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R. Fielding, Ed.
Adobe
J. Reschke, Ed.
greenbytes
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Hypertext Transfer Protocol (HTTP/1.1): Message Syntax and Routing

Abstract

The Hypertext Transfer Protocol (HTTP) is a stateless application-level protocol for distributed, collaborative, hypertext information systems. This document provides an overview of HTTP architecture and its associated terminology, defines the "http" and "https" Uniform Resource Identifier (URI) schemes, defines the HTTP/1.1 message syntax and parsing requirements, and describes related security concerns for implementations.

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1. Introduction

The Hypertext Transfer Protocol (HTTP) is a stateless application-level request/response protocol that uses extensible semantics and self-descriptive message payloads for flexible interaction with network-based hypertext information systems. This document is the first in a series of documents that collectively form the HTTP/1.1 specification:

1. "Message Syntax and Routing" (this document)
2. "Semantics and Content" [RFC7231]
3. "Conditional Requests" [RFC7232]
4. "Range Requests" [RFC7233]
5. "Caching" [RFC7234]
6. "Authentication" [RFC7235]

This HTTP/1.1 specification obsoletes RFC 2616 and RFC 2145 (on HTTP versioning). This specification also updates the use of CONNECT to establish a tunnel, previously defined in RFC 2817, and defines the "https" URI scheme that was described informally in RFC 2818.

HTTP is a generic interface protocol for information systems. It is designed to hide the details of how a service is implemented by presenting a uniform interface to clients that is independent of the types of resources provided. Likewise, servers do not need to be aware of each client's purpose: an HTTP request can be considered in isolation rather than being associated with a specific type of client or a predetermined sequence of application steps. The result is a protocol that can be used effectively in many different contexts and for which implementations can evolve independently over time.

HTTP is also designed for use as an intermediation protocol for translating communication to and from non-HTTP information systems. HTTP proxies and gateways can provide access to alternative information services by translating their diverse protocols into a hypertext format that can be viewed and manipulated by clients in the same way as HTTP services.

One consequence of this flexibility is that the protocol cannot be defined in terms of what occurs behind the interface. Instead, we are limited to defining the syntax of communication, the intent of received communication, and the expected behavior of recipients. If the communication is considered in isolation, then successful actions

ought to be reflected in corresponding changes to the observable interface provided by servers. However, since multiple clients might act in parallel and perhaps at cross-purposes, we cannot require that such changes be observable beyond the scope of a single response.

This document describes the architectural elements that are used or referred to in HTTP, defines the "http" and "https" URI schemes, describes overall network operation and connection management, and defines HTTP message framing and forwarding requirements. Our goal is to define all of the mechanisms necessary for HTTP message handling that are independent of message semantics, thereby defining the complete set of requirements for message parsers and message-forwarding intermediaries.

1.1. Requirements Notation

The key words "MUST", "MUST NOT", "REQUIRED", "SHALL", "SHALL NOT", "SHOULD", "SHOULD NOT", "RECOMMENDED", "MAY", and "OPTIONAL" in this document are to be interpreted as described in [RFC2119].

Conformance criteria and considerations regarding error handling are defined in Section 2.5.

1.2. Syntax Notation

This specification uses the Augmented Backus-Naur Form (ABNF) notation of [RFC5234] with a list extension, defined in Section 7, that allows for compact definition of comma-separated lists using a '#' operator (similar to how the '*' operator indicates repetition). Appendix B shows the collected grammar with all list operators expanded to standard ABNF notation.

The following core rules are included by reference, as defined in [RFC5234], Appendix B.1: ALPHA (letters), CR (carriage return), CRLF (CR LF), CTL (controls), DIGIT (decimal 0-9), DQUOTE (double quote), HEXDIG (hexadecimal 0-9/A-F/a-f), HTAB (horizontal tab), LF (line feed), OCTET (any 8-bit sequence of data), SP (space), and VCHAR (any visible [USASCII] character).

As a convention, ABNF rule names prefixed with "obs-" denote "obsolete" grammar rules that appear for historical reasons.

2. Architecture

HTTP was created for the World Wide Web (WWW) architecture and has evolved over time to support the scalability needs of a worldwide hypertext system. Much of that architecture is reflected in the terminology and syntax productions used to define HTTP.

2.1. Client/Server Messaging

HTTP is a stateless request/response protocol that operates by exchanging messages (Section 3) across a reliable transport- or session-layer "connection" (Section 6). An HTTP "client" is a program that establishes a connection to a server for the purpose of sending one or more HTTP requests. An HTTP "server" is a program that accepts connections in order to service HTTP requests by sending HTTP responses.

The terms "client" and "server" refer only to the roles that these programs perform for a particular connection. The same program might act as a client on some connections and a server on others. The term "user agent" refers to any of the various client programs that initiate a request, including (but not limited to) browsers, spiders (web-based robots), command-line tools, custom applications, and mobile apps. The term "origin server" refers to the program that can originate authoritative responses for a given target resource. The terms "sender" and "recipient" refer to any implementation that sends or receives a given message, respectively.

HTTP relies upon the Uniform Resource Identifier (URI) standard [RFC3986] to indicate the target resource (Section 5.1) and relationships between resources. Messages are passed in a format similar to that used by Internet mail [RFC5322] and the Multipurpose Internet Mail Extensions (MIME) [RFC2045] (see Appendix A of [RFC7231] for the differences between HTTP and MIME messages).

Most HTTP communication consists of a retrieval request (GET) for a representation of some resource identified by a URI. In the simplest case, this might be accomplished via a single bidirectional connection (===) between the user agent (UA) and the origin server (O).

```
      request  >
UA ===== 0
              <  response
```

A client sends an HTTP request to a server in the form of a request message, beginning with a request-line that includes a method, URI, and protocol version (Section 3.1.1), followed by header fields containing request modifiers, client information, and representation metadata (Section 3.2), an empty line to indicate the end of the header section, and finally a message body containing the payload body (if any, Section 3.3).

A server responds to a client's request by sending one or more HTTP response messages, each beginning with a status line that includes the protocol version, a success or error code, and textual reason phrase (Section 3.1.2), possibly followed by header fields containing server information, resource metadata, and representation metadata (Section 3.2), an empty line to indicate the end of the header section, and finally a message body containing the payload body (if any, Section 3.3).

A connection might be used for multiple request/response exchanges, as defined in Section 6.3.

The following example illustrates a typical message exchange for a GET request (Section 4.3.1 of [RFC7231]) on the URI "http://www.example.com/hello.txt":

Client request:

```
GET /hello.txt HTTP/1.1
User-Agent: curl/7.16.3 libcurl/7.16.3 OpenSSL/0.9.7l zlib/1.2.3
Host: www.example.com
Accept-Language: en, mi
```

Server response:

```
HTTP/1.1 200 OK
Date: Mon, 27 Jul 2009 12:28:53 GMT
Server: Apache
Last-Modified: Wed, 22 Jul 2009 19:15:56 GMT
ETag: "34aa387-d-1568eb00"
Accept-Ranges: bytes
Content-Length: 51
Vary: Accept-Encoding
Content-Type: text/plain
```

Hello World! My payload includes a trailing CRLF.

2.2. Implementation Diversity

When considering the design of HTTP, it is easy to fall into a trap of thinking that all user agents are general-purpose browsers and all origin servers are large public websites. That is not the case in practice. Common HTTP user agents include household appliances, stereos, scales, firmware update scripts, command-line programs, mobile apps, and communication devices in a multitude of shapes and sizes. Likewise, common HTTP origin servers include home automation

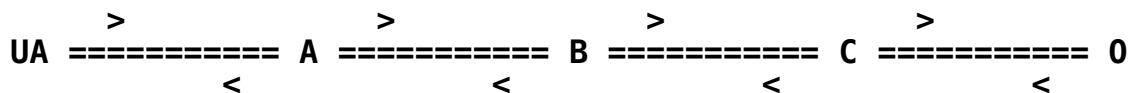
units, configurable networking components, office machines, autonomous robots, news feeds, traffic cameras, ad selectors, and video-delivery platforms.

The term "user agent" does not imply that there is a human user directly interacting with the software agent at the time of a request. In many cases, a user agent is installed or configured to run in the background and save its results for later inspection (or save only a subset of those results that might be interesting or erroneous). Spiders, for example, are typically given a start URI and configured to follow certain behavior while crawling the Web as a hypertext graph.

The implementation diversity of HTTP means that not all user agents can make interactive suggestions to their user or provide adequate warning for security or privacy concerns. In the few cases where this specification requires reporting of errors to the user, it is acceptable for such reporting to only be observable in an error console or log file. Likewise, requirements that an automated action be confirmed by the user before proceeding might be met via advance configuration choices, run-time options, or simple avoidance of the unsafe action; confirmation does not imply any specific user interface or interruption of normal processing if the user has already made that choice.

2.3. Intermediaries

HTTP enables the use of intermediaries to satisfy requests through a chain of connections. There are three common forms of HTTP intermediary: proxy, gateway, and tunnel. In some cases, a single intermediary might act as an origin server, proxy, gateway, or tunnel, switching behavior based on the nature of each request.



The figure above shows three intermediaries (A, B, and C) between the user agent and origin server. A request or response message that travels the whole chain will pass through four separate connections. Some HTTP communication options might apply only to the connection with the nearest, non-tunnel neighbor, only to the endpoints of the chain, or to all connections along the chain. Although the diagram is linear, each participant might be engaged in multiple, simultaneous communications. For example, B might be receiving requests from many clients other than A, and/or forwarding requests to servers other than C, at the same time that it is handling A's

request. Likewise, later requests might be sent through a different path of connections, often based on dynamic configuration for load balancing.

The terms "upstream" and "downstream" are used to describe directional requirements in relation to the message flow: all messages flow from upstream to downstream. The terms "inbound" and "outbound" are used to describe directional requirements in relation to the request route: "inbound" means toward the origin server and "outbound" means toward the user agent.

A "proxy" is a message-forwarding agent that is selected by the client, usually via local configuration rules, to receive requests for some type(s) of absolute URI and attempt to satisfy those requests via translation through the HTTP interface. Some translations are minimal, such as for proxy requests for "http" URIs, whereas other requests might require translation to and from entirely different application-level protocols. Proxies are often used to group an organization's HTTP requests through a common intermediary for the sake of security, annotation services, or shared caching. Some proxies are designed to apply transformations to selected messages or payloads while they are being forwarded, as described in Section 5.7.2.

A "gateway" (a.k.a. "reverse proxy") is an intermediary that acts as an origin server for the outbound connection but translates received requests and forwards them inbound to another server or servers. Gateways are often used to encapsulate legacy or untrusted information services, to improve server performance through "accelerator" caching, and to enable partitioning or load balancing of HTTP services across multiple machines.

All HTTP requirements applicable to an origin server also apply to the outbound communication of a gateway. A gateway communicates with inbound servers using any protocol that it desires, including private extensions to HTTP that are outside the scope of this specification. However, an HTTP-to-HTTP gateway that wishes to interoperate with third-party HTTP servers ought to conform to user agent requirements on the gateway's inbound connection.

A "tunnel" acts as a blind relay between two connections without changing the messages. Once active, a tunnel is not considered a party to the HTTP communication, though the tunnel might have been initiated by an HTTP request. A tunnel ceases to exist when both ends of the relayed connection are closed. Tunnels are used to extend a virtual connection through an intermediary, such as when Transport Layer Security (TLS, [RFC5246]) is used to establish confidential communication through a shared firewall proxy.

The above categories for intermediary only consider those acting as participants in the HTTP communication. There are also intermediaries that can act on lower layers of the network protocol stack, filtering or redirecting HTTP traffic without the knowledge or permission of message senders. Network intermediaries are indistinguishable (at a protocol level) from a man-in-the-middle attack, often introducing security flaws or interoperability problems due to mistakenly violating HTTP semantics.

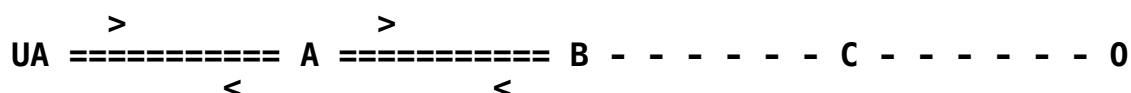
For example, an "interception proxy" [RFC3040] (also commonly known as a "transparent proxy" [RFC1919] or "captive portal") differs from an HTTP proxy because it is not selected by the client. Instead, an interception proxy filters or redirects outgoing TCP port 80 packets (and occasionally other common port traffic). Interception proxies are commonly found on public network access points, as a means of enforcing account subscription prior to allowing use of non-local Internet services, and within corporate firewalls to enforce network usage policies.

HTTP is defined as a stateless protocol, meaning that each request message can be understood in isolation. Many implementations depend on HTTP's stateless design in order to reuse proxied connections or dynamically load balance requests across multiple servers. Hence, a server **MUST NOT** assume that two requests on the same connection are from the same user agent unless the connection is secured and specific to that agent. Some non-standard HTTP extensions (e.g., [RFC4559]) have been known to violate this requirement, resulting in security and interoperability problems.

2.4. Caches

A "cache" is a local store of previous response messages and the subsystem that controls its message storage, retrieval, and deletion. A cache stores cacheable responses in order to reduce the response time and network bandwidth consumption on future, equivalent requests. Any client or server **MAY** employ a cache, though a cache cannot be used by a server while it is acting as a tunnel.

The effect of a cache is that the request/response chain is shortened if one of the participants along the chain has a cached response applicable to that request. The following illustrates the resulting chain if B has a cached copy of an earlier response from O (via C) for a request that has not been cached by UA or A.



A response is "cacheable" if a cache is allowed to store a copy of the response message for use in answering subsequent requests. Even when a response is cacheable, there might be additional constraints placed by the client or by the origin server on when that cached response can be used for a particular request. HTTP requirements for cache behavior and cacheable responses are defined in Section 2 of [RFC7234].

There is a wide variety of architectures and configurations of caches deployed across the World Wide Web and inside large organizations. These include national hierarchies of proxy caches to save transoceanic bandwidth, collaborative systems that broadcast or multicast cache entries, archives of pre-fetched cache entries for use in off-line or high-latency environments, and so on.

2.5. Conformance and Error Handling

This specification targets conformance criteria according to the role of a participant in HTTP communication. Hence, HTTP requirements are placed on senders, recipients, clients, servers, user agents, intermediaries, origin servers, proxies, gateways, or caches, depending on what behavior is being constrained by the requirement. Additional (social) requirements are placed on implementations, resource owners, and protocol element registrations when they apply beyond the scope of a single communication.

The verb "generate" is used instead of "send" where a requirement differentiates between creating a protocol element and merely forwarding a received element downstream.

An implementation is considered conformant if it complies with all of the requirements associated with the roles it partakes in HTTP.

Conformance includes both the syntax and semantics of protocol elements. A sender **MUST NOT** generate protocol elements that convey a meaning that is known by that sender to be false. A sender **MUST NOT** generate protocol elements that do not match the grammar defined by the corresponding ABNF rules. Within a given message, a sender **MUST NOT** generate protocol elements or syntax alternatives that are only allowed to be generated by participants in other roles (i.e., a role that the sender does not have for that message).

