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# Fault tolerant and integrated token ring network

Thomas Christopher Gilbar Florida International University

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# FLORIDA **INTERNATIONAL UNIVERSITY** Miami, Florida

Fault Tolerant and Integrated Token Ring Network

A thesis submitted in partial satisfaction of the requirements for the degree of Master of Science in Computer Engineering

**by**

Thomas Christopher Gilbar

To: Dean Gordon R. Hopkins College of Engineering and Design

This thesis, written **by** Thomas Christopher Gilbar, and entitled, Fault Tolerant an Integrated Token Ring Networks having been approved in respect to style an intellectual content, is referred to you for judgement.

We have read this thesis and recommend that it be approved.

John C. Comfort

Malcolm L. Heimer

Kang K. Yen

Wunnava V. Subbarao, Major Professor

Date of Defense: June 24, 1993

The thesis of Thomas Christopher Gilbar is approved.

Dean Gordon R. Hopkins College of Engineering and Design

Dr. Richard L. Campbell Dean of Graduate Studies

Florida International University, **1993**

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To my family and friends for their love and support

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#### ABSTRACT OF THE **THESIS**

Fault Tolerant and Integrated Token Ring Networks

**by**

Thomas Christopher Gilbar Florida International University, 1993 Miami, Florida

Professor Wunnava Subbarao, Major Professor

This thesis is a study of communication protocols (token ring, FDDI, and ISDN), microcontrollers (68HC 1EVB), and fault tolerance schemes. One of the major weaknesses of the token ring network is that if a single station fails, the entire system fails. **A** scheme involving a combination of hardware and timer interrupts in the software has been designed and implemented which deals with this risk. Software and protocols have been designed and applied to the network to reduce the chance of bit faults in communications. **ISDN** frame format proved to be exceptional in its capacity to carry echoed data and a large variety of tokens which could be used **by** the stations to test the data. **By** its very nature, the token ring supplied another major fault detection device **by** allowing the data to be returned and tested at its source. The resulting network was successful.

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# List of Symbols:

- BRI Basic Rate Interface<br>FDDI Fiber Distributed Da FDDI Fiber Distributed Data Interface<br>IEEE Institute of Electrical and Electro
- IEEE Institute of Electrical and Electronics Engineers<br>ISDN Integrated Services Digital Network
- ISDN Integrated Services Digital Network
- Primary Rate Interface

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#### Chapter 1

# **Introduction**

#### 1.0 A Brief History of Microcomputers, Networks, and Fault Tolerance:

#### 1.0.1 Development of the Computer

**Computers,** or **things that** can be loosely **defined as a computer,** have been around **for over a** hundred **years.** Changes **in** semiconductor technology **in the** past three decades have **allowed** the development **first** of small **scale** integration (SSI), then medium **scale** integration **(MSI),** large scale integration (LSI), and finally very large scale integration (VLSI). These developments have allowed more **and** more transistors, diodes, **and resistors to** be **placed on a** single chip, thus giving **rise to** microcomputers. A simple **lap top** computer **now has** more computing **power** than a computer which would have taken **up an entire roor in the** 1950's.

Intel Corporation developed a 4-bit programmable device called the Intel 4004 using the new semiconductor technology. **This** microprocessor **was** replaced **with the** Intel 8008, **an** 8-bit processor, which **was, in** turn, replaced with the Intel 8080. The Intel 8080 was **used in control** applications **and as** the CPU **in** small computers **in** the mid-1970's. This was the **advent of** what **is** now referred to as the microcomputer.

Motorola followed the example of the Intel 8080 by releasing the **6800** a few years later, a microprocessor designed with a different architecture and instruction set than its Intel equivalent, the **8085.** Both the **8085** and the **6800** were vastly superior to the **8080** in computing power. These new microprocessors were no longer used as simple programmable logic devices, they were now considered CPU's.

The **VLSI** technology led to the development of even more powerful microprocessors. Intel and Motorola developed 16-bit, 32 bit, and now even 64- bit microprocessors. These developments led to the single board microcomputer, which in turn led to cheaper microcomputers which were accessible **by** almost any person who desired to have a computer. Motorola released the MC68HC11, a powerful 2 MHz, 8-bit microcomputer with advanced on chip peripheral capabilities. Later Motorola released a microcomputer with the MC68020 for a CPU. The 68020 is a 33MHz, 32-bit microprocessor.

In the mid 1970's, IBM developed the **801** minicornputer, a computer with a microprocessor whose main function is to reduce the data path cycle time, regardless of whatever else it may need to

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**sacrifice. This** was the **first of what is called** Reduced Instruction **Set** Computers (RISC) processors. RISC processors are known for their speed. **To** achieve high speeds during processing, they have **several special characteristics. First, they have simplified instructions and** addressing modes. The RISC processors have **a** reduced instruction **set,** which results in **some of the more** basic, and most common, **processor** functions, such **as** loads **and stores,** being optimized. Next, multiple execution units allow parallel processing. The RISC machines **also** have **a** pipelined architecture. Finally, the RISC processors utilize **a** larger **set of** registers **to** minimize the usually time consuming memory **accesses.** However, with the reduced instruction **set** comes the need for compilers which **are** more complex.

Another type of processor recently developed is called the **Digital Signal** Processor. These devices **lean** more toward speeding **up arithmatic** functions rather than reducing data input/output time. The **faster** computation insures that the processing **rate is** limited only **by** the **speed of** data transfer between the processor and memory. DSP's have been optimized for high **speed** execution **of** signal processing functions such **as dot** products **and** Fast Fourier Transforms. To achieve the faster arithmetic, the DSP's include fast hardware multipliers **and totally** independent data and program storage areas. They also have fast interrupt switches for more efficient task switching.

Multiple ports **allow for** the design **of** efficient parallel processing architectures. A steady flow **of** data will **allow a DSP to** run as fast **as <sup>a</sup>**RISC machine. However, **any break in** the data **path can** result in performance loss.

#### **1.0.2** Networks

**With the** proliferation of computers came the need **to** allow communications between computers. Users needed **to** share **software,** data, and other forms **of** information, whether **it was** between computers on opposite sides **of the** world **or** between computers on the same **desk.**

No doubt the ideal **way to** communicate between computers would be to supply communications and control lines from every computer **to** every other computer **in** the world. **However,** setting up these dedicated **lines** would be expensive, time consuming, **and** would **take** up **too** much room. For example, stringing **a** dedicated **line** between two computers which **are several** thousand miles apart would be very expensive. Also, most computers require **access to** peripherals such as printers, hard drives outside of their own computers, modems, **etc.** However, these peripherals **are** expensive. If the user only needs **to use** these peripherals **occasionally,** the expense may be **excessive.**

A solution to this problem is to attach a computer to a network, so that while a user may not have a direct connection to a needed device, s/he can establish a connection **by** one of two ways:

- 1. First, by going through another computer on the same network that may have a direct connection with the device. For example, Figure 1.1 at the end of this section shows station **<sup>1</sup>** passing through station 2 to gain access to its printer.
- 2. By "borrowing" a line shared **by** several computers which accesses the device. In the case illustrated in Figure 1.2 (again at the end of this section), station 1 is accessing a printer **by** taking control of a shared line.

These solutions lead to other problems, however. In either case, the computers must share a common connection. The single line cannot handle more than one communication at a time. **A** scheme must be developed to decide who controls the line at any given moment. Further, each station on the network deserves its fair share of access time to the network.

In the 1970's and 1980's, computer science and data communications combined to come up with networking schemes.

Protocols and computer communication architectures were designed to meet the needs of various users. Ring, bus, and star topology networks were developed.

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Figure 1.1: Computer Accessing a Printer Through Another Computer



Figure 1.2: Computer Accessing a Printer By Seizing Control of a Common Line

#### 1.0.3 Token Ring

One scheme is the token ring network. Proposed in 1969, this is the oldest of the ring network schemes. The concept of the token ring is relatively simple: A token, a small packet of information different from any data that might be sent, is placed on the network. Only one of these tokens may be **present** on the network at a given time. If a station wishes **to** send data, it must wait for the token to arrive at its node on the network. The station removes the token from the network **and** sends **its message.** By removing the token from the network, no other station can receive the token, and therefore no other station **can send** data. Once the station finishes with its da transmission, it places the token back on the ring, potentially enabling further communications. By forcing a station to wait for the token before it can **send a** message, the token ring network removes the possibility of data corruption and collision. A more detailed description of the token ring will come later in this document.

#### 1.0.4 Fault Tolerance

One major problen in any network is ensuring that the information received is exactly the information sent. Any number of problems can occur in a system. The wrong address can be supplied,

thus sending **the** data **to** the wrong **place.** Both the sender and the **receiver need to** have ways **to** find this out. The data may **get** scrambled **while it** is on the **line.** One of the stations on the network may **go** down while the data **is** being sent, causing part or **all** of the message **to** be **lost.**

The solution **to** these **and** other problems **is to** make the networks **fault** tolerant. **In** other words, there must be **a way to** detect and **correct errors in** transmission. Several methods, **or** combinations of methods, may be **used:** Coding (Huffman, Hamming, parity generation **and** checking, etc), hardware modular redundancy, **acknowledgement signals, etc. Several** methods will be **reviewed** later in this document.

#### .L1 **Introduction to this document**

The main objective of **this** thesis **is to** develop **a** communications network which improves **a** popular LAN protocol by making **it** more **fault** tolerant and adapting **it to** modem technology. The token **ring** network **is** the basic protocol that will be used in **this** work. **This** topology **was** chosen for **several of its** characteristics, including that **it is** one **of** the three major LAN's, the innate security which comes from **its** circular layout, and the difficulty of creating **a** token ring network which **does** not completely fail when any of its **stations fail.** To develop this network, **a variety of** software **and** hardware techniques will be altered **and** combined. The result should be **a** more fault **tolerant** network which has been altered to compatible with modern protocols.

As **stated** above, the advent **of** computers **led to** the need **to** communicate between computers as well as computer peripherals. Simply connecting every computer **to every** other computer **and to its own** peripherals **is too** expensive **and** unrealistic. Therefore, fast, **efficient** networks must be developed **and** implemented.

This document will show the step by step **process of** designing **one** such **network:** The Token Ring. **Microcontrollers** such as the motorola MC68HC1 will **be** used *for* modeling **and** testing the system. The ring designed can **consist of** up **to** 256 **stations** (256 distinct addresses)(see Figure **1.3 on the** following page). The actual model developed will have three such **stations.** IBM **PC's will** be used **to** emulate the stations. Programming *for* the HC **11's will** be **in** assembly language **for** efficiency **and speed** of operation.

Simply building **a** token ring network **is not** enough. The **token** ring network must **also** be highly **fault** tolerant. **This will** be **achieved** through **a** variety of methods: address testing, encoding schemes, acknowledgements, **etc.** This document will focus **on** the **best way or** combination **of** ways **to** create a fault tolerant token ring network. Criteria for choosing the best way will include speed, efficiency, difficulty in installing, and of course, ultimate reliability. The concepts here can be extended to major fault tolerant data and computer communications.



Figure 1.3: 256 Station Token Ring Network

### 1.2 Chapter Organization

This document will be divided into 8 chapters and two appendices. Following the introduction, Chapter 2 will contain a detailed look at the considerations in the design of the system, which will include a discussion of the token ring network concepts and design and fault tolerance techniques. Chapter 3 will contain a discussion of the hardware used in this research. This will include a description of the 681C11 board, the IBM PC's, and how the network was developed with the microcontrollers. Chapter 4 will be a discussion of the system integration techniques, including protocols and methods. Chapter **5** will be a description of the software developed for this project. It will include a discussion of the particular needs of software for the token ring network (including fault tolerance) and the software needed to achieve communications between the various hardware elements of the system. Chapter 6 will show a variety of applications for the token ring network that will be developed. Chapter 7 will be the conclusions and will discuss areas for future study and experimentation as well as design alternatives and system enhancements. Diagrams and the software source code routines will be including in the appendices.

## Chapter 2

### Design Concepts for a Token Ring Network

#### 2.0 Introduction

Chapter 2 presents a study of the concepts used to develop the network for this thesis. Sections 2.1 through 2.4 of this chapter discuss the token ring network. Topics include the basic concept of token ring networks, official CCITT protocols, factors and methods used to design the token ring, and a comparison of the token ring network to other types of networks. Sections **2.5** and 2.6 of this chapter discuss fault tolerance. First, the basic concepts of fault tolerance will be discussed, including what it is, why it is necessary, and some basic fault tolerant schemes. Next, various fault tolerant schemes that could be used in this project will be discussed in detail.

# 2.1 Concept of Ring Networks

#### 2.1.1 Topologies

Networking computers has created some problems for designers. Directly connecting two or three computers is a relatively easy matter. Connecting four computers is a little more difficult, but still can be done. However, when the number of terminals increases

to five or more, there are just too many connections to be made. Also, **with** more connections, more **intelligence is** required. Each terminal needs to know exactly which connection leads to which terminal and needs to be able to listen in to all of the lines to see if data is being sent to its location. Intelligence and massive amounts **of** connections **are** both time consuming **and** expensive.

For these reasons, several networking topologies have been developed over the past couple of decades. The majority of the schemes **fall** under three basic topologies: The Star, **Tree,** and Ring. **A** special type of Tree, called a Bus topology, is also used quite often.

As seen in Fig. **2.1,** the terminals in the star network communicate through a switching element. This element establishes a dedicated path between the receiving terminal and the transmitting terminal. The star networks are easy to expand and require very few connections (each terminal **has** only **one** network connection, which **is to the central** node). Most **of** the intelligence **is** based **in** the **central** element, leaving very little for the terminals to do when it comes to networking. Since each terminal has its own line, the bandwidth requirements of the lines are quite small. The most difficult part of developing this topology is in developing ways of preventing data collisions. Data collisions occur when two or more terminals

simultaneously attempt to send a message on the same line. The messages will meet and the result will be the loss of data and the possible destruction of all messages. Two or more terminals can not simultaneously send data to a third terminal without risking the loss of some or all of the information. If anything but a full duplex line (a line which can send data in both directions concurrently) is being used, a terminal cannot transmit at the same time it is receiving. The more transmitters which are allowed to send simultaneously, the more intelligence and memory is required in the central element. Acknowledgement, request to send, and/or time division multiplexing schemes can be developed to alleviate these problems. A major weakness for this topology is the dependance on the central node. If it fails, the entire system will fail along with it.

Figure 2.2 shows a typical Tree topology. The bus topology is similar, but it does not have branches. Instead, it has one single "trunk" on which all of the stations are connected. For these topologies, a multi-point medium must be used. That is, several stations share the same transmission medium. Unfortunately, with only one communications medium, only two terminals can communicate at a time. Data collisions are difficult to avoid. **A** transmitting station must seize control of the line to keep other stations from attempting to communicate simultaneously. Again,

acknowledgement and request-to-send schemes must be developed. The required number of connections is minimized. Intelligence is located at the terminals; however, since each terminal has to watch only one line, the required intelligence is less than if every terminal was connected to every other terminal.

The topology which is the most inportant to this thesis is the Ring (Figure 2.3). Terminals are connected to the network through a repeating element. The network itself is a closed loop. This increases security **by** making it difficult for an unauthorized station to break into the network. Data circulate through the network using connections between the repeaters until it reaches the receiver. The receiver makes a copy of the data, and continues to forward it. The transmitter uses the returned message as a form of acknowledgernent. When the message returns to the source, the transmitter removes it from the network. Control of the network is then sent on to another station. Problems with this type of topology include setting up the order of the stations for sending messages, and removing a packet from circulating through the network, especially in the case of errors in addressing. These problems are solved **by** creating some type of control protocol. Another potential problem with the ring topology is that any break in the ring, or a failure of a repeater, will cause a failure in the whole network. Also, adding another terminal to a ring network **can** be difficult because **it involves** finding two **close stations that** are connected, breaking their connection, **and** then connecting each of them to the **new** station. Changes must be made in the basic ring design **to** deal with these problems.



**Figure 2.1** Example of a Star Topology

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**Figure 2.2** Example of Tree Topology



**Figure 2.3** Example of a Ring Topology

## 2.1.2 The Ring Repeater

The intelligence for this topology is mainly in the repeaters. For this thesis, the Motorola 68HC1 's will be used as the repeaters. These will be used for data encoding, storing data until it can be sent, error detection/correction, etc. Notice that any similar microcontroller, such as the Intel **8051,** could be used with similar results. The reasons for choosing the 68HC 11 will be **outlined later in** this chapter.

