802.11b/g is dominated by the ability of the receiver to synchronize
to the sender’s carrier. Recall that transmissions begin with the broadcast of a synchronizing preamble followed by the data. Thus, in order to decode the data, the receiver must first successfully synchronize. The experiments discussed here show that for a large number 802.11b/g bit-rates, transmission errors are mostly caused by synchronization errors. In 802.11b, there are two similarly performing synchronization schemes, and 802.11g provides a third scheme. Thus, while 802.11b/g provides a large number of bit-rates, it essentially only provides two synchronization schemes. As a result, in the interference regime, most of the 802.11g bit-rates have the same tolerance to interference. Therefore, in contrast to the noise limited regime, in the interference limited regime decreasing the data rate provides no added ability to decode frames. Furthermore, since lowering the data rate increases the duration of the transmission, it increases the possibility that a collision will occur. As a result, in many cases, lowering the data rate decreases the performance. This has important implications for selecting the bit-rate that minimizes the time until a transmission is successful. This issue is further discussed in Section 5.4.

A second finding of the experiments described here is that if synchronization is successful, then the packet error is independent of the packet size. This significantly differs from the noise limited case where the probability of successful transmission obeys \((1 - p_{BE})^Z\), where \(p_{BE}\) is the probability of bit-error and \(Z\) is the frame size. Thus, in the noise limited case, the probability of packet loss is exponential in \(Z\).

The remainder of this chapter proceeds as follows. The next section provides a brief discussion on the physical layer convergence procedure for 802.11b and 802.11g. In the section 5.2, the experiments and experimental set-up is described. Section 5.2.2 presents some experimental results. Section 5.3 discusses computation of transmission error probability from the data collected. Section 5.4 describes how the models presented in Section 5.3 can be used to determine the bit-rate that minimizes the expected time to successfully transmit a frame.
5.1 A brief primer on 802.11 b/g physical layer

Physical layer protocol data unit (PPDU) is the basic protocol data unit that is transmitted by any physical layer standard. For 802.11 b/g PPDU primarily consists of physical layer convergence procedure (PLCP) and the data (MPDU - MAC protocol data unit) is followed by it. The PLCP can be subdivided into two parts, the PLCP preamble and the PLCP header. PLCP preamble is as a set of symbols that are used by the receiver to train the demodulator to acquire the incoming BPSK, CCK or the OFDM signals. The PLCP header comes after the preamble and contains information such as the modulation scheme used and the length of the frame. Figure 5.1 shows the PPDU for 802.11b and 802.11g standards.

The top two frames show the PLCP preamble and the PLCP header for the 802.11b standard. Two types of preambles, short and long are defined for this standard. Most of the commercially available 802.11b cards and access points implement both these types of preambles. The short preamble is 72 bits long and the long preamble is 144 bits long. The synchronization field for the short preamble is 56 bits as against 128 bits for the long preamble. The start frame delimiter (SFD) is 16 bits long in either case. PLCP header primarily consists of 48 bits. First 8 bits of the PLCP header indicates the modulation scheme used for the MPDU that is to follow. The next 8 bits indicate the service field which is reserved for future use. The length field which is 16 bits indicate the length of the MPDU. By making use of the length and the type of modulation used, it is possible to determine the amount of time required for the current transmission to complete. Final 16 bits are reserved for CRC. 802.11b PLCP is always performed at 1Mbps (Barker code spreading with DBPSK modulation) regardless of the data rate (modulation scheme) used for transmitting MPDUs. An important thing to note about the short preamble is that after the preamble is transmitted the PLCP header is transmitted at 2Mbps (Barker code spreading with DQPSK). The 802.11b PLCP with long preamble takes 192μs and
Figure 5.1: Structure of PPDU for 802.11 b/g standards. The top two figures show the PPDU for 802.11 b with long and short preambles respectively. The bottom most figure shows the PPDU for the ERP OFDM PLCP used in 802.11g.