When a received protocol element is parsed, the recipient **MUST** be able to parse any value of reasonable length that is applicable to the recipient's role and that matches the grammar defined by the corresponding ABNF rules. Note, however, that some received protocol elements might not be parsed. For example, an intermediary

forwarding a message might parse a header-field into generic field-name and field-value components, but then forward the header field without further parsing inside the field-value.

HTTP does not have specific length limitations for many of its protocol elements because the lengths that might be appropriate will vary widely, depending on the deployment context and purpose of the implementation. Hence, interoperability between senders and recipients depends on shared expectations regarding what is a reasonable length for each protocol element. Furthermore, what is commonly understood to be a reasonable length for some protocol elements has changed over the course of the past two decades of HTTP use and is expected to continue changing in the future.

At a minimum, a recipient **MUST** be able to parse and process protocol element lengths that are at least as long as the values that it generates for those same protocol elements in other messages. For example, an origin server that publishes very long URI references to its own resources needs to be able to parse and process those same references when received as a request target.

A recipient **MUST** interpret a received protocol element according to the semantics defined for it by this specification, including extensions to this specification, unless the recipient has determined (through experience or configuration) that the sender incorrectly implements what is implied by those semantics. For example, an origin server might disregard the contents of a received Accept-Encoding header field if inspection of the User-Agent header field indicates a specific implementation version that is known to fail on receipt of certain content codings.

Unless noted otherwise, a recipient **MAY** attempt to recover a usable protocol element from an invalid construct. HTTP does not define specific error handling mechanisms except when they have a direct impact on security, since different applications of the protocol require different error handling strategies. For example, a Web browser might wish to transparently recover from a response where the Location header field doesn't parse according to the ABNF, whereas a systems control client might consider any form of error recovery to be dangerous.

2.6. Protocol Versioning

HTTP uses a "<major>.<minor>" numbering scheme to indicate versions of the protocol. This specification defines version "1.1". The protocol version as a whole indicates the sender's conformance with the set of requirements laid out in that version's corresponding specification of HTTP.

The version of an HTTP message is indicated by an HTTP-version field in the first line of the message. HTTP-version is case-sensitive.

```
HTTP-version = HTTP-name "/" DIGIT "." DIGIT
HTTP-name    = %x48.54.54.50 ; "HTTP", case-sensitive
```

The HTTP version number consists of two decimal digits separated by a "." (period or decimal point). The first digit ("major version") indicates the HTTP messaging syntax, whereas the second digit ("minor version") indicates the highest minor version within that major version to which the sender is conformant and able to understand for future communication. The minor version advertises the sender's communication capabilities even when the sender is only using a backwards-compatible subset of the protocol, thereby letting the recipient know that more advanced features can be used in response (by servers) or in future requests (by clients).

When an HTTP/1.1 message is sent to an HTTP/1.0 recipient [RFC1945] or a recipient whose version is unknown, the HTTP/1.1 message is constructed such that it can be interpreted as a valid HTTP/1.0 message if all of the newer features are ignored. This specification places recipient-version requirements on some new features so that a conformant sender will only use compatible features until it has determined, through configuration or the receipt of a message, that the recipient supports HTTP/1.1.

The interpretation of a header field does not change between minor versions of the same major HTTP version, though the default behavior of a recipient in the absence of such a field can change. Unless specified otherwise, header fields defined in HTTP/1.1 are defined for all versions of HTTP/1.x. In particular, the Host and Connection header fields ought to be implemented by all HTTP/1.x implementations whether or not they advertise conformance with HTTP/1.1.

New header fields can be introduced without changing the protocol version if their defined semantics allow them to be safely ignored by recipients that do not recognize them. Header field extensibility is discussed in Section 3.2.1.

Intermediaries that process HTTP messages (i.e., all intermediaries other than those acting as tunnels) **MUST** send their own HTTP-version in forwarded messages. In other words, they are not allowed to blindly forward the first line of an HTTP message without ensuring that the protocol version in that message matches a version to which that intermediary is conformant for both the receiving and sending of messages. Forwarding an HTTP message without rewriting the

HTTP-version might result in communication errors when downstream recipients use the message sender's version to determine what features are safe to use for later communication with that sender.

A client **SHOULD** send a request version equal to the highest version to which the client is conformant and whose major version is no higher than the highest version supported by the server, if this is known. A client **MUST NOT** send a version to which it is not conformant.

A client **MAY** send a lower request version if it is known that the server incorrectly implements the HTTP specification, but only after the client has attempted at least one normal request and determined from the response status code or header fields (e.g., Server) that the server improperly handles higher request versions.

A server **SHOULD** send a response version equal to the highest version to which the server is conformant that has a major version less than or equal to the one received in the request. A server **MUST NOT** send a version to which it is not conformant. A server can send a 505 (HTTP Version Not Supported) response if it wishes, for any reason, to refuse service of the client's major protocol version.

A server **MAY** send an HTTP/1.0 response to a request if it is known or suspected that the client incorrectly implements the HTTP specification and is incapable of correctly processing later version responses, such as when a client fails to parse the version number correctly or when an intermediary is known to blindly forward the HTTP-version even when it doesn't conform to the given minor version of the protocol. Such protocol downgrades **SHOULD NOT** be performed unless triggered by specific client attributes, such as when one or more of the request header fields (e.g., User-Agent) uniquely match the values sent by a client known to be in error.

The intention of HTTP's versioning design is that the major number will only be incremented if an incompatible message syntax is introduced, and that the minor number will only be incremented when changes made to the protocol have the effect of adding to the message semantics or implying additional capabilities of the sender. However, the minor version was not incremented for the changes introduced between [RFC2068] and [RFC2616], and this revision has specifically avoided any such changes to the protocol.

When an HTTP message is received with a major version number that the recipient implements, but a higher minor version number than what the recipient implements, the recipient **SHOULD** process the message as if it were in the highest minor version within that major version to which the recipient is conformant. A recipient can assume that a

message with a higher minor version, when sent to a recipient that has not yet indicated support for that higher version, is sufficiently backwards-compatible to be safely processed by any implementation of the same major version.

2.7. Uniform Resource Identifiers

Uniform Resource Identifiers (URIs) [RFC3986] are used throughout HTTP as the means for identifying resources (Section 2 of [RFC7231]). URI references are used to target requests, indicate redirects, and define relationships.

The definitions of "URI-reference", "absolute-URI", "relative-part", "scheme", "authority", "port", "host", "path-abempty", "segment", "query", and "fragment" are adopted from the URI generic syntax. An "absolute-path" rule is defined for protocol elements that can contain a non-empty path component. (This rule differs slightly from the path-abempty rule of RFC 3986, which allows for an empty path to be used in references, and path-absolute rule, which does not allow paths that begin with "///".) A "partial-URI" rule is defined for protocol elements that can contain a relative URI but not a fragment component.

```
URI-reference = <URI-reference, see [RFC3986], Section 4.1>
absolute-URI  = <absolute-URI, see [RFC3986], Section 4.3>
relative-part = <relative-part, see [RFC3986], Section 4.2>
scheme        = <scheme, see [RFC3986], Section 3.1>
authority     = <authority, see [RFC3986], Section 3.2>
uri-host      = <host, see [RFC3986], Section 3.2.2>
port          = <port, see [RFC3986], Section 3.2.3>
path-abempty  = <path-abempty, see [RFC3986], Section 3.3>
segment       = <segment, see [RFC3986], Section 3.3>
query         = <query, see [RFC3986], Section 3.4>
fragment      = <fragment, see [RFC3986], Section 3.5>

absolute-path = 1*( "/" segment )
partial-URI   = relative-part [ "?" query ]
```

Each protocol element in HTTP that allows a URI reference will indicate in its ABNF production whether the element allows any form of reference (URI-reference), only a URI in absolute form (absolute-URI), only the path and optional query components, or some combination of the above. Unless otherwise indicated, URI references are parsed relative to the effective request URI (Section 5.5).

2.7.1. http URI Scheme

The "http" URI scheme is hereby defined for the purpose of minting identifiers according to their association with the hierarchical namespace governed by a potential HTTP origin server listening for TCP ([RFC0793]) connections on a given port.

```
http-URI = "http:" "://" authority path-abempty [ "?" query ]  
          [ "#" fragment ]
```

The origin server for an "http" URI is identified by the authority component, which includes a host identifier and optional TCP port ([RFC3986], Section 3.2.2). The hierarchical path component and optional query component serve as an identifier for a potential target resource within that origin server's name space. The optional fragment component allows for indirect identification of a secondary resource, independent of the URI scheme, as defined in Section 3.5 of [RFC3986].

A sender **MUST NOT** generate an "http" URI with an empty host identifier. A recipient that processes such a URI reference **MUST** reject it as invalid.

If the host identifier is provided as an IP address, the origin server is the listener (if any) on the indicated TCP port at that IP address. If host is a registered name, the registered name is an indirect identifier for use with a name resolution service, such as DNS, to find an address for that origin server. If the port subcomponent is empty or not given, TCP port 80 (the reserved port for WWW services) is the default.

Note that the presence of a URI with a given authority component does not imply that there is always an HTTP server listening for connections on that host and port. Anyone can mint a URI. What the authority component determines is who has the right to respond authoritatively to requests that target the identified resource. The delegated nature of registered names and IP addresses creates a federated namespace, based on control over the indicated host and port, whether or not an HTTP server is present. See Section 9.1 for security considerations related to establishing authority.

When an "http" URI is used within a context that calls for access to the indicated resource, a client **MAY** attempt access by resolving the host to an IP address, establishing a TCP connection to that address on the indicated port, and sending an HTTP request message (Section 3) containing the URI's identifying data (Section 5) to the server. If the server responds to that request with a non-interim

HTTP response message, as described in Section 6 of [RFC7231], then that response is considered an authoritative answer to the client's request.

Although HTTP is independent of the transport protocol, the "http" scheme is specific to TCP-based services because the name delegation process depends on TCP for establishing authority. An HTTP service based on some other underlying connection protocol would presumably be identified using a different URI scheme, just as the "https" scheme (below) is used for resources that require an end-to-end secured connection. Other protocols might also be used to provide access to "http" identified resources -- it is only the authoritative interface that is specific to TCP.

The URI generic syntax for authority also includes a deprecated userinfo subcomponent ([RFC3986], Section 3.2.1) for including user authentication information in the URI. Some implementations make use of the userinfo component for internal configuration of authentication information, such as within command invocation options, configuration files, or bookmark lists, even though such usage might expose a user identifier or password. A sender **MUST NOT** generate the userinfo subcomponent (and its "@" delimiter) when an "http" URI reference is generated within a message as a request target or header field value. Before making use of an "http" URI reference received from an untrusted source, a recipient **SHOULD** parse for userinfo and treat its presence as an error; it is likely being used to obscure the authority for the sake of phishing attacks.

2.7.2. https URI Scheme

The "https" URI scheme is hereby defined for the purpose of minting identifiers according to their association with the hierarchical namespace governed by a potential HTTP origin server listening to a given TCP port for TLS-secured connections ([RFC5246]).

All of the requirements listed above for the "http" scheme are also requirements for the "https" scheme, except that TCP port 443 is the default if the port subcomponent is empty or not given, and the user agent **MUST** ensure that its connection to the origin server is secured through the use of strong encryption, end-to-end, prior to sending the first HTTP request.

```
https-URI = "https:" "://" authority path-abempty [ "?" query ]  
           [ "#" fragment ]
```

Note that the "https" URI scheme depends on both TLS and TCP for establishing authority. Resources made available via the "https" scheme have no shared identity with the "http" scheme even if their

resource identifiers indicate the same authority (the same host listening to the same TCP port). They are distinct namespaces and are considered to be distinct origin servers. However, an extension to HTTP that is defined to apply to entire host domains, such as the Cookie protocol [RFC6265], can allow information set by one service to impact communication with other services within a matching group of host domains.

The process for authoritative access to an "https" identified resource is defined in [RFC2818].

2.7.3. http and https URI Normalization and Comparison

Since the "http" and "https" schemes conform to the URI generic syntax, such URIs are normalized and compared according to the algorithm defined in Section 6 of [RFC3986], using the defaults described above for each scheme.

If the port is equal to the default port for a scheme, the normal form is to omit the port subcomponent. When not being used in absolute form as the request target of an OPTIONS request, an empty path component is equivalent to an absolute path of "/", so the normal form is to provide a path of "/" instead. The scheme and host are case-insensitive and normally provided in lowercase; all other components are compared in a case-sensitive manner. Characters other than those in the "reserved" set are equivalent to their percent-encoded octets: the normal form is to not encode them (see Sections 2.1 and 2.2 of [RFC3986]).

For example, the following three URIs are equivalent:

```
http://example.com:80/~smith/home.html
http://EXAMPLE.com/%7Esmith/home.html
http://EXAMPLE.com:/%7esmith/home.html
```

3. Message Format

All HTTP/1.1 messages consist of a start-line followed by a sequence of octets in a format similar to the Internet Message Format [RFC5322]: zero or more header fields (collectively referred to as the "headers" or the "header section"), an empty line indicating the end of the header section, and an optional message body.

```
HTTP-message  = start-line
                *( header-field CRLF )
                CRLF
                [ message-body ]
```

The normal procedure for parsing an HTTP message is to read the start-line into a structure, read each header field into a hash table by field name until the empty line, and then use the parsed data to determine if a message body is expected. If a message body has been indicated, then it is read as a stream until an amount of octets equal to the message body length is read or the connection is closed.

A recipient **MUST** parse an HTTP message as a sequence of octets in an encoding that is a superset of US-ASCII [USASCII]. Parsing an HTTP message as a stream of Unicode characters, without regard for the specific encoding, creates security vulnerabilities due to the varying ways that string processing libraries handle invalid multibyte character sequences that contain the octet LF (%x0A). String-based parsers can only be safely used within protocol elements after the element has been extracted from the message, such as within a header field-value after message parsing has delineated the individual fields.

An HTTP message can be parsed as a stream for incremental processing or forwarding downstream. However, recipients cannot rely on incremental delivery of partial messages, since some implementations will buffer or delay message forwarding for the sake of network efficiency, security checks, or payload transformations.

A sender **MUST NOT** send whitespace between the start-line and the first header field. A recipient that receives whitespace between the start-line and the first header field **MUST** either reject the message as invalid or consume each whitespace-preceded line without further processing of it (i.e., ignore the entire line, along with any subsequent lines preceded by whitespace, until a properly formed header field is received or the header section is terminated).

The presence of such whitespace in a request might be an attempt to trick a server into ignoring that field or processing the line after it as a new request, either of which might result in a security vulnerability if other implementations within the request chain interpret the same message differently. Likewise, the presence of such whitespace in a response might be ignored by some clients or cause others to cease parsing.

3.1. Start Line

An HTTP message can be either a request from client to server or a response from server to client. Syntactically, the two types of message differ only in the start-line, which is either a request-line (for requests) or a status-line (for responses), and in the algorithm for determining the length of the message body (Section 3.3).