Repeaters have three possible states: Listening, Transmitting, and Bypassing. In the listening state (Fig 2.4), the repeater receives data from the network, copies it, then puts it back on the network,. This **creates a** slight delay. At the beginning **of a** message, the repeater **tests** the destination address. If the repeater detects its own address, it remains in the listening state. If the destination address does not belong **to the** terminal, there **is no** longer **any need for the** repeater **to copy** the data for the station. It will transfer into the bypass state (Figure 2.5). In this state, the data is passed directly on to the next station on the network, removing the delay. The final state is the transmit state (Figure **2.6). In** this **state,** the repeater listens in one direction and sends in the other. The listening side is picking up the message that the transmitting side **is** sending, but **with at least a one bit** delay (the ammount of delay depends on the transmission flow). This serves as an acknowledgement. Sometimes certain bits are modified by the
**receiving station to** show that **the** packet **was,** indeed, copied.

# 2.1.3 Brief Introduction to the Token Ring

One **control** protocol **is** called **the** token ring. This technique **was first** proposed in 1969 **(it was** called the Newhall ring **at** the time), **making it the** oldest ring protocol. This protocol was developed to remove the risk **of** data **collisions, increase ring efficiency,** and **to** supply **a** means **of** determining **the** order **in** which the **stations will be allowed to access the network.** In **the next section, a detailed** description of how this **is done is** presented.



Figure 2.4 Repeater in the Listening State



Figure 2.5 Repeater in the Bypass State



Figure 2.6 Repeater in the Transmit State

### 2.2.1 General Token Ring Protocol

To solve some of the problems of a general ring network, the token ring circulates a small packet, called a token, around the ring. When a station receives the token, it tests to see if there is a message waiting to be sent from that station. If not, it sends the token on to the next station on the ring (Figure 2.7). If there is a message, the station changes the token to warn other terninals that the network is busy, and sends out the message immediately following the changed token (Figure **2.8).** All of the stations test the address on the packet. If the packet was sent to the terminal, the repeater will make a copy of the complete message and will send some type of acknowledgement (a separate packet, the whole packet itself with certain bits changed, part of the packet, etc.). Once the source has received an acknowledgement, the station changes the token back to its original form, and will send it to the next station on the network (Figure **2.9).** Since there is only one token, it is only on the ring if there is no message being sent, and a terminal cannot send unless it is holding the token, two terminals cannot send messages simultaneously.

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Figure 2.7 Free Token Circulates on a Token Ring Network



Figure 2.8 S1 Captures the Token, then Sends a Packet to S3



Figure 2.9 S1 Places the Free Token Back on the Network After It Receives the Acknowledgement from S3

#### 2.2.2 Priority Schemes

There are several priority schemes possible for the token ring. The easiest, of course, is to allow each station to send out one packet of a specified size or smaller, then pass the token on to the next station in the ring. This ensures that all of the stations have equal access to the ring. However, quite often some type of priority scheme is desired.

One such priority scheme is the IEEE **802** Ring **LAN** Standard. For this standard, the token is a packet of data which includes a byte (or more) for priority and a byte (or more) used to reserve the token. A station wishing to send a message waits for a free token with a priority less than its priority. Once it gets this, it can send its message, along with a busy token. While a station is waiting for a free token, it can reserve a passing busy token **by** setting the reserve field to its priority. **By** doing this, a station with a lower priority cannot take a free token before the station with a higher priority. If the reserve field has already been set to some value, the station tests that value and compares it to its priority. **If** the priority in the reserve field is less than its priority, the station clears the old priority and sets the reserve to its priority. However, if it is higher, it cannot set the reserve field and must wait for the higher priority station to send its message. When the busy token returns to the source, the source will

clear out the priority bit and then sends the now free token back on the network. The free token will circle the network until a station with a priority equal to or higher than the reserve priority picks it up, clears the reserve field, and sends a message. **If** the token is not reserved, any station can grab it and send a message. See Chapter 4 for more details on this protocol.

Another standard for prioritizing packets is the Fiber Distributed Data Interface (FDDI) Ring High Speed **LAN** standard. This method is very similar to the **IEEE** standard, but it is designed to utilize the high speed of the fiber more efficiently. The FDDI uses a timing algorithm which allows the stations to send larger packets. This algorithm takes into account the length of time that the station had to wait before it received the free token. With this method, the source is restricted **by** a time frame rather than a packet length. This allows larger, or more, packets to be sent, increasing the efficiency of the network.

### **2.3** Design of Token Ring System

There are several major considerations when designing a token ring network. A major one of these is fault tolerance. The biggest stumbling block for token ring, at least with the software, is improper maagement of

the token. **If a station** does **not** return **the free token to** the **network, or** does **not** return **ANY** token **to** the network, the **result** can be disastrous. Methods **for** avoiding this problem are discussed **later. Fault** tolerance **on** the hardware side **is** more difficult, but **must** be considered.

Other considerations **include** choosing an appropriate protocol **to follow.** For example, there **are** several priority schemes **that can be** followed. The designer must decide which **is the closest** to the **design** specifications. The designer **needs to know** if **one station has** higher priority **than** the others, **or if** certain types **of** messages have higher priority than others. **If** priority is necessary, the **best way for it to** be implemented while **still** giving **fair access to other stations** must be investigated.

Commnunication protocols rust **be** chosen. Frame format, encoding techniques, **etc.** must **all** be **chosen and** standardized throughout the network. Data **rates** must **be** chosen. All **of** these will be discussed **in** Chapter 4.

Hardware **is** another important **factor.** The design **of** the **stations** must be **set.** The **functions of** the repeaters, and the hardware **used for those** repeaters, must be chosen. Software must be written **so** that a large variety **of** hardware **can access** the network **using it. For** example, **a** good network **should** be **able to** handle both the Motorola and the **Intel** families **of** microprocessors, as well **as** allowing printers, plotters, **etc.** to be accessed by any of the stations.

Still another consideration is the structure of the token ring itself. The traditional token ring consists of a single, unidirectional path. Newer token rings, however, are bidirectional, or have more than one ring. The second direction or line can be used to choose the shortest path from the source to the destination, or for acknowledgements. Hardware and software limitations come into play when choosing the basic design of the token ring.

The specifics for the token ring which will be built for this project will be discussed in later sections and/or chapters.

### 2.4 Comparison of Token Ring to Other Networks

There are many other types of networks. This section will describe just a few of those to help outline the strength and weaknesses of the token ring network.

To begin with, consider the token bus. It is very similar to the token ring in its design, so more attention will be paid to it than the other network types. One of the major differences is the connections required. **A** bus needs a multi-point medium which can support any of a large number of devices. For the token ring, only point-to-point connections must be considered. Each

station needs only worry about the next station in the ring. This gives the designer more choices when it comes to designing the connections. Unfortunately, the point-to-point connections mean that there where will be a longer delay for a message sent on the token ring. For the token bus, the only delay that the sender needs to worry about is caused **by** propagation delay of the connections and distance between stations. A ring must deal with this delay, and also delays as the data is sent through all the other stations on the network. Each station adds at least one bit delay. Also, the message may have to travel a greater distance through a token ring since the token ring *is* unidirectional. Since all stations are connected to the same element on a token bus, the instant the message is sent it will go in all directions at once. For the token ring, if it needs to communicate with the station just before it in the network, the message will have to travel all the way around the ring.

The single direction of the token ring does have its advantages. It makes some forms of fault tolerance easier to implement than on the bus. For the bus, all stations have easy access to the same transmission line. This increases the odds of an error occurring, because two stations are trying to control the network simultaneously. The stations on the ring have such a set path and such limited access to the network that data collisions are far less frequent. Also, since each station on the token ring has two lines, one for input, and one for output, the stations can more readily send and receive

simultaneously without causing problems with the network. The same is not true for the bus. Unless a full duplex line is used, only one station can transmit at a time.

Another network is the pure **ALOHA** technique. This technique was designed to allow widely scattered terminals to access one central mainframe. To do this, the **ALOHANET** has two channels: One channel from the terminals to the mainframe, and another chanel from the mainframe to the terminals. When a station has a packet to be sent, it does so. This creates a free-for-all on the transmission line. The **ALOHA** technique is very simple, and the intelligence is minimized. However, the occurrence of data collision is astronomical compared to the token ring. Token rings are more difficult to implement, require more time, energy, and money, but the savings in fault tolerance and in transmission time due to increased throughput makes it well worth the effort.

Finally, consider the Carrier Sense Multiple Access with Collision Detection (CSMA/CD). This is the most commonly used control method used for bus topologies. This system is better than the **ALOHA** in that when a collision is detected, transmission is instantly terminated. This saves time in sending messages that are not going anywhere anyway. However, it still suffers from a great deal of collisions, again making it weaker than the token ring in that sense. Like the **ALOHA,** the CSMA/CD method is easier to implement than the token ring, and messages that actually make it through get to the destination faster.

### **2.5** Concept of Fault Tolerance

### 2.5.1 Introduction

Fault tolerance is a major concern for any system, whether it is part of a larger network or a simple terminal working on its own. Designers have been working on ways of building systems which can sustain small errors without losing the whole system, or errors can be made without losing data or passing on bad data. Many methods have been developed for both error detection and error correction. Some possible problems and a few possible fault tolerant solutions will be discussed in this section.

### 2.5.2 General Fault Tolerant Schemes

In more complicated systems, one piece of hardware having a fault can take the entire system off line. Sometimes diagnosing the problem is difficult, and fixing it is nearly impossible. One solution for a hardware fault is called Triple Modular Redundancy (TMR). This fault correction technique involves running three identical units simultaneously in parallel. The output of the units are then sent to a voter, and the voter outputs the majority decision. Some added hardware can also supply a warning that one of the units has become faulty so that the user is aware of the problem. This system has two benefits: first, the system will not automatically go down, giving the user time to get in and fix the problem while the system is still up and running. Second, the odds of two units having simultaneous faults are much smaller than of any single unit having a fault; therefore, the efficiency of a system increases as a whole. For more important pieces of a system, five, or even seven, redundant systems can be running simultaneously, further bringing down the odds of a fault occurring. The one major weakness of redundancy lies in the cost and the space taken up by adding the extra hardware. The importance of the unit and the cost of redundancy must be balanced before action is taken.

Another major concern for network faults is in accuracy of the transmission. In the time between the source sending the data and the destination receiving the data, many things can occur to cause errors to appear in the data. Methods for the destination to test data and to inform the source that the right data was received (or even that **ANY** data was received) must be developed. First, ways of checking (and possibly correcting) faulty data will be discussed. A second layer of TMR can be added to the output of three voters to deal with faults in the voter.

Parity generation and detection is probably the most popular error detection/correction method. In this method, the bits in a field of data (a byte, word, long word, etc.) are compared and used to generate an extra bit of data which is then added to the field. At the receiving end, the parity bit is regenerated using the data. If it matches the parity bit sent, the destination can be certain that at least a single bit error has not occurred. Unfortunately, a single parity bit can only detect a single error in the field. To increase the chances of detecting an error, multiple parity bits can be added by comparing certain bits within the field, rather than every bit with every other bit. Each added parity bit increases the chance of detecting multiple faults, and can even eventually be used for error correction. A second layer of parity can also be added. This second layer checks the other parity bits, further increasing the chance of detecting errors, this time in the parity generation itself. **Of** course, there is a cost. The more parity bits that need to be sent, the less information is sent in each field of data. This decreases the throughput of the communication system. Communication time is increased and more equipment is needed to create and test the parity. Again, the number of parity bits, and the benefits of error detection/correction, must be weighed against the cost. A single parity bit for a byte of data may not be used for correcting the data, but it will wam the user about single bit faults (the most common fault), and can be used to ask for a retransmission

of the data. The fewest number of bits that the code words differ by is called the Hamming Distance. The more bits used in the coding, the larger the Hamming Distance possible.

For example, assume that a system needs to encode two choices, A and B, and two-bit code words are chosen. Since only two words need to be encoded, then only two combinations are necessary, say 00 and 11. The following table results:



Notice that the chosen combinations differ by two bits; therefore, the Hamming Distance for this coding scheme is two. That is why 00 and 11 were chosen. Any other choice of combinations would result in only a one-bit difference. Single bit errors can now be detected. If the word 01 is received, the destination address knows that an error occurred. This is not a valid code. However, the destination address does not know if the right code was 00 or 11.

By increasing the Hamming Distance, errors can be corrected. For example, if 3 bits are now used, there are eight possible combinations. The following table results:



By choosing 000 and 111, the new Hamming Distance is three. If the receiver now receives 001, he not only knows that there has been an error, but also that the input was supposed to be 000, assuming single bit errors (111 would be a double bit error). Like parity coding, the more bits the better chance of error detection/correction, but it also takes longer to send the message thus, throughput goes down.

A major problem in any network is finding a way to indicate

to the source that the data has been received, or if the appropriate person received the data. One way to do this is to come up with an acknowledgement scheme. An acknowledgement can be anything from a single bit, to a byte, to an entire packet sent by the receiver back to the source once the transmission is completed. Sometimes the entire packet is sent back to the source with a single bit changed to indicate that it has been received. This has the added benefit of sending the exact same message back to the source so that the source can test for errors in transmission itself. If there is an error, the message can be resent. Of course, the larger the acknowledgement, the more time it takes for a transmission to be completed. Again, a balance must be struck. Longer transmissions sometimes need multiple acknowledgements, requiring the source to stop for a while, wait for the acknowledge, than resume sending. This takes up even more time.

#### 2.5.3 Fault Tolerance in Token Ring Networks

Faults can occur in many places in a ring network. In this section, one of the worst of these will be discussed: If one of the stations on the ring goes down, the entire ring can be brought to an abrupt halt. A traditional solution will be discussed, and will be compared to the solution which will actually be implemented later in this chapter.

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One traditional solution to this problem is to change the architecture of the ring itself. The ring can be made into what looks like a star ring. In this case, rather than connecting each station directly to the surrounding stations, the connections can be made to a central intelligence (Figure **2.10).** In other words, a station will send its data to a central station, which will than send the data to the next station on the ring, which will then send the data back to the central station, etc. Other than the central station, the network works identical to the token ring. The strength of this set up is that if a fault occurs, the central office will see it, know where it occurred, and skip over that station to send directly to the next station. Also, connection to the central intelligence allows for easier expansion, another serious problem with ring networks.

There are several weaknesses to this method. First, if the central intelligence fails, the whole system will probably also fail. Also, if any one station goes down, it may still create a delay or even temporarily disable the entire network. Finally, the single central intelligence adds delay to the network and may require a lot of extra cable to connect it to the other stations. However, unlike the pure star network, the central intelligence does not need to be very intelligent.

To help with these problems, added changes can be made. The

ring can be broken down into smaller rings, and the smaller rings are connected by what is called a bridge (Figure 2.11). With this added step, if one central intelligence, or a station, goes down, the entire network will not be affected. Delay and amount of cable may be reduced. However, the bridges must be quite intelligent to deal with the switching, and more central intelligence may be required.



Figure 2.10 Star-Ring Network



Figure 2.11 Star-Ring Network with Bridge

### 2.6 Possible Fault Tolerant Schemes in This Project

Problems with errors in received messages are relatively easy to deal with. Choosing an appropriate coding scheme will deal with any problems with bit errors. ISDN frame format will be put to use to deal with bit errors (See Chapter 4). The ISDN frame has two bytes of a Bl channel, two bytes of B2, and four bits of D. The BI channel will have the original data being sent. The B2 channel bytes will be a negated echo of the B1 channel bytes. The stations will be able to compare the data on B1 to the data on B2 to test for bit errors. This decreases throughput by about half, but it also greatly reduces the chances of bad data being received.

As mentioned earlier, the very nature of the token ring adds an extra measure of fault tolerance. The ISDN frame will be passed all the way through the network and return to the transmitting station. This will give the transmitting station the chance to test to make sure that all data received was the same as the data sent. Special tokens will deal with errors in the data sent. These will be described in later chapters.

To deal with problems with the token, a ring monitor system will be established. The first monitor will be the one defined as station 0. It will send out the start up token on a timed schedule. The time between these tokens will be based on a timer interrupt. The timer will be set to a time

based on the number of stations, data rate, number of bits in the frame (48), and a small amount of time added to offset processing time at each station. This process will be used to deal with the possibility that not all stations will come on line simultaneously.