Bottom frame of the figure 5.1 depicts the PLCP preamble and header for the 802.11g standard. Four different kinds of PLCP, ERP-DSSS/CCK, ERP-OFDM, ERP-PBCC and ERP-DSSS can be used for the 802.11g standard. When the service set contains clients that operate at 802.11b rates, ERP-DSSS/CCK PLCP is used. In this case either the short or the long preambles can be used. Of the other three mechanisms ERP-OFDM (extended rate phy - OFDM) is the one that is most widely implemented. ERP-OFDM allows for 7 bit rates (6,9,12,24,36 and 54Mbps) to be used. The PLCP header and the PLCP preamble of the 802.11g are always
transmitted at 6Mbps regardless of what physical rate is used for the data (MPDU) transmission. The 802.11g PLCP preamble is 16μs long with 12 symbols. Ten of these symbols are short and are used for AGC (automatic gain control), diversity selection and frequency offset estimation. The two symbols are long and are used for fine tuning of the frequency estimation. The way the signal is divided into groups of 1, 2, 4 or 6 bits depends on the data rate chosen. PLCP maps the data bits to BPSK, 16QAM or 64QAM depending on the data rate. The header of the PLCP consists primarily of fields that indicate data rate (modulation scheme), length of the MPDU frame and parity. There are other fields such as service and reserved which are left for future use. The PLCP header is 4μs long and hence the total time for PLCP procedure is 20μs in case of 802.11g.

5.2 Experiment Description
5.2.1 Experimental Setup and Protocol

Figure 5.2 depicts the block diagram of the experimental setup. The main aim of the setup is to precisely control of the channels between the transmitters and receivers and eliminate external interference.

The setup included of two access points, three laptops, and a controller computer. The two access points were the Cisco 1240 a/b/g [62] with Broadcom chipset. Prior work has shown that unlike some PCMCIA-based transmitters, the Cisco 1240 provides good transmit power stability. The sender and interferer controllers were used to adjust transmission bit-rate of the corresponding APs, transmit packets to the AP via the Ethernet (the AP then broadcasted these packets via the wireless transmitter), and to receive frame via the wireless transmitter (which was used for reference purposes). The receiver laptop was used to log all received frames. A modified MADWifi [63] driver was used to collect all frames received, including those received with bit-errors. The sender and interferer controllers and the receiver
**Figure 5.2:** The figure shows the block diagram of the experimental setup for all sets of measurements collected for analysis. The red lines indicate the RF cables used and the green thick line indicate the Ethernet cable connections used for controlling the experiment.
were equipped with Proxim Orinoco b/g Gold Cards with Atheros AR5212 chip-set [64, 65].

In order to conduct the experiments in a controlled and repeatable fashion, the receiver, the transmitter, and the interferer where isolated from each other. The isolation is achieved by using shielded wires to carry the "wireless" signals and by using attenuators between them. In order to prevent the RF leakage from the devices, all of the access points, laptops, splitters and connectors are wrapped with an RF resistant cloth. It was found that a single layer of the RF resistant cloth provides at least 40dB of attenuation, which was found to be sufficient to keep the weak RF leakage from affecting the experiments. Furthermore, the wireless transmissions used channel 1, while there were no nearby transmitters on channels 1-10.

Each attenuator was composed of calibrated Agilent 8495A and 8494A attenuators, providing repeatability within 0.3dB. These attenuators where used to control the received signal strengths, hence, the AP’s transmit power was not used. In all cases, the combined attenuation of attenuator A and B exceeded 130 dB. This is critical since the signal transmitted by the sender will partially reflect off of the receiver and be transmitter to the interferer. Thus, a combined attenuation of 130 dB ensured that the interferer is unable to detect the sender and vice versa. The level of attenuation and the inability to communication between sender and interferer when $C = -\infty$ dB was verified through experiments. Thus, the experimental setup resulted in the hidden node topology.

The objective of the experiment was to determine the probability of receiving a frame when the transmission is subject to interference. Since transmitting frames at precise times is difficult and error prone, frames were transmitted at random times so that collision occurred at random. Specifically, the sender transmitted packets at a fixed interval of approximately 12.4 msec between transmissions, while
the interferer transmitted packets randomly with the time between transmissions exponentially distributed with mean 31 msec. The number of interferer transmissions was recorded, and hence the total duration that the channel was occupied by the interferer could be determined. With this duration and the transmission duration, the probability that the frame experienced interference can be determined. The details of this are provided in Section 5.3.

Finally, each trial consisted of the sender broadcasting 10,000 frames with RTS/CTS disabled. The frames where broadcast, and hence, there were no retransmission nor ACKs. It was found that 10,000 frames resulted in a suitably small confidence interval.

5.2.2 Experimental results

As discussed above, the performance in the interference limited regime was investigated by transmitting packets so that collisions occurred at random. Some of the results of these experiments are shown in Figure 5.3. In this case, the average time between the beginning of the interferer’s frames was approximately 31 msec. The interferer frames were 576B and sent at 1Mbps. The sender’s frames were 1464B.