In theory, a client could receive requests and a server could receive responses, distinguishing them by their different start-line formats, but, in practice, servers are implemented to only expect a request (a response is interpreted as an unknown or invalid request method) and clients are implemented to only expect a response.

start-line = request-line / status-line

3.1.1. Request Line

A request-line begins with a method token, followed by a single space (SP), the request-target, another single space (SP), the protocol version, and ends with CRLF.

request-line = method SP request-target SP HTTP-version CRLF

The method token indicates the request method to be performed on the target resource. The request method is case-sensitive.

method = token

The request methods defined by this specification can be found in Section 4 of [RFC7231], along with information regarding the HTTP method registry and considerations for defining new methods.

The request-target identifies the target resource upon which to apply the request, as defined in Section 5.3.

Recipients typically parse the request-line into its component parts by splitting on whitespace (see Section 3.5), since no whitespace is allowed in the three components. Unfortunately, some user agents fail to properly encode or exclude whitespace found in hypertext references, resulting in those disallowed characters being sent in a request-target.

Recipients of an invalid request-line SHOULD respond with either a 400 (Bad Request) error or a 301 (Moved Permanently) redirect with the request-target properly encoded. A recipient SHOULD NOT attempt to autocorrect and then process the request without a redirect, since the invalid request-line might be deliberately crafted to bypass security filters along the request chain.

HTTP does not place a predefined limit on the length of a request-line, as described in Section 2.5. A server that receives a method longer than any that it implements SHOULD respond with a 501 (Not Implemented) status code. A server that receives a

request-target longer than any URI it wishes to parse MUST respond with a 414 (URI Too Long) status code (see Section 6.5.12 of [RFC7231]).

Various ad hoc limitations on request-line length are found in practice. It is RECOMMENDED that all HTTP senders and recipients support, at a minimum, request-line lengths of 8000 octets.

3.1.2. Status Line

The first line of a response message is the status-line, consisting of the protocol version, a space (SP), the status code, another space, a possibly empty textual phrase describing the status code, and ending with CRLF.

status-line = HTTP-version SP status-code SP reason-phrase CRLF

The status-code element is a 3-digit integer code describing the result of the server's attempt to understand and satisfy the client's corresponding request. The rest of the response message is to be interpreted in light of the semantics defined for that status code. See Section 6 of [RFC7231] for information about the semantics of status codes, including the classes of status code (indicated by the first digit), the status codes defined by this specification, considerations for the definition of new status codes, and the IANA registry.

status-code = 3DIGIT

The reason-phrase element exists for the sole purpose of providing a textual description associated with the numeric status code, mostly out of deference to earlier Internet application protocols that were more frequently used with interactive text clients. A client SHOULD ignore the reason-phrase content.

reason-phrase = *(HTAB / SP / VCHAR / obs-text)

3.2. Header Fields

Each header field consists of a case-insensitive field name followed by a colon (":"), optional leading whitespace, the field value, and optional trailing whitespace.

```
header-field    = field-name ":" OWS field-value OWS

field-name      = token
field-value     = *( field-content / obs-fold )
field-content   = field-vchar [ 1*( SP / HTAB ) field-vchar ]
field-vchar     = VCHAR / obs-text

obs-fold       = CRLF 1*( SP / HTAB )
                ; obsolete line folding
                ; see Section 3.2.4
```

The field-name token labels the corresponding field-value as having the semantics defined by that header field. For example, the Date header field is defined in Section 7.1.1.2 of [RFC7231] as containing the origination timestamp for the message in which it appears.

3.2.1. Field Extensibility

Header fields are fully extensible: there is no limit on the introduction of new field names, each presumably defining new semantics, nor on the number of header fields used in a given message. Existing fields are defined in each part of this specification and in many other specifications outside this document set.

New header fields can be defined such that, when they are understood by a recipient, they might override or enhance the interpretation of previously defined header fields, define preconditions on request evaluation, or refine the meaning of responses.

A proxy **MUST** forward unrecognized header fields unless the field-name is listed in the Connection header field (Section 6.1) or the proxy is specifically configured to block, or otherwise transform, such fields. Other recipients **SHOULD** ignore unrecognized header fields. These requirements allow HTTP's functionality to be enhanced without requiring prior update of deployed intermediaries.

All defined header fields ought to be registered with IANA in the "Message Headers" registry, as described in Section 8.3 of [RFC7231].

3.2.2. Field Order

The order in which header fields with differing field names are received is not significant. However, it is good practice to send header fields that contain control data first, such as Host on requests and Date on responses, so that implementations can decide when not to handle a message as early as possible. A server **MUST NOT** apply a request to the target resource until the entire request

header section is received, since later header fields might include conditionals, authentication credentials, or deliberately misleading duplicate header fields that would impact request processing.

A sender **MUST NOT** generate multiple header fields with the same field name in a message unless either the entire field value for that header field is defined as a comma-separated list [i.e., #(values)] or the header field is a well-known exception (as noted below).

A recipient **MAY** combine multiple header fields with the same field name into one "field-name: field-value" pair, without changing the semantics of the message, by appending each subsequent field value to the combined field value in order, separated by a comma. The order in which header fields with the same field name are received is therefore significant to the interpretation of the combined field value; a proxy **MUST NOT** change the order of these field values when forwarding a message.

Note: In practice, the "Set-Cookie" header field ([RFC6265]) often appears multiple times in a response message and does not use the list syntax, violating the above requirements on multiple header fields with the same name. Since it cannot be combined into a single field-value, recipients ought to handle "Set-Cookie" as a special case while processing header fields. (See Appendix A.2.3 of [Kri2001] for details.)

3.2.3. Whitespace

This specification uses three rules to denote the use of linear whitespace: OWS (optional whitespace), RWS (required whitespace), and BWS ("bad" whitespace).

The OWS rule is used where zero or more linear whitespace octets might appear. For protocol elements where optional whitespace is preferred to improve readability, a sender **SHOULD** generate the optional whitespace as a single SP; otherwise, a sender **SHOULD NOT** generate optional whitespace except as needed to white out invalid or unwanted protocol elements during in-place message filtering.

The RWS rule is used when at least one linear whitespace octet is required to separate field tokens. A sender **SHOULD** generate RWS as a single SP.

The BWS rule is used where the grammar allows optional whitespace only for historical reasons. A sender **MUST NOT** generate BWS in messages. A recipient **MUST** parse for such bad whitespace and remove it before interpreting the protocol element.


```
OWS          = *( SP / HTAB )  
              ; optional whitespace  
RWS          = 1*( SP / HTAB )  
              ; required whitespace  
BWS          = OWS  
              ; "bad" whitespace
```

3.2.4. Field Parsing

Messages are parsed using a generic algorithm, independent of the individual header field names. The contents within a given field value are not parsed until a later stage of message interpretation (usually after the message's entire header section has been processed). Consequently, this specification does not use ABNF rules to define each "Field-Name: Field Value" pair, as was done in previous editions. Instead, this specification uses ABNF rules that are named according to each registered field name, wherein the rule defines the valid grammar for that field's corresponding field values (i.e., after the field-value has been extracted from the header section by a generic field parser).

No whitespace is allowed between the header field-name and colon. In the past, differences in the handling of such whitespace have led to security vulnerabilities in request routing and response handling. A server **MUST** reject any received request message that contains whitespace between a header field-name and colon with a response code of 400 (Bad Request). A proxy **MUST** remove any such whitespace from a response message before forwarding the message downstream.

A field value might be preceded and/or followed by optional whitespace (OWS); a single SP preceding the field-value is preferred for consistent readability by humans. The field value does not include any leading or trailing whitespace: OWS occurring before the first non-whitespace octet of the field value or after the last non-whitespace octet of the field value ought to be excluded by parsers when extracting the field value from a header field.

Historically, HTTP header field values could be extended over multiple lines by preceding each extra line with at least one space or horizontal tab (obs-fold). This specification deprecates such line folding except within the message/http media type (Section 8.3.1). A sender **MUST NOT** generate a message that includes line folding (i.e., that has any field-value that contains a match to the obs-fold rule) unless the message is intended for packaging within the message/http media type.

A server that receives an obs-fold in a request message that is not within a message/http container **MUST** either reject the message by sending a 400 (Bad Request), preferably with a representation explaining that obsolete line folding is unacceptable, or replace each received obs-fold with one or more SP octets prior to interpreting the field value or forwarding the message downstream.

A proxy or gateway that receives an obs-fold in a response message that is not within a message/http container **MUST** either discard the message and replace it with a 502 (Bad Gateway) response, preferably with a representation explaining that unacceptable line folding was received, or replace each received obs-fold with one or more SP octets prior to interpreting the field value or forwarding the message downstream.

A user agent that receives an obs-fold in a response message that is not within a message/http container **MUST** replace each received obs-fold with one or more SP octets prior to interpreting the field value.

Historically, HTTP has allowed field content with text in the ISO-8859-1 charset [ISO-8859-1], supporting other charsets only through use of [RFC2047] encoding. In practice, most HTTP header field values use only a subset of the US-ASCII charset [USASCII]. Newly defined header fields **SHOULD** limit their field values to US-ASCII octets. A recipient **SHOULD** treat other octets in field content (obs-text) as opaque data.

3.2.5. Field Limits

HTTP does not place a predefined limit on the length of each header field or on the length of the header section as a whole, as described in Section 2.5. Various ad hoc limitations on individual header field length are found in practice, often depending on the specific field semantics.

A server that receives a request header field, or set of fields, larger than it wishes to process **MUST** respond with an appropriate 4xx (Client Error) status code. Ignoring such header fields would increase the server's vulnerability to request smuggling attacks (Section 9.5).

A client **MAY** discard or truncate received header fields that are larger than the client wishes to process if the field semantics are such that the dropped value(s) can be safely ignored without changing the message framing or response semantics.

3.2.6. Field Value Components

Most HTTP header field values are defined using common syntax components (token, quoted-string, and comment) separated by whitespace or specific delimiting characters. Delimiters are chosen from the set of US-ASCII visual characters not allowed in a token (DQUOTE and "(),/:;<=>?@[\\]{}").

```
token           = 1*tchar

tchar           = "!" / "#" / "$" / "%" / "&" / "'" / "*"
                  / "+" / "-" / "." / "^" / "_" / "`" / "|" / "~"
                  / DIGIT / ALPHA
                  ; any VCHAR, except delimiters
```

A string of text is parsed as a single value if it is quoted using double-quote marks.

```
quoted-string   = DQUOTE *( qdtext / quoted-pair ) DQUOTE
qdtext          = HTAB / SP / %x21 / %x23-5B / %x5D-7E / obs-text
obs-text        = %x80-FF
```

Comments can be included in some HTTP header fields by surrounding the comment text with parentheses. Comments are only allowed in fields containing "comment" as part of their field value definition.

```
comment         = "(" *( ctext / quoted-pair / comment ) ")"
ctext           = HTAB / SP / %x21-27 / %x2A-5B / %x5D-7E / obs-text
```

The backslash octet ("\") can be used as a single-octet quoting mechanism within quoted-string and comment constructs. Recipients that process the value of a quoted-string MUST handle a quoted-pair as if it were replaced by the octet following the backslash.

```
quoted-pair     = "\" ( HTAB / SP / VCHAR / obs-text )
```

A sender SHOULD NOT generate a quoted-pair in a quoted-string except where necessary to quote DQUOTE and backslash octets occurring within that string. A sender SHOULD NOT generate a quoted-pair in a comment except where necessary to quote parentheses ["(" and ")"] and backslash octets occurring within that comment.

3.3. Message Body

The message body (if any) of an HTTP message is used to carry the payload body of that request or response. The message body is identical to the payload body unless a transfer coding has been applied, as described in Section 3.3.1.

message-body = *OCTET

The rules for when a message body is allowed in a message differ for requests and responses.

The presence of a message body in a request is signaled by a Content-Length or Transfer-Encoding header field. Request message framing is independent of method semantics, even if the method does not define any use for a message body.

The presence of a message body in a response depends on both the request method to which it is responding and the response status code (Section 3.1.2). Responses to the HEAD request method (Section 4.3.2 of [RFC7231]) never include a message body because the associated response header fields (e.g., Transfer-Encoding, Content-Length, etc.), if present, indicate only what their values would have been if the request method had been GET (Section 4.3.1 of [RFC7231]). 2xx (Successful) responses to a CONNECT request method (Section 4.3.6 of [RFC7231]) switch to tunnel mode instead of having a message body. All 1xx (Informational), 204 (No Content), and 304 (Not Modified) responses do not include a message body. All other responses do include a message body, although the body might be of zero length.

3.3.1. Transfer-Encoding

The Transfer-Encoding header field lists the transfer coding names corresponding to the sequence of transfer codings that have been (or will be) applied to the payload body in order to form the message body. Transfer codings are defined in Section 4.

Transfer-Encoding = 1#transfer-coding

Transfer-Encoding is analogous to the Content-Transfer-Encoding field of MIME, which was designed to enable safe transport of binary data over a 7-bit transport service ([RFC2045], Section 6). However, safe transport has a different focus for an 8bit-clean transfer protocol. In HTTP's case, Transfer-Encoding is primarily intended to accurately delimit a dynamically generated payload and to distinguish payload encodings that are only applied for transport efficiency or security from those that are characteristics of the selected resource.

A recipient **MUST** be able to parse the chunked transfer coding (Section 4.1) because it plays a crucial role in framing messages when the payload body size is not known in advance. A sender **MUST NOT** apply chunked more than once to a message body (i.e., chunking an already chunked message is not allowed). If any transfer coding other than chunked is applied to a request payload body, the sender **MUST** apply chunked as the final transfer coding to ensure that the message is properly framed. If any transfer coding other than chunked is applied to a response payload body, the sender **MUST** either apply chunked as the final transfer coding or terminate the message by closing the connection.

For example,

Transfer-Encoding: gzip, chunked

indicates that the payload body has been compressed using the gzip coding and then chunked using the chunked coding while forming the message body.

Unlike Content-Encoding (Section 3.1.2.1 of [RFC7231]), Transfer-Encoding is a property of the message, not of the representation, and any recipient along the request/response chain **MAY** decode the received transfer coding(s) or apply additional transfer coding(s) to the message body, assuming that corresponding changes are made to the Transfer-Encoding field-value. Additional information about the encoding parameters can be provided by other header fields not defined by this specification.

Transfer-Encoding **MAY** be sent in a response to a HEAD request or in a 304 (Not Modified) response (Section 4.1 of [RFC7232]) to a GET request, neither of which includes a message body, to indicate that the origin server would have applied a transfer coding to the message body if the request had been an unconditional GET. This indication is not required, however, because any recipient on the response chain (including the origin server) can remove transfer codings when they are not needed.

A server **MUST NOT** send a Transfer-Encoding header field in any response with a status code of 1xx (Informational) or 204 (No Content). A server **MUST NOT** send a Transfer-Encoding header field in any 2xx (Successful) response to a CONNECT request (Section 4.3.6 of [RFC7231]).

Transfer-Encoding was added in HTTP/1.1. It is generally assumed that implementations advertising only HTTP/1.0 support will not understand how to process a transfer-encoded payload. A client **MUST NOT** send a request containing Transfer-Encoding unless it knows the

server will handle HTTP/1.1 (or later) requests; such knowledge might be in the form of specific user configuration or by remembering the version of a prior received response. A server **MUST NOT** send a response containing Transfer-Encoding unless the corresponding request indicates HTTP/1.1 (or later).