After the initialization token has been passed around, station 0 will no longer be the monitor. The system will be policing itself. Using timer interrupts, the stations will be able to determine if it has been too long since the last byte of data has passed through the network. Warning and error tokens will be used to determine if the failure was caused **by** a simple fault such as the last transmitting station failing to send a token or a station failing to pass a byte to the next station, or if a more serious hardware fault has occurred. Hardware faults can then be dealt with by skipping the station that has experienced the problem. The process will be described in more detail in later chapters.

To keep a station from sending messages that are too long and are unfairly tying up the network, the transmitting and receiving stations will both count the number of bytes or frames being sent. **If** the packet becomes too large, the transmitting station will be forced to stop sending. This will be done by using a special token. If this fails to stop a station, or if it is not the station that is transmitting that is sending out the endless message, a second timer interrupt, based on the time between new message tokens, will be set. In a procedure similar to that of a network which has gone quiet, this timer interrupt scheme will track down the station which is generating the data and skip over it. Again, see the software chapter for more details.

## 2.7 Considerations for Token Rings and Other LANS

Requirements of networks have been steadily increasing over the past several years. By 1995, audio and video will become just another data type. Multimedia is becoming more and more desirable, and networks need to keep up with the increased demand. Multi-media will require much more memory, storage capability, network bandwidth, and processing rates. 100 Mbps will become the baseline for networks, especially those requiring multi-media applications.

Early LAN schemes, such as Token Rings, FDDI, and Ethernet, were not designed to handle the requirements of multimedia. However, voice is not necessarily a constant stream of data. It is actually a mixture of talkspurts and silent periods. This allows for packetizing and sending of the data. Token Rings can take advantage of this fact. There are three current methods for voice-data integration on a Token Ring network.

First, distributed control can be used on the Token Ring. one such method removes the need for centralized control by giving each station a window in which it can transmit. **If** a station does not receive the token within this window, it is not allowed to send its packet. This allows the synchronous transmission of voice. This window can be either a fixed value, or can be based on the previous transmission.

Secondly, centralized control can be used. In this case, stations can be divided up into two categories: the ones which send only voice, and the ones that send only data. The central node sends out priority tokens at regular intervals. This token is sent around the ring to give all of the voice stations a chance to send a single voice packet. Once every station has had a chance, the central station sends out the regular token. If the regular token is busy, the central station must wait until a free token is sent before it can send the priority token.

Finally, a dual ring can be used. Once again a central station is used. In this case, three tokens are used. One is a null token to warn the stations that a new service cycle is beginning. The second token is used to indicate that only voice stations can send. The third token is used to indicate that only data stations can send.

Compared to other LAN's, Token Rings are preferable when trying to deal with the voice/data integration. For example, carrier-sense multiple access with collision detection (CSMA/CD), another major LAN, has many

disadvantages. One of these disadvantages is that making the necessary changes to the **CSMA** can be very expensive. Another disadvantage is that high utilization of this type of network causes the performance to degrade very quickly. Also, access delay is not guaranteed to be bounded for **CSMA,** causing probable real time delivery of voice packets and making **CSMA** not suitable for voice transmission.

Eventually, Asynchronous Transfer Mode, a special switch, will be used to integrate data, voice, and video. However, in the interim engineers have developed other methods of sending video. Video can presently be added to a network using a videoserver. This usually utilizes one or more computers to compress the frames of video signals so that the video can be integrated onto the network. Because it must handle both **A/D** conversion and data compression, a videoserver must have quite a bit of specialized hardware.

With the need for higher bandwidth on networks has come an increase in the development of viable options to make faster networks. Today's networks are on the verge of broadband (4 Mbps for Token Ring and 16 Mbps for Ethernet). Higher speeds are needed though. Proposals for new networks include **16** Mbps for Token Rings and 100Mbps FDDI (Fiber Distributed Data Networks).

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One method **of** increasing **speed** on **networks is the** Broadband ISDN. **BISDN** can use ATM to perform high speed switching (45Mbps to *155Mbps).* The push for this type of speed comes from the need to send large files and animated graphics over large distances. **BISDN** is also being used for basic consumer use in the form of high definition television (HDTV).

With the advent of ATM, variable length data frames becomes a **very** important and viable option to increasing network efficiency. The network will no longer have to wait on dummy information that had to be sent to **fulfil** the synchronous requirements. **In** older networks, **the** other **stations** had no way of determining when a transmitting station was finished, so that **packet had to be a definite length. This is a waste of** valuable **bus time. In** this project, variable length packets will be allowed. The maximum will be *256* bytes **due** to space **restrictions** on the 68HC11 board, but a **pakcket** can be **any size less** than that. The use **of** tokens will allow the **stations to realize** when the last frare of **a** transmission **is** arriving.

Frame relay, or the sending of variable length units of data through a network, was originally designed as a service for ISDN. Now extensive effort is being put into trying to make it a technical solution viable for any type of service.

Speed is not the only factor in developing technology. Fault tolerance must also be considered. Fault tolerance is not an all or nothing **consideration.** Each **LAN** must weigh the options and decide just how much fault tolerance is required. Network fault tolerance is dependent on three key aspects: equipment, design, and installation.

Fault tolerance **by** design is very complicated. In the design, consideration must be given to how to mix networks, hubs, protocols, and media. The scope of the network and importance of individual parts of the network **should be** taken **into** consideration. Splitting **up a** network **into sub** networks **can** help keep the **entire** network from failing when **a single user** fails. Parts of the network with more vital functions may require the extra cost and equipment necessary on developing a redundant system; however, this does not mean that every part of the network must use redundancy.

An example of fault tolerance **by** design is in the choice of network topology. **By** decreasing the number of hops (connection between networks) between links and workstations, a topology can decrease the chances of something going wrong in the transfer of data.

The network backbone is a prime candidate for redundancy since it is the link which ties subnetworks together. Redundancy can be accomplished **by** including a second cable which ensures that if one of the cables is broken, the network will **still** be **able** to communicate. Also, adding an **extra,** simple, redundant hub **can** keep separate hubs and subnetworks **communicating** when **one fails.**

**Since** most **errors in a** network **occur in** the hardware, equipment **is also** an important choice when considering fault tolerance **in a network. Power** supplies **are appropriate places for** adding redundancy **to** network components. **There** must **also** be equipment included which will allow switching **of control** when **a fault** occurs.

Of **course,** fault **tolerance on** the network **level is useless** without **fault** tolerance **on** the **station level.** More modern techniques **can target** specific **areas or** devices **for fault tolerance. For** example, **faults in control** computers **can** be monitored **using** band limiting **filters.** These **filters are** employed **to** monitor the propagation **of a** signal through **the** device. By comparing the **signal to a signal** from **a** another source, the system **can detect if** the **signal has exceeded a** threshold. **This** would indicate **a** possible fault **in** the system.

### Chapter 3

### System Hardware

### 3.0 Introduction

In this chapter, the key element of this project, the chosen hardware, will be discussed. The first section of the chapter will cover the architecture of the 68HC11 EVB board. After that will be a discussion of the RS-232C and its hardware. The next section will discuss the PC's which will be used as terminals on the network. The final section will be a discussion of using microcontrollers to make network gateways.

#### 3.1 The 68HC11 Microcontroller Unit

### 3.1.0 Introduction

In choosing the hardware for the repeaters, there are several considerations. The first consideration is the intelligence required. The most important function of these repeaters will, of course, be the sending and receiving of messages. This means dealing with source and destination addresses, changing the token between free token and busy token, dealing with priority schemes (if any), etc.. The repeaters for this project must deal with the possibility of being a monitor, which means that they must be given timing schemes and instructions

on skipping the next station if a fault is detected. They will also probably be used for coding and parity generation. For all of these reasons, the repeaters need to be reasonably intelligent.

The next factor is cost, reliability, and availability. The repeater must be made with accessible materials to make the network worthwhile. Designing with excessively high priced or hard to find materials would make building the network next to impossible for almost everyone who wishes to build it.

The final factors in choosing the hardware for the repeaters are convenience and flexibility. The hardware chosen should be relatively well known and user friendly. This helps with maintaining and managing the network. The more familiar a user is with the hardware the easier it will be to fix small problems with the network or to make any necessary changes to the network to make it fit the user's needs The user will not want to have to call in an expert for every problem that occurs.

The Motorola 68HC11 is well suited to implement this network. Introduced in the late 1970's, it has become one of the leading and standard microcontrollers in the industry. The 68HC11 is intelligent enough to deal with all of the functions that will be required of the network's repeaters. The microcontroller has **108** instructions, plus several versions of many of the instructions. This makes it very flexible and easy to program. Furthermore, the 68HC11 is easily obtained and is relatively inexpensive. As of the Spring of **1993,** the 68HC11 sells for just over \$130 for a single unit. When bought in quantity, the cost can drop to \$79 per unit. Since it has been around for almost two decades, the 68HC11 has been tested and updated enough to remove most of the major bugs. The 68HC11 is a common microcontroller, so many software and hardware Engineers have had at least some experience with it. Even without any experience, the 68HC11 is user friendly enough that anyone who has had any experience with microprocessor boards should not have much trouble in learning the system. Again, its age benefits by having reference material for the 68HC11 not only readily available, but also accurate and quite thorough.

### **<sup>1</sup>**General Information

The MC68HC11A8, the 68HC11 being used, is an 8-bit microcontroller with the ability to simulate some 16-bit functions (see the next section for details). The microcontroller is capable of average bus speeds of 2 Megahertz; however, it is also capable of bus speeds down to DC levels. This fact, combined with its **HCMOS** (high-density complementary metal-oxide semiconductor) **VLSI**

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design, means that the **68HC11** consumes little power compared to other microcontrollers. For on board memory, the MC68HC11A8 has 8 kilobytes of ROM (read-only memory), 256 bytes of RAM (randomaccess memory), and **512** bytes of EEPROM (electrically erasable programmable ROM). The board itself has an additional sixty-four kilobytes of address space, which can be used for storing programs, stacks, queues, etc.

The 68HC11 comes with some of its own on-board fault tolerance. There is an illegal operation code (opcode) circuit designed to detect any illegal opcodes and to cause a nonmaskable interrupt in case one is found. In case something happens to the clock, a monitor system is available to reset the system. Protection from software failures is provided **by** a watchdog system.

The diagram on the next page shows the general layout of the token ring designed for this experiment. Each station will consist of two HC 11's. The first will be connected to the user's computer and will also act as an input to the network from the station. This will be referred to as the inner HC11. The second HC11 has no direct contact with the computer and serves as the input to the station from the network. This will be referred to as the outer HC11.

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Figure 3.1 Network Design for this Experiment

The outer HC11's basic function is as an intelligent multiplexer. When the input from the **ACIA** RS-232C input (referred to as the primary line), connected to the previous station, is either bad or nonexistent, the outer HC11 will begin listening to its host RS-**232C** input (referred to as the secondary input), which is connected to the station before the previous station. This will cause the network to skip over the previous station. How this will be done is described in detail in the software chapter.

### .2 68HC11 Registers

The 68HC11 has seven registers (see Figure 3.2). This includes two 8-bit accumulator registers, called registers **A** and B. For many instructions, these accumulators can be combined to create a single 16- bit register, called the D register. This allows added 16 bit functions even though the **68HC** 11 actually has an 8-bit architecture.

In addition to the two accumulators, the **68HC11** has two 16 bit index registers, IX and IY, The second register helps to speed up the process **by** reducing the number of times the index register must be saved. These registers can be moved into the D register for 16-bit arithmetic operations.
The final three registers are a 16-bit stack pointer, a 16-bit program **counter,** and **a** standard 8-bit condition code register. The bits in the condition code register are defined as follows: zero (Z), negative (N), stop disable **(SD),** X interrupt mask (XI), I interrupt mask (IM), carry (C) and half carry **(HC),** and overflow bits **(0).**



Figure 3.2 Registers of the 68HC11



#### **3.1.3 68HC11** Ports

The **ICi1** EVB board has a total of 7 ports. **Of** these, four factor into this thesis: Port A, Port C, and the two RS-232C ports **(ACIA** and Host). These four ports will be discussed in this section. **All** registers to utilize the ports are memory mapped. This means that the programmer must manipulate certain addresses on the EVB board in order to use the ports. See Figure 3.3 on the following page for connections between the boards.

The ACIA is one of two asynchronous serial I/O ports. It is utilized with software by manipulating two registers: The **ACIA** status and data registers. If a byte of data has been input into the ACIA port, the first bit (LSB) of the ACIA status register will be changed to a logic 1. The data is read into the microprocessor **by** reading the ACIA data register. The first bit of the ACIA status register will remain a 1 until the status register and then the data register have been read. These registers are located at address locations **9800** and 9801 (hex), respectively. To output from this register, the second bit must be tested. **If it** is a zero, the port is not ready to output. If the second bit is a logic 1, the user can write to the data register and the data will be sent. For the inner HC11, this port will be used to communicate with the host computer. On the Outer board, the ACIA will act as the primary input from the network.



**igure 3.3** Station Hardware Design

The host port is the second asynchronous I/O port. It is utilized in a similar manner as the ACIA port. The registers used are the **SCSR** (Serial Communication Interface (SCI) Status Register), located at address 102E (hex), and the SCDR (SCI Data Register), located at address 102F (hex). The SCDR is actually two separate registers. One is used only on a read, and the other is used only on a write. If there has been an input to the host since the last read of the data register, the sixth bit (from the LSB) will be set. If the po is ready to send, the eighth bit (MSB) will be set.

The **C** and A ports will be used to communicate between the inner and outer **MC68HC11EVB** boards. A will be used for handshaking, C will be used to send the actual data.

The C port is an 8-bit parallel port which can function as either input or output, depending on the contents of the DDRC (Data Direction Register for port **C).** The DDRC is located at address **1007** (hex). The programmer can make a certain bit of the C register to be an input **by** writing a 0 to the appropriate bit of the DDRC. A logic 1 is used to make a bit an output. For example, writing hex OF to the DDRC will make the first 4 bits (LSB's) outputs and the other four inputs.

The C port has two data registers: The PORTC and PORTCL. The PORTCL data register actually latches the data after a clock edge (negative or positive edge can be chosen **by** the programmer) is applied to the STRA pin. The PORTC register changes along with the port **C** pins. PORTC is located at the address **1003** (hex), PORTCL at **1005.**

The **PIOC** register is the control register for the C port. To determine if a byte has been latched into PORTCL since the last read, the programmer can test the most significant bit of this register. If it is a logic 1, fresh data is in the latch. Otherwise, there are no new data.

Data will be transferred between the HC11's using the PORTCL register. The inner HC11 will have its C register set to input, the outer HC11 will use the C register as output.

Finally, the A register is half input, half output. The first three pins are input pins, the next four are output pins, and the last pin is an input or output, depending on the data direction register for **A.** Two output and one input pin will be used for the outer EVB. One input and one output will be used **by** the inner EVB. Reading or writing to these ports is accomplished through the data register for port **A, located at** address **1000** (hex).

## 3.1.4 **68HC11** Failure Statistics

Motorola supplies failure rate statistics for its microcontrollers and microprocessors in the Motorola Microprocessor, Microcontroller, and Peripheral Data data book. In this section, a very brief summary of these results will be supplied. See the bibliography for information **on the** data book.

Motorola runs several tests on its microprocessors for this **report.** One **of** these **is** the data retention **test** on the EPROM and EEPROM. This is a test of the EPROM and EEPROM's abilities to hold a charge over an extended period of time. The 68HC11A8 showed a failure rate of about 0.34% for up to **1008** hours at **150°C.**

Another test is the thermal shock test, in which the chip is tested **by** moving it directly from a fluorocarbon bath at **-65°C** to a bath of *150°C.* This test the stress caused **by** sudden changes in temperature **as** well **as** the increased conductivity due to a liquid environrent. This resulted in a failure rate of 0.13% for the **HCMOS** family of chips.

The third test is the high temperature operating life test, which

is performed to accelerate the failures due to the application of extreme conditions, including high input voltage and temperature. Tests ran for a maximum of **1008** hours with a voltage of 5.5 volts and temperature of **125°C.** These resulted in a failure rate of about **0.036%** for the **MC68HCI 1.**

The EEPROM read/write cycling test measures EEPROM cell operation over an expected lifetime. After running the 68HC11 at 5.5 volts and **85°C,** a failure rate of **0.87%** was found.

There are several other test results available in the data book, but they all point to the same thing: The 68HC11 has a small failure rate even under extreme conditions.