At high SIR, transmission errors occur with low probability. As expected, at low SIR, transmission errors occur regularly. It is important to note that the x-axis Figure 5.3 is the SIR when a collision occurs, but collisions do not always occur. Therefore, the probability of observing transmission error at low SIR is the probability of a collision occurring, i.e., the SIR is so low that if a collision occurs, then there is always an error. Since the SNR is high, when a collision does not occur, the frame is decoded with a high probability.

For 11, 12, and 24 Mbps, there is a plateau between the low SIR and the high SIR regions. Although not shown, this plateau also occurs at 5.5, 6, 9, and 18 Mbps. In the case of 6, 9, 12, 18, and 24 Mbps, this plateau ends at around 12
dB. For SIR above 12 dB, the observed probability of error is nearly the same for these bit-rates. 5.5 and 11 behave similarly, but the plateau ends at 7 dB. While not shown in Figure 5.3, 1 and 2 Mbps are similar to 11 Mbps in that they transition between a non-zero probability of transmission error and (nearly) zero probability of transmission error at 11 dB.

The reason that 6, 9, 12, 18, and 24 all have a plateau that ends at the same SIR is that they all use the same synchronization scheme and, apparently, this synchronization performs poorly when subjected to interference with SIR less than 12 dB. Similarly, 1, 2, 5.5, and 11 all use similar synchronization schemes\(^1\) that perform poorly when the SIR is less than 7 dB. Indeed, the height of the plateaus is the probability that the sender’s synchronization phase overlaps with the interferer’s transmission (See the next section for details). Thus, for 802.11g (802.11b), if the SIR during synchronization is below 12 dB (7dB), then synchronization will fail nearly every time. If the sender’s synchronization phase does not overlap with the interferer’s transmission, then there still is a possibility that the sender’s data transmission will overlap with the interferer’s transmission. The probability of incorrectly decoding the data part of the frame depends on the bit-rate. Figure 5.3 indicates that for many bit-rates, the SIR that results in errors in decoding the data part of the packet is considerably smaller than the SIR required to synchronize. For example, at 12Mbps, few data errors occur if the SIR exceeds 5 dB, but synchronization will always fail unless the SIR exceeds 12 dB. In summary, the region where synchronization always fails and data decoding always succeeds is exactly the plateau. Thus, the plateaus end at similar points when the synchronization schemes are the same. Since different modulation schemes have different tolerance to interference, the plateaus begin at different points. The height of each plateau

\(^1\) Our experiments have found no performance difference between 802.11b long and short preamble.
Figure 5.3: The observed probability of loss during the experiments described in Section 5.2.2. Note that not every frame experiences a collision, and hence, even at very low SIR, not all frames are lost. The fraction of frames lost depends on the probability that a frame experiences interference, the synchronization scheme used by the physical layer and the bit-rate used. For example, the probability of observing a transmission error at low SIR is equal to the probability of a collision occurring, i.e., every frame that experiences a collision is lost.

depends on the duration of the sender’s transmissions, which, of course, depends on the bit-rate used. Note that in the case of 36Mbps (and also 48Mbps and 54Mbps), synchronization is at least as tolerant to interference as decoding data is. Thus, in these cases, there is no plateau.

5.3 Probability of Transmission Error in the Interference Limited Regime

5.3.1 Analysis

When the received signal strength is sufficiently high, a transmission error can only occur when the sender’s transmission overlaps with an interferer’s transmission. In this case, there are three ways that the frame is lost. First, it is possible that synchronization will fail; we refer to this type of loss as a sync error (See Figure 5.4). The event where synchronization fails is denoted with $SE$. Second, a frame can be lost if there are bit-errors that cannot be recovered from FEC (if FEC is
used. Since these bit-errors are detected with CRC errors, the event that a frame is lost due to a CRC error is denoted with $CRCE$ (See Figure 5.5). The third way that packets are lost is that during the decoding of the packet header and payload, synchronization lock is lost (See Figure 5.5). A loss of lock event is denoted with $LL$. Note that $CRCE$ can only occur if synchronization succeeded. Similarly, a loss of lock can only occur if the initial lock occurs. Furthermore, bit errors are irrelevant if there was a loss of lock. Thus, the events, $CRCE$, $SE$ and $LL$ are mutually exclusive. The objective of this section is to determine $P(CRCE \mid SIR)$ and $P(SE \mid SIR)$, where these probabilities denote the probability of the event occurring when a collision occurs. Prior work has found that when the SIR is low enough that $P(LL \mid SIR) > 0$, bit errors are common and hence $P(CRCE \mid SIR) \approx 1$. Thus, $P(LL \mid SIR)$ has no impact on the probability of frame error. For this reason, $P(LL \mid SIR)$ is not investigated.