A server that receives a request message with a transfer coding it does not understand **SHOULD** respond with 501 (Not Implemented).

3.3.2. Content-Length

When a message does not have a Transfer-Encoding header field, a Content-Length header field can provide the anticipated size, as a decimal number of octets, for a potential payload body. For messages that do include a payload body, the Content-Length field-value provides the framing information necessary for determining where the body (and message) ends. For messages that do not include a payload body, the Content-Length indicates the size of the selected representation (Section 3 of [RFC7231]).

Content-Length = 1*DIGIT

An example is

Content-Length: 3495

A sender **MUST NOT** send a Content-Length header field in any message that contains a Transfer-Encoding header field.

A user agent **SHOULD** send a Content-Length in a request message when no Transfer-Encoding is sent and the request method defines a meaning for an enclosed payload body. For example, a Content-Length header field is normally sent in a POST request even when the value is 0 (indicating an empty payload body). A user agent **SHOULD NOT** send a Content-Length header field when the request message does not contain a payload body and the method semantics do not anticipate such a body.

A server **MAY** send a Content-Length header field in a response to a HEAD request (Section 4.3.2 of [RFC7231]); a server **MUST NOT** send Content-Length in such a response unless its field-value equals the decimal number of octets that would have been sent in the payload body of a response if the same request had used the GET method.

A server **MAY** send a Content-Length header field in a 304 (Not Modified) response to a conditional GET request (Section 4.1 of [RFC7232]); a server **MUST NOT** send Content-Length in such a response

unless its field-value equals the decimal number of octets that would have been sent in the payload body of a 200 (OK) response to the same request.

A server **MUST NOT** send a Content-Length header field in any response with a status code of 1xx (Informational) or 204 (No Content). A server **MUST NOT** send a Content-Length header field in any 2xx (Successful) response to a CONNECT request (Section 4.3.6 of [RFC7231]).

Aside from the cases defined above, in the absence of Transfer-Encoding, an origin server **SHOULD** send a Content-Length header field when the payload body size is known prior to sending the complete header section. This will allow downstream recipients to measure transfer progress, know when a received message is complete, and potentially reuse the connection for additional requests.

Any Content-Length field value greater than or equal to zero is valid. Since there is no predefined limit to the length of a payload, a recipient **MUST** anticipate potentially large decimal numerals and prevent parsing errors due to integer conversion overflows (Section 9.3).

If a message is received that has multiple Content-Length header fields with field-values consisting of the same decimal value, or a single Content-Length header field with a field value containing a list of identical decimal values (e.g., "Content-Length: 42, 42"), indicating that duplicate Content-Length header fields have been generated or combined by an upstream message processor, then the recipient **MUST** either reject the message as invalid or replace the duplicated field-values with a single valid Content-Length field containing that decimal value prior to determining the message body length or forwarding the message.

Note: HTTP's use of Content-Length for message framing differs significantly from the same field's use in MIME, where it is an optional field used only within the "message/external-body" media-type.

3.3.3. Message Body Length

The length of a message body is determined by one of the following (in order of precedence):

1. Any response to a HEAD request and any response with a 1xx (Informational), 204 (No Content), or 304 (Not Modified) status code is always terminated by the first empty line after the header fields, regardless of the header fields present in the message, and thus cannot contain a message body.
2. Any 2xx (Successful) response to a CONNECT request implies that the connection will become a tunnel immediately after the empty line that concludes the header fields. A client **MUST** ignore any Content-Length or Transfer-Encoding header fields received in such a message.
3. If a Transfer-Encoding header field is present and the chunked transfer coding (Section 4.1) is the final encoding, the message body length is determined by reading and decoding the chunked data until the transfer coding indicates the data is complete.

If a Transfer-Encoding header field is present in a response and the chunked transfer coding is not the final encoding, the message body length is determined by reading the connection until it is closed by the server. If a Transfer-Encoding header field is present in a request and the chunked transfer coding is not the final encoding, the message body length cannot be determined reliably; the server **MUST** respond with the 400 (Bad Request) status code and then close the connection.

If a message is received with both a Transfer-Encoding and a Content-Length header field, the Transfer-Encoding overrides the Content-Length. Such a message might indicate an attempt to perform request smuggling (Section 9.5) or response splitting (Section 9.4) and ought to be handled as an error. A sender **MUST** remove the received Content-Length field prior to forwarding such a message downstream.

4. If a message is received without Transfer-Encoding and with either multiple Content-Length header fields having differing field-values or a single Content-Length header field having an invalid value, then the message framing is invalid and the recipient **MUST** treat it as an unrecoverable error. If this is a request message, the server **MUST** respond with a 400 (Bad Request) status code and then close the connection. If this is a response message received by a proxy, the proxy **MUST** close the connection to the server, discard the received response, and send a 502 (Bad

Gateway) response to the client. If this is a response message received by a user agent, the user agent **MUST** close the connection to the server and discard the received response.

5. If a valid Content-Length header field is present without Transfer-Encoding, its decimal value defines the expected message body length in octets. If the sender closes the connection or the recipient times out before the indicated number of octets are received, the recipient **MUST** consider the message to be incomplete and close the connection.
6. If this is a request message and none of the above are true, then the message body length is zero (no message body is present).
7. Otherwise, this is a response message without a declared message body length, so the message body length is determined by the number of octets received prior to the server closing the connection.

Since there is no way to distinguish a successfully completed, close-delimited message from a partially received message interrupted by network failure, a server **SHOULD** generate encoding or length-delimited messages whenever possible. The close-delimiting feature exists primarily for backwards compatibility with HTTP/1.0.

A server **MAY** reject a request that contains a message body but not a Content-Length by responding with 411 (Length Required).

Unless a transfer coding other than chunked has been applied, a client that sends a request containing a message body **SHOULD** use a valid Content-Length header field if the message body length is known in advance, rather than the chunked transfer coding, since some existing services respond to chunked with a 411 (Length Required) status code even though they understand the chunked transfer coding. This is typically because such services are implemented via a gateway that requires a content-length in advance of being called and the server is unable or unwilling to buffer the entire request before processing.

A user agent that sends a request containing a message body **MUST** send a valid Content-Length header field if it does not know the server will handle HTTP/1.1 (or later) requests; such knowledge can be in the form of specific user configuration or by remembering the version of a prior received response.

If the final response to the last request on a connection has been completely received and there remains additional data to read, a user agent **MAY** discard the remaining data or attempt to determine if that

data belongs as part of the prior response body, which might be the case if the prior message's Content-Length value is incorrect. A client **MUST NOT** process, cache, or forward such extra data as a separate response, since such behavior would be vulnerable to cache poisoning.

3.4. Handling Incomplete Messages

A server that receives an incomplete request message, usually due to a canceled request or a triggered timeout exception, **MAY** send an error response prior to closing the connection.

A client that receives an incomplete response message, which can occur when a connection is closed prematurely or when decoding a supposedly chunked transfer coding fails, **MUST** record the message as incomplete. Cache requirements for incomplete responses are defined in Section 3 of [RFC7234].

If a response terminates in the middle of the header section (before the empty line is received) and the status code might rely on header fields to convey the full meaning of the response, then the client cannot assume that meaning has been conveyed; the client might need to repeat the request in order to determine what action to take next.

A message body that uses the chunked transfer coding is incomplete if the zero-sized chunk that terminates the encoding has not been received. A message that uses a valid Content-Length is incomplete if the size of the message body received (in octets) is less than the value given by Content-Length. A response that has neither chunked transfer coding nor Content-Length is terminated by closure of the connection and, thus, is considered complete regardless of the number of message body octets received, provided that the header section was received intact.

3.5. Message Parsing Robustness

Older HTTP/1.0 user agent implementations might send an extra CRLF after a POST request as a workaround for some early server applications that failed to read message body content that was not terminated by a line-ending. An HTTP/1.1 user agent **MUST NOT** preface or follow a request with an extra CRLF. If terminating the request message body with a line-ending is desired, then the user agent **MUST** count the terminating CRLF octets as part of the message body length.

In the interest of robustness, a server that is expecting to receive and parse a request-line **SHOULD** ignore at least one empty line (CRLF) received prior to the request-line.

Although the line terminator for the start-line and header fields is the sequence CRLF, a recipient MAY recognize a single LF as a line terminator and ignore any preceding CR.

Although the request-line and status-line grammar rules require that each of the component elements be separated by a single SP octet, recipients MAY instead parse on whitespace-delimited word boundaries and, aside from the CRLF terminator, treat any form of whitespace as the SP separator while ignoring preceding or trailing whitespace; such whitespace includes one or more of the following octets: SP, HTAB, VT (%x0B), FF (%x0C), or bare CR. However, lenient parsing can result in security vulnerabilities if there are multiple recipients of the message and each has its own unique interpretation of robustness (see Section 9.5).

When a server listening only for HTTP request messages, or processing what appears from the start-line to be an HTTP request message, receives a sequence of octets that does not match the HTTP-message grammar aside from the robustness exceptions listed above, the server SHOULD respond with a 400 (Bad Request) response.

4. Transfer Codings

Transfer coding names are used to indicate an encoding transformation that has been, can be, or might need to be applied to a payload body in order to ensure "safe transport" through the network. This differs from a content coding in that the transfer coding is a property of the message rather than a property of the representation that is being transferred.

```
transfer-coding      = "chunked" ; Section 4.1
                     / "compress" ; Section 4.2.1
                     / "deflate" ; Section 4.2.2
                     / "gzip" ; Section 4.2.3
                     / transfer-extension
transfer-extension = token *( OWS ";" OWS transfer-parameter )
```

Parameters are in the form of a name or name=value pair.

```
transfer-parameter = token BWS "=" BWS ( token / quoted-string )
```

All transfer-coding names are case-insensitive and ought to be registered within the HTTP Transfer Coding registry, as defined in Section 8.4. They are used in the TE (Section 4.3) and Transfer-Encoding (Section 3.3.1) header fields.

4.1. Chunked Transfer Coding

The chunked transfer coding wraps the payload body in order to transfer it as a series of chunks, each with its own size indicator, followed by an OPTIONAL trailer containing header fields. Chunked enables content streams of unknown size to be transferred as a sequence of length-delimited buffers, which enables the sender to retain connection persistence and the recipient to know when it has received the entire message.

```
chunked-body    = *chunk
                  last-chunk
                  trailer-part
                  CRLF
```

```
chunk           = chunk-size [ chunk-ext ] CRLF
                  chunk-data CRLF
```

```
chunk-size      = 1*HEXDIG
last-chunk      = 1*("0") [ chunk-ext ] CRLF
```

```
chunk-data      = 1*OCTET ; a sequence of chunk-size octets
```

The chunk-size field is a string of hex digits indicating the size of the chunk-data in octets. The chunked transfer coding is complete when a chunk with a chunk-size of zero is received, possibly followed by a trailer, and finally terminated by an empty line.

A recipient **MUST** be able to parse and decode the chunked transfer coding.

4.1.1. Chunk Extensions

The chunked encoding allows each chunk to include zero or more chunk extensions, immediately following the chunk-size, for the sake of supplying per-chunk metadata (such as a signature or hash), mid-message control information, or randomization of message body size.

```
chunk-ext       = *( ";" chunk-ext-name [ "=" chunk-ext-val ] )
```

```
chunk-ext-name  = token
chunk-ext-val    = token / quoted-string
```

The chunked encoding is specific to each connection and is likely to be removed or recoded by each recipient (including intermediaries) before any higher-level application would have a chance to inspect the extensions. Hence, use of chunk extensions is generally limited

to specialized HTTP services such as "long polling" (where client and server can have shared expectations regarding the use of chunk extensions) or for padding within an end-to-end secured connection.

A recipient **MUST** ignore unrecognized chunk extensions. A server ought to limit the total length of chunk extensions received in a request to an amount reasonable for the services provided, in the same way that it applies length limitations and timeouts for other parts of a message, and generate an appropriate 4xx (Client Error) response if that amount is exceeded.

4.1.2. Chunked Trailer Part

A trailer allows the sender to include additional fields at the end of a chunked message in order to supply metadata that might be dynamically generated while the message body is sent, such as a message integrity check, digital signature, or post-processing status. The trailer fields are identical to header fields, except they are sent in a chunked trailer instead of the message's header section.

trailer-part = *(header-field CRLF)

A sender **MUST NOT** generate a trailer that contains a field necessary for message framing (e.g., Transfer-Encoding and Content-Length), routing (e.g., Host), request modifiers (e.g., controls and conditionals in Section 5 of [RFC7231]), authentication (e.g., see [RFC7235] and [RFC6265]), response control data (e.g., see Section 7.1 of [RFC7231]), or determining how to process the payload (e.g., Content-Encoding, Content-Type, Content-Range, and Trailer).

When a chunked message containing a non-empty trailer is received, the recipient **MAY** process the fields (aside from those forbidden above) as if they were appended to the message's header section. A recipient **MUST** ignore (or consider as an error) any fields that are forbidden to be sent in a trailer, since processing them as if they were present in the header section might bypass external security filters.

Unless the request includes a TE header field indicating "trailers" is acceptable, as described in Section 4.3, a server **SHOULD NOT** generate trailer fields that it believes are necessary for the user agent to receive. Without a TE containing "trailers", the server ought to assume that the trailer fields might be silently discarded along the path to the user agent. This requirement allows intermediaries to forward a de-chunked message to an HTTP/1.0 recipient without buffering the entire response.

4.1.3. Decoding Chunked

A process for decoding the chunked transfer coding can be represented in pseudo-code as:

```
length := 0
read chunk-size, chunk-ext (if any), and CRLF
while (chunk-size > 0) {
    read chunk-data and CRLF
    append chunk-data to decoded-body
    length := length + chunk-size
    read chunk-size, chunk-ext (if any), and CRLF
}
read trailer field
while (trailer field is not empty) {
    if (trailer field is allowed to be sent in a trailer) {
        append trailer field to existing header fields
    }
    read trailer-field
}
Content-Length := length
Remove "chunked" from Transfer-Encoding
Remove Trailer from existing header fields
```

4.2. Compression Codings

The codings defined below can be used to compress the payload of a message.

4.2.1. Compress Coding

The "compress" coding is an adaptive Lempel-Ziv-Welch (LZW) coding [Welch] that is commonly produced by the UNIX file compression program "compress". A recipient SHOULD consider "x-compress" to be equivalent to "compress".

4.2.2. Deflate Coding

The "deflate" coding is a "zlib" data format [RFC1950] containing a "deflate" compressed data stream [RFC1951] that uses a combination of the Lempel-Ziv (LZ77) compression algorithm and Huffman coding.

Note: Some non-conformant implementations send the "deflate" compressed data without the zlib wrapper.

4.2.3. Gzip Coding

The "gzip" coding is an LZ77 coding with a 32-bit Cyclic Redundancy Check (CRC) that is commonly produced by the gzip file compression program [RFC1952]. A recipient **SHOULD** consider "x-gzip" to be equivalent to "gzip".