# **3.2** The RS-232C

## 3.2.0 Basic Information

Obviously a vital part of any network is the means **by** which the devices on the network will be connected together. For interconnection between stations on the network designed for this project, the RS-232C was chosen because it is the most common interface standard. In signaling, the RS-232C uses two voltage levels: A twelve (12) volt level is used to represent a logic 0, and a negative

twelve **(-12) volt is used to** represent logic **1. In** reality, voltage **is lost as a** signal **passes** through the connection line. Therefore, anything above three **(3) volts is** considered **a** logic **0, and** anything below **(more negative** than) negative three **(-3) volts is** considered logic 1.

The **RS-232C lines** are generally **good for signals** up **to 20Kbps.** Furthermore, distances **traversed** by **a** single **RS-232C line** should **not** be **greater** than 15 meters. **Voltage levels** begin **to** drop **too far and signals begin to get** garbled **for** signals **faster** than 20Kbps **or for** distances **greater** than **15 meters.** Distances **can be** made **greater** by boosting the **signal at** points **along** the **line. Faster** data **rates can** be achieved **with good** design, **including** data compression techniques. The data rate limit **is the** reason **for** the 16Kbps data **rate** limit **of** the **HC11.**

The **basic RS-232C** connector **has 25** pins; however, most applications **do not** require **all of** these **pins.** The HC11 RS-232C ports **use only a few of these pins for** comrunications. Figure **3.4 illustrates** the **pins used** by **the HC11's** two **RS-232C** ports. Table **3.1** briefly describes the use **of each** of these **pins.**



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# 3.2.1 Splitting the RS-232C

The most important fault tolerant characteristic of this design is the ability to skip a station which is no longer able to communicate. To do this, the output signal of each station's HC11 must go to two separate stations: The station directly after the present station, and the station after the next station (allowing the system to skip the next station if necessary).

Splitting the output of the host port of the inner HC11 will be achieved using a device called Multi-port. This device has 1 male/3 female connections, all wired in parallel. Up to two input connections can be connected directly to any one output connection. Any more than that creates too much loading on the output port.

Also, one of the ports that the output host port must be connected to is another host port. In order to do this, the Rx and Tx lines must be crossed to match the input to the output of the ports. This will be done using a device called a reverser.

# 3.3 IBM PC As a Workstation

As stated earlier, IBM Personal Computers will be used as the workstations on the network. Their main function will be to send data and programs to the 68HC1 's, which will be the repeaters of the system.

Programs for the 68HC11 (network, coding, etc.) were written using some type of word processing program. The programs were assembled on an assembler designed for the IBM PC by Motorola. Using a communications program, the assembled programs were then downloaded into the 68HC11. The final network function of the IBM workstation will be to start the network program on the 68HC1I running. Specific programs that will be used for communications and word processing will be discussed in the next chapter.

After this was done, the computers were used purely for data input/output. The PC's are used as dumb terminals; data are not sent into their memories; however, the data are echoed to the screen so the user can see what is being typed. Data are downloaded to the 68HC11's through the same communications program mentioned above. Data are sent directly from the Personal Computers' keyboards to the 68HC11 through one of the serial (COM) ports supplied on the PC's. Data received from other terminals are dealt with by the 68HC11's, then echoed on to the personal computers' screens.

# 3.4 Network Gateways By Microcontrollers

As stated earlier, the 68HC11's act as repeaters, which serve as the network gateways for the stations. This is accomplished by done using the two RS-232C compatible serial ports on the 68HC11 board. These were chosen instead of the other ports (there are five other ports on the 68HC11) because RS-232C is commonly used in computer communications.

One of these ports is used to connect the **68HC** 11 to the IBM Personal Computers. The communications is set at 9.6 Kilobits per second. Communications software on the IBM **PC** is utilized to establish and hold the link. Information on the communications software, exactly what the 68HC11 does with the IBM PC, etc. will be discussed in the next chapter.

The second serial port is used for communications between the 68HC11's. It can be seen from the token ring schematic in Chapter 2 that the repeater has to be connected to the network in two directions: One direction for data transmission, the other direction for data reception. For communications on the network, however, there is only one RS-232C port available. Therefore, this one port has to be divided into two. A discussion of how this is done can be found in section 3.2.1 Splitting the RS-232C. The data are sent through the network at the safest rate for RS-232C communications on the 68HC11, i.e. 9.6 Kilobits per second.

For **the** most part, the **rest** of the functions of the 68HC1 will be accomplished using software. These will be discussed in Chapter 5, which **is** the software chapter.

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# Chapter 4

# Communication Protocols

## 4.0 Introduction

There **are** certain requirements which are **vital to the** development **of any** network. One **of** these **is** the **need for** some type **of** protocol **to** keep **the** system from dissolving **into** anarchy. Another requirement **is** some **type of fault tolerance** scheme **to** keep **the** network up and running. After **all, if** the **data received on** the network are suspect, the network **itself is useless.** These requirements **can** be **met using** hardware, software, **or** combinations **of** both. This chapter will discuss these **vital** elements **and how** they have **been met in** this experiment. The **discussion** will include both **references to** methods **used in** other systems **and** the methods **used in** this experiment.

# 4.1 Other Communications Protocols

# 4.1L0 Introduction

**In** order **for a** network **to** be successful, **all** stations on **the** network must adhere **to a** common **set of rules** and standards. These protocols define everything **from** data format **to** the number **of** bytes which **can** be **sent** during **any** transmission. This **section** will discuss typical protocols used in token rings.

## 4.1.1 IEEE 802.5 Token Ring

One protocol designed for a token ring network is the IEEE 802.5 token ring. Special features of this protocol include a variable length frame, bits set aside for priority, and it allows reservation of the token by a station which wants to send a token.

Figure 4.1 shows the basic frame format of the IEEE 802.5 token ring. The token bit is a 0 if it is a token and a 1 if it is a free frame. The monitor bit is used to keep a frame from continuously circulating around the ring. The priority bits show the priority of the station which is taking the token. Only a station of higher priority may take the token. The reservation bits are for a station with higher priority to set aside the next free token. Once the previous message has been sent, the reservation bits are moved into the priority bit location so that only stations of higher or equal priority can take the token. The rest of the bits are self-explanatory. The IEEE 802.5 protocol delays transmission of the free token until the transmitting station has received the header of message. The token consists of a sequence of starting delimiter, access-control field, and ending delimiter. The information field begins with a header.



Figure 4.1 IEEE **802.5** Frame Format



The strength of this system **is** that **a** priority scheme **can** be designed. However, it **also lends itself to** the indefinite postponement of **a** message **for those** who **are** not fortunate enough to have **a** high priority. Limits should **also** be set on the information **area** of the **frame to** keep **one station from** keeping the **line** tied up **for too** long.

## 4.1.2 FDDI

A second protocol **is** the Fiber Distributed Data **Interface,** designed **for a** High-Speed Local Network (HSLN). This **protocol is** designed **to** take advantage **of the** high **speeds** that **fiber** networks **can attain** (100 MHZ).

**For this** protocol, limits **on** the data **sent** are determined by timing **rather** than **size** in **bits.** An **agreed** upon time limit **is placed** on sending. **If a** message goes beyond this time, the message **is not** cut-off. **It is** simply taken **away** from the **station until** the time **is** made up. Thus, the message length time averages **out to** be **the** time limit set.

Below **is** the frame format. Notice that **it is** similar **to** the frame format **of** the IEEE 802.5. **There** are **no access control bits, however.** The preamble **is a** sequence **of** idle symbols **used** for clocking purposes by the **receiving** stations.



Figure 4.2 FDDI Frame Format



# 4.2 Protocols Used in This Work

# 4.2.1 The ISDN Protocols

This experiment is designed to work on the ISDN frame format. This section will be a description of ISDN protocol.

The primary concept behind ISDN is allowing several channels on one line. Basic Rate Interface (BRI) consists of two "B" channels and one "D" channel. The B channels (referred to as B1 and B2) each run at 64 Kbps and are used for data, voice, etc. transfer. The D channel runs at 16Kbps and is usually used for control information but can also be used to send data. This gives a basic rate of 144kbps. Another 48Kbps is required to compensate for overhead bits, resulting in a final data rate of 192Kbps.

The basic frame format is shown below. The frame consists of two bytes of Bi data, two bytes of B2 data, four bits of D, and twelve overhead bits. Most of the overhead bits are for voltage level testing which is not necessary for this experiment. However, the bits were left in the frame to simulate the **ISDN** protocol as much as possible.

F and **Fa** are the framing and auxiliary framing bits. They are described as being positive zero. Without a negative zero, this does not really make sense. Since the RS-232C uses +12 volts for a logic 0, this bit was set to be a logic 0.

#### $FL$  $L$   $D$   $L$   $F_a$   $L$  $\overline{B1}$  $L|D|L|$  $L$  $D$  $L$  $B1$  $B2$  $LDL$  $B2$

Figure 4.3 ISDN Frame Format



The L bits are DC balancing bits and should be the opposite voltage of the F bits, which means that they are generally negative zeros.

The B1 and B2 channels can be used to send data (packet or circuit switched) to separate destinations. For example, one line may be connected to a fax and the other to a phone. In this case, the user can be receiving a fax from one location while speaking on the phone to another location.

**For this** work, **the B1** channel will be **the** primary data link. The B2 will be **used for** sending **a** negated echo **of the** data **on** the B1 channel. **In the** frame format, the data **on** the B1 channel will be echoed on the B2 channel immediately following it.

The D channel will be **used to** carry the token. This gives four **bits for** tokens. This provides **for a** maximum **of** sixteen different tokens. These **will** be **listed in** the **token ring protocol section.**

Limits **of the** HC1 1 **keep** this network from running **at the** 192Kbps. **The** network will be designed **to run at** 9.6Kbps, which **breaks** down **to** 800hz **for** the **D, 3.2Kbps for** each B, and **2.4Kbps** overhead maximum data **rate.** This slow **rate is one of** the weaknesses of **this** model, but the data **rate can** be **increased** by **using a** different **microcontroller board (see** recommendations **in** chapter **7).**

Another weakness **of using** the **ISDN** frame format **is** that **the** large amount **of** overhead further reduces the **rate of** the channels. This **is** especially bad **since** these **bits are** unnecessary **for** a **non-ISDN** system. However, this system **could eventually** be converted **to** communicate directly with another ISDN system.

The strength **of** the frame format **is** that each data frame sent

has its own control information. With the other protocols, the data are sent in one big clump. If there is an error in the data, it can propagate throughout the message and could lead to the entire message needing to be re-sent. For this system, there is no risk of this happening. The error can be spotted and re-sent before a pileup of data occurs. Also, since there are two channels, the system could eventually be developed to send two separate messages between two different pairs of stations simultaneously (see recommendations in chapter 7).

# 4.2.2 Fault Tolerant, Token Ring, and Other Protocols

This network follows the traditional rules of the token ring network. No station can send a message until it receives a free token, thus keeping more than one station from sending at a time. Each station inputs from one direction and outputs in the other. The entire network is a closed system and cannot be broken into once it is running.

Some of the features added to this system include a special "boot up" token which provides several special functions. When a station receives the token for the first time, it takes the number in the first byte of B1 in the frame and increments it. It then stores this value as its address. Since the data travels through both HCl1's in each station, both HC11's can find out their address in this without having to directly communicate. Next, each station knows the number of the previous station and the next station. This helps later on in the fault tolerance software. The start-up station (station 0) sends the token around a second time. Since the token had the stations counting off, this second trip around will provide the stations with the total number of stations on the network, which can help with the timer calculations (see software chapter) and can be used for security purposes.

Another feature incorporated is the use of several different tokens (which actually act as control bits). This helps with software and fault tolerance. The following table lists all of the possible tokens and a brief description of what each does. These tokens will be described in greater detail in the software chapter.

Messages will be limited in two ways. First, the packet length for the messages will be 256 bytes, or 128 frames. This will be maintained **by** counters in both the transmitter and the receiver. However, retransmissions will not be counted in the byte/frame counters. Therefore, to keep a station from continuously sending and/or retransmitting, the stations will also have a certain time limit. Transmitting stations will be cut off if they go beyond either the byte or time limits. To see how this will be done, see the software chapter.



# Chapter 5

## System Software

# 5.0 Introduction

Chapter 5 presents a study of the software modules necessary for this thesis. Topics will include communications between a station's inner and outer HC11's, between the inner HC11 of one station to the outer HC11 of the next (the actual network) and PC to HCi1 communications.

## *5.1* Token Ring Inner HC11 Software

## 5.1.0 Introduction

Since they act as the repeaters in the token ring, the heaviest part of the communications is dealt with by the inner HC11's. For this reason, the most complicated and longest programs are written for the 68HC11 microcontrollers. Since the Random Access Memory (RAM) space in these units is limited, the routines were written in assembly code (writing directly into assembly code makes for more efficient code than writing in a higher level language and compiling). In this section, the modules for each function of the inner HC11's will be described in detail.

# 5.1.1 Inner HC11 Main Program

The inner HC11's, which are the microcontrollers between the network **MC11's** and **the** computer, have the longest and most complicated programs of the system. The HC11's are programmed **to loop in a** main **routine** which polls the ACIA **for** input, polls the C po **for** data from the network, tests to **see** if something needs **to** be **sent to the screen,** and polls **to see if** something needs **to be sent to** the network. When **any of** these functions **are** applicable, **the MC11** program **then** proceeds **to** appropriate sub-routine. These **functions** will be described **in** detail **in** the following sections.

# 5.1.2 **Input From Terminal**

Once the network programs start **up,** the personal computers **used in** this thesis basically **act as** dumb terminals, doing **very** little on their own. Therefore, one **of** the main functions **of** the HC 11 **'s is to** gather the data from the terminal **and store it until it can** be **sent on** to **the** network.

This part of the program **uses** polling **of** the **ACIA to** input **from** the keyboard. When the terminal **serial** port (ACIA) receives data **in its** buffer, **it sets** the first **bit of its** status register. The character **is** then read from the data **register** of the ACIA. **If it is** the **first** character taken **since** the **last** queue **was sent** using the **"control** z," or not sent using the "control x," character (see below), the character is considered to be an integer and is used as the address of the destination of a message. Otherwise, depending on the data, the C 11 then performs one of several possible functions: Put the data into one of the two output queues after echoing it to the screen, ready the data for sending on to the network and wait for a free token before sending it, or "lose" the data (if the message has been aborted).

**If** the input is a character, the byte is added to a queue. As stated earlier, there are two output queues. The HC11 alternates between the two queues for storage. Since the MC11 must have the free token before it can send a message, there may be a short wait before the message can be sent. The two queues allow the user to begin a second message while the MC11 is waiting to send the first. Queues were chosen (as opposed to a queue) because of their First In, First Out (FIFO) system: The first character placed into the queue is the first character sent onto the network.

If the character is a "control z" (control button and z pressed simultaneously) the HC11 goes into Output to the Network mode. This is discussed in more detail in a later section. Meanwhile, the active queue moves to the second queue to deal with any further input from the terminal.

The third possibility is that the input is a "control x", which is the abort message signal. The **HCI** 1 places the head and tail of the queue to the same address, thus erasing the queue, and resets all functions so that it will be ready for the next message.

The final possibility is the back space character. **If** this character is typed, the last character placed in the queue is removed **by** moving the tail of the queue up one address. This character, **of** course, is not saved in the queue.

The inner HC11 is also used to set up the station's screen. The screen is split in half: The top half displays data from the station going to the network, the bottom half of the screen displays data coming in from the network. The screen setup is accomplished using the **ANSI** codes to place the cursor.

### 513 Output to Terminal

The HC11 is also responsible for the output of a received message to the screen. When a message is received, it is placed into a special queue set aside for network input (more on this queue later). As long as the data are valid, the HC11 begins removing the characters from the queue and sending them to the terminal.

Most of the network input is dealt with in the outer HC11. This HC11 uses the C port to pass the contents of the D and B1 channels to the inner HC11, which then proceeds to perform the function specified **by** the D control nibble. If the station is receiving and the data are valid, the D nibble may represent a 9, 11, 14, *or 15.* This indicates that the HC11 should add the 2 bytes of the B1 channel to the queue. In the case of the B and E control nibbles, the HC11 is programmed to set a special control variable which places the program into Output to Terminal mode, which prints the contents of the queue to the screen for the user to read.

Certain error messages (see fault tolerance section on the inner HC11 for details on these errors) can also be sent to the screen using the Output to Terminal mode of operation.

The ACIA output also uses polling. When the **ACIA** is ready to output the next byte, it sets the second bit of its status register. When this bit is set, the HC11 sends the next character to the queue to the **ACIA** data register, then move back into the main program. If the bit is not set, the HC11 returns to the main routine to perform any other functions.