Since the focus is on the interference limited regime and not on the noise limited regime, the received signal strength is high. Thus, an error can occur only when the sender is transmitting at the same time as the interferer. Specifically, a $SE$ can

**Figure 5.4:** Possibilities of a sender’s PLCP transmission overlapping with an interferer’s transmission
Figure 5.5: The possibilities of a sender’s payload overlapping with the interferer’s transmission. This can lead to CRC errors or loss of lock errors.

only occur if the sender’s synchronization phase overlaps with the interferer’s transmission. There are different types of overlaps. Specifically, the sender’s synchronization phase could completely or partially overlap with the interferer’s transmission. In the case of the experiments described in Section 5.2, the probability that the sender’s synchronization phase is entirely overlapped by the interferer’s transmission is 

\[ \frac{T_{\text{Sync,Interferer}} + T_{\text{Data,Interferer}} - T_{\text{Sync,Sender}}}{T_{\text{InterfererInterval}}} \]

where \( T_{\text{Sync,Interferer}} \) is the duration of the interferer’s synchronization, \( T_{\text{Data,Interferer}} \) is the duration that the interferer’s transmits data, \( T_{\text{InterfererInterval}} \) is the time between interferer’s transmissions. A partial overlap occurs when \( n \) synchronization symbols overlap with the interferer’s transmissions. In the experiments described in Section 5.2, the probability that exactly \( n \) symbols of the synchronization are overlapped with the interferer’s transmissions is 

\[ T_{\text{SyncSymbol,Sender}} / T_{\text{InterfererInterval}} \]

where \( T_{\text{SyncSymbol,Sender}} \) is the duration of a synchronization symbol. Thus, let \( P(\text{SE}|\text{SIR}, n) \) be the probability of a sync error when \( n \) of the synchronization symbols are overlapped with the interferer’s transmission. Then, the probability of observing a sync error in the experiments
described in Section 5.2 is

\[ P(\text{observed SE}|\text{SIR}) = P(\text{SE}|\text{SIR}, N_{\text{Sync Sym, Sender}}) \]
\[ \times \frac{T_{\text{Sync, Interferer}} + T_{\text{Data, Interferer}} - T_{\text{Sync, Sender}}}{T_{\text{Interferer Interval}}} \]
\[ + \sum_{n=1}^{N_{\text{Sync Symbol, Sender}} - 1} P(\text{SE}|\text{SIR}, n) \frac{T_{\text{Sync Symbol, Sender}}}{T_{\text{Interferer Interval}}}, \]

where \( P(\text{observed SE}|\text{SIR}) \) is the fraction of frames transmitted in the experiment that resulted in synchronization error and \( N_{\text{Sync Symbol, Sender}} \) is the number of symbols in the sender’s synchronization. Since the synchronization phase has a short duration, the probability of a partial overlap is much smaller than the probability of complete overlap. Thus,

\[ P(\text{observed SE}|\text{SIR}) \approx P(\text{SE}|\text{SIR}) \frac{T_{\text{Sync, Interferer}} + T_{\text{Data, Interferer}} - T_{\text{Sync, Sender}}}{T_{\text{Interferer Interval}}}, \]

where we drop the \( N_{\text{Sync Symbol, Sender}} \) from \( P(\text{SE}|\text{SIR}, N_{\text{Sync Symbol, Sender}}) \) since a complete overlap is the only type of collision that is significant.

\( P(\text{SE}|\text{SIR}) \) was estimated for 11 Mbps with long and short preamble and for 12 Mbps. For verification purposes, \( P(\text{SE}|\text{SIR}) \) was estimated for several other bit-rates. Figure 5.6 shows \( P(\text{SE}|\text{SIR}) \) for various types of synchronization types. It can be seen that 802.11b short preamble and long preamble behave nearly the same. This is surprising since the long preamble has 128 bits, while the short preamble has only 56. On the other hand, this behavior was also detected in the noise limited case (an analysis of the noise limited case is not included due to space limitations).