4.3. TE

The "TE" header field in a request indicates what transfer codings, besides chunked, the client is willing to accept in response, and whether or not the client is willing to accept trailer fields in a chunked transfer coding.

The TE field-value consists of a comma-separated list of transfer coding names, each allowing for optional parameters (as described in Section 4), and/or the keyword "trailers". A client **MUST NOT** send the chunked transfer coding name in TE; chunked is always acceptable for HTTP/1.1 recipients.

```
TE           = #t-codings
t-codings    = "trailers" / ( transfer-coding [ t-ranking ] )
t-ranking    = OWS ";" OWS "q=" rank
rank         = ( "0" [ "." 0*3DIGIT ] )
              / ( "1" [ "." 0*3("0") ] )
```

Three examples of TE use are below.

```
TE: deflate
TE:
TE: trailers, deflate;q=0.5
```

The presence of the keyword "trailers" indicates that the client is willing to accept trailer fields in a chunked transfer coding, as defined in Section 4.1.2, on behalf of itself and any downstream clients. For requests from an intermediary, this implies that either: (a) all downstream clients are willing to accept trailer fields in the forwarded response; or, (b) the intermediary will attempt to buffer the response on behalf of downstream recipients. Note that HTTP/1.1 does not define any means to limit the size of a chunked response such that an intermediary can be assured of buffering the entire response.

When multiple transfer codings are acceptable, the client **MAY** rank the codings by preference using a case-insensitive "q" parameter (similar to the qvalues used in content negotiation fields, Section

5.3.1 of [RFC7231]). The rank value is a real number in the range 0 through 1, where 0.001 is the least preferred and 1 is the most preferred; a value of 0 means "not acceptable".

If the TE field-value is empty or if no TE field is present, the only acceptable transfer coding is chunked. A message with no transfer coding is always acceptable.

Since the TE header field only applies to the immediate connection, a sender of TE MUST also send a "TE" connection option within the Connection header field (Section 6.1) in order to prevent the TE field from being forwarded by intermediaries that do not support its semantics.

4.4. Trailer

When a message includes a message body encoded with the chunked transfer coding and the sender desires to send metadata in the form of trailer fields at the end of the message, the sender SHOULD generate a Trailer header field before the message body to indicate which fields will be present in the trailers. This allows the recipient to prepare for receipt of that metadata before it starts processing the body, which is useful if the message is being streamed and the recipient wishes to confirm an integrity check on the fly.

Trailer = 1#field-name

5. Message Routing

HTTP request message routing is determined by each client based on the target resource, the client's proxy configuration, and establishment or reuse of an inbound connection. The corresponding response routing follows the same connection chain back to the client.

5.1. Identifying a Target Resource

HTTP is used in a wide variety of applications, ranging from general-purpose computers to home appliances. In some cases, communication options are hard-coded in a client's configuration. However, most HTTP clients rely on the same resource identification mechanism and configuration techniques as general-purpose Web browsers.

HTTP communication is initiated by a user agent for some purpose. The purpose is a combination of request semantics, which are defined in [RFC7231], and a target resource upon which to apply those semantics. A URI reference (Section 2.7) is typically used as an

identifier for the "target resource", which a user agent would resolve to its absolute form in order to obtain the "target URI". The target URI excludes the reference's fragment component, if any, since fragment identifiers are reserved for client-side processing ([RFC3986], Section 3.5).

5.2. Connecting Inbound

Once the target URI is determined, a client needs to decide whether a network request is necessary to accomplish the desired semantics and, if so, where that request is to be directed.

If the client has a cache [RFC7234] and the request can be satisfied by it, then the request is usually directed there first.

If the request is not satisfied by a cache, then a typical client will check its configuration to determine whether a proxy is to be used to satisfy the request. Proxy configuration is implementation-dependent, but is often based on URI prefix matching, selective authority matching, or both, and the proxy itself is usually identified by an "http" or "https" URI. If a proxy is applicable, the client connects inbound by establishing (or reusing) a connection to that proxy.

If no proxy is applicable, a typical client will invoke a handler routine, usually specific to the target URI's scheme, to connect directly to an authority for the target resource. How that is accomplished is dependent on the target URI scheme and defined by its associated specification, similar to how this specification defines origin server access for resolution of the "http" (Section 2.7.1) and "https" (Section 2.7.2) schemes.

HTTP requirements regarding connection management are defined in Section 6.

5.3. Request Target

Once an inbound connection is obtained, the client sends an HTTP request message (Section 3) with a request-target derived from the target URI. There are four distinct formats for the request-target, depending on both the method being requested and whether the request is to a proxy.

```
request-target = origin-form  
                / absolute-form  
                / authority-form  
                / asterisk-form
```

5.3.1. origin-form

The most common form of request-target is the origin-form.

origin-form = absolute-path ["?" query]

When making a request directly to an origin server, other than a CONNECT or server-wide OPTIONS request (as detailed below), a client MUST send only the absolute path and query components of the target URI as the request-target. If the target URI's path component is empty, the client MUST send "/" as the path within the origin-form of request-target. A Host header field is also sent, as defined in Section 5.4.

For example, a client wishing to retrieve a representation of the resource identified as

`http://www.example.org/where?q=now`

directly from the origin server would open (or reuse) a TCP connection to port 80 of the host "www.example.org" and send the lines:

```
GET /where?q=now HTTP/1.1
Host: www.example.org
```

followed by the remainder of the request message.

5.3.2. absolute-form

When making a request to a proxy, other than a CONNECT or server-wide OPTIONS request (as detailed below), a client MUST send the target URI in absolute-form as the request-target.

absolute-form = absolute-URI

The proxy is requested to either service that request from a valid cache, if possible, or make the same request on the client's behalf to either the next inbound proxy server or directly to the origin server indicated by the request-target. Requirements on such "forwarding" of messages are defined in Section 5.7.

An example absolute-form of request-line would be:

```
GET http://www.example.org/pub/WWW/TheProject.html HTTP/1.1
```

To allow for transition to the absolute-form for all requests in some future version of HTTP, a server **MUST** accept the absolute-form in requests, even though HTTP/1.1 clients will only send them in requests to proxies.

5.3.3. authority-form

The authority-form of request-target is only used for CONNECT requests (Section 4.3.6 of [RFC7231]).

authority-form = authority

When making a CONNECT request to establish a tunnel through one or more proxies, a client **MUST** send only the target URI's authority component (excluding any userinfo and its "@" delimiter) as the request-target. For example,

CONNECT www.example.com:80 HTTP/1.1

5.3.4. asterisk-form

The asterisk-form of request-target is only used for a server-wide OPTIONS request (Section 4.3.7 of [RFC7231]).

asterisk-form = "*"

When a client wishes to request OPTIONS for the server as a whole, as opposed to a specific named resource of that server, the client **MUST** send only "*" (%x2A) as the request-target. For example,

OPTIONS * HTTP/1.1

If a proxy receives an OPTIONS request with an absolute-form of request-target in which the URI has an empty path and no query component, then the last proxy on the request chain **MUST** send a request-target of "*" when it forwards the request to the indicated origin server.

For example, the request

OPTIONS http://www.example.org:8001 HTTP/1.1

would be forwarded by the final proxy as

OPTIONS * HTTP/1.1
Host: www.example.org:8001

after connecting to port 8001 of host "www.example.org".

5.4. Host

The "Host" header field in a request provides the host and port information from the target URI, enabling the origin server to distinguish among resources while servicing requests for multiple host names on a single IP address.

Host = uri-host [":" port] ; Section 2.7.1

A client **MUST** send a Host header field in all HTTP/1.1 request messages. If the target URI includes an authority component, then a client **MUST** send a field-value for Host that is identical to that authority component, excluding any userinfo subcomponent and its "@" delimiter (Section 2.7.1). If the authority component is missing or undefined for the target URI, then a client **MUST** send a Host header field with an empty field-value.

Since the Host field-value is critical information for handling a request, a user agent **SHOULD** generate Host as the first header field following the request-line.

For example, a GET request to the origin server for <http://www.example.org/pub/WWW/> would begin with:

```
GET /pub/WWW/ HTTP/1.1
Host: www.example.org
```

A client **MUST** send a Host header field in an HTTP/1.1 request even if the request-target is in the absolute-form, since this allows the Host information to be forwarded through ancient HTTP/1.0 proxies that might not have implemented Host.

When a proxy receives a request with an absolute-form of request-target, the proxy **MUST** ignore the received Host header field (if any) and instead replace it with the host information of the request-target. A proxy that forwards such a request **MUST** generate a new Host field-value based on the received request-target rather than forward the received Host field-value.

Since the Host header field acts as an application-level routing mechanism, it is a frequent target for malware seeking to poison a shared cache or redirect a request to an unintended server. An interception proxy is particularly vulnerable if it relies on the Host field-value for redirecting requests to internal servers, or for use as a cache key in a shared cache, without first verifying that the intercepted connection is targeting a valid IP address for that host.

A server **MUST** respond with a 400 (Bad Request) status code to any HTTP/1.1 request message that lacks a Host header field and to any request message that contains more than one Host header field or a Host header field with an invalid field-value.

5.5. Effective Request URI

Since the request-target often contains only part of the user agent's target URI, a server reconstructs the intended target as an "effective request URI" to properly service the request. This reconstruction involves both the server's local configuration and information communicated in the request-target, Host header field, and connection context.

For a user agent, the effective request URI is the target URI.

If the request-target is in absolute-form, the effective request URI is the same as the request-target. Otherwise, the effective request URI is constructed as follows:

If the server's configuration (or outbound gateway) provides a fixed URI scheme, that scheme is used for the effective request URI. Otherwise, if the request is received over a TLS-secured TCP connection, the effective request URI's scheme is "https"; if not, the scheme is "http".

If the server's configuration (or outbound gateway) provides a fixed URI authority component, that authority is used for the effective request URI. If not, then if the request-target is in authority-form, the effective request URI's authority component is the same as the request-target. If not, then if a Host header field is supplied with a non-empty field-value, the authority component is the same as the Host field-value. Otherwise, the authority component is assigned the default name configured for the server and, if the connection's incoming TCP port number differs from the default port for the effective request URI's scheme, then a colon (":") and the incoming port number (in decimal form) are appended to the authority component.

If the request-target is in authority-form or asterisk-form, the effective request URI's combined path and query component is empty. Otherwise, the combined path and query component is the same as the request-target.

The components of the effective request URI, once determined as above, can be combined into absolute-URI form by concatenating the scheme, "://", authority, and combined path and query component.

Example 1: the following message received over an insecure TCP connection

```
GET /pub/WWW/TheProject.html HTTP/1.1
Host: www.example.org:8080
```

has an effective request URI of

```
http://www.example.org:8080/pub/WWW/TheProject.html
```

Example 2: the following message received over a TLS-secured TCP connection

```
OPTIONS * HTTP/1.1
Host: www.example.org
```

has an effective request URI of

```
https://www.example.org
```

Recipients of an HTTP/1.0 request that lacks a Host header field might need to use heuristics (e.g., examination of the URI path for something unique to a particular host) in order to guess the effective request URI's authority component.

Once the effective request URI has been constructed, an origin server needs to decide whether or not to provide service for that URI via the connection in which the request was received. For example, the request might have been misdirected, deliberately or accidentally, such that the information within a received request-target or Host header field differs from the host or port upon which the connection has been made. If the connection is from a trusted gateway, that inconsistency might be expected; otherwise, it might indicate an attempt to bypass security filters, trick the server into delivering non-public content, or poison a cache. See Section 9 for security considerations regarding message routing.

5.6. Associating a Response to a Request

HTTP does not include a request identifier for associating a given request message with its corresponding one or more response messages. Hence, it relies on the order of response arrival to correspond exactly to the order in which requests are made on the same connection. More than one response message per request only occurs when one or more informational responses (1xx, see Section 6.2 of [RFC7231]) precede a final response to the same request.

5.7. Message Forwarding

An intermediary **MUST NOT** forward a message to itself unless it is protected from an infinite request loop. In general, an intermediary ought to recognize its own server names, including any aliases, local variations, or literal IP addresses, and respond to such requests directly.

5.7.1. Via

```
received-protocol = [ protocol-name "/" ] protocol-version
                    ; see Section 6.7
received-by       = ( uri-host [ ":" port ] ) / pseudonym
pseudonym        = token
```

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A proxy **MUST** send an appropriate Via header field, as described below, in each message that it forwards. An HTTP-to-HTTP gateway **MUST** send an appropriate Via header field in each inbound request message and **MAY** send a Via header field in forwarded response messages.

For each intermediary, the received-protocol indicates the protocol and protocol version used by the upstream sender of the message. Hence, the Via field value records the advertised protocol capabilities of the request/response chain such that they remain visible to downstream recipients; this can be useful for determining what backwards-incompatible features might be safe to use in response, or within a later request, as described in Section 2.6. For brevity, the protocol-name is omitted when the received protocol is HTTP.

The received-by portion of the field value is normally the host and optional port number of a recipient server or client that subsequently forwarded the message. However, if the real host is considered to be sensitive information, a sender **MAY** replace it with a pseudonym. If a port is not provided, a recipient **MAY** interpret that as meaning it was received on the default TCP port, if any, for the received-protocol.

A sender **MAY** generate comments in the Via header field to identify the software of each recipient, analogous to the User-Agent and Server header fields. However, all comments in the Via field are optional, and a recipient **MAY** remove them prior to forwarding the message.

For example, a request message could be sent from an HTTP/1.0 user agent to an internal proxy code-named "fred", which uses HTTP/1.1 to forward the request to a public proxy at p.example.net, which completes the request by forwarding it to the origin server at www.example.com. The request received by www.example.com would then have the following Via header field:

Via: 1.0 fred, 1.1 p.example.net

An intermediary used as a portal through a network firewall **SHOULD NOT** forward the names and ports of hosts within the firewall region unless it is explicitly enabled to do so. If not enabled, such an intermediary **SHOULD** replace each received-by host of any host behind the firewall by an appropriate pseudonym for that host.

An intermediary MAY combine an ordered subsequence of Via header field entries into a single such entry if the entries have identical received-protocol values. For example,

Via: 1.0 ricky, 1.1 ethel, 1.1 fred, 1.0 lucy

could be collapsed to

Via: 1.0 ricky, 1.1 mertz, 1.0 lucy

A sender SHOULD NOT combine multiple entries unless they are all under the same organizational control and the hosts have already been replaced by pseudonyms. A sender MUST NOT combine entries that have different received-protocol values.

5.7.2. Transformations

Some intermediaries include features for transforming messages and their payloads. A proxy might, for example, convert between image formats in order to save cache space or to reduce the amount of traffic on a slow link. However, operational problems might occur when these transformations are applied to payloads intended for critical applications, such as medical imaging or scientific data analysis, particularly when integrity checks or digital signatures are used to ensure that the payload received is identical to the original.

An HTTP-to-HTTP proxy is called a "transforming proxy" if it is designed or configured to modify messages in a semantically meaningful way (i.e., modifications, beyond those required by normal HTTP processing, that change the message in a way that would be significant to the original sender or potentially significant to downstream recipients). For example, a transforming proxy might be acting as a shared annotation server (modifying responses to include references to a local annotation database), a malware filter, a format transcoder, or a privacy filter. Such transformations are presumed to be desired by whichever client (or client organization) selected the proxy.