## 5.1.4 Output to Network

Data on the network is sent from the transmitting station's inner HC11's host port. When this port is ready to send something, the most significant bit of the SCSR status register is set to a 1. Therefore, the HC11 is programmed to first test to see if something needs to be sent onto the network using a special variable set aside for that purpose. Then, if there is something to be sent, test the first byte of the **SCSR** to see if the host port is ready to send something. If the bit is not set, the program waits for it to be set before sending. If they are set, the data are placed in the host data register, the SCDR and then the program moves back to the main routine.

If the station has a message to send and has received the free token (0 on the D nibble), it proceeds to Output to Network mode. First, it sends a frame consisting of a 1 on the D, the source address, and the destination address. After receiving the same data back as a positive acknowledgement (D nibble of 4) it begins sending it the data. The data are placed into ISDN frame format and then sent a byte at a time. As stated in the previous chapter, the ISDN frame is six bytes long and each frame carries two bytes of data. The entire message is be sent out in this way. Once the output queue is empty and all of the data have been tested (see fault tolerant protocols later in this chapter), the HC11 builds and sends a free token frame.

If the HC11 belongs to the receiving station, it converts the recently received data into **ISDN** frame format using a D nibble of 8. Then the station moves temporarily into Output to Terminal mode and send the frame back to the sender.

If the HC11 belongs to neither the transmitter nor the receiver, it receives the data from the network as an **ISDN** frame, become a transmitter temporarily, and sends the data on to the next station.

## **5.1.5** Receiving Data from the Network

The inner HC11 receives data from the network through the C port. The **C** port is a general purpose parallel input/output 8-bit port. For the inner HC11, it is set to input only using the data direction register, DDRC. Port A, with pins  $A_2 - A_0$  inputs, pins  $A_6 - A_3$ output, and pin  $A_7$  input or output, is used for handshaking between the two MC11's.

By placing a logic 1 on pin  $A_6$ , the inner HC11 indicates to the outer MC11 that it is ready to receive from the **C** port. This pin is connected to the  $A_0$  pin of the outer HC11. As long as this pin is 0, the outer HC11 does not send to the inner MC11.

Pin  $A_0$  of the HC11 is connected to pin  $A_5$  of the outer HC11.

This **pin is used to** indicate **if the** byte **sent is** D and B **1** data **(a** logic 0) or **ISDN** frame format (logic 1). If the data do not affect the station, the **outer HC11** sends **the** data directly through **to** the inner **HC11** as an ISDN frame so that the inner HC11 does not have to waste time converting the B and **D** data back into frame format.

Finally, the STRA pin of the inner HC11 is connected to the  $A_6$  pin of the outer **HC11.** When this pin receives a negative clock, the data on **the** C port **is** latched **into the** port C **latch register** (PORTCL).

After receiving the data and determining if it is ISDN frame format or **D** and B1 data, the inner HC11 performs the appropriate functions on the input. In the case of ISDN frame format, the HC11 reads the six bytes in, puts them into a queue, then goes to Output to Network mode to send them out. If it is not ISDN mode, the HC11 is programmed to test the first byte (which is the D nibble) sent over to determine the function it must perform. There are three queues for the input from the **C** port: One which holds the last complete message (waiting to be sent to the screen), one to hold the latest incoming message, and one small queue to hold a 6 byte ISDN frame which is passing through the inner HC11.

A summarized description of the tokens can be found in Table 5.1 on the following page. The following is a detailed description of each token and the action taken when each token is received.

If the D is a 0 (free token), the HC11 determines if there is a message **to** be **sent. If** there **is,** the HC11 puts **itself into** Output **to** Network mode, sends out a busy token frame with the source and destination addresses, and begins sending **its** message. **If there is no** message **to** be sent, **it sends** the **free token out.**

If the D is a 1 (busy token), the HC11 will prepare its input queue **to receive data** from the network, including pushing the **address** of the source into the first byte of the queue. Notice that the outer HC11 does not send this busy token to the HC11 unless it is for this station's address. The inner HC11 does not need to check the address. The HC 11 then makes **a** frame with a D of 4 (busy token received, proceed) and BI's containing source and destination address. This frame is then sent onto the network. **If** the transmitting station receives the 1 token back (signifying an error and that the intended receiver never picked the frame up), it retransmits the same frame and continue sending again.


If the **D** is a 3 (sent too many bytes or took too long, stop sending), the **HC 11** clears the queue **and** sends an **error** message **to the** screen (see inner HC11 fault tolerance for more). The HC11 then sends a free token onto the network.

**If** the D is a 4 (warning token), the HC11 creates a frame. The frame will have the 4 as the token and its address **as** the first byte. If the sending HC11 receives this frame back, it knows then that the system is operating and sends a free token onto the network.

If the D is a 5 (error, being removed from the network), the **HC11 sends** a message to the **user** and then shut down.

If the D is a 6 (start-up token), the HC11 tests to see if it is the first time it has received this frame. If it is the first time, increment the number in the first B1 byte, store it as the station's address, then pass it on to the next station. **If** it is the second time, store the first B1 byte as the number of stations, then pass it along. If the station is station 0, it will take the 6 token off of the network after the second **pass and** send out a free token.

If the D is a 7 (broadcast mode), the input queues will be

prepared and the HC11 is set to receive a message. The source address (on the first byte of B1) is placed into the queue and the same frame will be sent on to the next station. If the station is the station which sent the 7 token, the outer HC11 removes the token from the network. Meanwhile, the inner HC11 is sending the data.

If the D is an 8 (data received, send next two bytes), the receiving station compares the two bytes in the B1 channel to the bytes sent. If they do not match, the error variable is set causing any frames with a token of 8 to be ignored and the number of bytes to move back is sent on the first B1 byte with an error token (12) followed **by** the retransmission of the bytes. **If** they do match, the secondary queue pointer is incremented to point to the next two bytes on the output queue. For more information on the secondary queue pointer, see the fault tolerance section in this chapter.

If the D is a 9 (data transmit two bytes), the data is added to the input queue, then a frame consisting of the two B1 bytes and token 8 is made and sent onto the network.

If the D is a 10 (retransmit upon request of the receiver), the data are added to the input queue and a frame with token **15** and the two bytes is made and sent onto the network. The error byte (described later) is cleared to indicate to the receiver that it can begin listening to the network again.

If the D is an 11 (last frame, first byte only), the receiver queues the first byte of B1, goes to Output to Screen mode, and sends the 8 token frame back to the transmitter.

If the D is a 12 (error at receiver, retransmit), the transmitting station moves back the queue pointer to the secondary pointer and begins retransmitting. The first frame has a token 10 the rest will go back to a token of 9. This token is sent if the B1 and B2 bytes of the frame do not match. See Fault Tolerance section for more details. An error variable is set to indicate that the receiver should ignore any data flowing in until the retransmitted token (10) arrives.

If the D is a 13 (error at transmitter, retransmit), the receiving station moves back the number of bytes in its input queue specified by the first BI byte. The retransmitted data writes over the old. **If** a transmitter receives this token, it calculates the number of frames to move back, places that in byte one of B1, stores its secondary queue pointer into its primary queue pointer, then sends out the frame. The error variable is set to indicate to the transmitter that it should no longer check any returning data since the queue pointer is no longer

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up to the right point.

The 14 token (last frame, both bytes) is similar to the 12 token, except that the receiver takes both of the B1's instead of only the first.

The last token, 15 (retransmitted data received) is sent to the transmitter to indicate that the first frame of retransmitted data was received. This is used by the transmitter to clear out the error variables, which in turn signals the transmitter to start paying attention to the data that are passing through the network again.

# .1.6 Fault Tolerance Software in the Inner HC11

There are several fault tolerance schemes designed into the program of this system. One of these is the ability of the transritter to check the data as they return. To do this, the sending queue has two tail pointers: One which points to the next character to be sent (primary queue pointer), and one to point at the last character received back from the network. This allows the transmitter to make sure that what was sent is what was received. See Figure 5.2 for a diagram of the queue pointers (bad data received by the sender will be discussed with the outer HC11's). Several of the tokens are designed to allow for retransmission of bad data, thus making sure that the data transmitted matches the data received.



Figure **5.1** Queue Pointers for the Output to Network Queue

The system has three methods of keeping a station from sending a packet which is too large. The first is a count in the transmitter's program. **If** the sender tries to type in too many characters (more than 256), the **HC** 11 sends a warning message and refuses to take any more characters until it receives a back space, control z, or control x. If this fails, the receiver also counts the number of frames (of good data, retransmitted data not counted) sent. **If** the transmitter sends more than **128** frames (not counting the busy token frame), the receiver sends the 3 token to force it to stop sending. If both of these fail, a timer interrupt is set at both the receiver and transmitter, which is discussed in section 3.2.2.

Communication stopping completely on the network is handled by a timer on the outer HC11 (described later). The inner HC11 uses the error token to warn the user that he is being taken off of the systen and shut down.

Using multiple queues for the input functions (from port C and from the ACIA) can also be considered fault tolerance devices since they allow communication to continue and keep the data from being written over.

# *5.2* **Token Ring Outer HC11 Software**

#### 5.2.0 **Introduction**

The main purposes of the inner HC11s are to communicate with the network user and to output to the network. The functions of the outer HC 11 are to act as an intelligent multiplexer, to accept input from the network, as sensors, and as watchdogs to make sure that the network keeps busy. In this section, how these functions are performed on a software level is discussed.

# 5.2.1 Outer HC11 Input from the Network

The outer HC11's are the direct link of the stations to the network. All input from the network must pass through these stations first. This sets them up as natural editors for the inner HC11's. This saves time and programming space for the inner HC11 by dividing the function tasks, passing some of the most basic tasks onto a piece of hardware which is necessary for multiplexing purposes anyway.

The outer HC11 keeps track of the present function of the station using a function variable. This variable lets the HC11 know if the station is sending, receiving, or doing neither. This is important to know because the function of various tokens depends on this.

The outer HC11 takes the input from one of its two RS-232C ports (which will be discussed later in this chapter) in ISDN frame format. If the HC11 is in Skip mode, it moves to the second byte of the frare to test the first bit of the D nibble (see previous chapter for frame set up). All tokens which only deal with sending and receiving begin with a binary 1, so if the bit is a one, it can be sent straight to the inner HC11 in ISDN format. Since the frame does not concern the station, there is no need to break it down to the D and B channels. If the bit is a 0, the frame contains data which are important to all stations. In this case, and in the case of the station sending and receiving, the **outer** HC11 gets the D nibble **and** decides **what to** do depending **on** the value.

# 5.2.2 Output to the Inner HC11

Once **the Outer HC11 receives** the data from the network, **it** must **pass all** necessary information **on the** inner **HC11.** As described **in the** previous section, **the HG1** 's communicate **using** the C **port** for **data and** the A port **for** handshaking. The **outer HC11 uses** the **token on** the D channel **to** decide **what it** must **pass on to the** inner HC11.

The table **on** the **next page** describes the function **that** the HC11 performs **for each** token. Since most **of** the more complicated **functions are** shared by more **than one of** the **tokens,** the more detailed descriptions of the special **routines** focus **on** the function rather than the token.

For the receiving functions  $(D = 9, 10, 11, and 14)$ , the B1's are taken from the ISDN **frame** format **and** compared with the B2 bytes, which **should** be **a** negated echo **of** the **B1's.** If they do not match, the retransmit token **(12) is** sent to **the station to** be passed **on** to the transmitter. Transmitting function  $(D = 8)$  is also tested. If there **is no** match, the transmitting **error** token **(13) is** passed **to** the **station** to be dealt with **and sent on to** the receiving station. The busy

token  $(D = 1)$  is also tested and passed on if there is no match.

If the token does not apply to the station (Skip mode and nibble token beginning with a logic 1) the **ISDN** frame is not broken down to its B1, B2, and D components. In some cases  $(D = 0, 3, 15)$ and 12) only the D is needed, so the HC11 will not find the B channels. In most of the other cases, all B's and D's are found with an ISDN break down subroutine.



Data are transmitted to the station through the C port as described in the section **5.1.5.** For a better description of the warning token  $(D = 2)$  and the error token  $(D = 5)$ , see the next section (5.2.2).

# **5.2.3** Fault Tolerance in the Outer **HCi1**

The most critical part of this experiment is the network's ability to continue functioning even after a station has failed. This, as mentioned before, is accomplished **by** having each station connected to not only the previous station, but also the station before the previous station. This allows the network to skip over a problem station. This will be accomplished using two timer interrupts as described in the following paragraphs.

The first timer interrupt is based on the amount of time a device has been transritting. This timer is set to a value based on the number of stations, the baud rate, the maximum number of bits that can be sent (which includes an allowance for up to 20 retransmitted frames) and a small amount of time to allow for processing of the data at each station. The timer is set whenever a station receives a 0, 1, 2, 4, 5, 6 or 7 token (the tokens which indicate that a new message is starting or someone has recognized an error in the system). This means that the station immediately following the station which is sending too much (even if it is not the station presently officially

transmitting) **is** the **first to** have **its** timer **go off. If** the new message token does **not arrive** by the time the timer **goes off,** the station **then** sends **a** warning **token with its** address **as** byte 1 of B 1 **and listens to** both **its** secondary **and** primary input. If **the** warning returns **on** the primary input before the next timer interrupt, the **station sends** out **<sup>a</sup> free** token. If it hears the **warning on** the secondary channel, **it** sends out an **error** token **with its** address **and** starts **listening on** the **secondary** input. If **it** gets **nothing but** garbage, **it** assumes **that it has** an error and shuts itself down by sending out an error token with the **next** station's address.

Meanwhile, the **station** immediately following the **one** sending **out** the warning **listens to its** secondary **input. If it** gets the warning **token** back from the secondary input **and receives** the error **token** from **the** previous **station on its** primary **input** anyway, **it** remains **on the** secondary input **and increases** the address **on** the warning frame **to its** address. If **it gets a free token on its** primary, **it goes** back **to listening to the** primary input.

The second timer interrupt will be **based on** the time **between** receiving **any pair of** data bytes **.** The amount of time between these interrupts **is** calculated similar **to** the **factor for** the timer listed above, but **now it is** based **on only one** byte of data. If **a** byte does not arrive

within the specified time, the timer interrupts go off, beginning with the station immediately following the station which received the last byte of data (which is the station which is no longer sending). The steps that are taken are the same as those outlined in the previous paragraphs.

These interrupts are carefully planned so that the station which has the hardware problem is the one removed from the network and to make sure that it is tracked down as soon as possible. The next station is checking its secondary input so that it can determine if the previous station is receiving garbage or is not receiving or if the station before the previous has actually stopped sending or is sending garbage, which is a very important difference.

These two interrupts allow the software to enact the multiplexing fault tolerance which is the central concept behind the network designed in this experiment. However, the outer HC11 also has other fault tolerance capabilities. Foremost of these is the ability to check if the B2 bytes are the negated echo of the Bl bytes. This design is similar to the concept of parity, but it allows each bit to have its own check. These cause a lot of overhead and reduces throughput, but it greatly reduces the chances of a bad piece of data passing undetected. The negation of the data (as opposed to a simple echo) is performed to detect stuck at zero or one faults which would run through all the bytes of the data without otherwise being detected.

# 5.3 Terminal to Microcontroller Communications

Terminal to microcontroller (68HC11) communications is accomplished using the Terminal program in Windows". The programs are written on the PC, assembled, and the Terminal software is used to download the code to the memory of the microcontroller. Commands written directly into the software specify where to put the programs in the microcontroller memory.

The Terminal software takes care of all other functions. When the software is running, all data input to the terminal by the keyboard are automatically moved on to the microcontroller. The speed of transfer is determined at connection time by choosing one of the setups available in the Terminal program. Parity, word size, etc. between the terminal and the HC11 can also be chosen at this time.

# Chapter 6

#### Applications of a Fault Tolerant Token Ring Network

# 6.0 Introduction

This chapter will be a study of systems which could be possible applications of a network similar to the one developed in this research. Included in the study will be a description of what the system is, the benefits of using this network (including the use of the ISDN format), and some general outlines of how the systems will work. The three systems that will be discussed are: A medical network, teacher to students multimedia network, and a general office network.