Next, the probability of bit-errors occurring is examined. When the channel is noise limited, the probability of successfully transmitting a packet is given by \((1 - p_{BE})^Z\), where \( p_{BE} \) is the probability of bit error and \( Z \) is the frame size. However, the conclusion of extensive measurements is that when the transmission is interference limited, the probability of successfully transmitted a packet does not
obey \((1 - p_{BE})^Z\). Rather, \(P(CRCE|SIR)\) is independent of the frame length. On the other hand, if interference occurs randomly, then a longer frame is more likely to experience interference. Specifically, in the experiments described in Section 5.2, the probability that a frame experiences interference is \((Z_{Sender}/R_{Sender})/T_{InterfererInterval}\), where \(Z_{Sender}\) is size of the sender’s frame and \(R_{Sender}\) is the sender’s transmitted bit-rate. Thus, if \(P(CRCE|SIR)\) is independent of frame size, then

\[
P(\text{Observed CRCE}|SIR) = P(CRCE|SIR) \frac{Z_{Sender}/R_{Sender}}{T_{InterfererInterval}}, \tag{5.2}
\]

which is linear in \(Z\). Figure 5.8 shows \(P(\text{Observed CRCE}|SIR)\) as a function of \(Z\) for several bit-rates and confirms that \(P(\text{Observed CRCE}|SIR)\) is linear in \(Z\) and hence \(P(CRCE|SIR)\) is independent of the frame length.

Figure 5.8 shows the \(P(CRCE|SIR)\) for \(Z_{Sender} = 1464\)B and several modulation schemes. For reference, Figure 5.8 also show \(P(SE|SIR)\).

### 5.3.2 Discussion

Perhaps the most significant implications of Figure 5.8 is that when subject to interference, synchronization is considerably less robust to interference than decoding data. If the SNR is high and the short preamble is used, then 2, 5.5, and 11 Mbps all use the same synchronization. Thus, when subjected to interference, these modulation schemes will all have the same probability of error. Similarly,
Figure 5.7: Probability of CRC error observed during the experiments described in Section 5.2 as a function of frame size. The left-hand size is for 2Mbps while the right-hand size is for 11 and 12 Mbps, as indicated. Note that the probability of CRC error occurring is linear in the frame size.

Figure 5.8: Probability of error as a function of SIR for different types of modulation.
Table 5.1: SIR and SNR required for probability of error of 1/2.

<table>
<thead>
<tr>
<th>Bit rate (Mbps)</th>
<th>SNR</th>
<th>SIR</th>
</tr>
</thead>
<tbody>
<tr>
<td>2</td>
<td>0</td>
<td>-5</td>
</tr>
<tr>
<td>11</td>
<td>5</td>
<td>4.75</td>
</tr>
<tr>
<td>12</td>
<td>5</td>
<td>4</td>
</tr>
<tr>
<td>24</td>
<td>7</td>
<td>8</td>
</tr>
<tr>
<td>36</td>
<td>12</td>
<td>13</td>
</tr>
<tr>
<td>48</td>
<td>15</td>
<td>18</td>
</tr>
<tr>
<td>54</td>
<td>16</td>
<td>19</td>
</tr>
</tbody>
</table>

when subject to interference, 6, 9, 12, 18, 24, and 36 Mbps will all have the same probability of error. This behavior has implications for selecting a bit-rate. This issue is examined in more detail in the next section.

It is often assumed that the relationship between bit error and SNR is the same as the relationship between bit error and SIR or SNIR. To test this assumption, a larger number of experiments were performed to determine the noise limited performance of 802.11b/g. Table 5.1 used the results of these experiments to compare the SIR such that \( P(CRCE|SIR) = 0.5 \) to the SNR such that \( P(CRCE|SNR) = 0.5 \) under the assumption that the noise factor is 7dB, and hence the total noise (i.e., thermal noise plus the noise factor) is \(-93\)dB. Thus, excluding 2Mbps, 48Mbps, and 54Mbps, the assumption that relationship between SNR and bit-error is the same as the relationship between SIR and bit-error appears to approximately hold.

Therefore, on the one hand, SIR can be treated as SNR when considering decoding the data. On the other hand, since synchronization is sensitive to interference, the performance of 802.11 frame decoding significantly depends on whether the noise is Gaussian or is interference. Thus, the relationship between SNR and transmissions error is not the same as the relationship between SIR and transmissions error.