If a proxy receives a request-target with a host name that is not a fully qualified domain name, it MAY add its own domain to the host name it received when forwarding the request. A proxy MUST NOT change the host name if the request-target contains a fully qualified domain name.

A proxy **MUST NOT** modify the "absolute-path" and "query" parts of the received request-target when forwarding it to the next inbound server, except as noted above to replace an empty path with "/" or "*".

A proxy **MAY** modify the message body through application or removal of a transfer coding (Section 4).

A proxy **MUST NOT** transform the payload (Section 3.3 of [RFC7231]) of a message that contains a no-transform cache-control directive (Section 5.2 of [RFC7234]).

A proxy **MAY** transform the payload of a message that does not contain a no-transform cache-control directive. A proxy that transforms a payload **MUST** add a Warning header field with the warn-code of 214 ("Transformation Applied") if one is not already in the message (see Section 5.5 of [RFC7234]). A proxy that transforms the payload of a 200 (OK) response can further inform downstream recipients that a transformation has been applied by changing the response status code to 203 (Non-Authoritative Information) (Section 6.3.4 of [RFC7231]).

A proxy **SHOULD NOT** modify header fields that provide information about the endpoints of the communication chain, the resource state, or the selected representation (other than the payload) unless the field's definition specifically allows such modification or the modification is deemed necessary for privacy or security.

6. Connection Management

HTTP messaging is independent of the underlying transport- or session-layer connection protocol(s). HTTP only presumes a reliable transport with in-order delivery of requests and the corresponding in-order delivery of responses. The mapping of HTTP request and response structures onto the data units of an underlying transport protocol is outside the scope of this specification.

As described in Section 5.2, the specific connection protocols to be used for an HTTP interaction are determined by client configuration and the target URI. For example, the "http" URI scheme (Section 2.7.1) indicates a default connection of TCP over IP, with a default TCP port of 80, but the client might be configured to use a proxy via some other connection, port, or protocol.

HTTP implementations are expected to engage in connection management, which includes maintaining the state of current connections, establishing a new connection or reusing an existing connection, processing messages received on a connection, detecting connection failures, and closing each connection. Most clients maintain multiple connections in parallel, including more than one connection per server endpoint. Most servers are designed to maintain thousands of concurrent connections, while controlling request queues to enable fair use and detect denial-of-service attacks.

6.1. Connection

The "Connection" header field allows the sender to indicate desired control options for the current connection. In order to avoid confusing downstream recipients, a proxy or gateway **MUST** remove or replace any received connection options before forwarding the message.

When a header field aside from Connection is used to supply control information for or about the current connection, the sender **MUST** list the corresponding field-name within the Connection header field. A proxy or gateway **MUST** parse a received Connection header field before a message is forwarded and, for each connection-option in this field, remove any header field(s) from the message with the same name as the connection-option, and then remove the Connection header field itself (or replace it with the intermediary's own connection options for the forwarded message).

Hence, the Connection header field provides a declarative way of distinguishing header fields that are only intended for the immediate recipient ("hop-by-hop") from those fields that are intended for all recipients on the chain ("end-to-end"), enabling the message to be self-descriptive and allowing future connection-specific extensions to be deployed without fear that they will be blindly forwarded by older intermediaries.

The Connection header field's value has the following grammar:

```
Connection      = 1#connection-option
connection-option = token
```

Connection options are case-insensitive.

A sender **MUST NOT** send a connection option corresponding to a header field that is intended for all recipients of the payload. For example, Cache-Control is never appropriate as a connection option (Section 5.2 of [RFC7234]).

The connection options do not always correspond to a header field present in the message, since a connection-specific header field might not be needed if there are no parameters associated with a connection option. In contrast, a connection-specific header field that is received without a corresponding connection option usually indicates that the field has been improperly forwarded by an intermediary and ought to be ignored by the recipient.

When defining new connection options, specification authors ought to survey existing header field names and ensure that the new connection option does not share the same name as an already deployed header field. Defining a new connection option essentially reserves that potential field-name for carrying additional information related to the connection option, since it would be unwise for senders to use that field-name for anything else.

The "close" connection option is defined for a sender to signal that this connection will be closed after completion of the response. For example,

Connection: close

in either the request or the response header fields indicates that the sender is going to close the connection after the current request/response is complete (Section 6.6).

A client that does not support persistent connections **MUST** send the "close" connection option in every request message.

A server that does not support persistent connections **MUST** send the "close" connection option in every response message that does not have a 1xx (Informational) status code.

6.2. Establishment

It is beyond the scope of this specification to describe how connections are established via various transport- or session-layer protocols. Each connection applies to only one transport link.

6.3. Persistence

HTTP/1.1 defaults to the use of "persistent connections", allowing multiple requests and responses to be carried over a single connection. The "close" connection option is used to signal that a connection will not persist after the current request/response. HTTP implementations **SHOULD** support persistent connections.

A recipient determines whether a connection is persistent or not based on the most recently received message's protocol version and Connection header field (if any):

- o If the "close" connection option is present, the connection will not persist after the current response; else,
- o If the received protocol is HTTP/1.1 (or later), the connection will persist after the current response; else,
- o If the received protocol is HTTP/1.0, the "keep-alive" connection option is present, the recipient is not a proxy, and the recipient wishes to honor the HTTP/1.0 "keep-alive" mechanism, the connection will persist after the current response; otherwise,
- o The connection will close after the current response.

A client MAY send additional requests on a persistent connection until it sends or receives a "close" connection option or receives an HTTP/1.0 response without a "keep-alive" connection option.

In order to remain persistent, all messages on a connection need to have a self-defined message length (i.e., one not defined by closure of the connection), as described in Section 3.3. A server MUST read the entire request message body or close the connection after sending its response, since otherwise the remaining data on a persistent connection would be misinterpreted as the next request. Likewise, a client MUST read the entire response message body if it intends to reuse the same connection for a subsequent request.

A proxy server MUST NOT maintain a persistent connection with an HTTP/1.0 client (see Section 19.7.1 of [RFC2068] for information and discussion of the problems with the Keep-Alive header field implemented by many HTTP/1.0 clients).

See Appendix A.1.2 for more information on backwards compatibility with HTTP/1.0 clients.

6.3.1. Retrying Requests

Connections can be closed at any time, with or without intention. Implementations ought to anticipate the need to recover from asynchronous close events.

When an inbound connection is closed prematurely, a client MAY open a new connection and automatically retransmit an aborted sequence of requests if all of those requests have idempotent methods (Section 4.2.2 of [RFC7231]). A proxy MUST NOT automatically retry non-idempotent requests.

A user agent MUST NOT automatically retry a request with a non-idempotent method unless it has some means to know that the request semantics are actually idempotent, regardless of the method, or some means to detect that the original request was never applied. For example, a user agent that knows (through design or configuration) that a POST request to a given resource is safe can repeat that request automatically. Likewise, a user agent designed specifically to operate on a version control repository might be able to recover from partial failure conditions by checking the target resource revision(s) after a failed connection, reverting or fixing any changes that were partially applied, and then automatically retrying the requests that failed.

A client SHOULD NOT automatically retry a failed automatic retry.

6.3.2. Pipelining

A client that supports persistent connections MAY "pipeline" its requests (i.e., send multiple requests without waiting for each response). A server MAY process a sequence of pipelined requests in parallel if they all have safe methods (Section 4.2.1 of [RFC7231]), but it MUST send the corresponding responses in the same order that the requests were received.

A client that pipelines requests SHOULD retry unanswered requests if the connection closes before it receives all of the corresponding responses. When retrying pipelined requests after a failed connection (a connection not explicitly closed by the server in its last complete response), a client MUST NOT pipeline immediately after connection establishment, since the first remaining request in the prior pipeline might have caused an error response that can be lost again if multiple requests are sent on a prematurely closed connection (see the TCP reset problem described in Section 6.6).

Idempotent methods (Section 4.2.2 of [RFC7231]) are significant to pipelining because they can be automatically retried after a connection failure. A user agent SHOULD NOT pipeline requests after a non-idempotent method, until the final response status code for that method has been received, unless the user agent has a means to detect and recover from partial failure conditions involving the pipelined sequence.

An intermediary that receives pipelined requests MAY pipeline those requests when forwarding them inbound, since it can rely on the outbound user agent(s) to determine what requests can be safely pipelined. If the inbound connection fails before receiving a response, the pipelining intermediary MAY attempt to retry a sequence of requests that have yet to receive a response if the requests all have idempotent methods; otherwise, the pipelining intermediary SHOULD forward any received responses and then close the corresponding outbound connection(s) so that the outbound user agent(s) can recover accordingly.

6.4. Concurrency

A client ought to limit the number of simultaneous open connections that it maintains to a given server.

Previous revisions of HTTP gave a specific number of connections as a ceiling, but this was found to be impractical for many applications. As a result, this specification does not mandate a particular maximum number of connections but, instead, encourages clients to be conservative when opening multiple connections.

Multiple connections are typically used to avoid the "head-of-line blocking" problem, wherein a request that takes significant server-side processing and/or has a large payload blocks subsequent requests on the same connection. However, each connection consumes server resources. Furthermore, using multiple connections can cause undesirable side effects in congested networks.

Note that a server might reject traffic that it deems abusive or characteristic of a denial-of-service attack, such as an excessive number of open connections from a single client.

6.5. Failures and Timeouts

Servers will usually have some timeout value beyond which they will no longer maintain an inactive connection. Proxy servers might make this a higher value since it is likely that the client will be making more connections through the same proxy server. The use of persistent connections places no requirements on the length (or existence) of this timeout for either the client or the server.

A client or server that wishes to time out SHOULD issue a graceful close on the connection. Implementations SHOULD constantly monitor open connections for a received closure signal and respond to it as appropriate, since prompt closure of both sides of a connection enables allocated system resources to be reclaimed.

A client, server, or proxy MAY close the transport connection at any time. For example, a client might have started to send a new request at the same time that the server has decided to close the "idle" connection. From the server's point of view, the connection is being closed while it was idle, but from the client's point of view, a request is in progress.

A server SHOULD sustain persistent connections, when possible, and allow the underlying transport's flow-control mechanisms to resolve temporary overloads, rather than terminate connections with the expectation that clients will retry. The latter technique can exacerbate network congestion.

A client sending a message body SHOULD monitor the network connection for an error response while it is transmitting the request. If the client sees a response that indicates the server does not wish to receive the message body and is closing the connection, the client SHOULD immediately cease transmitting the body and close its side of the connection.

6.6. Tear-down

The Connection header field (Section 6.1) provides a "close" connection option that a sender SHOULD send when it wishes to close the connection after the current request/response pair.

A client that sends a "close" connection option MUST NOT send further requests on that connection (after the one containing "close") and MUST close the connection after reading the final response message corresponding to this request.

A server that receives a "close" connection option MUST initiate a close of the connection (see below) after it sends the final response to the request that contained "close". The server SHOULD send a "close" connection option in its final response on that connection. The server MUST NOT process any further requests received on that connection.

A server that sends a "close" connection option MUST initiate a close of the connection (see below) after it sends the response containing "close". The server MUST NOT process any further requests received on that connection.

A client that receives a "close" connection option MUST cease sending requests on that connection and close the connection after reading the response message containing the "close"; if additional pipelined requests had been sent on the connection, the client SHOULD NOT assume that they will be processed by the server.

If a server performs an immediate close of a TCP connection, there is a significant risk that the client will not be able to read the last HTTP response. If the server receives additional data from the client on a fully closed connection, such as another request that was sent by the client before receiving the server's response, the server's TCP stack will send a reset packet to the client; unfortunately, the reset packet might erase the client's unacknowledged input buffers before they can be read and interpreted by the client's HTTP parser.

To avoid the TCP reset problem, servers typically close a connection in stages. First, the server performs a half-close by closing only the write side of the read/write connection. The server then continues to read from the connection until it receives a corresponding close by the client, or until the server is reasonably certain that its own TCP stack has received the client's acknowledgement of the packet(s) containing the server's last response. Finally, the server fully closes the connection.

It is unknown whether the reset problem is exclusive to TCP or might also be found in other transport connection protocols.

6.7. Upgrade

The "Upgrade" header field is intended to provide a simple mechanism for transitioning from HTTP/1.1 to some other protocol on the same connection. A client MAY send a list of protocols in the Upgrade header field of a request to invite the server to switch to one or more of those protocols, in order of descending preference, before sending the final response. A server MAY ignore a received Upgrade header field if it wishes to continue using the current protocol on that connection. Upgrade cannot be used to insist on a protocol change.

Upgrade = 1#protocol

protocol = protocol-name ["/" protocol-version]

protocol-name = token

protocol-version = token

A server that sends a 101 (Switching Protocols) response MUST send an Upgrade header field to indicate the new protocol(s) to which the connection is being switched; if multiple protocol layers are being switched, the sender MUST list the protocols in layer-ascending order. A server MUST NOT switch to a protocol that was not indicated by the client in the corresponding request's Upgrade header field. A

server MAY choose to ignore the order of preference indicated by the client and select the new protocol(s) based on other factors, such as the nature of the request or the current load on the server.

A server that sends a 426 (Upgrade Required) response MUST send an Upgrade header field to indicate the acceptable protocols, in order of descending preference.

A server MAY send an Upgrade header field in any other response to advertise that it implements support for upgrading to the listed protocols, in order of descending preference, when appropriate for a future request.

The following is a hypothetical example sent by a client:

```
GET /hello.txt HTTP/1.1
Host: www.example.com
Connection: upgrade
Upgrade: HTTP/2.0, SHTTP/1.3, IRC/6.9, RTA/x11
```

The capabilities and nature of the application-level communication after the protocol change is entirely dependent upon the new protocol(s) chosen. However, immediately after sending the 101 (Switching Protocols) response, the server is expected to continue responding to the original request as if it had received its equivalent within the new protocol (i.e., the server still has an outstanding request to satisfy after the protocol has been changed, and is expected to do so without requiring the request to be repeated).

For example, if the Upgrade header field is received in a GET request and the server decides to switch protocols, it first responds with a 101 (Switching Protocols) message in HTTP/1.1 and then immediately follows that with the new protocol's equivalent of a response to a GET on the target resource. This allows a connection to be upgraded to protocols with the same semantics as HTTP without the latency cost of an additional round trip. A server MUST NOT switch protocols unless the received message semantics can be honored by the new protocol; an OPTIONS request can be honored by any protocol.

The following is an example response to the above hypothetical request:

```
HTTP/1.1 101 Switching Protocols
Connection: upgrade
Upgrade: HTTP/2.0
```

[... data stream switches to HTTP/2.0 with an appropriate response (as defined by new protocol) to the "GET /hello.txt" request ...]

When Upgrade is sent, the sender **MUST** also send a Connection header field (Section 6.1) that contains an "upgrade" connection option, in order to prevent Upgrade from being accidentally forwarded by intermediaries that might not implement the listed protocols. A server **MUST** ignore an Upgrade header field that is received in an HTTP/1.0 request.