### 6.1 Medical Network

#### 6.1.0 Introduction:

Lack of communications is probably one of the leading factors in the problems with the United States' present medical system. Patients, especially the elderly and the employed, have trouble getting to their doctors. Pharmacies have trouble determining if the voice on the other side of the line is really a Doctor or a person trying to get his/her hands on drugs. Ambulance drivers and paramedics are hampered by the time it takes to determine any possible allergies or other illnesses in a patient, and may still end up having to guess in the end and hope that they are not wrong. Communication with insurance companies is slow and involves too much paper work. People on vacation have trouble getting to prescriptions or getting in touch with their Doctors.

The list goes on, but the result is the same: The lack of a strong *network* between Doctors, hospitals, pharmacies, ambulances, insurance companies, and patients handicaps the American medical system.

# 6.1.1 The Benefits of the Medical Network

A single network which connects all of the major participants in the medical system (pharmacies, Doctors, patients, etc.) would greatly improve the system.

A patient to Doctor network would allow the patient to get follow ups to tests without having to go back to the Doctor. The ISDN format would allow for several channels, each one carrying a different type of information. For example, with the basic ISDN BRI format, 3 channels are available. One channel could be used for the voice. The second channel could be used to send graphics (X-rays, CAT scans, etc.) to a patients with ISDN format and a computer. The third channel (the D channel) could be used for communications overhead and other data. This third channel can also be used for security. Some type of medical card can be designed which will have specific information about the patient so that the Doctor can be sure that the patient is who s/he claims to be.

Figure 6.1 on the following page illustrates this connection. The patient's computer, card reader/writer, and ISDN phone can be put into ISDN frame format by the network connector. The network connector will also send the message on to the token ring. Since there will be a large number of people on the network and token ring networks get slower with the number of stations, several token ring networks can be tied together. One of the stations on the ring can act as a gateway to another token ring network. It will have the same opportunity to receive and send as any of the other stations on the network.

On the receiving end, the Doctor will have his/her phone and computer. The Doctor's network connector will take the **ISDN** formatted data and separate it out into phone and computer data sending it to the appropriate place.

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Figure 6.1 Patient to Doctor Network Connection

Another benefit of this ISDN system is that the computer to computer communications will allow the deaf to communicate with their Doctors.

In emergencies, time is of the essence. Ambulance personnel should have access to a patient's records at their finger tips. They could then almost instantly know allergies, medical history, and any other information that is necessary to determine the appropriate treatment. Again, a medical card may be useful in looking up a patient's records. Presently, the medical alert bracelets and necklaces are really the only contact that the paramedics have with a patient. These alerts, however, can not hold very much information, and they can not be easily changed with a patient's changing history. Data on this network can be changed automatically, so paramedics will be sure that they are getting the most up to date information about the patient.

This also helps Doctors keep better records on their patients. Every visit to a hospital, a different doctor, a pharmacy, or, for Doctors with multiple offices, a visit to a different office can be placed directly into the patient's records.

The network can add security to the prescription process. Presently, Doctors call the pharmacies and use a code number read to the pharmacist to leave a prescription. There are three problems with this: Anyone can call a pharmacy, once someone finds out a Doctor's code there is no way to find out that the code has been compromised or track the culprit down, and many pharmacists/clerks are too busy to check the code.

The first benefit of the network designed in this research is that token ring networks are notoriously difficult to break into. Also, the D channel allows Doctors to send specific security codes automatically, adding more security to the system and speeding up the system as a whole. Pharmacists no longer need to take the time to answer the phone because the data can be sent straight to a computer screen. Each time a prescription is refilled, the pharmacist can update the patients records, letting the Doctor know the length of time between refills. For drugs which run out of refills, the pharmacist can send the information and requests for more refills directly to the Doctor without having to make the phone call.

Insurance companies can also be added to the network. This will allow claims to be instantly filed, speeding up the process and decreasing the wait before the Doctor, hospital, or pharmacy is compensated or the patient is reimbursed. Since a large percentage of health care cost comes as a result of administrative costs, making the

system more efficient would make health care more affordable.

Figure 6.2 shows the total network connections, including hospitals, Doctors, pharmacies, patients, insurance companies, and ambulances. These will all work similar to the patient to Doctor network described earlier. Again, the large number of stations would probably result in the need for several separate ring networks. Each station is given its own network in the drawing, but this does not necessarily have to be true. For example, a patient may be on the same network as his/her Doctor depending on the location. Hospitals may want their own network so that communications between its labs, Doctors, pharmacy, nurses, etc. may be faster. The ambulance can be connected to the network through a wireless communications scheme, giving them access to the patients records.

To increase security, each station can be given different security clearance. For example, Patient's may not be able to contact a pharmacy by computer at all. Meanwhile, the pharmacy may be able to access only information pertaining to medication, and then only to update prescription dates, not change the medication itself. Security methods can be achieved on the D channel, saving the two B channels for sending data (voice and computer data).

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At the present time, most people do not have access to this type of equipment. However, the price of computers and **ISDN** equipment should eventually drop to a level which will make them affordable to most people.



Figure 6.2 Conceptual Integrated Medical Network

### 6.2.0 Introduction

**With all of the advances in technology, the chalk board lecture** method **of teaching is out of date.** Computer **graphics,** CD ROM, **laser** disc players, sound blasters, etc. can make learning faster, easier, and **more entertaining.** The **combination of visual and auditory stimulus in** learning **increases student** comprehension **and retention.** Computer learning programs **greatly increase student interaction in the** learning process, **which in turn sparks** more interest and **concentration in** what **it being taught.**

New classes **which** consist of **the mixture of visual and** auditory stimulus, called **electronic classrooms, are being developed (Figure** 6.3). **Students will work** from **their own** PC. These **PC's will be able to** run **lessons which** combine computer **graphics, sound, and live action video.** Multi -media **also allows for** more **graphic illustration of subjects.** For example, instead of having a teacher attempt to describe a cube by drawing **it on a two** dimensional **black** board **and then** making **odd** hand movements, **an electronic** classroom would be **able to use three** dimensional graphics **to** bring the **cube to life,** spinning **it** around and more **effectively showing** the students **just** what **a cube is.**



Figure 6.3 Interactive and Intelligent Electronic Classroom

# **6.2.1**  Networking in Multimedia

Probably one of the most important uses of networking in multimedia is to allow the teacher to evaluate the progress of the students. Using the network, teachers will be able to go into a student's "account" to see the student's scores on the exercises in the lesson. Information on how the students are performing in certain areas of the lesson can assist teachers in determining where there are weaknesses in the program, or at the very least indicate to a teacher which topics need to be further explored.

A student could also send messages and questions to the teacher if s/he needs individual attention. This is especially beneficial in cases where teachers may not be in the same room, or even the same state or country, as the student. This gives students flexibility in when and where they study the lessons. For example, students who work during the day can perform the lessons by going to the classroom or logging on the network from home in the evenings. Any questions can be sent to the teacher through the network, to be answered at the instructor's convenience.

The network also allows instructors to create lessons which require interaction between two or more students. This allows students at different locations to interact in solving difficult problems and/or doing problems which focus on increasing their teamwork skills.

Figure 6.4 illustrates the token ring network as a tool in an electronic classroom. Notice that any station can also be a remote station set up on some type of modem. Since most multi-media applications take up so much memory space, a single mainframe with a large amount of memory on the network can reduce the amount of memory required at the individual stations.

Security is important when it comes to students' grades and scores. Again, token ring networks are by nature relatively secure. Along with this, the network can allow students to have individual accounts, protected by passwords, which will store these scores and the present location of the student in the lessons. Teachers can be given special codes which will allow them to gain access to the students' records to evaluate scores on the various exercises.

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Figure 6.4 Network Diagram for an Electronic Classroom

# 6.3.0 Introduction

Inter-activity between workers in an office environment is often vital to the smooth operation of the office. Traditional methods of sending memos create piles of paper which are wasted and can get lost. The need for quick access to data makes some type of electronic interaction even more desirable.

For example, imagine if one secretary goes on vacation and the boss suddenly needs a paper/letter/etc. that the absent secretary typed on the computer. Searching through the secretary's desk hoping to find the disk which contains the necessary document can be time consuming and fruitless. A network (and a well organized filling system of course) can make finding this missing paper much easier.

Figure 6.5 shows a typical office setting with the token ring set up. The wires can be run in the walls as much as possible to reduce the amount of hanging wires.



Figure 6.5 Five Station Office Token Ring Network

# **6.3.1** Description of an Office Network

A fault tolerant token ring network would fit into a typical office format quite easily. The network would allow the free flow of data between the employees. Security can be added as needed to keep unauthorized employees from sensitive documents by creating levels of security clearance.

There are several uses for this network. To begin with, papers or letters written **by** a secretary or receptionist can be sent to one of the other employees for proof reading without having to waste paper **by** printing the document. Next, an electronic mailing system would be easy to set up after the network is already in place. This would allow messages to be passed between employees. Third, several people can work on the same paper simultaneously. For example, the Boss may be working on the economic side of a proposal while the Engineer is putting together the technical aspects of the same proposal. Once they are finished, the papers can be put on the network, sent to one of the two, and combined.

Figure **6.6** is a conceptual diagram of the token ring for the office shown in Figure **6.5.** Notice that there is an added station which is a gateway to further networks outside of this one. Since these employees probably work together more often than with other

employees **outside** of **this** immediate **office or** someone **outside** of the company **all** together, they **can be** given **their own** sub-network with **which to** work. This will **speed** up **their** communications **to** each other and isolate them from **other** sub-networks in **case** some part **of the** system elsewhere **in the** company **goes down.**

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Figure 6.6 Diagram of an Office Token Ring Network

#### Chapter 7

# Results and Conclusions

# 7.0 Introduction

**This chapter is a final discussion of this experiment.** A **discussion** of the performance **of** the resulting network follows the first **section.** The **conclusions** and recommendations are **in the final section.** The recommendations and conclusions are divided **into** hardware, software, **and** communications protocols.

# 7.1 **Results**

# 7.1.0 Introduction

This **section is** a **discussion of the** performance **of** the designed network. Topics will include speed and efficiency resulting from the combination **of** the protocols, **software,** and hardware. Most **of** the **data** are difficult **to** confirm. For example, the number **of** instructions that **need to** be **executed for each** fuction **is** difficult **to** determine because looping and polling can create many uncertainties. However, the numbers **used in** this section came from **close** study **of** the final program.

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# **7..1** Speed of the System

The **HC1** 1 ports were set to **9600** baud. This is the maximum baud for reliable communications between the PC and the ACIA. Greater speeds begin to increase the corruption of data sent. To remain consistent, the entire system, including the host ports, was set to **9600** baud. Greater baud rates can be achieved through changes in registers and hardware, but this baud rate served its purpose.

The RS-232 ports send one byte of data at a time. Along with this data, the ports send a start bit and a stop bit. This increases each transmission to ten bits, and each frame to sixty bits. Therefore, one frame takes approximately **6.25** ms (60bits/9600 baud) to move from one HC11 to another on the network. Therefore, a frame would have  $n*6.25$  ms, where n is the number of stations, of travel time to work its way through the network. For example, the network built has three stations. This means that a frame would take **18.75** ms of travel time to move through the network.

This, however, does not include execution time at each station. Each type of token can mean different things to on outer HC11. In fact, each frame of data can mean different things to different stations on the network. For example, if station 1 is sending to station 3, station 2 should be in skip mode. Station 1 must pull the data from the queue, put it into **ISDN** frame format, and send it. Meanwhile, it must also listen to the network waiting for the acknowledgement, test the returned data for errors, then either retransmit or move on to the next test. Station 3 must decode the D channel, decode the B channels, test for errors, queue the data (including incrementing counters), insert the appropriate acknowledgement token, put the frame into **ISDN** frame format, and send the frame. Station 2, however, will pass the data straight through its system as an ISDN frame format. This means that station 2 has a shorter execution time than the other two stations.

Table **7.1** shows some approximate execution times. The number of instructions executed is an approximation based on careful study of the program. An average of 4 cycles per command was chosen because much of the program consists of loads and stores of immediate numbers (2 cycles each), load and store effective address (4 cycles each) decrements and increments (mostly 2 but sometimes 3 cycles, depending on the register), branches (3 cycles), **jumps** and returns (6 cycles and **5** cycles), and load index registers (5 cycles). The HCI1 has a clock of about 2 MHz.


Notice that all of these times are quite a bit smaller than the 6.25 ms transmission time. This shows that the stations are done processing and sending out the frame before the next frame arrives at the station. This results in overlapping of execution time and transmitting time, which reduces the amount of time that the system sees for sending a packet of information. This means the frames themselves take the same amount of time, but with the overlap the packet as a whole moves faster.

Table 7.2 shows the approximate transmitting time, from the transmitting station back to the transmitting station, of frames/packets through the system. This is based on packet size, three stations, and the assumption that only the first and the last execution times are seen by the network due to the function overlap. Free tokens are assumed to take about 200 instructions to execute.



The broadcast mode takes more time because all stations are receiving. Non-broadcast has at least one station which is skipped. Notice that the transfer time of a whole packet is much less than the sum of its parts. The table below illustrates the overlap and how it cuts down the transrnission time.



Notice that although **there** are 9 frames **sent** to **3 stations, only <sup>11</sup>**transmission times **are seen** rather than 27. This is a **great** saving **in** time.

### **7.1.2** Efficiency of the Network

For every 6 bytes (frame) sent, only 2 of those bytes are actual data. This means that only **33%** (16/48) of the information sent is data. When the start and stop bits of the RS-232 ports are added, this percentage drops even further to about **27% (16/60).** This gives an actual data rate of only 2592 bps. For free token frames, the percentage drops even more since the only necessary data is on the **D** channel  $(4/60 = 7\%)$ .

However, these numbers are a little misleading. For each frame, 20 bits can be considered used for fault tolerance  $(20/60 =$ 33%). Since one of the focuses of this research is on fault tolerance, this loss is worthwhile. The only actual, unused bits are the twelve overhead bits of the ISDN frame and the stop and start bits of the RS-**232,** which totals twenty four of the sixty bits, or about 40%. Unfortuneatly, to keep with ISDN frame format and RS-232, this large percentage must be endured.

For any network, efficient use of the line is desirable. The least efficient time for the ring network is when the token is traveling through the network. In this case, the line is used for **18.75** msec in the **19.95** msec it takes the token to travel around the network. This leaves the line idle for 1.2 msec, or about **6%** of the time. When the percentage of actual data rate is figured in, this drops the network down to carry worthwhile data about 6.2% of the time.

The network becomes more efficient when a packet is being sent. As seen earlier, the overlap makes the network move smoothly so that there is almost always something on the line, The larger the packet, the more the line idle percentage drops toward 0.0%. This means that the RS-232 line is busy nearly all of the time. When the percentage of useful data is factored in, this gives a total efficiency of about 60% if fault tolerant bits are factored in, or about 20% if they are not.

### 7.2 Conclusions and Recommendations for Future Study

## 7.2.1 **Hardware Recommendations and Conclusions**

In the early stages of this research, the choice of microcontroller to use to implement the network design was a major decision. In the end, materials available became the strongest argument for the 68HC11EVB board. As the work progressed and more was learned about the board, the pros and cons of the 68HC11EVB became more apparent.

The 68HC11EVB microcontroller is very flexible and

performed well in this experiment. Debugging programs can be difficult, especially for the outer 68HC 11EVB. With no direct contact to the outer, the limited debugging abilities of the board became very apparent. However, researchers would probably have problems debugging an outer board regardless of the microcontroller.

One feature on a microcontroller that a future researcher may want to look for is more RS-232 ports on a single board. A microcontroller with four RS-232 ports would be ideal. These ports would supply the two inputs from the network, one output to the network, and one port to communicate with the computer. This would eliminate the need for an outer microcontroller and save the programmer quite a bit of work.

More advanced microcontrollers, such as the MVME133 VME board (68020 microprocessor based mircocomputer), would add a few benefits to the system. The **68020VME** is a 32-bit microcontroller with a coprocessor. With its eight 32-bit data registers and **7** address registers, the 68020 processor is more flexible than the 68HC11 with its two 8-bit accumulators and two sixteen bit index registers. These added registers would make the programming much easier and faster. By giving some of the variables which are most often used their own accumulators, the expanded register set would reduce the need for loads and stores.

The benefits of the register set of the **68020** is outweighed **by** its limited number of ports. Also, the HC1I instruction set is good enough to perform all of the functions for this experiment. It is doubtful that an expanded instruction set would simplify the program.