The behavior at 2Mbps is intriguing and required a huge number of experiments to verify. On the one hand, at high SIR (e.g., 7dB), 2Mbps performed worst than 11Mbps. However, at -5dB, 2Mbps significantly outperforms 11Mbps. The
causes of this behavior are currently under investigation.

5.4 Bit-rate Selection in the Interference Limited Regime

The poor performance of 802.11b/g synchronization in the interference limited regime has significant implications for bit-rate selection. For example, if a channel has a SNR that can support 36Mbps, then reducing the bit-rate to 6, 9, 12, 18, or 24 will not impact the ability to synchronize and not increase robustness to interference. Furthermore, decreasing the bit-rate increases the transmission time and hence increases the possibility of experiencing interference. Hence, arguably, the highest bit-rate that the channel can support should be used regardless of the frequency of transmission errors. This contradicts the common approach to ARF that reduces bit-rate when packet losses are detected [66]. This section examines bit-rate selection in more detail.

Since interference can result in transmission errors, the behavior of the 802.11 backoff algorithm must be included. Specifically, for transmission at bit-rate $R$ of a frame of size $Z$ with probability of success $p$, the expected time to transmit is

$$E{T_{tx}}(Z, R, p) = T_{\text{slot}} \left( \sum_{j=1}^{6} \frac{(2^{j+4} - 1)}{2} (1 - p)^j + \frac{1023}{2} \sum_{j=7}^{\infty} (1 - p)^j \right) + \frac{1}{1 - p} (DIFS + SIFS + T_{\text{ACK}} + T_{\text{Sync}} + T_{\text{PPLCPHeader}} + \frac{Z}{R}),$$

where $T_{\text{Slot}}$ is the duration of a time-slot, DIFS and SIFS are the durations of the DCF and short interface frame spacings, respectively, $T_{\text{Sync}}$ is the time it takes to synchronize, $T_{\text{PPLCPHeader}}$ is the time to transmit the PCLP header, $T_{\text{ACK}}$ is the duration required to send an ACK or, in case that the transmission failed,
$T_{ACK} + DIFS$ is the time that the transmitter waits before beginning to decrement the backoff timer. Here, it is assumed that the ACK is transmitted at 2Mbps, which is the default value used 802.11 APs such as the Cisco 1240. The constants $T_{Sync}$ and $T_{PLCPHeader}$ depend on whether 802.11b or g is used and in the case of 802.11b, they depend on whether the long preamble or short preamble is used. Here, it is assumed that the short preamble is used with 802.11b and the 802.11g PLCP header is used with 802.11g bit-rates. In (5.3), it is assumed that the initial value of the contention window is 31 (i.e., $2^5 - 1$) and the maximum value is 1023, i.e., $CW_{\min} = 31$ and $CW_{\max} = 1023$.

Under the assumption that RTS/CTS eliminates the majority of interference, the time to transmit a frame when RTS/CTS is used is

$$ETx (Z, R, 1) + 2 \times SIFS + T_{RTS} + T_{CTS},$$

(5.4)

where $T_{RTS}$ and $T_{CTS}$ are the times to transmit the RTS and CTS, respectively. Again, in the results that follow, it is assume that control packets are sent at 2Mbps.

When the transmission is interference limited, the probability of successful transmission depends on how often a collision occurs. Suppose that the interference is such that the fraction of time that the channel is occupied is $\rho$ and the duration of the silent times between interferer transmissions is exponentially distributed with mean $\lambda$. Thus, we have the following.
Proposition 1  The probability of a transmission failure is

\begin{align*}
P(\text{frame error} | SIR) &= (\rho + (1 - \rho)(1 - \exp(-\lambda T_{\text{sync}}))) P(\text{SE}|SIR) \tag{5.5} \\
&+ (1 - \rho) \exp(-\lambda T_{\text{sync}}) \tag{5.6} \\
&\times (1 - \exp(-\lambda (Z/R))) P(\text{CRC}E|SIR) \tag{5.7} \\
&+ (1 - (1 - \rho) \exp(-\lambda T_{\text{sync}})) \tag{5.8} \\
&\times (1 - P(\text{SE}|SIR)) P(\text{CRC}E|SIR) \tag{5.9} \\
&\times (1 - \exp(-\lambda T_{\text{sync}})) \tag{5.10}
\end{align*}

Proof. There are three ways in which a failure can occur. First, there could be a sync error. A sync error can only occur if the channel is busy during synchronization, which occurs with probability \( (\rho + (1 - \rho)(1 - \exp(-\lambda T_{\text{sync}}))) \). To see this, note that the channel is busy at the beginning of the synchronization phase with probability \( \rho \). If the channel is not busy at the beginning of the synchronization phase (which occurs with probability \( 1 - \rho \)), then it will become busy during synchronization with probability \( 1 - \exp(-\lambda T_{\text{sync}}) \). Thus, (5.6) is the probability of synchronization failure.