A client cannot begin using an upgraded protocol on the connection until it has completely sent the request message (i.e., the client can't change the protocol it is sending in the middle of a message). If a server receives both an Upgrade and an Expect header field with the "100-continue" expectation (Section 5.1.1 of [RFC7231]), the server **MUST** send a 100 (Continue) response before sending a 101 (Switching Protocols) response.

The Upgrade header field only applies to switching protocols on top of the existing connection; it cannot be used to switch the underlying connection (transport) protocol, nor to switch the existing communication to a different connection. For those purposes, it is more appropriate to use a 3xx (Redirection) response (Section 6.4 of [RFC7231]).

This specification only defines the protocol name "HTTP" for use by the family of Hypertext Transfer Protocols, as defined by the HTTP version rules of Section 2.6 and future updates to this specification. Additional tokens ought to be registered with IANA using the registration procedure defined in Section 8.6.

7. ABNF List Extension: #rule

A #rule extension to the ABNF rules of [RFC5234] is used to improve readability in the definitions of some header field values.

A construct "#" is defined, similar to "*", for defining comma-delimited lists of elements. The full form is "<n>#<m>element" indicating at least <n> and at most <m> elements, each separated by a single comma (",") and optional whitespace (OWS).

In any production that uses the list construct, a sender **MUST NOT** generate empty list elements. In other words, a sender **MUST** generate lists that satisfy the following syntax:

`1#element => element *(OWS "," OWS element)`

and:

`#element => [1#element]`

and for $n \geq 1$ and $m > 1$:

`<n>#<m>element => element <n-1>*<m-1>(OWS "," OWS element)`

For compatibility with legacy list rules, a recipient **MUST** parse and ignore a reasonable number of empty list elements: enough to handle common mistakes by senders that merge values, but not so much that they could be used as a denial-of-service mechanism. In other words, a recipient **MUST** accept lists that satisfy the following syntax:

`#element => [("," / element) *(OWS "," [OWS element])]`

`1#element => *("," OWS) element *(OWS "," [OWS element])`

Empty elements do not contribute to the count of elements present. For example, given these ABNF productions:

`example-list = 1#example-list-elmt`
`example-list-elmt = token ; see Section 3.2.6`

Then the following are valid values for `example-list` (not including the double quotes, which are present for delimitation only):

`"foo,bar"`
`"foo ,bar,"`
`"foo , ,bar,charlie "`

In contrast, the following values would be invalid, since at least one non-empty element is required by the `example-list` production:

`""`
`" "`
`"',"`

Appendix B shows the collected ABNF for recipients after the list constructs have been expanded.

8. IANA Considerations

8.1. Header Field Registration

HTTP header fields are registered within the "Message Headers" registry maintained at <http://www.iana.org/assignments/message-headers/>.

This document defines the following HTTP header fields, so the "Permanent Message Header Field Names" registry has been updated accordingly (see [BCP90]).

Header Field Name	Protocol	Status	Reference
Connection	http	standard	Section 6.1
Content-Length	http	standard	Section 3.3.2
Host	http	standard	Section 5.4
TE	http	standard	Section 4.3
Trailer	http	standard	Section 4.4
Transfer-Encoding	http	standard	Section 3.3.1
Upgrade	http	standard	Section 6.7
Via	http	standard	Section 5.7.1

Furthermore, the header field-name "Close" has been registered as "reserved", since using that name as an HTTP header field might conflict with the "close" connection option of the Connection header field (Section 6.1).

Header Field Name	Protocol	Status	Reference
Close	http	reserved	Section 8.1

The change controller is: "IETF (iesg@ietf.org) - Internet Engineering Task Force".

8.2. URI Scheme Registration

IANA maintains the registry of URI Schemes [BCP115] at <http://www.iana.org/assignments/uri-schemes/>.

This document defines the following URI schemes, so the "Permanent URI Schemes" registry has been updated accordingly.

URI Scheme	Description	Reference
http	Hypertext Transfer Protocol	Section 2.7.1
https	Hypertext Transfer Protocol Secure	Section 2.7.2

8.3. Internet Media Type Registration

IANA maintains the registry of Internet media types [BCP13] at <http://www.iana.org/assignments/media-types/>.

This document serves as the specification for the Internet media types "message/http" and "application/http". The following has been registered with IANA.

8.3.1. Internet Media Type message/http

The message/http type can be used to enclose a single HTTP request or response message, provided that it obeys the MIME restrictions for all "message" types regarding line length and encodings.

Type name: message

Subtype name: http

Required parameters: N/A

Optional parameters: version, msgtype

version: The HTTP-version number of the enclosed message (e.g., "1.1"). If not present, the version can be determined from the first line of the body.

msgtype: The message type -- "request" or "response". If not present, the type can be determined from the first line of the body.

Encoding considerations: only "7bit", "8bit", or "binary" are permitted

Security considerations: see Section 9

Interoperability considerations: N/A

Published specification: This specification (see Section 8.3.1).

Applications that use this media type: N/A

Fragment identifier considerations: N/A

Additional information:

 Magic number(s): N/A

 Deprecated alias names for this type: N/A

 File extension(s): N/A

 Macintosh file type code(s): N/A

Person and email address to contact for further information:
 See Authors' Addresses section.

Intended usage: COMMON

Restrictions on usage: N/A

Author: See Authors' Addresses section.

Change controller: IESG

8.3.2. Internet Media Type application/http

The application/http type can be used to enclose a pipeline of one or more HTTP request or response messages (not intermixed).

Type name: application

Subtype name: http

Required parameters: N/A

Optional parameters: version, msgtype

 version: The HTTP-version number of the enclosed messages (e.g., "1.1"). If not present, the version can be determined from the first line of the body.

msgtype: The message type -- "request" or "response". If not present, the type can be determined from the first line of the body.

Encoding considerations: HTTP messages enclosed by this type are in "binary" format; use of an appropriate Content-Transfer-Encoding is required when transmitted via email.

Security considerations: see Section 9

Interoperability considerations: N/A

Published specification: This specification (see Section 8.3.2).

Applications that use this media type: N/A

Fragment identifier considerations: N/A

Additional information:

Deprecated alias names for this type: N/A

Magic number(s): N/A

File extension(s): N/A

Macintosh file type code(s): N/A

Person and email address to contact for further information:
See Authors' Addresses section.

Intended usage: COMMON

Restrictions on usage: N/A

Author: See Authors' Addresses section.

Change controller: IESG

8.4. Transfer Coding Registry

The "HTTP Transfer Coding Registry" defines the namespace for transfer coding names. It is maintained at <http://www.iana.org/assignments/http-parameters>.

8.4.1. Procedure

Registrations **MUST** include the following fields:

- o Name
- o Description
- o Pointer to specification text

Names of transfer codings **MUST NOT** overlap with names of content codings (Section 3.1.2.1 of [RFC7231]) unless the encoding transformation is identical, as is the case for the compression codings defined in Section 4.2.

Values to be added to this namespace require IETF Review (see Section 4.1 of [RFC5226]), and **MUST** conform to the purpose of transfer coding defined in this specification.

Use of program names for the identification of encoding formats is not desirable and is discouraged for future encodings.

8.4.2. Registration

The "HTTP Transfer Coding Registry" has been updated with the registrations below:

Name	Description	Reference
chunked	Transfer in a series of chunks	Section 4.1
compress	UNIX "compress" data format [Welch]	Section 4.2.1
deflate	"deflate" compressed data ([RFC1951]) inside the "zlib" data format ([RFC1950])	Section 4.2.2
gzip	GZIP file format [RFC1952]	Section 4.2.3
x-compress	Deprecated (alias for compress)	Section 4.2.1
x-gzip	Deprecated (alias for gzip)	Section 4.2.3

8.5. Content Coding Registration

IANA maintains the "HTTP Content Coding Registry" at <http://www.iana.org/assignments/http-parameters>.

The "HTTP Content Coding Registry" has been updated with the registrations below:

Name	Description	Reference
compress	UNIX "compress" data format [Welch]	Section 4.2.1
deflate	"deflate" compressed data ([RFC1951]) inside the "zlib" data format ([RFC1950])	Section 4.2.2
gzip	GZIP file format [RFC1952]	Section 4.2.3
x-compress	Deprecated (alias for compress)	Section 4.2.1
x-gzip	Deprecated (alias for gzip)	Section 4.2.3

8.6. Upgrade Token Registry

The "Hypertext Transfer Protocol (HTTP) Upgrade Token Registry" defines the namespace for protocol-name tokens used to identify protocols in the Upgrade header field. The registry is maintained at <http://www.iana.org/assignments/http-upgrade-tokens>.

8.6.1. Procedure

Each registered protocol name is associated with contact information and an optional set of specifications that details how the connection will be processed after it has been upgraded.

Registrations happen on a "First Come First Served" basis (see Section 4.1 of [RFC5226]) and are subject to the following rules:

1. A protocol-name token, once registered, stays registered forever.
2. The registration MUST name a responsible party for the registration.
3. The registration MUST name a point of contact.
4. The registration MAY name a set of specifications associated with that token. Such specifications need not be publicly available.
5. The registration SHOULD name a set of expected "protocol-version" tokens associated with that token at the time of registration.

6. The responsible party MAY change the registration at any time. The IANA will keep a record of all such changes, and make them available upon request.
7. The IESG MAY reassign responsibility for a protocol token. This will normally only be used in the case when a responsible party cannot be contacted.

This registration procedure for HTTP Upgrade Tokens replaces that previously defined in Section 7.2 of [RFC2817].

8.6.2. Upgrade Token Registration

The "HTTP" entry in the upgrade token registry has been updated with the registration below:

Value	Description	Expected Version Tokens	Reference
HTTP	Hypertext Transfer Protocol	any DIGIT.DIGIT (e.g, "2.0")	Section 2.6

The responsible party is: "IETF (iesg@ietf.org) - Internet Engineering Task Force".

9. Security Considerations

This section is meant to inform developers, information providers, and users of known security considerations relevant to HTTP message syntax, parsing, and routing. Security considerations about HTTP semantics and payloads are addressed in [RFC7231].

9.1. Establishing Authority

HTTP relies on the notion of an authoritative response: a response that has been determined by (or at the direction of) the authority identified within the target URI to be the most appropriate response for that request given the state of the target resource at the time of response message origination. Providing a response from a non-authoritative source, such as a shared cache, is often useful to improve performance and availability, but only to the extent that the source can be trusted or the distrusted response can be safely used.

Unfortunately, establishing authority can be difficult. For example, phishing is an attack on the user's perception of authority, where that perception can be misled by presenting similar branding in

hypertext, possibly aided by userinfo obfuscating the authority component (see Section 2.7.1). User agents can reduce the impact of phishing attacks by enabling users to easily inspect a target URI prior to making an action, by prominently distinguishing (or rejecting) userinfo when present, and by not sending stored credentials and cookies when the referring document is from an unknown or untrusted source.

When a registered name is used in the authority component, the "http" URI scheme (Section 2.7.1) relies on the user's local name resolution service to determine where it can find authoritative responses. This means that any attack on a user's network host table, cached names, or name resolution libraries becomes an avenue for attack on establishing authority. Likewise, the user's choice of server for Domain Name Service (DNS), and the hierarchy of servers from which it obtains resolution results, could impact the authenticity of address mappings; DNS Security Extensions (DNSSEC, [RFC4033]) are one way to improve authenticity.

Furthermore, after an IP address is obtained, establishing authority for an "http" URI is vulnerable to attacks on Internet Protocol routing.

The "https" scheme (Section 2.7.2) is intended to prevent (or at least reveal) many of these potential attacks on establishing authority, provided that the negotiated TLS connection is secured and the client properly verifies that the communicating server's identity matches the target URI's authority component (see [RFC2818]). Correctly implementing such verification can be difficult (see [Georgiev]).

9.2. Risks of Intermediaries

By their very nature, HTTP intermediaries are men-in-the-middle and, thus, represent an opportunity for man-in-the-middle attacks. Compromise of the systems on which the intermediaries run can result in serious security and privacy problems. Intermediaries might have access to security-related information, personal information about individual users and organizations, and proprietary information belonging to users and content providers. A compromised intermediary, or an intermediary implemented or configured without regard to security and privacy considerations, might be used in the commission of a wide range of potential attacks.

Intermediaries that contain a shared cache are especially vulnerable to cache poisoning attacks, as described in Section 8 of [RFC7234].

Implementers need to consider the privacy and security implications of their design and coding decisions, and of the configuration options they provide to operators (especially the default configuration).

Users need to be aware that intermediaries are no more trustworthy than the people who run them; HTTP itself cannot solve this problem.

9.3. Attacks via Protocol Element Length

Because HTTP uses mostly textual, character-delimited fields, parsers are often vulnerable to attacks based on sending very long (or very slow) streams of data, particularly where an implementation is expecting a protocol element with no predefined length.

To promote interoperability, specific recommendations are made for minimum size limits on request-line (Section 3.1.1) and header fields (Section 3.2). These are minimum recommendations, chosen to be supportable even by implementations with limited resources; it is expected that most implementations will choose substantially higher limits.

A server can reject a message that has a request-target that is too long (Section 6.5.12 of [RFC7231]) or a request payload that is too large (Section 6.5.11 of [RFC7231]). Additional status codes related to capacity limits have been defined by extensions to HTTP [RFC6585].

Recipients ought to carefully limit the extent to which they process other protocol elements, including (but not limited to) request methods, response status phrases, header field-names, numeric values, and body chunks. Failure to limit such processing can result in buffer overflows, arithmetic overflows, or increased vulnerability to denial-of-service attacks.

9.4. Response Splitting

Response splitting (a.k.a, CRLF injection) is a common technique, used in various attacks on Web usage, that exploits the line-based nature of HTTP message framing and the ordered association of requests to responses on persistent connections [Klein]. This technique can be particularly damaging when the requests pass through a shared cache.

Response splitting exploits a vulnerability in servers (usually within an application server) where an attacker can send encoded data within some parameter of the request that is later decoded and echoed within any of the response header fields of the response. If the decoded data is crafted to look like the response has ended and a

subsequent response has begun, the response has been split and the content within the apparent second response is controlled by the attacker. The attacker can then make any other request on the same persistent connection and trick the recipients (including intermediaries) into believing that the second half of the split is an authoritative answer to the second request.

For example, a parameter within the request-target might be read by an application server and reused within a redirect, resulting in the same parameter being echoed in the Location header field of the response. If the parameter is decoded by the application and not properly encoded when placed in the response field, the attacker can send encoded CRLF octets and other content that will make the application's single response look like two or more responses.

A common defense against response splitting is to filter requests for data that looks like encoded CR and LF (e.g., "%0D" and "%0A"). However, that assumes the application server is only performing URI decoding, rather than more obscure data transformations like charset transcoding, XML entity translation, base64 decoding, sprintf reformatting, etc. A more effective mitigation is to prevent anything other than the server's core protocol libraries from sending a CR or LF within the header section, which means restricting the output of header fields to APIs that filter for bad octets and not allowing application servers to write directly to the protocol stream.