Due to limits in communication rates with PC's, a **9600** baud rate was chosen for all communications. With the limited number of stations and small packets allowed, this baud rate was enough. However, a network designed with a higher baud rate would be desirable for larger networks. Transmission time through the RS-232 ports takes most of the packet communication time. Compared to the transmission time, the instruction execution was very small. This means that the frames could be sent continuously and processing could occur between frames. However, at higher baud rates this may not be true. Eventually, the HC11 would not be able to keep up with the transmissions and delays would have to be added. Therefore, for higher baud rates a faster microprocessor is desirable.

However, for this experimental network, the 68HClIEVB turned out to be a good choice. Increased knowledge of microcontrollers in general and the 68HC11EVB in particular resulted

from implementing the network. Efficient use of ports, instruction code, and registers were three of the benefits. Also, the importance in timing for networks became apparent. Theoretical research often downplays the importance of timing. Building the network showed just how important timing actually is.

The hardware used to skip a station proved adequate to the situation. The reverser and the multi-port devices performed even better than expected. The outer HC11 worked well as a software driven multiplexer, again justifying its use in this project.

### 722 Protocol Conclusions and Recommendations

With an experimental network such as this, a mixture of many protocols can be achieved. In this case, the **ISDN** frame forrnat was combined with the token ring protocol and a very modified version of the FDDI timing protocol. This turned out to be a good mixture. It supplied all the necessary components to result in a successful network.

Since one of the purposes of this research was to learn more about the protocols used, these recommendations concentrate more on applying the protocols chosen in different ways than in applying other protocols.

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The **ISDN** frame format fulfilled its purpose well. The three channels naturally supplied capability for a large variety of tokens (D channel) and fault tolerance in the form of echoing the data (the B2 channel). The overhead made it a little inefficient, but the benefits far outweighed this.

One of the benefits of **ISDN** is the ability to send to two separate stations simultaneously. This ability could also be applied here. The frame could be broken up into its two channels: B1 going to one station, B2 to another. For example, byte one of B1 in an ISDN frame would carry data from station 1 to station **3.** Byte two of B1 could be used for the negated echo. If less fault tolerance is acceptable, it could be used for a second byte of data. Meanwhile, B2 could perform the same function for a communication being sent from station 2 to station 1. The software for this would be much more complicated. The D channel would have to be split if it were to still hold token information. Two bits would be used for one channel, two for the other. This would limit the variety of tokens allowed and further decrease fault tolerance. However, the ability of ISDN to send several communications simultaneously would be exploited.

In designing and building the token ring, its strengths and weaknesses had to be completely explored. Designing the software

for the network illustrated that the token ring is much more complicated than it sounds. Keeping track of the token and keeping the ring running is difficult. However, once it is running, it is very naturally fault tolerant. The concept of the token proved to be an effective method for keeping the network from dissolving into a freefor-all. The circular nature of the ring creates a serious risk of a byte or frame of data endlessly looping through the network. This *makes* program ing more difficult, but the single direction of the token ring compensated for this. The final organized network made the effort worthwhile.

The timing methods which are very loosely based on the FDDI protocol should prove *very* effective in isolating and overcoming faults. These methods keep the time that a network is down due to a station fault to a minimum. Also, they are effective in keeping any one station from tying up the line. Any station which does not abide **by** the rules can be removed from the network.

These three protocols turned out to be complementary. By taking what was necessary, an effective protocol was developed and implemented. Without a strong protocol, the entire network would have fallen apart.

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# 23 Software Conclusions and Recommendations

This project supplied an excellent opportunity to learn how to write code which must be a strong combination of speed and efficient use of memory.

The limited RAM space of the HC11 and the amount of room that was necessary for the queues made limiting the program a major objective. Loops and sub routines were used wherever possible to decrease the amount of repeated code. The programs turned out to be well within the RAM limits of the HC11. The assembly coding kept the code efficient. However, for future study, programming in a higher level language such as **C** would be less time consuming.

As with any other network, execution speed of the code is important. Most of the code was limited to comands which took very few cycles. Loading and storing accumulators are probably the two most common functions, taking up between three and five cycles depending on addressing used.

Interrupts were kept to a minimum to make the program easier to follow. However, for faster networks interrupt schemes for the ACIA and the host ports of both the inner and outer HC11's should be implemented. This would increase the speed of execution. The timing interrupt would be an excellent method for minimizing the threat of failure. Future research should study the possibility of implementing this timing method. With the 68HC11, the host and ACIA ports have different handshaking requirements; therefore, implementing the timing proved too difficult. Other microcontrollers may not have this problem.

Other than this, the code worked well. The programming was a good learning experience. For example, converting to and from **ISDN** was a wonderful exercise for using the logical shift commands. It was also a good exercise in alternating between the A and B accumulators and the **D** accumulator.

## 7.2.4 Final Remarks

This project turned out to be a good exercise in programming, applying the HCl1 to a situation, designing and building a network, and designing fault tolerant schemes.

The greatest benefits came from studying a network as close as was required. This exercise increased awareness **of** how complicated and fascinating networking methods are.

Also, the great amount of time spent considering fault tolerant

options made the project worthwhile. The experiment led to greater understanding of both the traditional methods of fault tolerance and applying software and hardware (such as the multi-port and the timer interrupts) to come up with new methods.

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Appendix:

Program Listings

\* This is the program for the inner HC <sup>11</sup>



\* Setting constants for various ASCI character codes



\* Setting starting address for the program

ORG \$C000

\* Initializing stack pointers

 $\bar{\bar{\gamma}}$ 



\* **Iniatializing** ACIA port for poling



\* **Clearing** counters and function indicators



\* Setting other **ports**



\* **Setting** basic address to FF **to** start



\* Setting **the stations to** receiving **to** start

LDAA #\$01 STAA RECEIVING \* Startup routine **...** Asks user if they are station 0

- STARTUP EQU \* JSR **CLEARSC** LDY #MSG1 LDAA **#\$01** STAA ERRMESS JSR WAITUP BRA ANSWER  $\sim$
- WAITUP EQU \* JSR PRINTSCREEN **LDAA** ERRMESS **BNE** WAITUP RTS
- \* Clears the screen
- CLEARSC EQU **PSHA** PSHB LDY #CLEAR **LDAA** #\$01 STAA ERRMESS JSR WAITUP PULB **PULA** RTS
- \* Waiting for an answer  $\ldots$  Yes (Y) station 0, or No (N) not station 0.
- \* Otherwise, wait for a valid answer

ANSWER EQU<sup>\*</sup>

**LDAA** ACIA BITA **#\$01** BEQ ANSWER LDAA ACIA+l JSR **CLEARSC** CMPB #'Y' BNE ISNO CLR MYADDRESS **LDAA #\$06 STAA DC**

CLR BYTE1 LDAA #\$FF STAA BYTE2 JSR MAKEFRAME LDAA #\$01 STAA SENDSING BRA **MAIN**

ISNO EQU CMPB #'N'<br>BEQ MAIN MAIN BRA STARTUP

\* Main **loop**

 $\epsilon$ 

MAIN EQU

**\* Test for** input from the **ACIA (from** user)

LDAA ACIA **BITA** #\$01 **BNE INACIA1**

\* **Test** for input from port C **(outer** HC11)



\* **Test to see** if sending a **single** byte of **data**

LDAA SENDSING **BNE** SENDONE

**\*** Test to **see if** sending a packet

LDAA SENDING BITA #\$O1 BNE SEND1

\* Test to **see if in the process of** sending **an** error message

**LDAA ERRMESS BNE** SENDERROR

- \* Test to see if presently sending to the screen from the network. If not,<br>\* all tests done so return, to beginning of leap. Otherwise, output
- \* all tests done so return to beginning of loop. Otherwise, output<br>\* next character to gazen if gazen by for not fall
- next character to screen if screen buffer not full.
	- **LDAA** OUT2SC **BEQ MAIN** LDX TOSCREENT LDAB **0,X** LDAA ACIA BITA #\$02 **BEQ** MAIN **STAB ACIA+1** CMPB #CR
- \* If output to screen is carriage return, add a line feed



- \* Move output to screen queue, if it is empty, signal no longer outputing,
- \* otherwise, continue loop

STX TOSCREENT CPX TOSCREENH **BNE** MAIN CLR OUT2SC BRA MAIN

\* Subroutine to output a line feed



\* Sends program to sub routine which deals with an input from the **ACIA**



- \* Sends program to sub routine which deals with an input from **C,** then
- \* sends message to outer HC11 that data has been received

INC1 EQU \* JSR INC LDAA #\$40 STAA PORTA BRA MAIN

\* Sends program **to** sub **routine** which sends a **single** byte on to network

SENDONE EQU JSR SCHAR BRA MAIN

\* Continues sending an error message

SENDERROR EQU JSR PRINTSCREEN BRA MAIN

\* Sends program **to** sub **routine** which **deals** with **continuing output** of a packet

SEND1 EQU JSR SEND BRA MAIN

\* Sub **routine** to clear **the screen**

CLSC EQU LDAA #\$O1 STAA ERRMESS LDY #CLEAR RTS

\* **Sub** routine which gets an input from the ACIA (User)



\* Test input

CHECK EQU<sup>\*</sup> CMPB #QT BEQ DUMP CMPB #ET BEQ BEGINSEND CMPB #BS



\* If **not a** special character, **test** counter. Too many characters, send **error**

- \* message. Otherwise, add to queue, increment appropriate counters,<br>\* echo character to screen, and roturn to main program
- \* echo character to **screen, and return to main** program.
- ADDTOQ EQU LDAA SCOUNT CMPA #\$FF BEQ ERRORMESS LDX AIHEAD STAB O,X INX STX AIHEAD INC SCOUNT PSC EQU \*
- JSR ECHO CMPB #CR BEQ LINEF RTS
- \* Add a **line** feed **it it is a** character **return being** echoed
- LINEF EQU \* LDAB #LF BRA PSC

\* If **it is** a back space, clear previous character, **echo to screen** decrement

- \* counters, and return **to** main program
- REMOVE EQU<sup>\*</sup> LDX AIHEAD DEX STX AIHEAD DEC SCOUNT JSR ECHO RTS

**ECHO** 

EQU \* LDAA ACIA BITA #\$02 BEQ ECHO STAB ACIA+I RTS

\* If too many characters (>256) input, ready **approp. error** message

ERRORMESS EQU \* LDY #MSG2 LDAA #\$01 STAA **ERESS** JSR PRINTSCREEN RTS

\* **If it is** the **first character in a** message, **it is the destination** address.

DESADD EQU \* SUBB **#\$30** STAB DESTINATION LDAA #\$01 STAA ERRMESS LDY #CLEARIN LDAA #\$08 STAA SENDING RTS

\* If **it is the** quit character, **clear** input queue, counters, and **function**

- \* **indicator, then** return **to** main program
- DUMP EQU \* LDX AITAIL STX AIHEAD CLR SCOUNT CLR SENDING CLR DESTINATION RTS

\* **If it is exit** character, **set the** system up to send the message

BEGINSEND EQU \*

\* If nothing to send, ignore the exit command



\* If the previous message is still being sent, ignore the exit command



- \* If everything is okay, clear counter and move ACIA input queue to
- \* network output queue and vice versa



\* Swaps queue pointers



\* Indicating ready to send in function indicator



\* Send error message is can't send yet

**CANT** EQU LDY **#MSG3**



- \* Make **first** frame with **approp. D,** source address in B1, and destination
- \* address **in** B2. **If** destination address **greater than the** number of
- \* addresses, send **as** broadcast.



\* Make frame, then return **to** main program



\* If function indicator **indicates** that **in** the process of sending, send character

SEND EQU \* LDAB **SENDING** BITB #\$F6 BEQ SCHAR

\* Testing where **station is in** sending

DEALTOKEN EQU<sup>\*</sup>

\* If there is still more of an ISDN frame to send, send next byte



\* If last byte of free token just sent, clear sending function

CMPB #\$03 BEQ THATSALL

\* Otherwise, make free token and prepare to send it



\* Send next byte of ISDN frame

SENDIT EQU<sup>\*</sup>  $LDAB$   $0,X$ INX STX ISDNTAIL JSR HOST RTS

\* When host RS-232 ready, send a byte

HOST EQU JSR DELAY LDAA SCSR BITA #\$80 BEQ HOST STAB SCDR RTS

\* After last byte of free token sent, clear out everything and return to main





\* This sub routine sends ISDN frames from the Host RS-232

SCHAR EQU \*

\* **Load and** send **next** byte **in** ISDN queue

LDX ISDNTAIL LDAB 0,X<br>JSR HOST **HOST** INX STX ISDNTAIL CPX ISDNHEAD BNE OUTTAHERE

\* If there is no more **bytes** in the ISDN frae, **test if** sending **single** frame

LDAA SENDSING BNE ALLDONE

\* **If not sending single** frame, **test if resending**



\* If **not** resending, **get next character and put into** B1

LDX ASTAIL LDAB O,X STAB BYTEl INX CPX ASHEAD BEQ LASTONE

\* If **not last** character **in** queue, **get** next character **in** queue **and put in** B2

 $LDAB$  0,X STAB BYTE2 INX CPX ASHEAD

### BEQ LASTTWO

\* If that **is not** the **last** character, **set** D to 9

LDAB #\$09 STAB DC BRA ALLSET

\* **If** moving back, **set continue** sending

MOVINGBACK EQU \* LDAA #\$01 STAA SENDING JSR MAKEFRAME RTS

\* **If last** character, set B2 to FF and D to 11 (indicating **last** character being **sent)**

LASTONE EQU<sup>\*</sup> LDAB #\$FF STAB BYTE2 LDAB #\$0B STAB DC LDAB #\$05 STAB SENDING BRA ALLSET

\* **If last** two characters, **set** D to **14**

LASTTWO EQU<sup>\*</sup> LDAB #\$OE STAB DC LDAB *#\$05* STAB SENDING

\* Make frame with B1, B2, and D set in above routines, then return to main

ALLSET EQU STX ASTAIL JSR MAKEFRAME RTS

\* When finished sending a single frame, reset pointers **and return to** main

ALLDONE EQU



\* This routine rmakes **BI, B2** and D **into an** ISDN frame format

MAKEFRAME EQU \*

\* First byte of ISDN



- \* Second byte
- TBA LDAB DC BITB #\$08 BEQ OTHER<br>ORAA #\$3A ORAA BRA FBYTE2
- OTHER EQU \* ORAA #\$2A
- FBYTE2 EQU \*
- \* Getting B1 **to put into** second byte of B1

LDAB BYTE1 EORB #\$FF LSRA LSLD JSR STORE

\* Third byte

\* Fourth byte

 $\hat{\mathcal{A}}$ 



\* Fifth byte



OTHER2  $\begin{tabular}{ll} EQU & * \\ ORAA & #\$38 \end{tabular}$ 



\* Last byte



\* Routine **to store** byte into ISDN frame

STORE **EQU**  LDX ISDNHEAD  $STAA$  0,X INX STX **ISDNHEAD** RTS

\* This routine **takes care** of an input from C

INC EQU

\* If **A is 0,** then input **it in B1, B2,** and D **form**

LDAA PORTA BITA #\$O1 BNE ISDNFORM

\* **Load first** byte, **store it in D, and** signal **to** outer **IC11** that data **is** read

LDAA PORTCL CLR PORTA STAA DC BNE COMP1

\* If D **=0 (free** token), **test if** there **is** something **to** send

LDAA SENDING CMPA #\$04 BNE ELSE1

\* If there **is,** make the **first** frame **and begin** sending

```
LDAA #$01
STAA SENDING
JSR MAKEFIRST
RTS
```
\* Otherwise, send **free** token **back** on to **network**

- ELSE1 EQU \* LDAA #\$FF STAA BYTE1 STAA BYTE2 JSR MAKEFRAME LDAA #\$01 STAA SENDSING RTS
- \* If A **not 0, then** the data **being sent is in ISDN** form. Send **it** straight **on,** not for this station



\* **If** MSB **of** D **is a** one, **go to** upper **test** (8 thru **15)**

BITA #\$08 BNE GOCOMP8 CMPA #\$01 BNE COMP3

\* If D **= 1** (busy token), **test if** this **station is** sending. If **it is, an** error **has** occured

LDAA SENDING CMPA #\$01 BEQ NOONEGOT

- \* **If** this **station is not** sending, **it** must be **a** message **for** this **station.**
- \* Load acknowledgement  $(D = 0)$ , this station's address into B2, and
- \* **get source** address from port C (put **into** input from network stack)

LDAA #\$04 STAA DC LDAA MYADDRESS STAA BYTE2 JSR GETBYTE STAA BYTE1 ADDA #\$30 JSR ADDINC RTS

- \* **If noone got** the **first** busy token, **send** it again **and begin resending**
	- \* everything **sent so** far.