The second way a transmission failure can occur is if the channel is not busy during synchronization, but becomes busy during the transmission of the data, and then a bit-errors occur. The probability of this type of error occurring is given in (5.7-5.8).

The final way that a transmission error can occur is when the channel is busy during synchronization, but the synchronization succeeds. However, the bit-errors occur during data transmission phase. The probability of this type of error is given in (5.9-5.10).

With (5.3) and (5.5), the time required to transmit a frame in the face of occasional interference can be determined. Figure 5.9 shows the average time to transmit a 40B and 1400B frame when the channel is purely interference limited.
Figure 5.9: Average time require to complete a transmission of a frame when $\rho = 0.3$ and $\lambda = 1/10 \ m \ sec^{-1}$ and when the frame is 40B (left-hand side) and 1400B (right-hand side).

and the channel utilization is $\rho = 0.3$. As expected, the fastest bit-rates are either 11Mbps, 36Mbps, 48Mbps, or 54Mbps. Which of these rates is best depends on the packet size and SIR. Note that 48Mbps is nearly the same as 54Mbps, hence, little performance is lost if only 11Mbps, 36Mbps and 54Mbps are considered. We see that for small size frames, and $SIR = 10 \ dB$, 11 Mbps is the best. Furthermore, at this SIR, 11Mbps is faster than 36 and 54Mbps by a factor of five, or $300 \mu s$. Similarly, when the frame is 1400B and the $SIR = 16 \ dB$, 36 Mbps is the fastest and is faster than 54Mbps by a factor of 2 or $400 \mu s$.

Figure 5.9 shows that in the interference limited case, the bit-rate that results in the smallest expected time to successful transmission depends on the frame size, the SIR, and the channel utilization. Figure 5.10 shows the bit-rate for a wide range of frame sizes, SIRs, and channel utilizations. Figure 5.10 also shows where RTS/CTS leads to the smallest expected transmission time. In 802.11, frames larger than RTSThreshold use RTS/CTS. However, the optimal value the RTSThreshold is unknown. Figure 5.10 shows that the optimal value of RTSThreshold depends on the channel utilization (at $\rho \leq 0.5$, RTSThreshold$>1500B$; $\rho = 0.2$, RTSThreshold$=500B$ and at $\rho = 0.3$, RTSThreshold$=0$). It is possible to compute the optimal
Figure 5.10: Optimal Rate Regions. For a given combination of channel utilization, $\rho$, frame size, and SIR, there exists a particular bit-rate that results in the smallest delay. The above shows the optimal bit-rate in each region of the packet size/SIR plane. The region where a particular bit-rate is optimal is marked with a particular color. In these plots, $\lambda = 1/10 \text{ m sec}^{-1}$. 

value of RTSThreshold as a function of channel utilization. It is also possible to explore the optimal bit-rate and RTSThreshold when the channel is slightly SNR limited. For example, if the channel can only support bit-rates of 24Mbps or less.

5.5 Related work

There has been considerable effort focused on understanding the behavior of 802.11. In [67], measurements are used to explore the types of transmission errors at various bit-rates. Furthermore, [67] provides a useful explanation of many subtleties of transmission and decoding in 802.11. However, [67] does not examine the interference limited regime. In [15], the 802.11b nodes were examined in a rooftop network setting where it was found that packet error was not closely correlated with SNR. One possible explanation of this behavior is that the channel suffered from
delay spread. However, the transmissions could have also suffered from interference. One important drawback of the work presented in this paper is that the combined impacts of delay spread and interference was not studied. Further investigation in this issue is required. In [68], a PHY receiver model for decoding frames in the presence of interference was developed from measurement. While the developed model had some predictive power, it did not include the impact of synchronization, but rather relied on the replacing SNR with SNIR in the relationship between SNR and bit-error (See Section 5.3.2 above). In [69], capture was studied and a simple model for capture was developed from measurements. Contrary to findings presented here, [69] found that synchronization and decoding could as long as $SIR > 0$ dB.