9.5. Request Smuggling

Request smuggling ([Linhart]) is a technique that exploits differences in protocol parsing among various recipients to hide additional requests (which might otherwise be blocked or disabled by policy) within an apparently harmless request. Like response splitting, request smuggling can lead to a variety of attacks on HTTP usage.

This specification has introduced new requirements on request parsing, particularly with regard to message framing in Section 3.3.3, to reduce the effectiveness of request smuggling.

9.6. Message Integrity

HTTP does not define a specific mechanism for ensuring message integrity, instead relying on the error-detection ability of underlying transport protocols and the use of length or chunk-delimited framing to detect completeness. Additional integrity mechanisms, such as hash functions or digital signatures applied to the content, can be selectively added to messages via extensible

metadata header fields. Historically, the lack of a single integrity mechanism has been justified by the informal nature of most HTTP communication. However, the prevalence of HTTP as an information access mechanism has resulted in its increasing use within environments where verification of message integrity is crucial.

User agents are encouraged to implement configurable means for detecting and reporting failures of message integrity such that those means can be enabled within environments for which integrity is necessary. For example, a browser being used to view medical history or drug interaction information needs to indicate to the user when such information is detected by the protocol to be incomplete, expired, or corrupted during transfer. Such mechanisms might be selectively enabled via user agent extensions or the presence of message integrity metadata in a response. At a minimum, user agents ought to provide some indication that allows a user to distinguish between a complete and incomplete response message (Section 3.4) when such verification is desired.

9.7. Message Confidentiality

HTTP relies on underlying transport protocols to provide message confidentiality when that is desired. HTTP has been specifically designed to be independent of the transport protocol, such that it can be used over many different forms of encrypted connection, with the selection of such transports being identified by the choice of URI scheme or within user agent configuration.

The "https" scheme can be used to identify resources that require a confidential connection, as described in Section 2.7.2.

9.8. Privacy of Server Log Information

A server is in the position to save personal data about a user's requests over time, which might identify their reading patterns or subjects of interest. In particular, log information gathered at an intermediary often contains a history of user agent interaction, across a multitude of sites, that can be traced to individual users.

HTTP log information is confidential in nature; its handling is often constrained by laws and regulations. Log information needs to be securely stored and appropriate guidelines followed for its analysis. Anonymization of personal information within individual entries helps, but it is generally not sufficient to prevent real log traces from being re-identified based on correlation with other access characteristics. As such, access traces that are keyed to a specific client are unsafe to publish even if the key is pseudonymous.

To minimize the risk of theft or accidental publication, log information ought to be purged of personally identifiable information, including user identifiers, IP addresses, and user-provided query parameters, as soon as that information is no longer necessary to support operational needs for security, auditing, or fraud control.

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Appendix A. HTTP Version History

HTTP has been in use since 1990. The first version, later referred to as HTTP/0.9, was a simple protocol for hypertext data transfer across the Internet, using only a single request method (GET) and no metadata. HTTP/1.0, as defined by [RFC1945], added a range of request methods and MIME-like messaging, allowing for metadata to be transferred and modifiers placed on the request/response semantics. However, HTTP/1.0 did not sufficiently take into consideration the effects of hierarchical proxies, caching, the need for persistent connections, or name-based virtual hosts. The proliferation of incompletely implemented applications calling themselves "HTTP/1.0" further necessitated a protocol version change in order for two communicating applications to determine each other's true capabilities.

HTTP/1.1 remains compatible with HTTP/1.0 by including more stringent requirements that enable reliable implementations, adding only those features that can either be safely ignored by an HTTP/1.0 recipient or only be sent when communicating with a party advertising conformance with HTTP/1.1.

HTTP/1.1 has been designed to make supporting previous versions easy. A general-purpose HTTP/1.1 server ought to be able to understand any valid request in the format of HTTP/1.0, responding appropriately with an HTTP/1.1 message that only uses features understood (or safely ignored) by HTTP/1.0 clients. Likewise, an HTTP/1.1 client can be expected to understand any valid HTTP/1.0 response.

Since HTTP/0.9 did not support header fields in a request, there is no mechanism for it to support name-based virtual hosts (selection of resource by inspection of the Host header field). Any server that implements name-based virtual hosts ought to disable support for HTTP/0.9. Most requests that appear to be HTTP/0.9 are, in fact, badly constructed HTTP/1.x requests caused by a client failing to properly encode the request-target.

A.1. Changes from HTTP/1.0

This section summarizes major differences between versions HTTP/1.0 and HTTP/1.1.

A.1.1. Multihomed Web Servers

The requirements that clients and servers support the Host header field (Section 5.4), report an error if it is missing from an HTTP/1.1 request, and accept absolute URIs (Section 5.3) are among the most important changes defined by HTTP/1.1.

Older HTTP/1.0 clients assumed a one-to-one relationship of IP addresses and servers; there was no other established mechanism for distinguishing the intended server of a request than the IP address to which that request was directed. The Host header field was introduced during the development of HTTP/1.1 and, though it was quickly implemented by most HTTP/1.0 browsers, additional requirements were placed on all HTTP/1.1 requests in order to ensure complete adoption. At the time of this writing, most HTTP-based services are dependent upon the Host header field for targeting requests.

A.1.2. Keep-Alive Connections

In HTTP/1.0, each connection is established by the client prior to the request and closed by the server after sending the response. However, some implementations implement the explicitly negotiated ("Keep-Alive") version of persistent connections described in Section 19.7.1 of [RFC2068].

Some clients and servers might wish to be compatible with these previous approaches to persistent connections, by explicitly negotiating for them with a "Connection: keep-alive" request header field. However, some experimental implementations of HTTP/1.0 persistent connections are faulty; for example, if an HTTP/1.0 proxy server doesn't understand Connection, it will erroneously forward that header field to the next inbound server, which would result in a hung connection.

One attempted solution was the introduction of a Proxy-Connection header field, targeted specifically at proxies. In practice, this was also unworkable, because proxies are often deployed in multiple layers, bringing about the same problem discussed above.

As a result, clients are encouraged not to send the Proxy-Connection header field in any requests.

Clients are also encouraged to consider the use of Connection: keep-alive in requests carefully; while they can enable persistent connections with HTTP/1.0 servers, clients using them will need to monitor the connection for "hung" requests (which indicate that the client ought stop sending the header field), and this mechanism ought not be used by clients at all when a proxy is being used.

A.1.3. Introduction of Transfer-Encoding

HTTP/1.1 introduces the Transfer-Encoding header field (Section 3.3.1). Transfer codings need to be decoded prior to forwarding an HTTP message over a MIME-compliant protocol.

A.2. Changes from RFC 2616

HTTP's approach to error handling has been explained. (Section 2.5)

The HTTP-version ABNF production has been clarified to be case-sensitive. Additionally, version numbers have been restricted to single digits, due to the fact that implementations are known to handle multi-digit version numbers incorrectly. (Section 2.6)

Userinfo (i.e., username and password) are now disallowed in HTTP and HTTPS URIs, because of security issues related to their transmission on the wire. (Section 2.7.1)

The HTTPS URI scheme is now defined by this specification; previously, it was done in Section 2.4 of [RFC2818]. Furthermore, it implies end-to-end security. (Section 2.7.2)

HTTP messages can be (and often are) buffered by implementations; despite it sometimes being available as a stream, HTTP is fundamentally a message-oriented protocol. Minimum supported sizes for various protocol elements have been suggested, to improve interoperability. (Section 3)

Invalid whitespace around field-names is now required to be rejected, because accepting it represents a security vulnerability. The ABNF productions defining header fields now only list the field value. (Section 3.2)

Rules about implicit linear whitespace between certain grammar productions have been removed; now whitespace is only allowed where specifically defined in the ABNF. (Section 3.2.3)

Header fields that span multiple lines ("line folding") are deprecated. (Section 3.2.4)

The NUL octet is no longer allowed in comment and quoted-string text, and handling of backslash-escaping in them has been clarified. The quoted-pair rule no longer allows escaping control characters other than HTAB. Non-US-ASCII content in header fields and the reason phrase has been obsoleted and made opaque (the TEXT rule was removed). (Section 3.2.6)

Bogus Content-Length header fields are now required to be handled as errors by recipients. (Section 3.3.2)

The algorithm for determining the message body length has been clarified to indicate all of the special cases (e.g., driven by methods or status codes) that affect it, and that new protocol

elements cannot define such special cases. CONNECT is a new, special case in determining message body length. "multipart/byteranges" is no longer a way of determining message body length detection. (Section 3.3.3)

The "identity" transfer coding token has been removed. (Sections 3.3 and 4)

Chunk length does not include the count of the octets in the chunk header and trailer. Line folding in chunk extensions is disallowed. (Section 4.1)

The meaning of the "deflate" content coding has been clarified. (Section 4.2.2)

The segment + query components of RFC 3986 have been used to define the request-target, instead of abs_path from RFC 1808. The asterisk-form of the request-target is only allowed with the OPTIONS method. (Section 5.3)

The term "Effective Request URI" has been introduced. (Section 5.5)

Gateways do not need to generate Via header fields anymore. (Section 5.7.1)

Exactly when "close" connection options have to be sent has been clarified. Also, "hop-by-hop" header fields are required to appear in the Connection header field; just because they're defined as hop-by-hop in this specification doesn't exempt them. (Section 6.1)

The limit of two connections per server has been removed. An idempotent sequence of requests is no longer required to be retried. The requirement to retry requests under certain circumstances when the server prematurely closes the connection has been removed. Also, some extraneous requirements about when servers are allowed to close connections prematurely have been removed. (Section 6.3)

The semantics of the Upgrade header field is now defined in responses other than 101 (this was incorporated from [RFC2817]). Furthermore, the ordering in the field value is now significant. (Section 6.7)

Empty list elements in list productions (e.g., a list header field containing ", ,") have been deprecated. (Section 7)

Registration of Transfer Codings now requires IETF Review (Section 8.4)

This specification now defines the Upgrade Token Registry, previously defined in Section 7.2 of [RFC2817]. (Section 8.6)

The expectation to support HTTP/0.9 requests has been removed. (Appendix A)

Issues with the Keep-Alive and Proxy-Connection header fields in requests are pointed out, with use of the latter being discouraged altogether. (Appendix A.1.2)

Appendix B. Collected ABNF

BWS = OWS

Connection = *("," OWS) connection-option *(OWS "," [OWS connection-option])

Content-Length = 1*DIGIT

HTTP-message = start-line *(header-field CRLF) CRLF [message-body]

HTTP-name = %x48.54.54.50 ; HTTP

HTTP-version = HTTP-name "/" DIGIT "." DIGIT

Host = uri-host [":" port]

OWS = *(SP / HTAB)

RWS = 1*(SP / HTAB)

TE = [("," / t-codings) *(OWS "," [OWS t-codings])]

Trailer = *("," OWS) field-name *(OWS "," [OWS field-name])

Transfer-Encoding = *("," OWS) transfer-coding *(OWS "," [OWS transfer-coding])

URI-reference = <URI-reference, see [RFC3986], Section 4.1>

Upgrade = *("," OWS) protocol *(OWS "," [OWS protocol])

Via = *("," OWS) (received-protocol RWS received-by [RWS comment]) *(OWS "," [OWS (received-protocol RWS received-by [RWS comment])])

absolute-URI = <absolute-URI, see [RFC3986], Section 4.3>

absolute-form = absolute-URI

absolute-path = 1*("/" segment)

asterisk-form = "*"

authority = <authority, see [RFC3986], Section 3.2>

authority-form = authority

```

chunk = chunk-size [ chunk-ext ] CRLF chunk-data CRLF
chunk-data = 1*OCTET
chunk-ext = *( ";" chunk-ext-name [ "=" chunk-ext-val ] )
chunk-ext-name = token
chunk-ext-val = token / quoted-string
chunk-size = 1*HEXDIG
chunked-body = *chunk last-chunk trailer-part CRLF
comment = "(" *( ctext / quoted-pair / comment ) ")"
connection-option = token
ctext = HTAB / SP / %x21-27 ; '!'-' '
      / %x2A-5B ; '*'-'['
      / %x5D-7E ; ']'-'~'
      / obs-text

field-content = field-vchar [ 1*( SP / HTAB ) field-vchar ]
field-name = token
field-value = *( field-content / obs-fold )
field-vchar = VCHAR / obs-text
fragment = <fragment, see [RFC3986], Section 3.5>

header-field = field-name ":" OWS field-value OWS
http-URI = "http://" authority path-abempty [ "?" query ] [ "#"
  fragment ]
https-URI = "https://" authority path-abempty [ "?" query ] [ "#"
  fragment ]

last-chunk = 1*"0" [ chunk-ext ] CRLF

message-body = *OCTET
method = token

obs-fold = CRLF 1*( SP / HTAB )
obs-text = %x80-FF
origin-form = absolute-path [ "?" query ]

partial-URI = relative-part [ "?" query ]
path-abempty = <path-abempty, see [RFC3986], Section 3.3>
port = <port, see [RFC3986], Section 3.2.3>
protocol = protocol-name [ "/" protocol-version ]
protocol-name = token
protocol-version = token
pseudonym = token

qdtext = HTAB / SP / "!" / %x23-5B ; '#'-'['
      / %x5D-7E ; ']'-'~'
      / obs-text
query = <query, see [RFC3986], Section 3.4>
quoted-pair = "\" ( HTAB / SP / VCHAR / obs-text )

```

quoted-string = DQUOTE *(qdtext / quoted-pair) DQUOTE

rank = ("0" ["." *3DIGIT]) / ("1" ["." *3"0"])

reason-phrase = *(HTAB / SP / VCHAR / obs-text)

received-by = (uri-host [":" port]) / pseudonym

received-protocol = [protocol-name "/"] protocol-version

relative-part = <relative-part, see [RFC3986], Section 4.2>

request-line = method SP request-target SP HTTP-version CRLF

request-target = origin-form / absolute-form / authority-form /
asterisk-form

scheme = <scheme, see [RFC3986], Section 3.1>

segment = <segment, see [RFC3986], Section 3.3>

start-line = request-line / status-line

status-code = 3DIGIT

status-line = HTTP-version SP status-code SP reason-phrase CRLF

t-codings = "trailers" / (transfer-coding [t-ranking])

t-ranking = OWS ";" OWS "q=" rank

tchar = "!" / "#" / "\$" / "%" / "&" / "'" / "*" / "+" / "-" / "." /
"^" / " " / "`" / "|" / "~" / DIGIT / ALPHA

token = 1*tchar

trailer-part = *(header-field CRLF)

transfer-coding = "chunked" / "compress" / "deflate" / "gzip" /
transfer-extension

transfer-extension = token *(OWS ";" OWS transfer-parameter)

transfer-parameter = token BWS "=" BWS (token / quoted-string)

uri-host = <host, see [RFC3986], Section 3.2.2>

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Authors' Addresses

Roy T. Fielding (editor)
Adobe Systems Incorporated
345 Park Ave
San Jose, CA 95110
USA

E-Mail: fielding@gbiv.com
URI: <http://roy.gbiv.com/>

Julian F. Reschke (editor)
greenbytes GmbH
Hafenweg 16
Muenster, NW 48155
Germany

E-Mail: julian.reschke@greenbytes.de
URI: <http://greenbytes.de/tech/webdav/>