- \* Adds source address taken in from C port into appropriate Queue, sets
- \* appropriate **counters, and** sends acknowledgement



\* Goes **to** compare upper **D's**



- \* Takes a byte in from port C
- GETBYTE EQU \* LDAA #\$40 STAA PORTA
- WAIT EQU \* LDAA PIOC BITA #\$80 BEQ WAIT LDAA PORTCL CLR PORTA RTS
- \* If D = 3, sent **too** many bytes. Clear sending queue, send **error** message
- \* to screen, send free token
- COMP3 EQU CMPA #\$03 BNE COMP5 LDY #MSG4 LDAA #\$01 STAA ERRMESS LDX ASTAIL STX ASHEAD STX SASTAIL CLR DC LDAA #\$O1 STAA SENDSING LDAA #\$FF STAA BYTE1 STAA BYTE2 LDX ISDNHEAD CPX ISDNTAIL BNE WAITALLDONE CLR SENDING JSR MAKEFRAME RTS
- \* If ISDN fr **e has** not been completely sent, wait until it **is** finished
- \* **to** stop **sending**

WAITALLDONE EQU LDAA #\$09 STAA SENDING RTS

\* If  $D = 5$ , send error message to screen, then shut down from network

- COMP5 EQU CMPA #\$05 BNE COMP6 LDY #MSG5 WAITDONE EQU<sup>\*</sup> JSR PRINTSCREEN LDAA ERRMESS **BNE WAITDONE END**
- \* If  $D = 6$ , startup token
- COMP6 EQU CMPA #\$06 **BNE** COMP7 JSR GETBYTE STAA BYTE1 **LDAA** MYADDRESS **BEQ** ZERO **LDAA** #\$02 ZERO EQU
	- ORAA **COUNT BEQ** LTCOMP CMPA #\$01 BNE NXCOMP
- \* If this is the second time station 0 has gotten the startup token, send free token
	- **LDAA** #\$00 **STAA DC LDAA** #\$FF **STAA** BYTE1 STAA BYTE2 BRA **ALLDONE2**
- \* If this is the first time a non-0 station has gotten startup, get B1 from
- port C, increment the value, and put that number into address.

NXCOMP EQU<sup>\*</sup>
CMPA #\$02 BNE LTCOMP LDAA BYTE1 INCA STAA BYTE1 STAA MYADDRESS BRA ALLDONE2

\* If second time **non-O** or first time **for station** 0, **get B1** and **store it** as \* the number **of stations**

LTCOMP EQU<sup>\*</sup> LDAA BYTE1 INCA STAA NOSTATS

\* Make a frame and **send** it **out**



\* If  $D = 7$ , broadcast mode. Ready for input

COMP7 EQU LDAA #\$FF STAA BYTE2 JSR GETBYTE STAA BYTE1 ADDA #\$30 JSR ADDINC LDAA #\$01 STAA BROADCAST RTS

\* If D **8,** then get B1 and B2 that were **received** and compare **to what was sent**

COMP8 EQU BITA #\$04 BNE GOCOMP12 CMPA #\$8



\* Once **the last** byte or two **bytes** come back, send **free token**



\* If sent **does** not **equal** received, **send** the error token **and begin resendin** \* from **last** byte **received back**



RTS

MOVEBACK EQU \* LDX SASTAIL LDAA 0,X STAA BYTE1 INX LDAA 0,X<br>STAA BYT BYTE2 INX STX ASTAIL INC ERROR LDX ISDNHEAD CPX ISDNTAIL BNE NOTYET JSR MAKEFRAME RTS

\* **If** in the middle of sending **a** frame, **wait before** resending

NOTYET EQU \* LDAA #\$09 STAA SENDING RTS

\* **If** this **station** is sending, broadcast, **and a** 9 **is** received, **treat** it **as an** <sup>8</sup>



- \* If D **=** 9, **data** being **sent.** Get **B1** and B2 from port C. **If** not in error mode,
- \* add **to** Queue. **If** broadcast, send data out with D = **9, if not,** send
- \* data out with  $D = 8$ .

COMP9 EQU CMPA #\$09 BNE COMP1O LDAA BROADCAST CMPA #\$02 BEQ IMSEND JSR GETBYTE STAA BYTE1 JSR GETBYTE STAA BYTE2



- \* Adds data from port C to input Queue. If too many characters **sent,**
- send error token

**IGNORE** 



\* If D = 10, data **resent.** Begin getting **data** again. Send okay token.

COMP1O EQU CMPA #\$0A BNE COMPil JSR GETBYTE STAA BYTE1 JSR GETBYTE STAA BYTE2 CLR ERROR

- LDAA #\$OF STAA DC JSR ADD2Q JSR MAKEFRAME RTS
- \* If D = **11, sent last** byte
- COMP11 **EQU**  LDAA BROADCAST CMPA #\$02 BEQ IMSEND2 JSR GETBYTE STAA BYTE1 CLR BYTE2 JSR READYOUT RTS

\* **If** I'm sending broadcast, **test** what was **sent to what was** received.

- IMSEND2 EQU CLR BROADCAST JSR GETBYTE STAA B1 JMP COMPARETOSEND
- \* **If** this station **is receiving,** begin outputing **to screen**
- READYOUT EQU<sup>\*</sup> CLR RECEIVING JSR ADD2Q LDAA #\$O1 STAA OUT2SC STAA ERRMESS LDY #CLEARREC LDX INCHEAD STX TOSCREENH LDX INCTAIL STX TOSCREENT CPX QUEUE3 BNE MOVE LDX QUEUE4
- \* Move around queues to begin inputing again, and send okay token

MOVE EQU \* STX INCTAIL STX INCHEAD LDAA BROADCAST BNE FINISH LDAA #\$08 STAA DC

\* Make appropriate token and send it **out.**

FINISH EQU \* JSR MAKEFRAME LDAA #\$01 STAA SENDSING CLR INFRAME CLR BROADCAST RTS

\*  $D = 12$ , begin retransmitting

COMP12 EQU CMPA #\$OC BNE COMP13 LDAA RECEIVING BNE SE LDAA #\$O1 STAA ERROR STAA SENDSING JSR MAKEFRAME RTS

SE EQU \* LDAA #\$OA STAA DC JSR MOVEBACK RTS

\* If D = 13, error. **If** receiving, nove back and **cut** out appropriate characters.

COMP13 EQU CMPA #\$OD BNE COMP14 LDAA RECEIVING BEQ TRANS



- \* **If** transmitting, get **a new** frame together with the number of bad frares **that**
- \* have been **sent**
- TRANS EQU LDD ASTAIL SUBD SASTAIL. STAB BYTE1 LDAA #\$FF STAA BYTE2 LDX SASTAIL STX ASTAIL LDAA #\$09 STAA SENDING RTS

\* If  $D = 14$ , broadcast, and I am sending, stop sending and treat as  $D = 8$ 

- IMSEND3 EQU LDAA #\$08 CLR BROADCAST JMP COMP8
- \* If D **=** 14, get **last** 2 bytes **and finish**
- COMP14 EQU CMPA #\$OE BNE COMP15  $\bar{\chi}$ LDAA BROADCAST CMPA #\$02

BEQ IMSEND3 JSR GETBYTE STAA BYTE1 JSR GETBYTE STAA BYTE2 JSR READYOUT RTS

\* If D = 15, clear **error** and **begin** transmitting **again**

COMP15 EQU CLR ERROR LDAA #\$08 JSR COMP8 RTS

\* If data coming **in** ISDN form, **get** 6 bytes, put into **ISDN** queue, **and** send **it on**

- ISDNFORM1 EQU LDAA #\$40 STAA PORTA LDAA #\$06 STAA COUNT LDX #ISDN STX ISDNHEAD STX ISDNTAIL GETIN EQU<sup>\*</sup> LDAA PIOC BITA #\$80 BEQ GETIN LDAA PORTCL CLR PORTA JSR STORE LDAA #\$40 STAA PORTA DEC COUNT BNE GETIN LDAA #\$01 STAA SENDSING RTS
- \* Outputing message **to screen**

PRINTSCREEN EQU<sup>\*</sup>





\* Short **delay**



\* Variables:





\* Pointers:



\* Setting up bytes necessary for clearing screen:



\* Clear receiving side:





**\*** Clearing input side:





\* Setting up bytes for messages:



**\*** Setting up **roor** for **queues:**



**\* setting** up room for ISDN queue:

ISDN RMB **256**

**\*** Setting up stack for system:

STACK RMB 1

- \* This is the program **for** the outer **HC11's**
- \* Inititializing port addresses:



\* Initializing constants



\* Starting program at \$COOO in memory

ORG \$C000

\* Initializing **stack** pointers



\* Initializing ports:



\* Clearing variables and initializing ports:

LDAA #\$00 STAA SECOND STAA NOSTATS STAA SCCR1 STAA PIOC STAA PORTA

## STAA COUNT STAA DC

Initializing function **to** skip:

LDAA #\$02 STAA FUNCTION

\* Initializing ports:



Clearing ISDN bytes:



- Main loops
- MAIN EQU
- **Test and** input from network



If **all** 6 bytes of **a** frame have **not** been received, loop **back to** main

CMPA #\$06 **BNE MAIN LDAA FUNCTION** CMPA #\$2

\* If sending or receiving, send test input



\* If first bit of  $D = 1$  and skip mode, loop back to main. Otherwise, test  $D$ 

**LDAA** 0,X DEX BITA **#\$10 BEQ** TESTD1 JSR SENDISDN BRA MAIN

\* Get D, test it, etc. Return by cleaning out queue and counter and return to main \* loop



\* Pull D out of ISDN frame and test it to do appropriate functions



\* If  $D = 0$ , Send D only to inner HC11 and put self into skip mode

**COMP0** EQU \* CMPA #\$00 BNE COMPI JSR SENDCHAR LDAA #\$02 STAA FUNCTION RTS

\* If D = 1, get B1 and **82,** then **test to see** if **it** is **to** this station

- COMPI EQU CMPA #01 BNE COMP3 JSR GETBS LDAA BYTE2A CMPA MYADDRESS BNE DUMPIT JSR COMPARE LDAA DC CMPA #\$01 BNE DUMPIT2
- \* If message **is to this station,** send D and source **address** to inner HC <sup>1</sup>

JSR SENDCHAR LDAA BYTE1A JSR SENDCHAR CLR FUNCTION RTS

\* **If** not **to** this **station,** go into skip mode

- DUMPIT EQU \* LDAA #\$02 STAA FUNCTION
- \* And send to inner HC 1 in ISDN frame format



 $*$  If  $D = 3$ , too many characters sent. If sending, go into skip mode and pass \* **Don** to inner HC11.

COMP3 EQU \* LDAA FUNCTION CMPA #\$01 BNE NOTMINE LDAA DC JSR SENDCHAR LDAA #\$02 STAA FUNCTION RTS

- \* If  $D = 4$ , test to see if this station is sending the message. If yes, set to  $*$  sending mode. If no nass it on sending mode. If no, pass it on
- COMP4 EQU \* CMPA #\$04 BNE COMP5 JSR GETBS LDAA BYTE1A CMPA MYADDRESS BNE NOTMINE LDAA #\$01 STAA FUNCTION RTS

\* **If** message does **not concern** this station, **pass it** on as **an** ISDN frame

NOTMINE EQU<sup>\*</sup> JSR SENDISDN LDAA #\$02 STAA FUNCTION RTS

\* If D **= 5, go to** receiving mode

- COMPS EQU CMPA #\$05 BNE COMP6 CLR FUNCTION RTS
- \* **If** startup token, **get B1. If first** time getting 6, increment **and store**
- \* **as my** address. If second time, **increment and store as** number of
- \* stations. **Pass** both on **to** inner Hell

COMP6 EQU

CMPA #\$06 BNE COMP7 JSR GETBS LDAA DC JSR SENDCHAR LDAA BYTE1A JSR SENDCHAR LDAA SECOND BNE SECONDTIME LDAA #\$01 STAA SECOND LDAA BYTE1A INCA STAA MYADDRESS RTS

SECONDTIME EQU<sup>\*</sup> LDAA BYTE1A INCA

\* If number of stations  $=$  my address, this station must be station 0

STAA NOSTATS CMPA MYADDRESS BNE RETURN LDAA #\$OO STAA MYADDRESS STAA SECOND

RETURN EQU RTS

\* If D = 7, test if it belongs **to** me. If it **does,** go **to send** mode

COMP7 EQU JSR GETBS LDAA BYTEIA CMPA MYADDRESS BNE NOTMINE2 LDAA #\$01 STAA FUNCTION BRA THATSIT

\* If **not, go to receive**



\* If  $D = 8$ , test bytes for accuracy. Send D and both B's on to inner Hcl1



GO12 EQU<br>JMP  $\ast$ COMP12

\* If  $D = 9$ , check accuracy then send D, B1 and B2 to inner if okay.

- \* If not, **only** send D
- COMP9 EQU CMPA #\$09 BNE COMP1O JSR GETBS JSR COMPARE LDAA DC JSR SENDCHAR LDAA DC CMPA #\$09

BNE OUTTAHERE LDAA BYTElA JSR SENDCHAR LDAA BYTE2A JSR SENDCHAR

OUTTAHERE EQU<sup>\*</sup> RTS

\* If  $D = 10$ , check accuracy of B's. Send D to inner. If accuracy okay, send both \* **B's to** inner

COMP10 EQU \* CMPA #\$0A BNE COMP11 JSR GETBS JSR COMPARE LDAA DC JSR SENDCHAR CMPA #\$0A BNE OUTTAHERE LDAA BYTE1A JSR SENDCHAR LDAA BYTE2A JSR SENDCHAR RTS

\* If  $D = 11$ , test B's. Send D. Send B1 if okay.



RTS

- \* If  $D = 12$ , send  $D$  to inner and test  $B$ 's. If  $B$ 's okay, send first to inner.
- COMP12 **EQU**  BITA #\$02 **BNE** COMP14 JSR GETBS **JSR** COMPARE **LDAA DC** JSR SENDCHAR CMPA #\$OC **BNE** RET **LDAA** BYTElA **JSR** SENDCHAR RTS
- \* **If** D = 14, test B's. Okay, send all 3. Otherwise, send only **D.**
- \* Go to skip mode
- COMP14 EQU<sup>\*</sup> CMPA #\$OE **BNE** COMP15 JSR GETBS **JSR** COMPARE **LDAA DC** JSR SENDCHAR **LDAA** DC CMPA #\$OE **BNE** GONE **LDAA** BYTElA JSR SENDCHAR **LDAA** BYTE2A JSR SENDCHAR **LDAA** #\$02 STAA FUNCTION

GONE EQU<sup>\*</sup> RTS

\* If D **15,** test B's. okay, send all three, otherwise, send only D

 $\ast$ **COMP15** EQU\* **GETBS JSR** COMPARE LDAA DC JSR **SENDCHAR**

CMPA #\$OF **BNE GONE** LDAA BYTElA JSR SENDCHAR **LDAA** BYTE2A SENDCHAR RTS



\* Does **the** handshaking and sends characters between inner and **outer HC11's**



\* Gets **B1 and** B2 from **the** ISDN frame



- LDD 0,X LSLD LSLD STAA BYTE2A LSRB LSRB TBA INX INX LDAB 0,X LSRD LSRD LSRD STAB BYTE2B RTS
- \* Compares **the first** byte of **B1** to the second byte of **B1 to** make **sure**
- \* **they** are opposites. Does the same **for** B2 bytes. Changes D **if**
- \* there **is** an **error.**
- COMPARE EQU LDAB BYTElA BITB BYTEIB BNE MISTAKE LDAB BYTE2A BITB BYTE2B BNE MISTAKE RTS
- MISTAKE EQU LDAA #\$OC ADDA FUNCTION STAA DC RTS
- \* Pulls the 4 **bits** of D from the ISDN frame
- GETD EQU \* LDX TAIL INX LDAA O,X ANDA #\$10 LSRA STAA DC CLRA

INX INX LDAB O,X ANDB #\$80 LSLD LSLD LSLD ORAA DC STAA DC INX LDAA O,X ANDA #\$1O LSRA LSRA LSRA ORAA DC STAA DC IN X  $LDAA$   $0,X$ ANDA #\$02 LSRA ORAA DC STAA DC RTS

\* **Sends** the 6 **bytes of** ISDN **frame** to inner HC11





\* Variables:



\* Pointers:



**\*** Room **for** ISDN **queue**

